

Yealink SIP-T2 Series/T19P/T4 Series IP Phones Auto Provisioning Guide

Table of Contents

Table of Contents	iii
Summary of Changes	vii
Changes for Release 72, Guide Version 72.2	vii
Changes for Release 72, Guide Version 72.1	vii
Changes for Release 71, Guide Version 71.165	vii
Changes for Release 71, Guide Version 71.140	viii
Changes for Release 71, Guide Version 71.125	viii
Changes for Release 71, Guide Version 71.120	viii
Changes for Release 71, Guide Version 71.110	viii
Changes for Release 70, Guide Version 1.3	
Introduction	1
Getting Started	3
Obtaining Configuration Information	3
Obtaining Configuration Files	3
Obtaining Phone Information	4
Managing Configuration Files	4
Editing Common CFG File	5
Editing MAC-Oriented CFG File	7
Encrypting Configuration Files	14
Customizing Resource Files	14
Customizing a Ring Tone	14
Customizing an LCD Language	15
Customizing an LCD Logo	16
Customizing a Wallpaper	
Customizing a Local Contact File	
Customizing a Replace Rule File Customizing a Dial-now File	
Customizing a Diai-now File	
Customizing a Super Search Template	
Upgrading Firmware	
Configuring a TFTP Server	27

Preparing a Root Directory	27
Configuring a TFTP Server	28
Obtaining the Address of Provisioning Server	29
Zero Touch	29
Plug and Play (PnP) Server	31
DHCP Options	32
Phone Flash	33
Update Mode	35
Power On	35
Repeatedly	36
Weekly	37
Auto Provision Now	38
Multi-mode Mixed	39
SIP NOTIFY Message	39
Downloading and Verifying Configurations	41
Downloading Configuration Files	41
Resolving and Updating Configurations	41
Verifying Configurations	42
Troubleshooting	45
Glossary	47
Appendix	49
Configuring an FTP Server	49
Preparing a Root Directory	49
Configuring an FTP server	50
Configuring an HTTP Server	52
Preparing a Root Directory	52
Configuring an HTTP Server	52
Configuring a DHCP server	55
Customizing a Ring Tone Using Cool Edit Pro	63
Customizing a Logo File Using PictureExDemo	64
Description of Configuration Parameters in CFG Files	65
Time 7ones	225

BLF LED Mode

Summary of Changes

This section describes the changes to this guide for each release and guide version.

Changes for Release 72, Guide Version 72.2

This version is updated to incorporate SIPT48G IP phones. The following sections are new for this version:

- Customizing a Directory Template on page 23
- Customizing a Super Search Template on page 24

Major updates have occurred to the following sections:

- Editing Common CFG File on page 5
- Editing MAC-Oriented CFG File on page 7
- Customizing Resource Files on page 14
- Description of Configuration Parameters in CFG Files on page 65
- BLF LED Mode on page 228

Changes for Release 72, Guide Version 72.1

This version is updated to incorporate SIPT46G, SIPT42G and SIPT41P IP phones. The following sections are new for this version:

- Time Zones on page 225
- BLF LED Mode on page 228

Major updates have occurred to the following sections:

- Editing Common CFG File on page 5
- Editing MAC-Oriented CFG File on page 7
- Customizing Resource Files on page 14
- Description of Configuration Parameters in CFG Files on page 65

Changes for Release 71, Guide Version 71.165

Documentations of the newly released SIP-T21P and SIP-T19P IP phones have also been added.

Changes for Release 71, Guide Version 71.140

Major updates have occurred to the following sections:

- Editing Common CFG File on page 5
- Editing MAC-Oriented CFG File on page 7
- Encrypting Configuration Files on page 14
- Customizing an LCD Logo on page 16
- Customizing a Local Contact File on page 18
- Description of Configuration Parameters in CFG Files on page 65

Changes for Release 71, Guide Version 71.125

Major updates have occurred to the following section:

Customizing an LCD Logo on page 16

Changes for Release 71, Guide Version 71.120

Major updates have occurred to the following section:

• Description of Configuration Parameters in CFG Files on page 65

Changes for Release 71, Guide Version 71.110

The following sections are new for this version:

- Encrypting Configuration Files on page 14
- Update Mode on page 35
- SIP NOTIFY Message on page 39
- Resolving and Updating Configurations on page 41
- Description of Configuration Parameters in CFG Files on page 65

Major updates have occurred to the following sections:

- Customizing a Local Contact File on page 18
- Customizing a Replace Rule File on page 21
- Customizing a Dial-now File on page 22

Changes for Release 70, Guide Version 1.3

The following sections are new for this version:

- Customizing a Wallpaper on page 17
- Customizing a Replace Rule File on page 21
- Customizing a Dial-now File on page 22

Major updates have occurred to the following sections:

- Customizing a Local Contact File on page 18
- Upgrading Firmware on page 25

Introduction

Yealink IP phones are full-featured telephones that can be plugged directly into an IP network and can be used easily without manual configuration.

This guide provides instructions on how to provision Yealink IP phones with the minimum settings required. Yealink IP phones support FTP, TFTP, HTTP, and HTTPS protocols for auto provisioning and are configured by default to use the TFTP protocol.

The purpose of this guide is to serve as a basic guidance for provisioning Yealink IP phones, including:

- Yealink SIP-T28P
- Yealink SIP-T26P
- Yealink SIP-T22P
- Yealink SIP-T21P
- Yealink SIP-T20P
- Yealink SIP-T19P
- Yealink SIP-T48G
- Yealink SIP-T46G
- Yealink SIP-T42G
- Yealink SIP-T41P

The auto provisioning process outlined in this guide applies to Yealink IP phones running firmware V72 or later. We recommend that IP phones running firmware V72 or later CANNOT be downgraded to an earlier firmware version. The new firmware is compatible with old configuration parameters, but not vice versa.

This guide is also applicable to SIP-T21/T20/T19 IP phones.

Getting Started

This section provides instructions on how to get ready for auto provisioning. The auto provisioning process discussed in this guide uses the TFTP server as the provisioning server.

To begin the auto provisioning process, the following steps are required:

- Obtaining Configuration Information
- Managing Configuration Files

Obtaining Configuration Information

Obtaining Configuration Files

Before beginning provisioning, you need to obtain configuration files. There are two configuration files both of which are CFG-formatted. We call these two files Common CFG file and MAC-Oriented CFG file. The phone tries to download these CFG files from the server during provisioning.

The MAC-Oriented CFG file is only effectual for the specific phone. It uses the 12-digit MAC address of the phone as the file name. For example, if the MAC address of the phone is 0015651130F9, the MAC-Oriented CFG file name must be 0015651130F9.cfg. However, the Common CFG file is effectual for all the phones with the same model. It uses a fixed name "y0000000000XX.cfg" or "y0000000000X.cfg" as the file name, where "XX" or "X" equals to the first two digits or the first digit (except 0 for SIP-T28P) of the hardware version of the phone model.

The names of the Common CFG file for each phone model are:

Phone Model	Common CFG File
SIP-T28P	y00000000000.cfg
SIP-T26P	y00000000004.cfg
SIP-T22P	y00000000005.cfg
SIP-T21P	y00000000034.cfg
SIP-T20P	y00000000007.cfg
SIP-T19P	y00000000031.cfg
SIP-T48G	y00000000035.cfg
SIP-T46G	y00000000028.cfg

Phone Model	Common CFG File
SIP-T42G	y00000000029.cfg
SIP-T41P	y00000000036.cfg

You can ask the distributor or Yealink FAE for configuration files. The IP phones running firmware version 71 or later can only recognize configuration files using UTF-8 or ANSI encoding.

Obtaining Phone Information

Before beginning provisioning, you also need the phone information. For example, MAC address and the SIP account information of the phone.

MAC Address: The unique 12-digit serial number of the phone. You can obtain it from the bar code on the back of the phone.

SIP Account Information: This may include SIP credentials such as user name, password and IP address of the SIP server. Ask your system administrator for SIP account information.

Managing Configuration Files

Auto provisioning enables Yealink IP phones to update themselves automatically via downloading Common CFG and MAC-Oriented CFG files. Before beginning provisioning, you may need to edit and customize your configuration files. For more information on configuration parameters in configuration files, refer to Description of Configuration Parameters in CFG Files on page 65.

Editing Common CFG File

Common CFG file contains configuration parameters which apply to phones with the same model, such as language and volume.

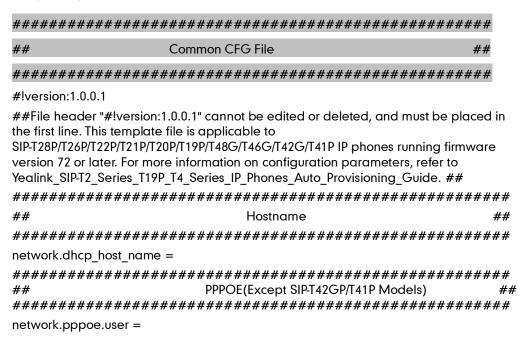
The following figure shows a portion of the common CFG file:

```
##File header "#!version:1.0.0.1" can not be edited or deleted, and must be placed in the first line.##
##This template file is applicable to SIP-T28P/T26P/T22P/T20P/T21P/T19P/T48G/T46G/T42G/T41P IP phones running
##For more information on configuration parameters, refer to Yealink_SIP-T2_Series_T19P_T4_Series_IP_Phones_Au
network.dhcp host name =
## PPPOE(Except T41P/T42G Models) ##
network.pppoe.user =
network.pppoe.password =
network.bridge_mode =
network.pc_port.ip
network.pc_port.mask =
network.dhcp.start_ip =
network.dhcp.end_ip =
network.pc_port.enable =
network.internet_port.speed_duplex =
network.pc_port.speed_duplex =
network.pc_port.dhcp_server =
network.static_dns_enable =
network.ipv6_static_dns_enable =
```

The line beginning with "#" is considered to be a comment.

The file header "#!version:1.0.0.1" is not a comment and must be placed in the first line. It cannot be edited or deleted.

The partial parameters in the Common CFG file are described as follows:



```
network.pppoe.password =
PC Port(Only SIP-T28P/T26P/T22P/T20P Models support)
##
network.bridge_mode =
network.pc_port.ip =
network.pc port.mask =
network.dhcp.start ip =
network.dhcp.end_ip =
##
              Network Advanced
network.pc port.enable =
network.internet_port.speed_duplex =
network.pc_port.speed_duplex =
network.pc port.dhcp server =
network.static dns enable =
network.ipv6 static dns enable =
##Only T41P/T42G/T46G/T48G Models support this parameter##
network.vlan.pc_port_mode =
##
                   VIAN
network.vlan.internet port enable =
network.vlan.internet port vid =
network.vlan.internet_port_priority =
network.vlan.pc port enable =
network.vlan.pc port vid =
network.vlan.pc port priority =
network.vlan.dhcp enable =
network.vlan.dhcp option =
WEB Port
network.port.http =
network.port.https =
wui.https_enable =
wui.http_enable =
QoS
network.qos.rtptos =
network.gos.signaltos =
802.1X
##
network.802_1x.mode =
network.802 1x.identity =
network.802 1x.md5 password =
```

Editing MAC-Oriented CFG File

MAC-Oriented CFG file contains configuration parameters which are expected to be updated per phone, such as the registration information.

The following figure shows a portion of the MAC-Oriented CFG file:

```
##This template file is applicable to SIP-T28P/T26P/T22P/T20P/T21P/T19P/T48G/T46G/T42G/T41P IP phones running
##For more information on configuration parameters, refer to Yealink_SIP-T2_Series_T19P_T4_Series_IP_Phones_A
## Accountl Basic Settings ##
account.1.enable =
 account.1.label =
account.1.display_name =
account. 1. auth name =
 account.1.user_name =
account.1.password =
account. 1. outbound_proxy_enable = account. 1. outbound_host = account. 1. outbound_port =
account.1.sip_listen_port = account.1.transport =
## Failback ##
account.1.reregister_enable =
account.1.naptr_build =
account.1.fallback.redundancy_type =
account.1.fallback.timeout =
account. 1. fallback.timeout =
account. 1. sip_server.1. address =
account. 1. sip_server.1. port =
account. 1. sip_server.1. retry_counts =
account. 1. sip_server.1. retry_counts =
account. 1. sip_server.1. failback_mode =
account. 1. sip_server.1. failback_timeout =
account. 1. sip_server.1. register_on_enable =
account. 1. sip_server. 2. address = account. 1. sip_server. 2. port = account. 1. sip_server. 2. expires =
account.1.sip_server.2.retry_counts =
account.1.sip_server.2.retry_counts =
account.1.sip_server.2.failback_mode =
account.1.sip_server.2.failback_timeout =
account.1.sip_server.2.register_on_enable =
```

The partial parameters in the MAC-Oriented CFG file are described as follows:

```
##
            MAC-Oriented CFG File
                                                      ##
#!version:1.0.0.1
##File header "#Iversion:1.0.0.1" cannot be edited or deleted, and must be placed in
the first line. This template file is applicable to SIP-T28P/T26P/T22P/T21P/T20P/T19P/
T48G/T46G/T42G/T41P IP phones running firmware version 72 or later. For more
information on configuration parameters, refer to
Yealink_SIP-T2_Series_T19P_T4_Series_IP_Phones_Auto_Provisioning_Guide. ##
##
                     Account 1 Basic Settings
                                                        ##
account.1.enable =
account.1.label =
account.1.display name =
account.1.auth name =
account.1.user_name =
account.1.password =
account.1.outbound_proxy_enable =
account.1.outbound host =
account.1.outbound_port =
account.1.sip_listen_port =
account.1.transport =
##
                    Failback
account.1.reregister enable =
account.1.naptr build =
account.1.fallback.redundancy type =
account.1.fallback.timeout =
account.1.sip_server.1.address =
account.1.sip_server.1.port =
account.1.sip_server.1.expires =
account.1.sip_server.1.retry_counts =
account.1.sip server.1.failback mode =
account.1.sip_server.1.failback_timeout =
account.1.sip_server.1.register_on_enable =
account.1.sip server.2.address =
```

```
account.1.sip_server.2.port =
account.1.sip_server.2.expires =
account.1.sip_server.2.retry_counts =
account.1.sip_server.2.failback_mode =
account.1.sip_server.2.failback_timeout =
account.1.sip_server.2.register_on_enable =
##
             Account 2 Basic Settings (except SIP-T19P Model)
account.2.enable =
account.2.label =
account.2.display_name =
account.2.auth_name =
account.2.user_name =
account.2.password =
account.2.outbound_proxy_enable =
account.2.outbound_host =
account.2.outbound_port =
account.2.sip_listen_port =
account.2.transport =
Failback
account.2.reregister_enable =
account.2.naptr_build =
account.2.fallback.redundancy_type =
account.2.fallback.timeout =
account.2.sip_server.1.address =
account.2.sip_server.1.port =
account.2.sip_server.1.expires =
account.2.sip_server.1.retry_counts =
account.2.sip_server.1.failback_mode =
account.2.sip_server.1.failback_timeout =
account.2.sip_server.1.register_on_enable =
account.2.sip_server.2.address =
account.2.sip_server.2.port =
account.2.sip_server.2.expires =
account.2.sip_server.2.retry_counts =
```

```
account.2.sip_server.2.failback_mode =
account.2.sip_server.2.failback_timeout =
account.2.sip server.2.register on enable =
Account3 Basic Settings (except SIP-T21P/T20P/T19P Models)
account.3.enable =
account.3.label =
account.3.display name =
account.3.auth_name =
account.3.user name =
account.3.password =
account.3.outbound proxy enable =
account.3.outbound host =
account.3.outbound port =
account.3.sip listen port =
account.3.transport =
##
                      Failback
                                                             ##
account.3.reregister_enable =
account.3.naptr_build =
account.3.fallback.redundancy type =
account.3.fallback.timeout =
account.3.sip_server.1.address =
account.3.sip server.1.port =
account.3.sip_server.1.expires =
account.3.sip_server.1.retry_counts =
account.3.sip server.1.failback mode =
account.3.sip server.1.failback timeout =
account.3.sip_server.1.register_on_enable =
account.3.sip_server.2.address =
account.3.sip server.2.port =
account.3.sip_server.2.expires =
account.3.sip_server.2.retry_counts =
account.3.sip_server.2.failback_mode =
account.3.sip_server.2.failback_timeout =
```

```
account.3.sip_server.2.register_on_enable =
Account4 Basic Settings (for SIP-T28P/T48G/T46G models)
##
account.4.enable =
account.4.label =
account.4.display_name =
account.4.auth name =
account.4.user_name =
account.4.password =
account.4.outbound proxy enable =
account.4.outbound_host =
account.4.outbound_port =
account.4.sip_listen_port =
account.4.transport =
##
                      Failback
                                                            ##
account.4.reregister_enable =
account.4.naptr_build =
account.4.fallback.redundancy_type =
account.4.fallback.timeout =
account.4.sip_server.1.address =
account.4.sip_server.1.port =
account.4.sip_server.1.expires =
account.4.sip_server.1.retry_counts =
account.4.sip_server.1.failback_mode =
account.4.sip_server.1.failback_timeout =
account.4.sip server.1.register on enable =
account.4.sip_server.2.address =
account.4.sip_server.2.port =
account.4.sip_server.2.expires =
account.4.sip_server.2.retry_counts =
account.4.sip_server.2.failback_mode =
account.4.sip_server.2.failback_timeout =
account.4.sip_server.2.register_on_enable =
```

```
##
             Account 5 Basic Settings (for SIP-T28P/T48G/T46G models)
                                                           ##
account.5.enable =
account.5.label =
account.5.display name =
account.5.auth name =
account.5.user_name =
account.5.password =
account.5.outbound proxy enable =
account.5.outbound_host =
account.5.outbound_port =
account.5.sip_listen_port =
account.5.transport =
Failback
##
                                                           ##
account.5.reregister_enable =
account.5.naptr_build =
account.5.fallback.redundancy_type =
account.5.fallback.timeout =
account.5.sip_server.1.address =
account.5.sip server.1.port =
account.5.sip_server.1.expires =
account.5.sip_server.1.retry_counts =
account.5.sip server.1.failback mode =
account.5.sip_server.1.failback_timeout =
account.5.sip_server.1.register_on_enable =
account.5.sip server.2.address =
account.5.sip server.2.port =
account.5.sip_server.2.expires =
account.5.sip_server.2.retry_counts =
account.5.sip server.2.failback mode =
account.5.sip_server.2.failback_timeout =
account.5.sip_server.2.register_on_enable =
```

```
##
               Account Basic Settings (for SIP-T28P/T48G/T46G models)
account.6.enable =
account.6.label =
account.6.display_name =
account.6.auth_name =
account.6.user_name =
account.6.password =
account.6.outbound_proxy_enable =
account.6.outbound_host =
account.6.outbound_port =
account.6.sip_listen_port =
account.6.transport =
Failback
account.6.reregister_enable =
account.6.naptr build =
account.6.fallback.redundancy_type =
account.6.fallback.timeout =
account.6.sip_server.1.address =
account.6.sip server.1.port =
account.6.sip_server.1.expires =
account.6.sip_server.1.retry_counts =
account.6.sip_server.1.failback_mode =
account.6.sip_server.1.failback_timeout =
account.6.sip_server.1.register_on_enable =
account.6.sip_server.2.address =
account.6.sip_server.2.port =
account.6.sip_server.2.expires =
account.6.sip_server.2.retry_counts =
account.6.sip_server.2.failback_mode =
account.6.sip_server.2.failback_timeout =
account.6.sip_server.2.register_on_enable =
```

Encrypting Configuration Files

To protect against unauthorized access and tampering of sensitive information (e.g., login password, registration information), you can encrypt configuration files using Yealink Configuration Encryption Tool. AES keys must be 16 characters and the supported characters contain: $0 \sim 9$, $A \sim Z$, $a \sim z$. For more information on how to encrypt configuration files, refer to *Yealink Configuration Encryption Tool User Guide*.

Customizing Resource Files

When configuring some particular features, you may need to upload resource files to IP phones, such as personalized ring tone file, language package file and logo file. Yealink supplies some resource file templates for the particular features. Ask the distributor or Yealink FAE for resource file templates. The following provides information on how to customize resource files and specify the access URL for the resource files.

Customizing a Ring Tone

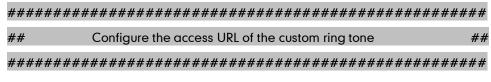
Yealink IP phones have built-in system ring tones. You can change the ring type, or customize a ring tone and upload it to the phone via auto provisioning.

The ring tone file must meet the following:

Phone Model	File Format	Single File Size	Total Files Size
SIP-T28P/T26P/T22P/T20P	.wav	<=100KB	<=100KB
SIP-T21P/T19P	.wav	<=100KB	<=200KB
SIP-T48G/T46G	.wav	<=8MB	<=20MB
SIP-T42G/T41P	.wav	<=100KB	<=100KB

The ring tone file must be PCMU audio format, mono channel, 8K sample rate and 16 bit resolution.

For more information on customizing a ring tone file, refer to Customizing a Ring Tone Using Cool Edit Pro on page 63.



ringtone.url =

For example, enter "tftp://192.168.1.100/Ring1.wav" in the "ringtone.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the ring tone file "Ring1.wav".

#Delete all custom ring tones.

ringtone.delete = http://localhost/all

Customizing an LCD Language

You can modify the language translation for the phone user interface. For SIP-T4X IP phones, you can also add a new language to IP phones.

The following table lists available languages and the associated language files for the phone user interface:

Available Language	Associated Language Pack for SIP-T28P/T26P/T22P/T20P/ T42G/T41P	Associated Language Pack for SIP-T21P/T19P/T48G/T46G
English	lang+English.txt	lang+English.txt
Chinese_S	1	lang-Chinese_S.txt
Chinese_T	1	lang-Chinese_T.txt
German	lang-German.txt	lang-German.txt
French	lang-French.txt	lang-French.txt
Italian	lang-Italian.txt	lang-Italian.txt
Polish	lang-Polish.txt	lang-Polish.txt
Portuguese	lang-Portuguese.txt	lang-Portuguese.txt
Spanish	lang-Spanish.txt	lang-Spanish.txt
Turkish	lang-Turkish.txt	lang-Turkish.txt

The following figure shows a portion of a txt formatted English language file:

Configure the access URL of the LCD language file

gui_lang.url =

If you want to modify the translation of an existing language, configure the parameter "gui_lang.url =" in the configuration file, for example:

gui lang.url = tftp://192.168.1.100/lang+English.txt.

During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the language file "lang+English.txt".

If you want to add a new language "Russian" to SIP-T4X IP phones, configure the parameter "gui_lang.url =" in the configuration file, for example:

gui_lang.url = tftp://192.168.1.100/lang-Russian.txt

During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the language file "lang-Russian.txt". After update, you will find a new language selection "Russian" on the phone user interface:

Menu->Basic->Language.

#Delete all custom languages.

gui lang.delete = http://localhost/all

SIPT48G/T46G IP phones also support to add a custom or modify an existing LCD language with an X.GUI.Y.lang (e.g., 0.GUI.English.lang) formatted language file. "X" is an integer which starts from 0. "Y" means the name of the language. X must start from 10

Available languages may vary between different firmware versions.

English language file name must be "lang+English.txt".

To modify translation of an existing language, do not rename the language file.

Customizing an LCD Logo

Yealink IP phones allow you to customize the logo displayed on the LCD screen. SIP-T20P IP phones only support a text logo. Logo is not applicable to SIP-T48G/T46G IP phones. These two IP phone models use the wallpaper instead.

The following table lists the supported logo file format and resolution for each phone model:

Phone Model	Logo File Format	Resolution
SIP-T28P	.dob	<=236*82 2 gray scale
SIP-T26P	.dob	<=132*64 2 gray scale
SIP-T22P/T21P/T19P	.dob	<=132*64 2 gray scale
SIP-T42G/T41P	.dob	<=192*64 2 gray scale

For more information on customizing a logo file, refer to Customizing a Logo File Using PictureExDemo on page 64.

Ask the distributor or Yealink FAE for the logo file, or you can customize a *.dob logo file. Upload the logo file to the provisioning server and then specify the access URL in configuration files:

Configure the access URL of the Logo File

##

#(not applicable to SIP-T20P IP phones)

lcd_logo.url =

For example, enter "tftp://192.168.1.100/logo.dob" in the "lcd_logo.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the logo file "logo.dob".

To use the custom logo, you also need to configure the following parameter:

#Configure the logo mode (not applicable to SIP-T20P IP phones).

#0-Disabled (Except for SIP-T28P IP phones), 1-System logo, 2-Custom logo

phone_setting.lcd_logo.mode = 2

To configure a text logo, you need to configure the following parameter:

#Enable or disable a text logo (only applicable to SIP-T20P IP phones).

#0-Disabled, 1-Enabled

phone_setting.lcd_logo.mode = 1

#Configure a text logo (only applicable to SIP-T20P IP phones).

phone_setting.lcd_logo.text = Yealink

After auto provisioning, you will find that the custom logo or text logo appears on the LCD screen.

#Delete all custom logo files (not applicable to SIP-T20P IP phones).

lcd logo.delete = http://localhost/all

Customizing a Wallpaper

Yealink SIP-T48G and SIP-T46G IP phones allow you to customize the wallpaper displayed on the LCD screen.

The following table lists the supported wallpaper image format and resolution for SIP-T48G and SIP-T46G IP phones:

Phone Model	Format	Resolution	Single File Size	Total File Size
SIP-T46G	.jpg/.png/.bmp	<=480*272	<=5MB	<=20MB
SIP-T48G	.jpg/.png/.bmp	<=800*480	<=5MB	<=20MB

Upload the wallpaper image to the provisioning server and then specify the access URL in configuration files:

For example, enter "tftp://192.168.1.100/wallpaper.jpg" in the "wallpaper_upload.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the wallpaper image "wallpaper.jpg".

To use the custom wallpaper, you also need to configure the following parameter:

#Configure the custom image (e.g., wallpaper.jpg) as phone wallpaper. phone_setting.backgrounds = Config:wallpaper.jpg

Customizing a Local Contact File

Yealink IP phones allow you to upload contact data in batch via auto provisioning. You can create multiple contacts using the provided local contact template file. The existing local contacts on the phones will be overwritten by the downloaded local contacts.

Yealink IP phones support both *.xml and *.csv formats.

When editing the local contact template file, learn the following:

- Add groups between <root_group> and </root_group>.
- At most 5 groups (including the default groups) can be stored on SIP-T19/T2xP IP phones.
- At most 48 groups (including the default groups) can be stored on SIP-T4X IP phones.
- Add local contacts between <root_contact> and </root_contact>.
- At most 1000 local contacts can be added to IP phones.
- When specifying a desired line for a contact, valid values are -1~6. This is not applicable to SIP-T19P IP phones.

(For SIP-T21P/T20P, valid values are $0\sim2$. For SIP-T26P/T22P, valid values are $0\sim3$. For SIP-T28P, valid values are $0\sim6$. 0 stands for Auto (the first registered line). $1\sim6$ stand for line $1\sim1$ ine 6.

For SIP-T42G/T41P, valid values are -1 \sim 2. For SIP-T48G/T46G, valid values are -1 \sim 5. -1 stands for Auto (the first registered line). 0 \sim 5 stand for line1 \sim line6. Multiple line IDs are separated by commas.

When specifying a ring tone for a contact, valid values are Auto,
 Resource:RingN.wav (system ring tone, integer N ranges from 1 to 5 for

- SIP-T2xP/T19P/T42G/T41P and from 1 to 8 for SIP-T48G/T46G) and Custom:Name.wav (custom ring tone).
- When specifying a group for a contact, valid values are the group names (built-in or custom groups).
- When specifying an avatar for a contact, valid values are "Default: avatar name" (for the built-in avatar) and "Config: avatar name" (for the custom avatar). This is only applicable to SIP-T48G/T46G IP phones.

To customize a local contact file:

- 1. Open the template file using an ASCII editor.
- 2. For each group that you wish to add, add the following string to the file. Each starts on a separate line:

```
<group display_name="" ring=""/>
```

Where:

```
display_name="" specifies the name of the group.
```

ring="" specifies the ring tone for this group.

3. For each contact that you wish to add, add the following string to the file. Each starts on a separate line:

```
<contact display_name="" office_number="" mobile_number="" other_number=""
line="" ring="" group_id_name="" default_photo=""/>
```

Where:

display_name="" specifies the name of the contact (This value cannot be blank or duplicated).

office number="" specifies the office number of the contact.

mobile_number="" specifies the mobile number of the contact.

other_number="" specifies the other number of the contact.

line="" specifies the line for the contact.

ring="" specifies the ring tone for the contact.

group_id_name="" specifies the group you want to add the contact to.

default_photo="" specifies the avatar for the contact (for SIP-T48G/T46G IP phones).

- 4. Specify the values within double quotes.
- **5.** Save the change.

After editing the local contact template file, upload it to the provisioning server and then specify the access URL in configuration files.

The following shows an example of a local contact file used for SIP-T2xP IP phones:

```
<root group>
   <group display name="All Contacts" ring=""/>
   <group display name="Family" ring="Resource:Ring1.wav"/>
   <group display name="Friend" ring="Auto"/>
</root group>
<root contact>
   <contact display name="Mary" office number="123" mobile number="456"</pre>
   other number="2201" line="0" ring="Auto" group id name="Family"/>
   <contact display name="Damy" office number="124" mobile number="789"</pre>
   other number="2202" line="1" ring="Resource:Ring2.wav"
   group id name=""/>
   <contact display name="Jack" office number="125" mobile number="234"</pre>
   other number="2203" line="2" ring="Custom:lin.wav"
   group_id_name="Family"/>
   <contact display_name="Ada" office number="8800"</pre>
  mobile number="1234" other number="0000" line="0" ring=""
  group id name=""/>
</root contact>
```


local_contact.data.url =

file "Contact.tar".

For example, enter "tftp://192.168.1.100/contact_list.xml" in the "local_contact.data.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the contact file "contact_list.xml".

Customizing a Replace Rule File

You can create replace rules directly in configuration files, or create multiple replace rules using the supplied replace rule template file. The existing replace rules on the phones will be overwritten by the downloaded replace rules.

When editing the replace rule template file, learn the following:

- <DialRule> indicates the start of the template file and </DialRule> indicates the
 end of the template file.
- Create replace rules between < DialRule > and </ DialRule >.
- When specifying the desired line(s) to apply the replace rule, valid values are 0
 and line ID. The digit 0 stands for all lines. Multiple line IDs are separated by
 commas. Specifying the desired line(s) to apply replace rules is not applicable to
 SIP-T19P IP phones.
- At most 100 replace rules can be added to the IP phone.
- For the basic expression syntax of the replace rule, refer to Yealink phone-specific user guide.

To customize a replace rule file:

- 1. Open the template file using an ASCII editor.
- 2. For each replace rule you wish to add, add the following string to the file. Each starts on a separate line:

```
<Data Prefix="" Replace="" LineID=""/>
```

Where:

Prefix="" specifies the numbers to be replaced.

Replace="" specifies the alternate string.

LineID="" specifies the desired line(s) for this rule. When you leave it blank or enter 0, this replace rule will apply to all lines.

- 3. Specify the values within double quotes.
- Save the change.

The following shows an example of a replace rule file:

```
<DialRule>
     <Data Prefix="1" Replace="05928665234" LineID=""/>
     <Data Prefix="2(xx)" Replace="002$1" LineID="0"/>
     </DialRule>
```

################	#################################	##
##	Upload replace rule file	##
#################	################################	##
dialplan_replace_rule.url =		

For example, enter "tftp://192.168.1.100/DialPlan.xml" in the "dialplan_replace_rule.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the replace rule file "DialPlan.xml".

Customizing a Dial-now File

You can create dial-now rules directly in configuration files, or create multiple dial-now rules using the supplied dial-now rule template file. The existing dial-now rules on the phones will be overwritten by the downloaded dial-now rules.

When editing a dial-now file, learn the following:

- <DialNow> indicates the start of the template file and </DialNow> indicates the
 end of the template file.
- Create dial-now rules between < DialNow > and </DialNow >.
- When specifying the desired line(s) for the dial-now rule, valid values are 0 and line ID. The digit 0 stands for all lines. Multiple line IDs are separated by commas.
 Specifying the desired line(s) to apply dial-now rules is not applicable to SIP-T19P IP phones.
- At most 100 dial-now rules can be added to the IP phone.
- For the basic expression syntax of the dial-now rule, refer to Yealink phone-specific user guide.

To customize a dial-now file:

- 1. Open the template file using an ASCII editor.
- 2. For each dial-now rule you wish to add, add the following string to the file. Each starts on a separate line:

```
<Data DialNowRule="" LineID=""/>
```

Where:

DialNowRule=""/ rule="" specifies the dial-now rule.

LineID=""/ lines="" specifies the desired line(s) for this rule. When you leave it blank or enter 0, this dial-now rule will apply to all lines.

- **3.** Specify the values within double quotes.
- 4. Save the change.

The following shows an example of a dial-now file:

```
<DialNow>
  <Data DialNowRule="1234" LineID="1"/>
  <Data DialNowRule="52[0-6]" LineID="1"/>
  <Data DialNowRule="xxxxxx" LineID=""/>
  </DialNow>
```

For example, enter "tftp://192.168.1.100/DialNow.xml" in the "dialplan_dialnow.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the dial-now file "DialNow.xml".

Customizing a Directory Template

Directory provides easy access to frequently used lists. You can access lists by pressing the Directory soft key when the IP phone is idle. The lists may contain Local Directory, History, Remote Phone Book and LDAP (LDAP list is not applicable to SIP-T19P IP phones). You can add the desired list(s) to Directory using the supplied directory template (favorite_setting.xml). After setup, place the directory template to the provisioning server and specify the access URL in the configuration files. Directory is not applicable to SIP-T20P IP phones.

When editing a directory template, learn the following:

- Do not rename the directory template.
- <root_favorite_set> indicates the start of a template and </root_favorite_set> indicates the end of a template.
- The default display names of directory lists are Local Directory, History, Remote Phone Book and LDAP.
- When specifying the display priority of the directory list, the valid values are 1, 2, 3 and 4. 1 is the highest priority, 4 is the lowest.
- When enabling or disabling the desired directory list for Directory, the valid values are 0 and 1. 0 stands for Disabled, 1 stands for Enabled.

To customize a directory template:

- Open the template file using an ASCII editor.
- 2. For each directory list that you want to configure, edit the corresponding string in the file. For example, you want to configure the local directory list, edit the following strings:

```
<item id_name="localdirectory" display_name="Local Directory" priority="1" enable="1" />
```

Where:

id_name="" specifies the directory list (id_name = "localdirectory" specifies the local directory list). Do not edit this field.

display_name="" specifies the display name of the directory list. We recommend you do not edit this field.

priority="" specifies the display priority of the directory list. enable="" enables or disables the directory list for Directory.

- 3. Edit the values within double quotes.
- 4. Place this file to the provisioning server.

The following is an example of a directory template:

```
<root_favorite_set>
    <item id_name="localdirectory" display_name="Local Directory" priority="1"
    enable="1" />
        <item id_name="history" display_name="History" priority="2" enable="0" />
        <item id_name="remotedirectory" display_name="Remote Phone Book"
        priority="3" enable="0" />
        <item id_name="ldap" display_name="LDAP" priority="4" enable="0" />
        </root favorite set>
```

Customizing a Super Search Template

Search source list in dialing allows the IP phone to search for entries from the desired lists when the phone is in the dialing screen, and then the user can select the desired entry to dial out quickly. The lists may contain Local Directory, History, Remote Phone Book and LDAP (LDAP list is not applicable to SIP-T19P IP phones). You can configure the search source list in dialing using the supplied super search template (super_search.xml). After setup, place the super search template to the provisioning server and specify the access URL in the configuration files. Search source list in dialing is not applicable to SIP-T20P IP phones.

When editing a super search template, learn the following:

- Do not rename the super search template.
- <root_super_search> indicates the start of a template and </root_super_search> indicates the end of a template.
- The default display names of directory lists are Local Directory, History, Remote Phone Book and LDAP.
- When specifying the priority of search results, the valid values are 1, 2, 3 and 4. 1 is the highest priority, 4 is the lowest.

• When enabling or disabling the phone to search the desired directory list, the valid values are 0 and 1.0 stands for Disabled, 1 stands for Enabled.

To customize a super search template:

- 1. Open the template file using an ASCII editor.
- 2. For each directory list that you want to configure, edit the corresponding string in the file. For example, you want to configure the local directory list, edit the following strings:

```
<item id_name="local_directory_search" display_name="Local Directory" priority="1" enable="1" />
```

Where:

id_name="" specifies the directory list (id_name = "local_directory_search" specifies the local directory list). Do not edit this field.

display_name="" specifies the display name of the directory list. We recommend you do not edit this field.

priority="" specifies the priority of search results.

enable="" enables or disables the phone to search the directory list.

- 3. Edit the values within double quotes.
- 4. Place this file to the provisioning server.

The following is an example of a super search template:

```
<root_super_search>
    <item id_name="local_directory_search" display_name="Local Directory"
    priority="1" enable="1" />
        <item id_name="calllog_search" display_name="History" priority="2"
        enable="1" />
        <item id_name="remote_directory_search" display_name="Remote Phone
        Book" priority="3" enable="0" />
        <item id_name="ldap_search" display_name="LDAP" priority="4" enable="0" />
        </root_super_search>
```

Upgrading Firmware

Yealink IP phones allow you to upgrade firmware manually via web user interface, or upgrade firmware in batch via auto provisioning.

The following table lists the firmware name for each phone model (X is replaced by the actual firmware version):

Phone Model	Firmware Name
SIP-T28P	2.x.x.rom

Phone Model	Firmware Name
SIP-T26P	6.x.x.rom
SIP-T22P	7.x.x.rom
SIP-T21P	34.x.x.x.rom
SIP-T20P	9.x.x.x.rom
SIP-T19P	31.x.x.x.rom
SIP-T48G	35.x.x.x.rom
SIP-T46G	28.x.x.x.rom
SIP-T42G	29.x.x.x.rom
SIP-T41P	36.x.x.x.rom

To upgrade the phones' firmware in batch via auto provisioning, ask the distributor for the firmware file, upload it to the provisioning server, and then specify the access URL in configuration files.

#######	<i>\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\\</i>	#####
##	Configure the access URL of the firmware file	##
#######	+++++++++++++++++++++++++++++++++++++++	#####

firmware.url =

For example, enter "tftp://admin:password@192.168.1.100/2.72.0.1.rom" ("admin" is replaced by the authentication user name and "password" is replaced by the authentication password) in the "firmware.url =" field. During the auto provisioning process, the phone connects to the provisioning server "192.168.1.100", and downloads the firmware file "2.72.0.1.rom".

Configuring a TFTP Server

Yealink IP phones support using FTP, TFTP, HTTP and HTTPS protocols to download configuration files. You can use one of these protocols for provisioning. The TFTP protocol is used by default. The following section provides instructions on how to configure a TFTP server.

We recommend that you use 3CDaemon or TFTPD32 as a TFTP server. 3CDaemo and TFTPD32 are free applications for Windows. You can download 3CDaemon online: http://www.oldversion.com/3Com-Daemon.html and TFTPD32 online: http://tftpd32.jounin.net/.

For more information on how to configure FTP and HTTP servers, refer to Configuring an FTP server on page 49 and Configuring an HTTP Server on page 52.

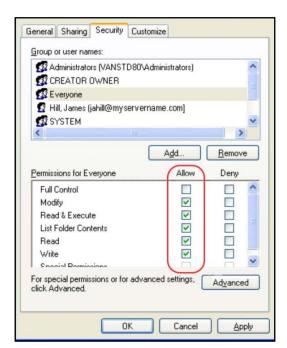
Preparing a Root Directory

To prepare a root directory:

- 1. Create a TFTP root directory on the local system.
- 2. Place configuration files to this root directory.
- 3. Set security permissions for the TFTP directory folder.

You need to define a user or a group name, and set the permissions: read, write or modify. Security permissions vary by organizations.

An example of configuration on the Windows platform is shown as below:

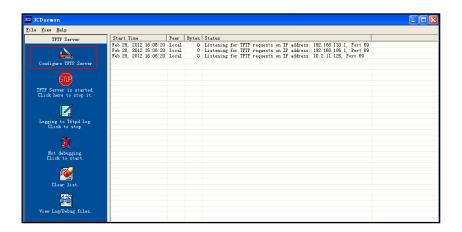


Configuring a TFTP Server

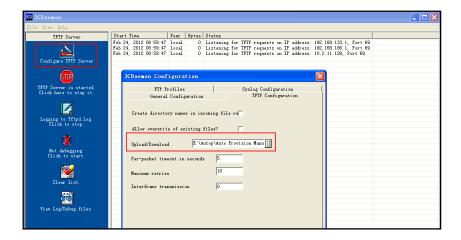
If you have a 3CDaemon application installed on your local system, use it directly. Otherwise, download and install it.

To configure a TFTP server:

1. Double click 3CDaemon.exe to start the application. A configuration page is shown as below:



2. Select **Configure TFTP Server**. Click the button to locate the TFTP root directory from your local system:



3. Click the Confirm button to finish configuring the TFTP server.

The server URL "tftp://IP/" (Here "IP" means the IP address of the provisioning server, for example, "tftp://192.168.1.100/") is where the phone downloads configuration files from.

Obtaining the Address of Provisioning Server

Yealink IP phones support obtaining the provisioning server address in the following ways:

- Zero Touch
- Plug and Play (PnP) Server
- DHCP Options
- Phone Flash

The priority of obtaining the provisioning server address is as follows: Zero Touch-->PnP Server-->DHCP Options (Custom option-->option 66-->option 43) -->Phone Flash.

The following sections detail the process of each way (take the SIP-T28P IP phone as an example).

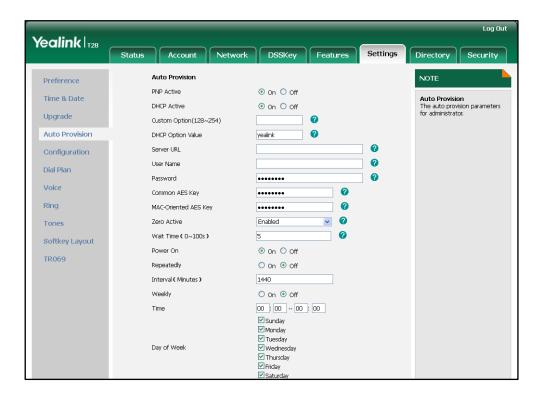
Zero Touch

Zero Touch allows you to configure the network parameters and provisioning server address via phone user interface during startup. This feature is helpful when there is a system failure on the phone. To use Zero Touch, make sure this feature is enabled.

To configure the Zero Touch via web user interface:

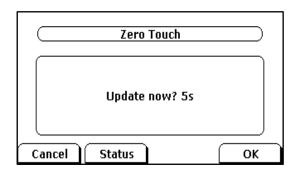
- 1. Click on Settings->Auto Provision.
- 2. Select **Enabled** from the pull-down list of **Zero Active**.

3. Configure the wait time in the Wait Time (0~100s) field.



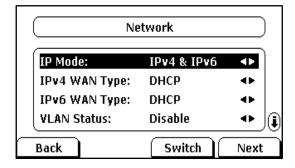
4. Click Confirm to accept the change.

When Zero Touch is enabled, there will be a configuration wizard during startup:



Press the **OK** soft key.

The network parameters are configurable via phone user interface:

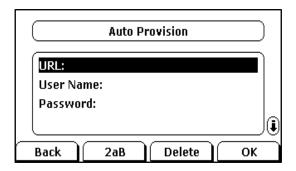


Press the **Next** soft key after finishing network setting.

Configure the provisioning server address, authentication user name (optional) and

password (optional) in the Auto Provision screen.

An example of screenshot is shown as below:

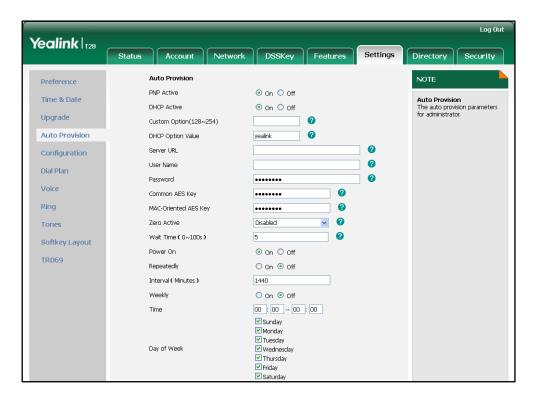


Plug and Play (PnP) Server

Yealink IP phones support obtaining the provisioning server address from the PnP server. The phone broadcasts the PnP SUBSCRIBE message to obtain the provisioning server address during startup. To use Plug and Play, make sure this feature is enabled.

To configure PnP via web user interface:

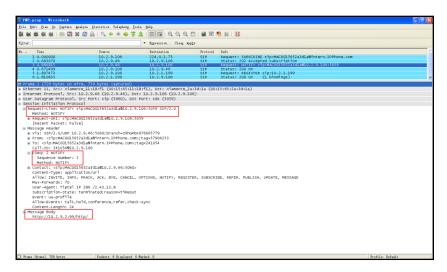
- 1. Click on Settings->Auto Provision.
- 2. Mark the On radio box in the PNP Active field.



3. Click Confirm to accept the change.

Any PnP server activated in the network responses with a **SIP NOTIFY** message, and an address of the provisioning server is contained in the message body. Then the phone

can connect to the provisioning server and perform the auto provisioning process.



DHCP Options

Yealink IP phones support obtaining the provisioning server address from DHCP options. You can configure the phone to obtain the provisioning server address from a custom DHCP option, or the phone will automatically detect the Option 66 and Option 43. The Option 66 is used to identify the TFTP server. To obtain the provisioning server address by a custom DHCP option, make sure the DHCP option is set properly.

The custom DHCP option must be in accordance with the one defined in the DHCP server. For more information on configuring a DHCP server, refer to Configuring a DHCP server on page 55.

To configure the DHCP option via web user interface:

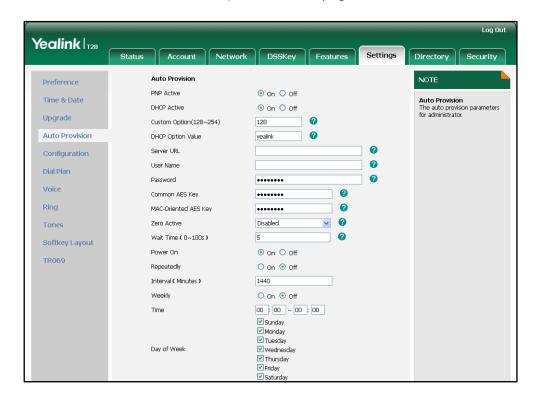
- 1. Click on Settings->Auto Provision.
- 2. Mark the On radio box in the DHCP Active field.
- 3. Enter the desired value in the Custom Option (128~254) field.

4. Enter the desired value in the **DHCP Option Value** field.

The default value is yealink.

5. Configure the desired update mode.

For more information, refer to Update Mode on page 35.



6. Click Confirm to accept the change.

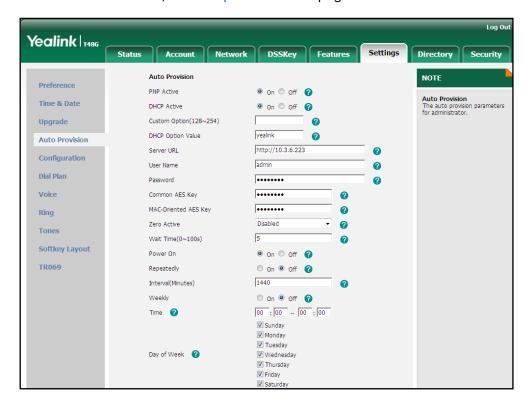
Phone Flash

Yealink IP phones support obtaining the provisioning server address from the phone flash. To obtain the provisioning server address by reading the phone flash, make sure the configuration is set properly.

To configure the Phone Flash via web user interface:

1. Click on **Settings**->**Auto Provision**.

- Enter the URL, user name and password of the provisioning server in the Server
 URL, User Name and Password fields (the user name and password are optional).
- Configure the desired update mode.
 For more information, refer to Update Mode on page 35.



4. Click Confirm to accept the change.

Update Mode

The update mode is used to set the desired time for the phone to perform the auto provisioning process. This chapter introduces the following update modes in detail:

- Power On
- Repeatedly
- Weekly
- Auto Provision Now
- Multi-mode Mixed
- SIP NOTIFY Message

When there is an active call on the phone during provisioning, the auto provisioning process will detect the call status every 30 seconds. If the call is released within 2 hours, the auto provisioning process will be performed normally. Otherwise, the process will be completed, due to timeout.

Power On

The phone performs the auto provisioning process when the phone is powered on.

To activate the Power On mode via a web user interface:

1. Click on **Settings**->**Auto Provision**.

Yealink | 128 Status Network DSSKey **Auto Provision** NOTE Preference PNP Active ● On ○ Off ? Time & Date ● On ○ Off ? DHCP Active Upgrade Custom Option(128~254) yealink DHCP Option Value Auto Provision http://10.3.6.223/ Configuration User Name 0 Voice Common AES Key ••••• MAC-Oriented AES Key ••••• Ring Zero Active Disabled Tones Wait Time(1~100s) Softkey Layout Power On ● On ○ Off ? Repeatedly On Off ? 1440 Interval(Minutes) ○ On ● Off ? Weekly 00 : 00 -- 00 : 00 ✓ Sunday ✓ Monday Tuesday Day of Week 🕜 ✓ Wednesday ✓ Friday✓ Saturday Autoprovision Now ?

2. Mark the On radio box in the Power On field.

3. Click **Confirm** to accept the change.

Repeatedly

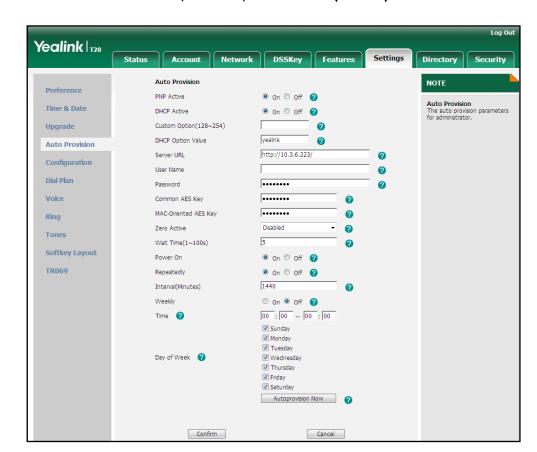
The phone performs the auto provisioning process at regular intervals. You can configure the interval for the Repeatedly mode. The default interval is 1440 minutes.

Cancel

To activate the Repeatedly mode via web user interface:

Confirm

- 1. Click on Settings->Auto Provision.
- 2. Mark the On radio box in the Repeatedly field.



3. Enter the interval time (in minutes) in the Interval (Minutes) field.

Click Confirm to accept the change.

Weekly

The phone performs the auto provisioning process at the fixed time every week. You can configure what time of the day and which day of the week to trigger the phone to perform the auto provisioning process. For example, you can configure the phone to check and update new configuration between 2 to 3 o'clock every Friday and Sunday.

To activate the Weekly mode via web user interface:

- 1. Click on Settings->Auto Provision.
- 2. Mark the On radio box in the Weekly field.
- 3. Enter the desired time in the **Time** field.

Yealink | 128 Status Directory Security **Auto Provision** NOTE Preference PNP Active On Off Time & Date DHCP Active ● On ○ Off ? Upgrade Custom Option(128~254) yealink DHCP Option Value a Auto Provision http://10.3.6.223/ n Configuration User Name 0 Voice Common AES Key ••••• MAC-Oriented AES Key ••••• Ring Zero Active Disabled 0 Tones 5 Wait Time(1~100s) Softkey Layout Power On On ○ Off ? TR069 Repeatedly On Off ? 1440 Interval(Minutes) Weekly ● On ○ Off ? 00 : 00 -- 00 : 00 Time 🕜 ✓ Sunday Monday Tuesday Day of Week 🕜 ✓ Wednesday Thursday ✓ Friday✓ Saturday Autoprovision Now ?

4. Mark one or more checkboxes in the **Day of Week** field.

5. Click **Confirm** to accept the change.

Auto Provision Now

You can use Auto Provision Now mode to manually trigger the phone to perform the auto provisioning process immediately.

Cancel

To use the Auto Provision Now mode via web user interface:

Confirm

1. Click on Settings->Auto Provision.

Yealink 128 Network DSSKey Features Settings Status Directory PNP Active On ○ Off ? Auto Provision The auto provision parameters for administrator. Time & Date DHCP Active On ○ Off ? Upgrade Custom Option(128~254) DHCP Option Value **Auto Provision** http://10.3.6.223/ Configuration User Name a Dial Plan Common AES Key Tones **Softkey Layout** Power On ● On ○ Off ? TR069 Repeatedly ⊚ on ⊚ off ? Interval(Minutes) 1440 Weekly ⊙ On ⊚ Off ? 00 : 00 -- 00 : 00 Time 🕜 Sunday ✓ Monday Day of Week ▼ Wednesday Thursday Friday ✓ Saturday Autoprovision Now ?

2. Click Autoprovision Now.

The phone will perform the auto provisioning process immediately.

Confirm

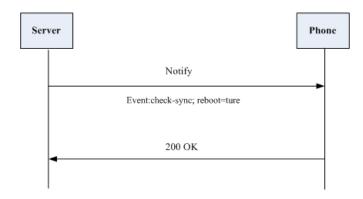
Multi-mode Mixed

You can activate more than one update mode for auto provisioning. For example, you can activate the "Power On" and "Repeatedly" modes simultaneously. The phone will perform the auto provisioning process when it is powered on and at a specified interval.

SIP NOTIFY Message

The phone will perform the auto provisioning process when receiving a SIP NOTIFY message which contains the header "Event: check-sync". If the header of the SIP NOTIFY message contains an additional string "reboot=true", the phone will reboot immediately and then perform the auto provisioning process. This update mode requires server support.

The following figure shows the message flow:



Downloading and Verifying Configurations

Downloading Configuration Files

Once obtaining a provisioning server address in one of the ways introduced above, the phone will connect to the provisioning server and download configuration files. During the auto provisioning process, the phone will try to download the Common CFG file firstly, and then try to download the MAC-Oriented CFG file from the provisioning server. If resource files need to be updated and the access URLs have been specified in configuration files, the phone will then try to download and update the resource files.

Resolving and Updating Configurations

After downloading, the phone resolves the configuration files, downloads the resource files requested in the configuration files, and then updates the configurations and resource files to the phone flash. Generally, updated configurations will automatically take effect after the auto provisioning process is completed. For update of some specific configurations which require a reboot before taking effect, for example, network configurations, the phone will reboot to make the configurations effective after the auto provisioning process is completed.

The phone calculates the MD5 values of the downloaded files. If the MD5 values of the Common and MAC-Oriented configuration files are the same as those of the last downloaded configuration files, this means these two configuration files on the provisioning server are not changed. The phone will complete the auto provisioning without repeated update. This is used to avoid unnecessary restart and impact of phone use.

If configuration files have been AES-encrypted, the phone will decrypt them after downloading the configuration files. For more information on how the phone decrypts configuration files, refer to *Yealink Configuration Encryption Tool User Guide*.

The phone only reboots when there is at least a specific configuration requiring a reboot during auto provisioning.

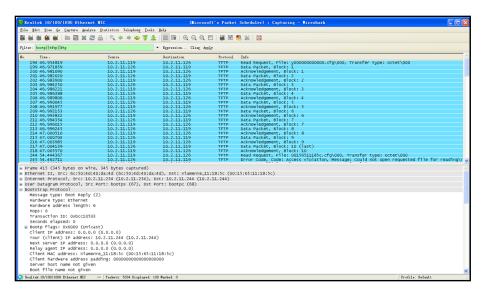
For more information on the specific configurations which require a reboot during auto provisioning, refer to Description of Configuration Parameters in CFG Files on page 65.

Verifying Configurations

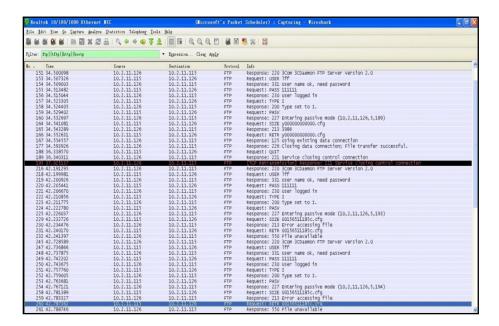
After auto provisioning, you can then verify the update via phone user interface, or you can verify it via web user interface of the phone. For more information, refer to Yealink phone-specific user guide.

During the auto provisioning process, you can monitor the downloading requests and response messages by a WinPcap tool. The following shows some examples.

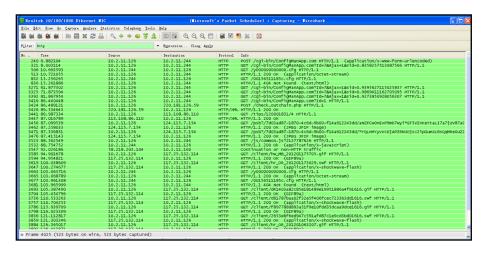
Example1: Yealink SIP-T28P IP phone downloads configuration files from the TFTP server.



Example 2: Yealink SIP-T28P IP phone downloads configuration files from the FTP server.



Example 3: Yealink SIP-T28P IP phone downloads configuration files from the HTTP server.



Troubleshooting

This chapter provides general troubleshooting information to help you solve problems you might encounter when deploying phones.

If you require additional information or assistance with the deployment, contact your system administrator.

Why does the phone fail to download configuration files?

- Ensure that auto provisioning feature is configured properly.
- Ensure that the provisioning server and network are reachable.
- Ensure that authentication credentials configured on the phone are correct.
- Ensure that configuration files exist on the provisioning server.

Why does the provisioning server return HTTP 404?

- Ensure that the provisioning server is properly set up.
- Ensure that the access URL is correct.
- Ensure that the requested files exist on the provisioning server.

Why does the phone display "Network Unavailable"?

- Ensure that the Ethernet cable is plugged into the Internet port on the phone and the Ethernet cable is not loose.
- Ensure that the switch or hub in your network is operational.
- Ensure that the configurations of network are properly set in the configuration files.

Why is the permission denied when uploading files to the root directory of the FTP server?

- Ensure that the complete path to the root directory of the FTP server is authorized.
- Check security permissions on the root directory of the FTP server, if necessary, change the permissions.

Why doesn't the phone obtain the IP address from the DHCP server?

- Ensure that settings are correct on the DHCP server.
- Ensure that the phone is configured to obtain the IP address from the DHCP server.

Why doesn't the phone download the ring tone?

- Ensure that the file format of the ring tone is *.wav.
- Ensure that the size of the ring tone file is no larger than that the phone supports.
- Ensure that the properties of the ring tone for the phone are correct.
- Ensure that the network is available and the root directory is right for downloading.
- Ensure that the ring tone file exists on the provisioning server.

Why doesn't the phone update configurations?

- Ensure that the configuration files are different from the last ones.
- Ensure that the phone has downloaded the configuration files.
- Ensure that the parameters are correctly set in the configuration files.

Glossary

MAC Address: A Media Access Control address (MAC address) is a unique identifier assigned to network interfaces for communications on the physical network segment.

MD5: The MD5 Message-Digest Algorithm is a widely used cryptographic hash function that produces a 128-bit (16-byte) hash value.

DHCP: Dynamic Host Configuration Protocol (DHCP) is a network configuration protocol for hosts on Internet Protocol (IP) networks. Computers that are connected to IP networks must be configured before they can communicate with other hosts.

FTP: File Transfer Protocol (FTP) is a standard network protocol used to transfer files from one host to another host over a TCP-based network, such as the Internet. It is often used to upload web pages and other documents from a private development machine to a public web-hosting server.

HTTP: The Hypertext Transfer Protocol (HTTP) is an application protocol for distributed, collaborative, hypermedia information systems. HTTP is the foundation of data communication for the World Wide Web.

HTTPS: Hypertext Transfer Protocol Secure (HTTPS) is a combination of Hypertext Transfer Protocol (HTTP) with SSL/TLS protocol. It provides encrypted communication and secure identification of a network web server.

TFTP: Trivial File Transfer Protocol (TFTP) is a simple protocol to transfer files. It has been implemented on top of the User Datagram Protocol (UDP) using port number 69.

AES: Advanced Encryption Standard (AES) is a specification for the encryption of electronic data.

URL: A uniform resource locator or universal resource locator (URL) is a specific character string that constitutes a reference to an Internet resource.

XML: Extensible Markup Language (XML) is a markup language that defines a set of rules for encoding documents in a format that is both human-readable and machine-readable.

Appendix

Configuring an FTP Server

This section provides instructions on how to configure an FTP server using 3CDaemon. You can download the 3CDaemon software online:

http://www.oldversion.com/3Com-Daemon.html.

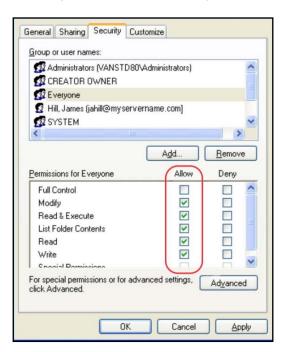
Preparing a Root Directory

To prepare a root directory:

- 1. Create an FTP root directory on the local system.
- 2. Place the configuration files to this root directory.
- 3. Set the security permissions for the FTP directory folder.

You need to define a user or group name, and set the permissions: read, write, and modify. Security permissions vary by organizations.

An example of configuration on the Windows platform is shown as below:



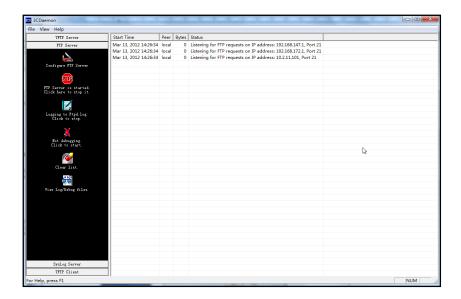
Configuring an FTP server

If you have a 3CDaemon application installed on your local system, use it directly. Otherwise, download and install it.

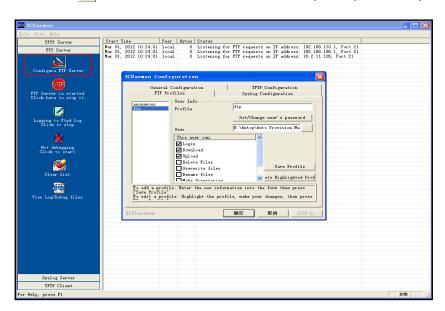
To configure an FTP server:

- 1. Double click the 3CDaemon.exe to start the application.
- 2. Click the FTP Server button on the left of the main page.

A configuration page is shown as below:



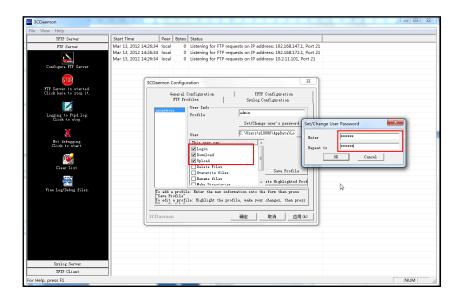
- 3. Select Configure FTP Server.
- **4.** Click the button to locate the FTP root directory from your local system:



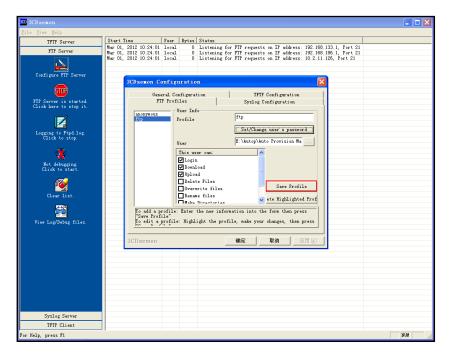
- 5. Enter the new authentication user name in the **Profile** field.
- 6. Click the Set/Change user's password button to set the password in the pop-up

dialogue box.

- 7. Click the **OK** button to save.
- Mark the check boxes of Login, Download and Upload to make sure the FTP user has the login, download and upload permission.



9. Click the **Save Profile** button to save the settings and finish the configurations.



10. Click the Confirm button to finish configuring the FTP server.

The server URL "ftp://username:password@IP/" (Here "IP" means the IP address of the provisioning server, "username" and "password" are the authentication for FTP download. For example, "ftp://admin:123456@192.168.1.100/") is where the phone downloads configuration files from.

Configuring an HTTP Server

This section provides instructions on how to configure an HTTP server using HFS tool. You can download the HFS software online: http://www.snapfiles.com/get/hfs.html.

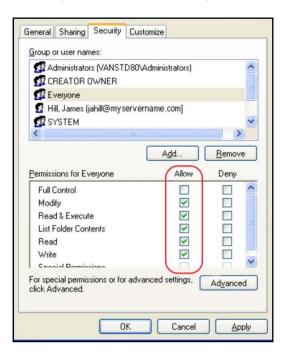
Preparing a Root Directory

To prepare a root directory:

- 1. Create an HTTP root directory on the local system.
- 2. Place configuration files to this root directory.
- **3.** Set the security permissions for the HTTP directory folder.

You need to define a user or group name and set the permissions: read, write, and modify. Security permissions vary by organizations.

An example of configuration on the Windows platform is shown as below:



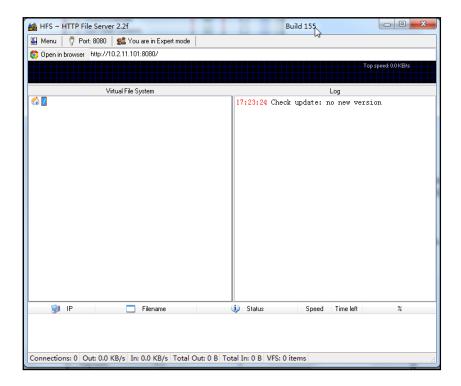
Configuring an HTTP Server

HFS tool is an executable application, so you don't need to install it.

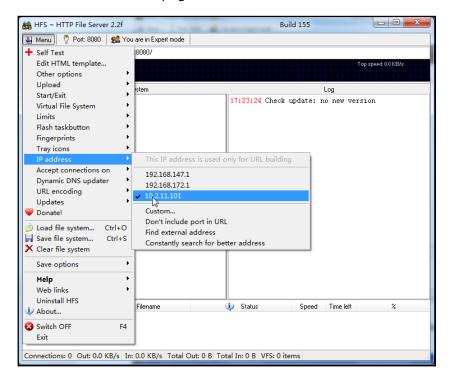
To configure an HTTP server:

1. Download the application file to your local directory, double click the hfs.exe.

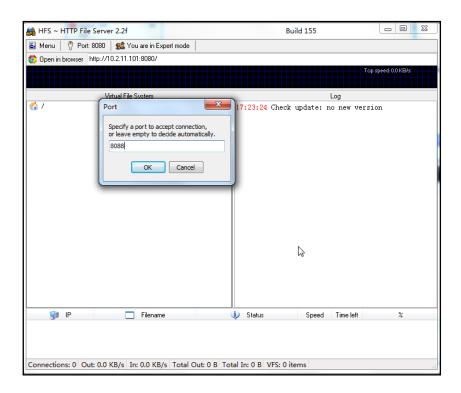
The main configuration page is shown as below:



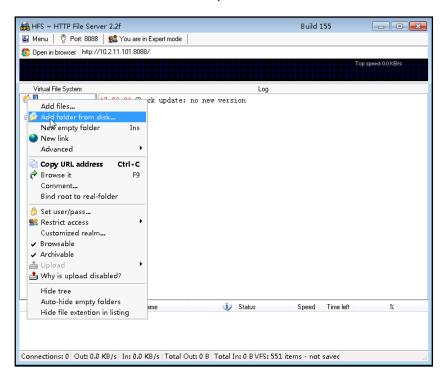
2. Click Menu in the main page and select the IP address of the PC from IP address.



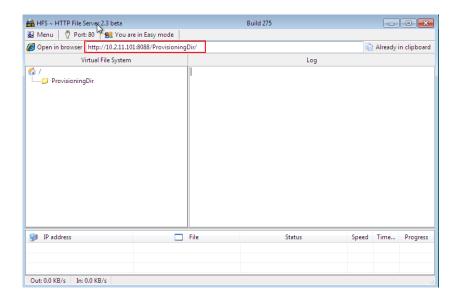
The default HTTP port is 8080. You can also reset the HTTP port (make sure there is no port conflict).



3. Right click the icon on the left of the main page, select Add folder from disk to add the HTTP Server root directory.



4. Locate the root directory from your local system.



5. Check the server URL (e.g., http:// 10.2.11.101:8088/ProvisioningDir/) by clicking "Open in browser".

Yealink IP phones also support the Hypertext Transfer Protocol with SSL/TLS (HTTPS) protocol for auto provisioning. HTTPS protocol provides the encrypted communication and secure identification. For more information on installing and configuring an Apache HTTPS Server, refer to the network resource.

Configuring a DHCP server

This section provides instructions on how to configure a DHCP server for Windows using DHCP Turbo. You can download this software online:

http://www.tucows.com/preview/265297 and install it following the setup wizard.

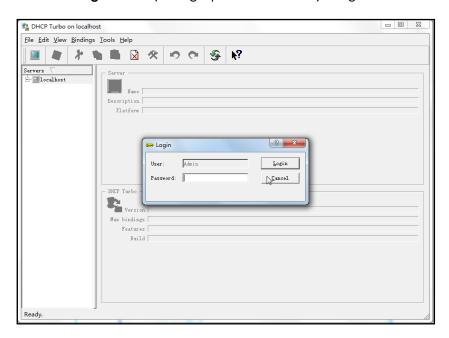
Before configuring the DHCP Turbo, make sure:

- The firewall on the PC is disabled.
- There is no DHCP server in your local system.

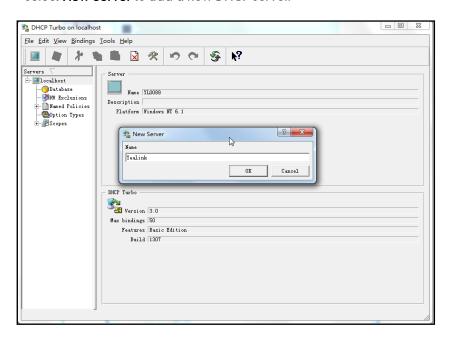
To configure the DHCP Turbo:

1. To start the DHCP Turbo application, double click localhost.

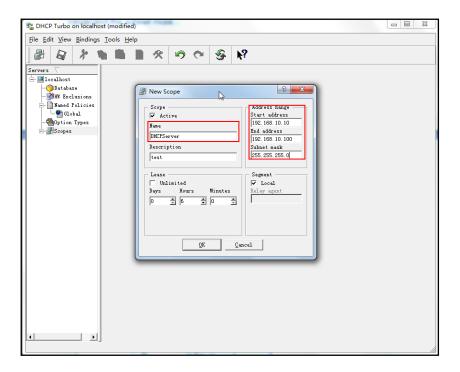
2. Click the **Login** button (the login password is blank) to log in.



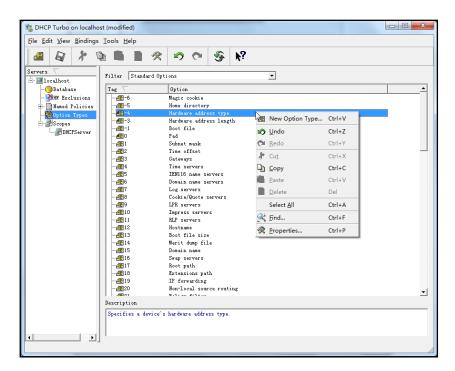
3. You can then edit the existing DHCP server, or you can right click **localhost** and select **New Server** to add a new DHCP server.



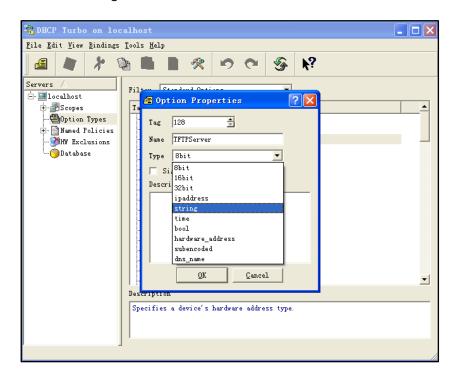
- 4. Right click **Scopes** and select **New Scope**.
- 5. Configure the DHCP server name, the DHCP IP range and the subnet mask.
- 6. Click **OK** to accept the change.



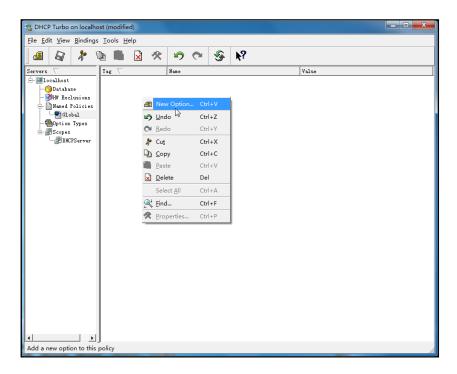
7. You can add a custom option via DHCP Turbo. Select **Option Types**, right click one of the options on the right of the main page, and then select **New Option Type**.

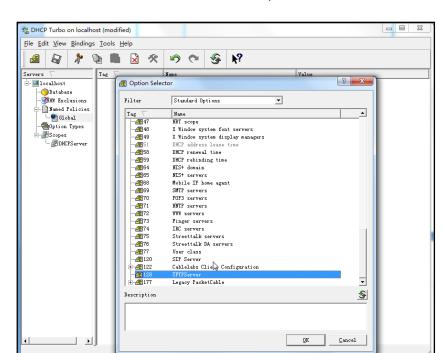


8. Set the custom DHCP option (custom DHCP option tag number ranges from 128 to 254) and select the option type (Yealink supports **String** and **IP Address** option types only). Click the **OK** button to finish setting the option properties. Click save the change.



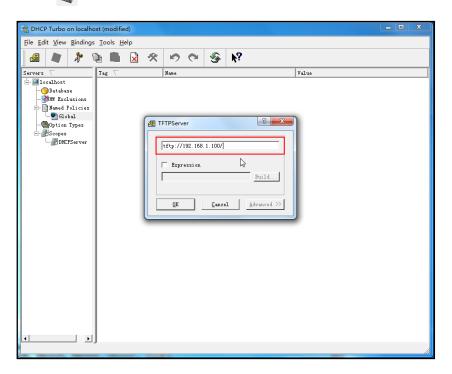
9. Click **Named Policies**-->**Global**, right click the blank area on the right of the main page and then select **New Option**.





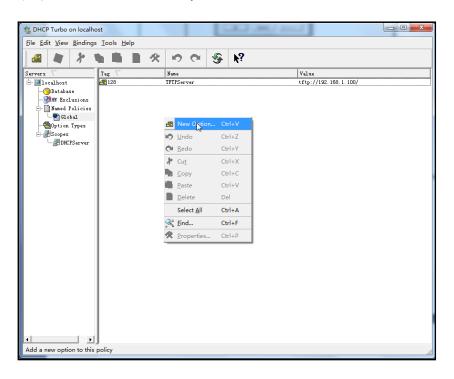
10. Scroll down and double click the custom option 128.

- 11. Fill the provisioning server address in the input field.
- 12. Click the \mathbf{OK} button to finish setting a custom option.
- 13. Click 🔊 to save the change.

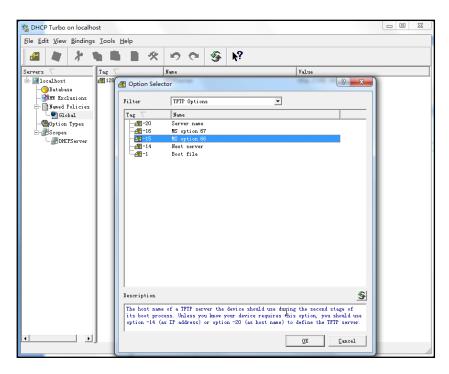


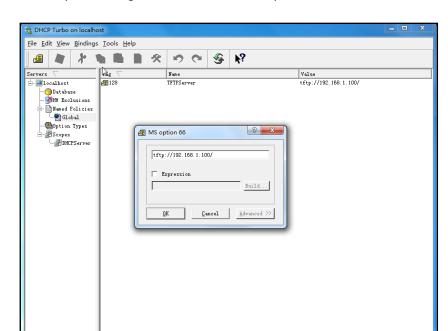
You can add the option 66 via DHCP Turbo. The following shows the detailed processes.

1. Click **Named Policies**-->**Global**, right click the blank area on the right of the main page and then select **New Option**.



- 2. Select TFTP Options from the pull-down list of Filter.
- 3. Scroll down and double click MS option 66.



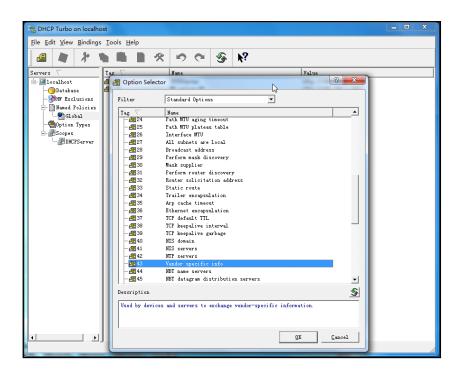


4. Fill the provisioning server address in the input field.

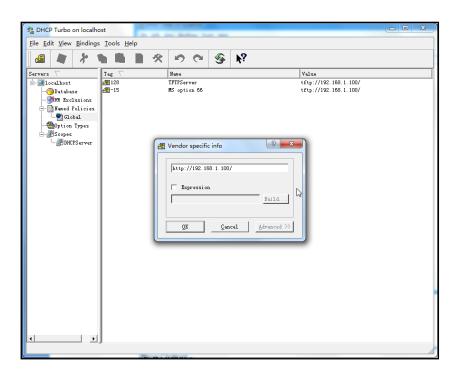
- 5. Click the **OK** button to finish setting a custom option.
- 6. Click 🔝 to save the change.

You also can add the option 43. The following shows the detailed processes.

- 1. Click **Named Policies**-->**Global**, right click the blank area on the right of the main page and then select **New Option**.
- 2. Select the Standard Options from the pull-down list of Filter.
- 3. Scroll down and double click 43.



4. Fill the provisioning server address in the input field.



- 5. Click the \mathbf{OK} button to finish setting a custom option.
- 6. Click 📓 to save the change.

Customizing a Ring Tone Using Cool Edit Pro

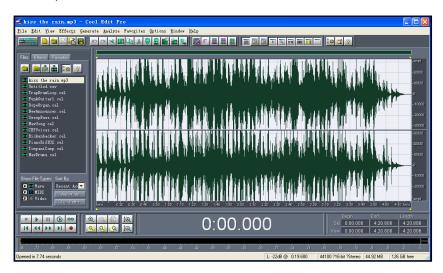
If you have installed the Cool Edit application, double click to open it. Otherwise, you can download the installation package online:

http://www.toggle.com/lv/group/view/kl36218/Cool_Edit_Pro.htm and install it.

To customize a ring tone using Cool Edit Pro:

- 1. Open the Cool Edit Pro application.
- 2. Click File to open an audio file.
- **3.** Locate the ring tone file, click **Open**, the file is uploaded as follows.

A sample audio file loaded is shown as below:



- 4. Select and copy the audio waveform.
- Select File->New to create a new file, set the channels as Mono, the sample rate as 8000 and the resolution as 16-bit.
- 6. Paste the audio waveform to the new file.



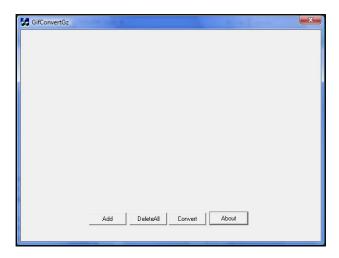
7. Select File->Save as to save the new audio file. On the Save waveform page,

select the file format as A/mu-law wave.

Customizing a Logo File Using PictureExDemo

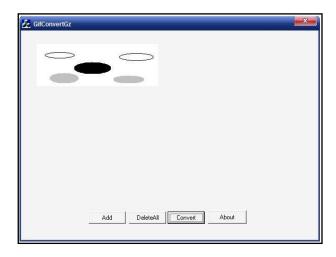
The original picture format must be *.bmp or *.gif. We recommend placing all files and the PictureExDemo application to the root directory of the PC.

1. Double click the PictureExDemo.exe.



Click Add button to open a *.bmp or *.gif file.
 You can repeat the second step to add multiple original picture files.

3. Click the Convert button.



Then you can find the **DOB** logo files in the **adv** directory.

Description of Configuration Parameters in CFG Files

If you want to reset the configuration of a parameter, set the value of the parameter to !NULL! or %NULL%. For example, local_time.ntp_server1 = %NULL%. After the auto provisioning process is completed, the NTP server 1 will be reset to "cn.pool.ntp.org".

Parameter	Permitted Values	Descriptions	Web Setting Path
network.ip_ad dress_mode =	0, 1 or 2	It configures the IP address mode. 0-IPv4 1-IPv6 2-IPv4&IPv6 The default value is 0. It takes effect after a reboot.	Network->Basic-> Internet Port-> Mode (IPv4/IPv6)
network.intern et_port.type =	0, 1 or 2	It configures the Internet (WAN) port type for IPv4 when the IP address mode is configured as IPv4 or IPv4&IPv6. 0-DHCP 1-PPPoE (not applicable to SIPT42G/T41P IP phones) 2-Static IP Address The default value is 0. It takes effect after a reboot.	Network->Basic-> IPv4 Config
network.static _dns_enable =	0 or 1	It enables or disables the phone to use manually configured static IPv4 DNS when Internet (WAN) port type for IPv4 is configured as DHCP. O-Disabled (use the IPv4 DNS obtained by DHCP) 1-Enabled The default value is 0. It takes effect after a reboot.	Network->Basic-> IPv4 Config->Static DNS
network.intern et_port.ip =	IPv4 address	It configures the IPv4 address when the IP address mode is configured as IPv4 or IPv4&IPv6, and the Internet (WAN) port type for IPv4 is configured as Static IP Address. The default value is blank.	Network->Basic-> IPv4 Config->Static IP Address->IP Address

Parameter	Permitted Values	Descriptions	Web Setting Path
		It takes effect after a reboot.	
network.intern et_port.mask =	Subnet Mask	It configures the IPv4 subnet mask when the IP address mode is configured as IPv4 or IPv4&IPv6, and the Internet (WAN) port type for IPv4 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv4 Config->Static IP Address-> Subnet Mask
network.intern et_port.gatew ay =	IPv4 address	It configures the IPv4 default gateway when the IP address mode is configured as IPv4 or IPv4&IPv6, and the Internet (WAN) port type for IPv4 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv4 Config->Static IP Address-> Gateway
network.prima ry_dns =	IPv4 address	It configures the primary IPv4 DNS server when the IP address mode is configured as IPv4 or IPv4&IPv6, and the Internet (WAN) port type for IPv4 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv4 Config->Static IP Address-> Primary DNS
network.secon dary_dns =	IPv4 address	It configures the secondary IPv4 DNS server when the IP address mode is configured as IPv4 or IPv4&IPv6, and the Internet (WAN) port type for IPv4 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv4 Config->Static IP Address-> Secondary DNS
network.pppo e.user = (not applicable to SIPT42G/T41P IP phones)	String within 32 characters	It configures the user name for PPPoE connection. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv4 Config-> PPPoE->User Name
network.pppo e.password =	String within 99	It configures the password for PPPoE connection.	Network->Basic-> IPv4 Config->

Parameter	Permitted Values	Descriptions	Web Setting Path
(not applicable to SIPT42G/T41P IP phones)	characters	The default value is blank. It takes effect after a reboot.	PPPoE->Password
network.ipv6_i nternet_port.ty pe =	0 or 1	It configures the Internet (WAN) port type for IPv6 when the IP address mode is configured as IPv6 or IPv4&IPv6. 0-DHCP 1-Static IP Address The default value is 0. It takes effect after a reboot.	Network->Basic-> IPv6 Config
network.ipv6_ static_dns_en able =	0 or 1	It enables or disables the phone to use manually configured static IPv6 DNS when Internet (WAN) port type for IPv6 is configured as DHCP. O-Disabled (use the IPv6 DNS obtained by DHCP) 1-Enabled The default value is 0. It takes effect after a reboot.	Network->Basic-> IPv6 Config->IPv6 Static DNS
network.ipv6_ prefix =	Integer from 0 to 128	It configures the IPv6 prefix when the IP address mode is configured as IPv6 or IPv4&IPv6, and the Internet (WAN) port type for IPv6 is configured as Static IP Address. The default value is 64. It takes effect after a reboot.	Network->Basic-> IPv6 Config->Static IP Address->IPv6 Prefix (0~128)
network.ipv6_i nternet_port.i p =	IPv6 address	It configures the IPv6 address when the IP address mode is configured as IPv6 or IPv4&IPv6, and the Internet (WAN) port type for IPv6 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv6 Config->Static IP Address->IP Address
network.ipv6_i nternet_port.g	IPv6 address	It configures the IPv6 default gateway when the IP address mode is	Network->Basic-> IPv6 Config->Static

Parameter	Permitted Values	Descriptions	Web Setting Path
ateway =		configured as IPv6 or IPv4&IPv6, and the Internet (WAN) port type for IPv6 is configured as Static IP Address.	IP Address-> Gateway
		The default value is blank. It takes effect after a reboot.	
network.ipv6_ primary_dns =	IPv6 address	It configures the primary IPv6 DNS server when the IP address mode is configured as IPv6 or IPv4&IPv6, and the Internet (WAN) port type for IPv6 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network->Basic-> IPv6 Config->Static IP Address-> Primary DNS
network.ipv6_ secondary_dn s =	IPv6 address	It configures the secondary IPv6 DNS server when the IP address mode is configured as IPv6 or IPv4&IPv6, and the Internet (WAN) port type for IPv6 is configured as Static IP Address. The default value is blank. It takes effect after a reboot.	Network-> Basic-> IPv6 Config->Static IP Address-> Secondary DNS
network.ipv6_i cmp_v6.enabl e = (only applicable to SIPT48G/T46G IP phones)	0 or 1	It enables or disables the phone to obtain IPv6 network settings by ICMPv6. 0-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	Network->Advanc ed->ICMPv6 Status->Active
network.bridg e_mode = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	0 or 1	It configures the PC (LAN) port type. 0-Router 1-Bridge The default value is 1. It takes effect after a reboot. Note: SIP-T19P/T21P/T48G/T46G/ T42G/T41P/ IP phones only support bridge mode for PC connection.	Network->PC Port ->PC Port Config
network.pc_p ort.enable =	0 or 1	It enables or disables the PC port. 0 -Disabled	Network->PC Port ->PC Port Active

Parameter	Permitted Values	Descriptions	Web Setting Path
		1-Auto Negotiation The default value is 1. It takes effect after a reboot.	
network.pc_p ort.ip = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	IP address	It configures the IP address of the PC (LAN) port when the PC (LAN) port is configured as Router. The default value is 10.0.0.1. It takes effect after a reboot.	Network->PC Port ->PC Port Config ->As Router->IP Address
network.pc_p ort.mask = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	IP address	It configures the mask of the PC (LAN) port when the PC (LAN) port is configured as Router. The default value is 255.255.255.0. It takes effect after a reboot.	Network->PC Port ->PC Port Config ->As Router ->Subnet Mask
network.pc_p ort.dhcp_serv er = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	0 or 1	It enables or disables the phone to act as a DHCP server when the PC (LAN) port is configured as Router. 0-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	Network->PC Port ->PC Port Config ->As Router ->Enable DHCP Server
network.dhcp. start_ip = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP	IP address	It configures the start IP address of the DHCP IP segment. The default value is 10.0.0.10. It takes effect after a reboot.	Network->PC Port ->PC Port Config ->As Router->Start IP Address

Parameter	Permitted Values	Descriptions	Web Setting Path
phones)			
network.dhcp. end_ip = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	IP address	It configures the end IP address of the DHCP IP segment. The default value is 10.0.0.100. It takes effect after a reboot.	Network->PC Port ->PC Port Config ->As Router->End IP Address
network.intern et_port.speed _duplex =	0, 1, 2, 3, 4 or 5	It configures the transmission mode and speed of the Internet (WAN) port. 0-Auto negotiate 1-Full duplex 10Mbps 2-Full duplex 100Mbps 3-Half duplex 10Mbps 4-Half duplex 100Mbps 5-Full duplex 1000Mbps (only applicable to SIPT48G, SIPT46G and SIPT42G IP phones) The default value is 0. It takes effect after a reboot.	Network-> Advanced->Port Link-> WAN Port Link
network.pc_p ort.speed_dup lex =	0, 1, 2, 3, 4 or 5	It configures the transmission mode and speed of the PC (LAN) port. 0-Auto negotiate 1-Full duplex 10Mbps 2-Full duplex 10Mbps 3-Half duplex 10Mbps 4-Half duplex 100Mbps 5-Full duplex 1000Mbps (only applicable to SIPT48G, T46G and T42G IP phones) The default value is 0. It takes effect after a reboot.	Network-> Advanced->Port Link->PC Port Link
network.vlan.i nternet_port_e	0 or 1	It enables or disables VLAN of the Internet (WAN) port.	Network-> Advanced->VLAN ->WAN Port->

Parameter	Permitted Values	Descriptions	Web Setting Path
nable =		0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	Active
network.vlan.i nternet_port_v id =	Integer from 1 to 4094	It configures VLAN ID of the Internet (WAN) port. The default value is 1. It takes effect after a reboot.	Network-> Advanced->VLAN ->WAN Port->VID (1-4094)
network.vlan.i nternet_port_p riority =	Integer from 0 to 7	It configures VLAN priority of the Internet (WAN) port. The default value is 0. It takes effect after a reboot.	Network-> Advanced->VLAN ->WAN Port-> Priority
network.vlan. pc_port_enabl e =	0 or 1	It enables or disables VLAN of the PC (LAN) port. 0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	Network-> Advanced->VLAN >PC Port->Active
network.vlan. pc_port_vid =	Integer from 1 to 4094	It configures VLAN ID of the PC (LAN) port. The default value is 1. It takes effect after a reboot.	Network-> Advanced->VLAN >PC Port->VID (1-4094)
network.vlan. pc_port_priorit y =	Integer from 0 to 7	It configures VLAN priority of the PC (LAN) port. The default value is 0. It takes effect after a reboot.	Network-> Advanced->VLAN >PC Port->Priority
network.vlan. pc_port_mode = (only	0 or 1	It configures the way the phone processes packets sent from the PC port when VLAN is enabled on the PC port. 0 -Forward the packets directly	
applicable to SIPT4X IP phones)		1-Tag and then forward the packets if there is no VLAN tag on the packets, otherwise, forward the packets directly.	

Parameter	Permitted Values	Descriptions	Web Setting Path
		The default value is 0.	
		It takes effect after a reboot.	
network.vlan. dhcp_enable =	0 or 1	It enables or disables the phone to obtain VLAN by DHCP. 0-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	Network-> Advanced->VLAN >DHCP VLAN-> Active
network.vlan. dhcp_option =	Integer from 128 to 254	It configures the DHCP option from which the phone will obtain the VLAN settings. You can configure at most five DHCP options and separate them by commas. The default value is 132. It takes effect after a reboot.	Network-> Advanced->VLAN >DHCP VLAN-> Option
network.dhcp _host_name =	String within 99 characters	It configures the client host name for DHCP option 12. For SIPT28P IP phones: The default value is SIPT28P. For SIPT26P IP phones: The default value is SIPT26P. For SIPT22P IP phones: The default value is SIPT22P. For SIPT21P IP phones: The default value is SIPT21P. For SIPT20P IP phones: The default value is SIPT20P. For SIPT19P IP phones: The default value is SIPT19P. For SIPT48G IP phones: The default value is SIPT48G. For SIPT46G IP phones: The default value is SIPT46G. For SIPT42G IP phones: The default value is SIPT46G.	Features->General Information-> DHCP Hostname

Parameter	Permitted Values	Descriptions	Web Setting Path
		For SIPT41P IP phones:	
		The default value is SIP-T41P.	
		It takes effect after a reboot.	
		It enables or disables the HTTP	
		protocol for web server access.	
wui.http_enab		0-Disabled	Network->
le =	0 or 1	1-Enabled	Advanced->Web
		The default value is 1.	Server->HTTP
		It takes effect after a reboot.	
		It enables or disables the HTTPS	
		protocol for web server access.	
wui.https_ena		0 -Disabled	Network->
ble =	0 or 1	1-Enabled	Advanced->Web
		The default value is 1.	Server->HTTPS
		It takes effect after a reboot.	
		It configures the HTTP port for web	Network->
network.port.h	Integer from	server access.	Advanced->Web
ttp =	1 to 65535	The default value is 80.	Server->HTTP Port
		It takes effect after a reboot.	(1~65535)
		It configures the HTTPS port for web	Network->
network.port.h	Integer from	server access.	Advanced->Web
ttps =	1 to 65535	The default value is 443.	Server->HTTPS Port
		It takes effect after a reboot.	(1~65535)
		It configures the maximum local RTP	Network->
network.port.	Integer from 1024 to	port.	Advanced->Local
max_rtpport =	65535	The default value is 11800.	RTP Port-> Max RTP
		It takes effect after a reboot.	Port (1024~65535)
	Into see from	It configures the minimum local RTP	Network->
network.port.	Integer from 1024 to	port.	Advanced->Local
min_rtpport =	65535	The default value is 11780.	RTP Port->Min RTP
		It takes effect after a reboot.	Port (1024~65535)
notwork ass #	Intogor from	It configures the voice QoS.	Network->
network.qos.rt ptos =	Integer from 0 to 63	The default value is 46.	Advanced->Voice
P.00	0.000	It takes effect after a reboot.	QoS (0~63)

Parameter	Permitted Values	Descriptions	Web Setting Path
		It configures the SIP QoS.	Network->
network.qos.si gnaltos =	Integer from 0 to 63	The default value is 26.	Advanced->SIP
griditos –	0 10 00	It takes effect after a reboot.	QoS (0~63)
		It configures the 802.1x mode.	
		0 -Disabled	
		1-EAP-MD5	
network.802_1	0, 1, 2, 3 or 4	2-EAP-TLS	Network-> Advanced->802.1x
x.mode =	0, 1, 2, 3 01 4	3 -PEAP-MSCHAPV2	->802.1x Mode
		4-EAP-TTLS/EAP-MSCHAPv2	
		The default value is 0.	
		It takes effect after a reboot.	
		It configures the user name for 802.1x	
network.802_1	String within 32 characters	authentication.	Network-> Advanced->802.1x ->Identity
x.identity =		The default value is blank.	
		It takes effect after a reboot.	,
	Otalia are italia	It configures the password for 802.1x	Nationalis
network.802_1 x.md5_passw	String within 32	authentication.	Network-> Advanced->802.1x
ord =	characters	The default value is blank.	->MD5 Password
		It takes effect after a reboot.	
		It configures the access URL of the CA	
		certificate when the 802.1x mode is	
network.802_1	URL within	configured as EAP-TLS, PEAP-MSCHAPV2 or	Network->
x.root_cert_url	511 characters	EAP-TTLS/EAP-MSCHAPV2.	Advanced->802.1x ->CA Certificates
=	characters	The default value is blank.	->CA Certificates
		It takes effect after a reboot.	
		It configures the access URL of the	
network.802_1	URI within	device certificate when the 802.1x	Network->
x.client_cert_u	511	mode is configured as EAPTLS.	Advanced->802.1x
rl =	characters	The default value is blank.	->Device
		It takes effect after a reboot.	Certificates
network.vpn_		It enables or disables OpenVPN	
enable =		feature.	Network->
(not	0 or 1	0 -Disabled	Advanced->VPN
applicable to		1-Enabled	->Active

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T19P IP		The default value is 0.	
phones)		It takes effect after a reboot.	
openvpn.url = (not applicable to SIPT19P IP phones)	URL within 511 characters	It configures the access URL of the *.tar file for OpenVPN. The default value is blank.	Network-> Advanced->VPN- >Upload VPN Config
network.lldp.e nable =	0 or 1	It enables or disables LLDP feature. 0-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	Network-> Advanced->LLDP- >Active
network.lldp.p acket_interval =	Integer from 1 to 3600	It configures the interval (in seconds) for the phone to broadcast the LLDP request. The default value is 60. It takes effect after a reboot.	Network-> Advanced->LLDP- >Packet Interval (1~3600s)
network.span_ to_pc_port =	0 or 1	It enables or disables the phone to span data packets received in the WAN port to the PC port. If it is enabled, all packets from WAN port can be received by PC port. 0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	Network-> Advanced->Span to PC->Span to PC Port
sip.reg_surge_ prevention =	Integer from 0 to 60	It configures the maximum duration (in seconds) for account register after startup. The default value is 0. It takes effect after a reboot.	Network-> Advanced-> Registration Random-> Registration Random (0~60s)
sip.send_resp onse_by_requ est =	0 or 1	It configures from where the phone retrieves the destination address for response. The phone will then send all SIP response messages to the destination address.	

Parameter	Permitted Values	Descriptions	Web Setting Path
		0 -from VIA header in the request message	
		1-from source address of the request message	
		The default value is 1.	
		It takes effect after a reboot.	
		It configures the phone behavior when receiving a SIP NOTIFY message which contains the header "Event: check-sync".	
sip.notify_rebo ot_enable =	0, 1or 2	0 -The phone will reboot only if the SIP NOTIFY message contains an additional string "reboot=true".	
		1-The phone will be forced to reboot.	
		2 -The phone will ignore the SIP NOTIFY message.	
		The default value is 1.	
		It configures the phone to export log files to a syslog server or the local system.	Comingra
syslog.mode	0 or 1	0-Local	Settings-> Configuration->Ex
=		1-Server	port System Log
		The default value is 0.	
		It takes effect after a reboot.	
	IP address	It configures the IP address or domain name of the syslog server when exporting log to the syslog server.	Settings->
syslog.server =	or domain	It takes effect only if the parameter	Configuration->
	name	"syslog.mode" is configured as Server. The default value is blank.	Server Name
		It takes effect after a reboot.	
syslog.log_lev	Integer from 0 to 6	It configures the detail level of syslog information to be exported. O means nothing and 6 means all.	Settings-> Configuration->
		The default value is 3. It takes effect after a reboot.	System Log Level

Parameter	Permitted Values	Descriptions	Web Setting Path
auto_provision .power_on =	0 or 1	It enables or disables the phone to perform an auto provisioning process when powered on. 0-Disabled 1-Enabled The default value is 1. Note: The old parameter "auto_provision.mode" is also applicable to IP phones.	Settings->Auto Provision->Power On
auto_provision .pnp_enable =	0 or 1	It enables or disables Plug and Play feature. If it is enabled, the phone will broadcast PnP SUBSCRIBE messages to obtain a provisioning server address after startup. 0-Disabled 1-Enabled The default value is 1.	Settings->Auto Provision->PNP Active
auto_provision .weekly.enabl e =	0 or 1	It enables or disables the phone to check new configuration weekly. 0-Disabled 1-Enabled The default value is 0.	Settings->Auto provision->Weekly
auto_provision .weekly.dayof week =	0,1,2,3,4,5,6 or a combination of these digits	It configures the days of the week for the phone to check new configuration weekly. 0-Sunday 1-Monday 2-Tuesday 3-Wednesday 4-Thursday 5-Friday 6-Saturday Example: auto_provision.weekly.dayofweek = 01 means the phone will check the new configuration every Sunday and	Settings->Auto provision->Day of week

Parameter	Permitted Values	Descriptions	Web Setting Path
auto_provision .weekly.begin _time =	Time from 00:00 to 23:59	Monday. The default value is 0123456. Note: The old parameters "auto_provision.weekly.mask" is also applicable to SIP-T21P/T19P/T4X IP phones and "auto_provision.schedule.dayofweek". is also applicable to SIP-T28P/T26P/T22P/T20P IP phones. It configures the begin time of the day for the phone to check new configuration weekly. The default value is 00:00. Note: The old parameter "auto_provision.schedule.time_from" is also applicable to SIP-T28P/T26P/T22P/T20P IP phones.	Settings->Auto provision->Time
auto_provision .weekly.end_ti me =	Time from 00:00 to 23:59	It configures the end time of the day for the phone to check new configuration weekly. The default value is 00:00. Note: The old parameter "auto_provision.schedule.time_to" is also applicable to SIP-T28P/T26P/T22P/T20P IP phones.	Settings->Auto provision->Time
auto_provision .repeat.enabl e =	0 or 1	It enables or disables the phone to check new configuration repeatedly. O-Disabled 1-Enabled The default value is 0.	Settings->Auto provision-> Repeatedly
auto_provision .repeat.minut es =	Integer from 1 to 43200	It configures the interval (in minutes) for the phone to check new configuration repeatedly. The default value is 1440. Note: The old parameter "auto_provision.schedule.periodic_min ute" is also applicable to SIP-T28P/T26P/T22P/T20P IP phones.	Settings->Auto provision->Interval (minutes)

Parameter	Permitted Values	Descriptions	Web Setting Path
auto_provision .dhcp_option. enable =	0 or 1	It enables or disables the phone to obtain the provisioning server address by detecting DHCP options. 0-Disabled 1-Enabled The default value is 1.	Settings->Auto Provision->DHCP Active
auto_provision .dhcp_option.l ist_user_optio ns =	Integer from 128 to 254	It configures the custom DHCP option for provisioning server address. The default value is blank.	Settings->Auto Provision->Custom Option (128~254)
auto_provision .server.url =	URL within 511 characters	It configures the access URL of configuration files. The default value is blank.	Settings->Auto Provision->Server URL
auto_provision .server.userna me =	String within 32 characters	It configures the user name for authentication during auto provisioning. The default value is blank.	Settings->Auto Provision->User Name
auto_provision .server.passw ord =	String within 32 characters	It configures the password for authentication during auto provisioning. The default value is blank.	Settings->Auto Provision-> Password
auto_provision .dhcp_option. option60_valu e =	String within 99 characters	It configures the value (vendor name of the device) of DHCP option 60. The default value is yealink.	Settings->Auto Provision->DHCP Option Value
auto_provision .aes_key_16.c om =	16 characters	It configures the AES key for decrypting the Common CFG file. The valid characters contain: $0 \sim 9$, A $\sim Z$, a $\sim z$. The default value is blank.	Settings->Auto Provision-> Common AES Key
auto_provision .aes_key_16.m ac =	16 characters	It configures the AES key for decrypting the MAC-Oriented CFG file. The valid characters contain: $0 \sim 9$, A \sim Z, a \sim z. The default value is blank.	Settings->Auto Provision-> MAC-Oriented AES Key

Parameter	Permitted Values	Descriptions	Web Setting Path
auto_provision .aes_key_in_fil e =	0 or 1	It enables or disables the phone to decrypt configuration files using the encrypted AES keys. 0-Disabled 1-Enabled If it is set to 1, the phone will download <y0000000000xx_security>.enc and <mac_security>.enc files during auto provisioning, and then decrypts these files into the plaintext keys (e.g., key2, key3) respectively using the phone built-in key (e.g., key1). The IP phone then decrypts the encrypted configuration files using corresponding key (e.g., key2, key3). The default value is 0.</mac_security></y0000000000xx_security>	
autoprovision. X.name = (X ranges from 1 to 50)	String within 64 characters	It configures the name of the code for triggering auto provisioning. The default value is blank. It takes effect after a reboot.	
autoprovision. X.code = (X ranges from 1 to 50)	String	It configures the code for triggering auto provisioning. Valid characters are digits, # and *. Example: autoprovision.1.code = *99 The default value is blank. It takes effect after a reboot.	
autoprovision. X.url = (X ranges from 1 to 50)	URL within 511 characters	It configures the access URL of auto provisioning server. The default value is blank. It takes effect after a reboot.	
autoprovision. X.user = (X ranges from 1 to 50)	String within 64 characters	It configures the user name for authentication during auto provisioning. The default value is blank. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
autoprovision. X.password = (X ranges from 1 to 50)	String within 32 characters	It configures the password for authentication during auto provisioning. The default value is blank. It takes effect after a reboot.	
autoprovision. X.com_aes = (X ranges from 1 to 50)	16 characters	It configures the AES key for decrypting the Common CFG file. The default value is blank. It takes effect after a reboot.	
autoprovision. X.mac_aes = (X ranges from 1 to 50)	16 characters	It configures the AES key for decrypting the MAC-Oriented CFG file. The default value is blank. It takes effect after a reboot.	
sip.use_23_as _pound =	0 or 1	It enables or disables the phone to reserve the pound sign when dialing out. O-Disabled (convert the pound sign into "%23") 1-Enabled The default value is 1.	Features->General Information-> Reserve # in User Name
sip.rfc2543_ho ld =	0 or 1	It enables or disables the phone to support RFC 2543 hold (c=0.0.0.0). 0 -Disabled 1 -Enabled The default value is 0.	Features->General Information->RFC 2543 Hold
sip.use_out_b ound_in_dialo g =	0 or 1	It enables or disables the phone to keep sending SIP requests to the outbound server in a dialog. O-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	Features->General Information->Use Outbound Proxy In Dialog
watch_dog.en able =	0 or 1	It enables or disables Watch Dog feature. If it is enabled, the phone will reboot automatically when the system	Settings-> Preference-> Watch Dog

Parameter	Permitted Values	Descriptions	Web Setting Path
		is broken down. 0 -Disabled 1 -Enabled The default value is 1.	
managements erver.enable = (not applicable to SIPT42G and SIPT41P IP phones)	0 or 1	It enables or disables TR069 feature. 0 -Disabled 1 -Enabled The default value is 0.	Settings->TR069-> Enable TR069
managements erver.usernam e = (not applicable to SIPT42G and SIPT41P IP phones)	String within 128 characters	It configures the user name for the phone to authenticate with the ACS. The default value is blank.	Settings->TR069-> ACS Username
managements erver.passwor d = (not applicable to SIPT42G and SIPT41P IP phones)	String within 64 characters	It configures the password for the phone to authenticate with the ACS. The default value is blank.	Settings->TR069-> ACS Password
managements erver.url = (not applicable to SIPT42G and SIPT41P IP phones)	URL within 511 characters	It configures the access URL of the ACS. The default value is blank.	Settings->TR069-> ACS URL
managements erver.periodic _inform_enabl	0 or 1	It enables or disables the phone to report its configuration to the ACS. 0 -Disabled	Settings->TR069-> Enable Periodic Inform

Parameter	Permitted Values	Descriptions	Web Setting Path
e = (not applicable to SIPT42G and SIPT41P IP phones)		1-Enabled The default value is 1.	
managements erver.periodic _inform_interv al = (not applicable to SIPT42G and SIPT41P IP phones)	Integer from 5 to 4294967295	It configures the interval (in seconds) for the phone to report its configuration to the ACS. The default value is 60.	Settings->TR069-> Periodic Inform Interval (seconds)
managements erver.connecti on_request_us ername = (not applicable to SIPT42G and SIPT41P IP phones)	String within 128 characters	It configures the user name for the phone to authenticate the connection requests. The default value is blank.	Settings->TR069-> Connection Request Username
managements erver.connecti on_request_p assword = (not applicable to SIPT42G and SIPT41P IP phones)	String within 64 characters	It configures the password for the phone to authenticate the connection requests. The default value is blank.	Settings->TR069-> Connection Request Password
transfer.semi_ attend_tran_e nable =	0 or 1	It enables or disables the transferee party's phone to prompt a missed call on the LCD screen before displaying the caller ID when performing a semi-attended transfer. 0-Enabled	Features->Transfer ->Semi-Attend Transfer

Parameter	Permitted Values	Descriptions	Web Setting Path
		1-Disabled The default value is 1.	
transfer.blind_ tran_on_hook_ enable =	0 or 1	It enables or disables the phone to complete the blind transfer through on-hook. O-Disabled 1-Enabled The default value is 1.	Features->Transfer ->Blind Transfer On Hook
transfer.on_ho ok_trans_ena ble =	0 or 1	It enables or disables the phone to complete the attended transfer through on-hook. O-Disabled 1-Enabled The default value is 1.	Features->Transfer ->Semi Attend Transfer On Hook
transfer.dsske y_deal_type =	0, 1 or 2	It configures the DSS key behavior during an active call when user presses the DSS key and the DSS key is configured as a speed dial, transfer or BLF/BLF list key. 0-New Call 1-Attended Transfer 2-Blind Transfer The default value is 2.	Features->Transfer ->Transfer Mode Via Dsskey
transfer.multi_ call_trans_ena ble = (not applicable to SIPT20P and SIPT4X IP phones)	0 or 1	It enables or disables the phone to enter Transfer to menu during multiple calls when pressing the transfer soft key or TRAN key. 0-Disabled 1-Enabled The default value is 1.	Features->General Information-> Allow Trans Exist Call
transfer.tran_o thers_after_co nf_enable =	0 or 1	It enables or disables the phone to transfer call to the two parties after a local conference call hangs up. 0-Disabled 1-Enabled	Features->Transfer ->Transfer on Conference Hang up

Parameter	Permitted Values	Descriptions	Web Setting Path
		The default value is 0.	
voice.vad =	0 or 1	It enables or disables VAD (Voice Activity Detection) feature. 0-Disbaled 1-Enabled The default value is 0.	Settings->Voice-> Echo Cancellation ->VAD
voice.cng =	0 or 1	It enables or disables CNG (Comfortable Noise Generator) feature. 0-Disabled 1-Enabled The default value is 1.	Settings->Voice-> Echo Cancellation ->CNG
voice.echo_ca ncellation =	0 or 1	It enables or disables AEC (Acoustic Echo Canceller) feature. O-Disabled 1-Enabled The default value is 1.	Settings->Voice-> Echo Cancellation ->ECHO
voice.jib.adap tive =	0 or 1	It configures the type of jitter buffer. 0-Fixed 1-Adaptive The default value is 1.	Settings->Voice-> JITTER BUFFER-> Type
voice.jib.min	Integer from 0 to 400	It configures the minimum delay (in milliseconds) of jitter buffer. The default value is 60.	Settings->Voice-> JITTER BUFFER-> Min Delay
voice.jib.max	Integer from 0 to 400	It configures the maximum delay (in milliseconds) of jitter buffer. The default value is 300.	Settings->Voice-> JITTER BUFFER-> Max Delay
voice.jib.norm al =	Integer from 0 to 400	It configures the normal delay (in milliseconds) of jitter buffer. The default value is 120.	Settings->Voice-> JITTER BUFFER-> Normal
voice.tone.cou ntry =	Custom, Australia, Austria, Brazil, Belgium,	It configures the country tone for the phone. The default value is Custom.	Settings->Tones-> Select Country

Parameter	Permitted Values	Descriptions	Web Setting Path
	China,		
	Czech,		
	Denmark,		
	Finland,		
	France,		
	Germany,		
	Great		
	Britain,		
	Greece,		
	Hungary,		
	Lithuania,		
	India, Italy,		
	Japan,		
	Mexico,		
	New		
	Zealand,		
	Netherlands,		
	Norway,		
	Portugal,		
	Spain,		
	Switzerland,		
	Sweden,		
	Russia,		
	United		
	States,		
	Chile, Czech		
	ETSI		
		It customizes the dial tone when the	
		parameter "voice.tone.country" is set	
		to Custom.	
		tonelist = element[,element]	
		[,element]	
		Where	
voice.tone.dia	String		Settings->Tones->
I =	Sung	element =	Dial
		[!]Freq1[+Freq2][+Freq3][+Freq4]	
		/Duration	
		Freq: the frequency of the tone (ranges	
		from 200 to 7000 Hz). If it is set to 0Hz, it	
		means silence. A tone is comprised of	
		at most four different frequencies.	

Parameter	Permitted Values	Descriptions	Web Setting Path
		Duration : the duration (in milliseconds) of the dial tone, ranges from 0 to 30000ms.	
		You can configure at most eight different tones for one condition, and separate them by commas. (e.g., 250/200, 0/1000, 200+300/500, 600+700+800+1000/2000). If you want the phone to play tones once, add an exclamation mark "!" before tones (e.g., !250/200, 0/1000, 200+300/500, 600+700+800+1000/2000).	
		The default value is blank. It customizes the ringback tone when the parameter "voice.tone.country" is	
voice.tone.rin g =	String	set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial".	Settings->Tones-> Ring Back
		The default value is blank.	
voice.tone.bus y =	String	It customizes the busy tone when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial".	Settings->Tones-> Busy
		The default value is blank.	
voice.tone.con gestion =	String	It customizes the tone for network congestion when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format,	Settings->Tones-> Congestion
		refer to the parameter "voice.tone.dial". The default value is blank.	

Parameter	Permitted Values	Descriptions	Web Setting Path
voice.tone.call waiting =	String	It customizes the call waiting tone when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial". The default value is blank.	Settings->Tones-> Call Waiting
voice.tone.dia Irecall =	String	It customizes the call back tone when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial". The default value is blank.	Settings->Tones-> Dial Recall
voice.tone.inf o =	String	It customizes the info tone when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial". The default value is blank.	Settings->Tones-> Info
voice.tone.stut ter =	String	It customizes the stutter tone when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial". The default value is blank.	Settings->Tones-> Stutter
voice.tone.me ssage = (not applicable to SIPT20P, SIPT42G and	String	It customizes the message tone when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter	Settings->Tones-> Message

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT41P IP phones)		"voice.tone.dial". The default value is blank.	
voice.tone.aut oanswer =	String	It customizes the warning tone for auto answer when the parameter "voice.tone.country" is set to Custom. The value format is Freq/Duration. For more information on the value format, refer to the parameter "voice.tone.dial". The default value is blank.	Settings->Tones-> Auto Answer
voice.group_s pk_vol =	Integer from 1 to 15	It configures the receiving volume of the group listening mode. The default value is 8.	
voice.ring_vol	Integer from 1 to 15	It configures the volume of ringer. The default value is 8.	
voice.handfre e.spk_vol =	Integer from 1 to 15	It configures the receiving volume of speaker. The default value is 8. It takes effect after a reboot (only for SIP-T28P/T26P/T22P/T20P IP phones).	
voice.handset .spk_vol =	Integer from 1 to 15	It configures the receiving volume of handset. The default value is 8. It takes effect after a reboot (only for SIPT28P/T26P/T22P/T20P IP phones).	
voice.headset .spk_vol =	Integer from 1 to 15	It configures the receiving volume of headset. The default value is 8. It takes effect after a reboot (only for SIP-T28P/T26P/T22P/T20P IP phones).	
voice.handfre e.tone_vol =	Integer from 1 to 15	It configures the dial tone volume of speaker. The default value is 8. It takes effect after a reboot (only for SIP-T28P/T26P/T22P/T20P IP phones).	

Parameter	Permitted Values	Descriptions	Web Setting Path
voice.handset .tone_vol =	Integer from 1 to 15	It configures the dial tone volume of handset. The default value is 8. It takes effect after a reboot (only for SIP-T28P/T26P/T22P/T20P IP phones).	
voice.headset .tone_vol =	Integer from 1 to 15	It configures the dial tone volume of headset. The default value is 8. It takes effect after a reboot (only for SIP-T28P/T26P/T22P/T20P IP phones).	
voice.handfre e_send = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	Integer from 1 to 53	It configures the sending volume of speaker. The default value is 35. It takes effect after a reboot.	Features-> Audio->Handfree Send Volume (1~53)
voice.handset _send = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	Integer from 1 to 53	It configures the sending volume of handset. The default value is 25. It takes effect after a reboot.	Features-> Audio->Handset Send Volume (1~53)
voice.headset _send = (only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT20P IP phones)	Integer from 1 to 53	It configures the sending volume of headset. The default value is 30. It takes effect after a reboot.	Features-> Audio->Headset Send Volume (1~53)

Parameter	Permitted Values	Descriptions	Web Setting Path
security.trust_c ertificates =	0 or 1	It enables or disables the phone to only accept the certificates in the Trusted Certificates list. 0-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	Security->Trusted Certificates->Only Accept Trusted Certificates
security.ca_ce rt =	0, 1 or 2	It configures the source certificates for the phone to authenticate for TLS connection. 0-Default certificates 1-Custom certificates 2-All certificates The default value is 2. It takes effect after a reboot.	Security->Trusted Certificates->CA Certificates
security.cn_val idation =	0 or 1	It enables or disables the phone to mandatorily validate the CommonName or SubjectAltName of the certificate received from the connecting server. 0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	Security->Trusted Certificates->Com mon Name Validation
security.dev_c ert =	0 or 1	It configures the device certificates for the phone to send for TLS authentication. 0-Default certificates 1-Custom certificates The default value is 0. It takes effect after a reboot.	Security->Server Certificates->Devi ce Certificates
security.user_n ame.user = security.user_n	String within 32 characters String within	It configures the user name of the user for web server access. The default value is user. It configures the user name of the	
ame.admin =	32	administrator for web server access.	

Parameter	Permitted Values	Descriptions	Web Setting Path
security.user_n ame.var =	characters String within 32 characters	The default value is admin. It configures the user name of the var for web server access. The default value is var.	
security.user_ password =	String within 32 characters	It configures the password of the user, var and administrator. The valid value format is username:new password. Example: security.user_password = admin:password123 means setting the password of administrator (current user name is "admin") to password123. The default value is blank.	Security->Passwor d
security.var_e nable =	0 or 1	It enables or disables the 3-level permissions (admin, user, var). 0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	
web_item_lev el.url =	URL within 511 characters	It configures the access URL of the file, which defines 3-level access permissions. The default value is blank. It takes effect after a reboot.	
phone_setting .custom_softke y_enable = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables customizing the soft keys layout. 0-Disabled 1-Enabled The default value is 0.	Settings->Softkey Layout->Custom Softkey
custom_softke y_call_failed.u rl = (not	URL within 511 characters	It configures the access URL of the file for custom soft keys layout on the LCD screen when Call failed.	

Parameter	Permitted Values	Descriptions	Web Setting Path
applicable to SIPT20P IP phones)		The default value is blank.	
custom_softke y_call_in.url = (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the file for custom soft keys layout on the LCD screen when Call in. The default value is blank.	
custom_softke y_connecting. url = (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the file for custom soft keys layout on the LCD screen when Connecting. The default value is blank.	
custom_softke y_dialing.url = (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the file for custom soft keys layout on the LCD screen when Dialing. The default value is blank.	
custom_softke y_ring_back.ur I = (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the file for custom soft keys layout on the LCD screen when Ringback. The default value is blank.	
custom_softke y_talking.url = (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the file for custom soft keys layout on the LCD screen when Talking. The default value is blank.	
memorykey.X. type = (X ranges	Integer	It configures the desired feature for memory key X. Valid values are:	DSSKey->Memory Key->Memory KeyX->Type

Parameter	Permitted Values	Descriptions	Web Setting Path
from 1 to 10)		0-N/A	
(only		1-Conference	
applicable to		2-Forward	
SIPT28P and		3 -Transfer	
SIP-T26P IP phones)		4-Hold	
pe.,		5-DND	
		7-Call Return	
		8-SMS	
		9-Directed Pickup	
		10-Call Park	
		11-DTMF	
		12-Voice Mail	
		13-Speed Dial	
		14-Intercom	
		15-Line	
		16 -BLF	
		17 -URL	
		18-Group Listening	
		20 -Private Hold	
		22-XML Group	
		23-Group Pickup	
		24-Multicast Paging	
		25-Record	
		27-XML Browser	
		34 -Hot Desking	
		35 -URL Record	
		38 -LDAP	
		39 -BLF List	
		40 -Prefix	
		41-Zero Touch	
		42 -ACD	
		45 -Local Group	
		46 -Network Group	
		49-Custom Button	
		50 -Keypad Lock	

Parameter	Permitted Values	Descriptions	Web Setting Path
memorykey.X.I ine = (X ranges from 1 to 10) (only applicable to SIPT28P and SIPT26P IP phones) memorykey.X.	Integer from 1 to 6	55-Meet-Me Conference 56-Retrieve Park 57-Hoteling 58-ACD Trace 59-Disp Code 60-Emergency 61-Directory The default value is 0. It configures the desired line to apply the key feature. 1-Line 1 2-Line 2 3-Line 3 4-Line 4 5-Line 5 6-Line 6 The default value is 1.	DSSKey->Memory Key->Memory KeyX->Line
value = (X ranges from 1 to 10) (only applicable to SIP-T28P and SIP-T26P IP phones)	String within 99 characters	It configures the value of the memory key feature. For example, when the key feature is BLF, it is used to configure the number of the monitored user. The default value is blank.	DSSKey->Memory Key->Memory KeyX->Value
memorykey.X. pickup_value = (X ranges from 1 to 10) (only applicable to SIPT28P and SIPT26P IP phones)	String within 256 characters	It configures the pickup code for BLF feature or conference ID followed by the # sign for Meet-Me conference feature. It only applies to BLF and Meet-Me conference features. The default value is blank.	DSSKey->Memory Key->Memory KeyX->Extension

Parameter	Permitted Values	Descriptions	Web Setting Path
memorykey.X. xml_phonebo ok = (X ranges from 1 to 10) (only applicable to SIPT28P and SIPT26P IP phones)	Integer from 0 to 5	It configures the desired local group/XML group/network group for the memory key X. It only applies to the Local Group, XML Group and Network Group features. When the key feature is configured as Local Group, valid values are: 0-All contacts 1-First local group 2-Second local group 3-Third local group 5-Fifth local group When the key feature is configured as XML Group (remote phone book), valid values are: 0-First XML group 1-Second XML group 2-Third XML group 4-Fifth XML group When the key feature is configured as Network Group, valid values are: 0-All contacts 1-Group 2-Enterprise 3-GroupCommon 4-EnterpriseCommon 5-Personal The default value is 0.	DSSKey->Memory Key->Memory KeyX->Line
linekey.X.type = (SIP-T28P: X ranges from 1 to 6. SIP-T26P/T22P:	Integer	It configures the key feature for the line key X. Valid values are: 0-N/A (only applicable to SIP-T4X IP phones) 1-Conference	DSSKey->Line Key->Line KeyX->Type

Parameter	Permitted Values	Descriptions	Web Setting Path
X ranges from		2-Forward	
1 to 3.		3 -Transfer	
SIP-T21P/T20P:		4 -Hold	
X ranges from		5-DND	
1 to 2.		7-Call Return	
SIPT48G: X ranges from 1 to 29.		8 -SMS (not applicable to SIP-T20P, SIP-T48G, SIP-T42G and SIP-T41P IP phones)	
SIP-T46G: X ranges from 1		9 -Directed Pickup	
to 27.		10-Call Park	
SIP-T42G/T41P:		11-DTMF	
X ranges from		12-Voice Mail	
1 to 15.)		13-Speed Dial	
(not		14-Intercom	
applicable to		15-Line	
SIP-T19P IP phones)		16 -BLF	
		17 -URL (only applicable to SIP-T4X IP phones)	
		18-Group Listening	
		20 -Private Hold	
		22-XML Group (not applicable to SIP-T20P IP phones)	
		23 -Group Pickup	
		24-Multicast Paging	
		25-Record	
		27-XML Browser	
		34 -Hot Desking	
		35 -URL Record	
		38 -LDAP (not applicable to SIPT20P IP phones)	
		39 -BLF List	
		40 -Prefix	
		41 -Zero Touch	
		42 -ACD (not applicable to SIPT48G/T46G)	
		45-Local Group	

Parameter	Permitted Values	Descriptions	Web Setting Path
		46 -Network Group (not applicable to SIPT20P IP phones)	
		49 -Custom Button (not applicable to SIPT4X IP phones)	
		50 -Keypad Lock (not applicable to SIPT48G IP phones)	
		55-Meet-Me Conference	
		56 -Retrieve Park	
		57-Hoteling	
		58 -ACD Trace	
		59 -Disp Code	
		60 -Emergency	
		61-Directory	
		62 -Network Favorite (only applicable to SIPT48G and SIPT46G IP phones)	
		63 -UC Favorite (only applicable to SIPT48G and SIPT46G IP phones)	
		64 -Buddies (only applicable to SIPT48G and SIPT46G IP phones)	
		65 -My Status (only applicable to SIPT48G and SIPT46G IP phones)	
		For SIP- T2xP/T19P IP phones:	
		The default value is 15.	
		For SIPT48G IP phones:	
		The default value of the line key 1-6 is 15, and the default value of the line keys 7-29 is 0.	
		For SIPT46G IP phones:	
		The default value of the line key 1-6 is 15, and the default value of the line keys 7-27 is 0.	
		For SIPT42G/T41P IP phones:	
		The default value of the line key 1-3 is 15, and the default value of the line keys 4-15 is 0.	
linekey.X.line	Integer from 1 to 6	It configures the desired line to apply the key feature.	DSSKey->Line Key->Line

Parameter	Permitted Values	Descriptions	Web Setting Path
(SIP-T28P: X ranges from 1 to 6. SIP-T26P/T22P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T48G: X ranges from 1 to 29. SIP-T46G: X ranges from 1 to 27. SIP-T42G/T41P: X ranges from 1 to 15.) (not		1-Line 1 2-Line 2 3-Line 3 4-Line 4 5-Line 5 6-Line 6 When X=1, the default value is 1. When X=2, the default value is 2. When X=3, the default value is 3. When X=4, the default value is 4. When X=5, the default value is 5. When X=6, the default value is 6. Line 3 is only applicable to SIPT28P, SIPT26P, SIPT22P and SIPT4X IP phones. Lines 4-6 are only applicable to SIPT28P, SIPT28P, SIPT48G and SIPT46G IP phones.	KeyX->Line
(not applicable to SIP-T19P IP phones) linekey.X.valu e = (SIP-T28P: X ranges from 1 to 6. SIP-T26P/T22P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T48G: X ranges from 1 to 29. SIP-T46G: X ranges from 1 to 27.	String within 99 characters	It configures the value of the line key feature. For example, when setting the key feature to BLF, it configures the number of the monitored user. The default value is blank.	DSSKey->Line Key->Line KeyX->Value

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT42G/T41P: X ranges from 1 to 15.) (not applicable to SIPT19P IP phones) linekey.X.pick up_value = (SIPT28P: X ranges from 1 to 6. SIPT26P/T22P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT48G: X ranges from 1 to 29. SIPT46G: X ranges from 1 to 27. SIPT42G/T41P: X ranges from 1 to 15.) (not applicable to SIPT19P IP phones)	String within 256 characters	It configures the pickup code for BLF feature or conference ID followed by the # sign for Meet-Me conference feature. It only applies to BLF and Meet-Me conference features. The default value is blank.	DSSKey->Line Key->Line KeyX->Extension
linekey.X.xml_phonebook = (SIP-T28P: X ranges from 1 to 6. SIP-T26P/T22P: X ranges from 1 to 3.	Integer from 0 to 48	It configures the desired local group/XML group/network group for the line key X. It only applies to the Local Group, XML Group and Network Group features. XML Group and Network Group features are not applicable to SIP-T20P IP phones.	DSSKey->Line Key->Line KeyX->Line

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T21P/T20P:		When the key feature is configured as	
X ranges from		Local Group, valid values are:	
1 to 2.		0 -All contacts	
SIPT48G: X		1-First local group	
ranges from 1		2 -Second local group	
to 29.		3 -Third local group	
SIPT46G: X ranges from 1		4-Fourth local group	
to 27.		5-Fifth local group	
SIP-T42G/T41P:			
X ranges from		48- Forty-eighth local group	
1 to 15.)		Local groups 6-48 are only applicable	
(not		to SIPT4X IP phones.	
applicable to		When the key feature is configured as	
SIPT19P IP		XML Group (remote phone book),	
phones)		valid values are:	
		0 -First XML group	
		1-Second XML group	
		2-Third XML group	
		3- Fourth XML group	
		4- Fifth XML group	
		When the key feature is configured as	
		Network Group, valid values are:	
		0 -All contacts	
		1-Group	
		2-Enterprise	
		3 -GroupCommon	
		4-EnterpriseCommon	
		5-Personal	
		The default value is 0.	
linekey.X.label			
=			
(SIP-T28P: X	String within	It configures the label displayed on the	DSSKey->Line
ranges from 1	99	LCD screen for each line key.	Key->Line
to 6.	characters	The default value is blank.	KeyX->Label
SIP-T26P/T22P:			
X ranges from			

Parameter	Permitted Values	Descriptions	Web Setting Path
1 to 3.			
SIP-T21P/T20P:			
X ranges from			
1 to 2.			
SIP-T48G: X			
ranges from 1			
to 29.			
SIPT46G: X			
ranges from 1 to 27.			
SIP-T42G/T41P:			
X ranges from			
1 to 15.)			
(not			
applicable to			
SIP-T19P IP			
phones)			
		It configures the key feature for the	
programablek		programmable key X.	
ey.X.type =		Valid values are:	
(SIP-T28/T26P:		0-N/A	
X ranges from		2-Forward	
1 to 14;		5-DND	
SIP-T22P:		7 -Call Return	
X=1-10, 12-14;		8 -SMS (not applicable to SIP-T20P,	
SIP-T21P:		SIPT41P, SIPT42G and SIPT48G IP	DSSKey->
X=1-10, 14;	Integer	phones)	Programmable
SIP-T20P:	3	9-Directed Pickup	Key->Type
X=5-12, 14;		13-Spead Dial	
SIP-T19P:		22-XML Group (not applicable to	
X=1-9, 13, 14.		SIPT19P IP phones)	
SIP-T48G/		23-Group Pickup	
T46G: X=1-10, 12-14;		27-XML Browser	
SIP-T42G/T41P:		28-History	
X=1-10, 13.)		30 -Menu	
,		31 -Switch Account (not applicable to SIP-T19P and SIP-T4X IP phones)	

Parameter	Permitted Values	Descriptions	Web Setting Path
		32 -New SMS (not applicable to SIPT20P, SIPT41P, SIPT42G and SIPT48G IP phones)	
		33 -Status	
		34 -Hot Desking (only applicable to SIPT19P, SIPT48G and SIPT46G IP phones)	
		38 -LDAP (not applicable to SIP-T19P and SIP-T20P IP phones)	
		40 -Prefix (not applicable to SIP-T20P IP phones)	
		41 -Zero Touch (not applicable to SIPT4X IP phones)	
		43-Local Directory	
		44-Network Directory	
		45 -Local Group	
		46 -Network Group	
		47 -XML Directory (not applicable to SIP-T20P IP phones)	
		50 -Keypad Lock (not applicable to SIPT48G IP phones)	
		51 -Switch Account Up (only applicable to SIPT4X IP phones)	
		52 -Switch Account Down (only applicable to SIPT4X IP phones)	
		55-Meet-Me Conference	
		61-Directory	
		64 -Buddies (only applicable to SIPT48G and SIPT46G IP phones)	
		65 -My Status (only applicable to SIPT48G and SIPT46G IP phones)	
programablek ey.X.line =		It configures the desired line to apply the programmable key feature.	
(SIP-T28/T26P:	Integer from	1-Line 1	DSSKey->
X ranges from	1 to 6	2 -Line 2	Programmable Key->Line
1 to 14;		3 -Line 3	Ney-Zuile
SIPT21P:		4 -Line 4	

Parameter	Permitted Values	Descriptions	Web Setting Path
X=1-10, 14;		5 -Line 5	
SIP-T20P:		6 -Line 6	
X=5-12, 14;		Line 3 is only applicable to SIPT28P,	
SIP-T19P:		SIPT26P, SIPT22P and SIPT4X IP phones.	
X=1-9, 13, 14.		Lines 4-6 are only applicable to	
SIP-T48G/		SIPT28P, SIPT48G and SIPT46G IP	
T46G: X=1-10,		phones.	
12-14;		The default value is 1.	
SIP-T42G /T41P:			
X=1-10, 13.)			
programablek			
ey.X.value =			
(SIP-T28/T26P:			
X ranges from			
1 to 14;			
SIP-T22P: X=1-10, 12-14;			
		It configures the value of the programmable key feature.	
SIP-T21P: X=1-10, 14;	String within		DSSKey->
SIP-T20P:	99	For example, when configuring the key feature to be Speed Dial, it is used to	Programmable
X=5-12, 14;	characters	configure the number.	Key->Value
SIP-T19P:		The default value is blank.	
X=1-9, 13, 14.			
SIPT48G/			
T46G: X=1-10,			
12-14;			
SIP-T42G /T41P:			
X=1-10, 13.)			
programablek		It configures the desired local	
ey.X.xml_phon		group/XML group/network group for	
ebook =		the programmable key.	
(SIP-T28/T26P:		It only applies to the Local Group, XML	DOCK
X ranges from	Integer from	Group and Network Group features.	DSSKey-> Programmable
1 to 14;	0 to 48	XML Group and Network Group	Key->Line
SIP-T22P:		features are not applicable to SIP-T20P	.,
X=1-10, 12-14;		IP phones.	
SIP-T21P:		When the key feature is configured as	
X=1-10, 14;		Local Group, valid values are:	

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T20P:		0-All contacts	
X=5-12, 14;		1-First local group	
SIP-T19P:		2-Second local group	
X=1-9, 13, 14.		3 -Third local group	
SIP-T48G/		4 -Fourth local group	
T46G: X=1-10, 12-14;		5-Fifth local group	
SIP-T42G/T41P: X=1-10, 13.)		48- Forty-eighth local group	
1		Local groups 6-48 are only applicable	
		to SIP-T4X IP phones.	
		When the key feature is configured as	
		XML Group (remote phone book),	
		valid values are:	
		0-First XML group	
		1-Second XML group	
		2-Third XML group	
		3-Fourth XML group	
		4-Fifth XML group	
		When the key feature is configured as Network Group, valid values are:	
		0 -All contacts	
		1-Group	
		2-Enterprise	
		3 -GroupCommon	
		4-EnterpriseCommon	
		5 -Personal	
		The default value is 0.	
programablek			
ey.X.history_ty			
pe =		It configures the history type of	
(SIP-T28/T26P:		programmable key.	DSSKey->
X ranges from	Integer	0 -Local History	Programmable
1 to 14;		1-Network History	Key->Line
SIP-T22P:		The default value is 0.	
X=1-10, 12-14;			
SIP-T21P:			

Parameter	Permitted Values	Descriptions	Web Setting Path
X=1-10, 14; SIPT19P: X=1-9, 13, 14. SIPT48G/ T46G: X=1-10, 12-14; SIPT42G/T41P: X=1-10, 13.) (not applicable to SIPT20P IP phones)			
programablek ey.X.label = (X ranges from 1 to 4) (not applicable to SIPT20P IP phones)	String within 99 characters	It configures the label displayed on the LCD screen for each soft key. The default value is blank.	DSSKey-> Programmable Key->Label
programablek ey.X.pickup_v alue = (SIP-T28/T26P: X ranges from 1 to 14; SIP-T22P: X=1-10, 12-14; SIP-T21P: X=1-10, 14; SIP-T20P: X=5-12, 14; SIP-T19P: X=1-9, 13, 14. SIP-T48G/ T46G: X=1-10, 12-14; SIP-T42G /T41P:	String within 256 characters	It configures conference ID followed by the # sign for Meet-Me conference feature. It only applies to Meet-Me conference feature. The default value is blank.	DSSKey->Program mable Key->Extension

Parameter	Permitted Values	Descriptions	Web Setting Path
expansion_mo dule.X.key.Y.ty pe = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P: X ranges from 1 to 3. Y ranges from 1 to 40) (only applicable to SIP-T28P, SIP-T26P, SIP-T26P, SIP-T48G and SIP-T46G IP phones)	Integer	It configures the key feature of the expansion module X key Y. Valid values are: 0-NA 1-Conference 2-Forward 3-Transfer 4-Hold 5-DND 7-Call Return 8-SMS (not applicable to SIP-T48G IP phones) 9-Directed Pickup 10-Call Park 11-DTMF 12-Voice Mail 13-Speed Dial 14-Intercom 15-Line 16-BLF 17-URL 18-Group Listening 20-Private Hold 22-XML Group 23-Group Pickup 24-Multicast Paging 25-Record 27-XML Browser 34-Hot Desking 35-URL Record 37-Switch (not applicable to SIP-T48G/T46G IP phones) 38-LDAP 39-BLF List	DSSKey->Ext Key->Type

Parameter	Permitted Values	Descriptions	Web Setting Path
		40 -Prefix	
		41 -Zero Touch	
		42 -ACD (not applicable to SIPT48G/T46G)	
		45 -Local Group	
		46 -Network Group	
		49-Custom Button	
		50 -Keypad Lock (not applicable to SIPT48G IP phones)	
		55-Meet-Me Conference	
		56 -Retrieve Park	
		57-Hoteling	
		58-ACD Grace	
		59 -Disp Code	
		60 -Emergency	
		61 -Directory	
		62 -Network Favorite (only applicable to SIPT48G and SIPT46G IP phones)	
		63 -UC Favorite (only applicable to SIPT46G IP phones)	
		64 -Buddies (only applicable to SIPT48G and SIPT46G IP phones)	
		65 -My Status (only applicable to SIPT48G and SIPT46G IP phones)	
		For SIPT28P/T26P IP phones:	
		The default value of the expansion keys 2-20, 22-40 is 0, and the default value of the expansion keys 1, 21 is 37.	
		For SIPT48G/T46G IP phones:	
		The default value is 0.	
expansion_mo dule.X.key.Y.lin		It configures the desired line to apply the expansion module key feature.	
e =	Integer from	1-Line 1	DSSKey->Ext
(SIP-T28P/T48G	1 to 6	2 -Line 2	Key->Line
/T46G: X ranges from 1		3 -Line 3	
to 6.		4 -Line 4	

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT26P: X ranges from 1 to 3. Y ranges from 1 to 40) (only applicable to SIPT28P, SIPT26P, SIPT48G and SIPT46G IP phones) expansion_mo		5-Line 5 6-Line 6 Lines 4-6 are only applicable to SIPT28P, SIP-T48G and SIP-T46G IP phones. The default value is 1.	
dule.X.key.Y.v alue = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P: X ranges from 1 to 3. Y ranges from 1 to 40) (only applicable to SIP-T28P, SIP-T26P, SIP-T26P, SIP-T26P, SIP-T46G IP phones)	String within 99 characters	It configures the value of the expansion module key feature. The default value is blank.	DSSKey->Ext Key->Value
expansion_mo dule.X.key.Y.pi ckup_value = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P: X	String within 255 characters	It configures the pickup code for BLF feature or conference ID followed by the # sign for Meet-Me conference feature. It only applies to BLF and Meet-Me conference features. The default value is blank.	DSSKey->Ext Key->Extension

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 3. Y ranges from 1 to 40) (only applicable to SIPT28P, SIPT26P, SIPT48G and SIPT46G IP phones)			
expansion_mo dule.X.key.Y.la bel = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P: X ranges from 1 to 3. Y ranges from 1 to 40) (only applicable to SIP-T28P, SIP-T26P, SIP-T26P, SIP-T48G and SIP-T46G IP phones)	String within 99 characters	It configures the label displayed on the LCD screen of the expansion module for each key. The default value is blank.	DSSKey->Ext Key->Label
expansion_mo dule.X.key.Y.x ml_phoneboo k = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P: X	Integer from 0 to 48	It only applies to the Local Group, XML Group and Network Group features. When the key feature is configured as Local Group, valid values are: 0-All contacts 1-First local group 2-Second local group 3-Third local group	DSSKey->Ext Key->Ext KeyX->Line

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 3. Y ranges from 1 to 40) (only applicable to SIPT28P, SIPT26P, SIPT46G IP phones)	Values	4-Fourth local group 5-Fifth local group 48-Forty-eighth local group Local groups 6-48 are only applicable to SIPT48G/T46G IP phones. When the key feature is configured as XML Group (remote phone book), valid values are: 0-First XML group 1-Second XML group 2-Third XML group 3-Fourth XML group 4-Fifth XML group When the key feature is configured as Network Group, valid values are: 0-All contacts 1-Group 2-Enterprise 3-GroupCommon 4-EnterpriseCommon 5-Personal The default value is 0.	
phone_setting .page_tip = (only applicable to SIPT46G, SIPT42G and SIPT41P IP phones)	0 or 1	It enables or disables the phone to display page icons on LCD screen and flash page switch key LED to indicate different states of line keys. 0-Disabled 1-Enabled The default value is 0.	DSSKey->Line key->Enable Page Tips
phone_setting .backgrounds = (only applicable to SIPT48G and	Resource:X (Valid values of X are: Default.jpg, 1.png, 2.png,	It configures the wallpaper displayed on the phone. Example: To configure a phone built-in picture (e.g., 1.png) to be wallpaper, the value format is:	Settings-> Preference-> Wallpaper

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T46G IP	3.png,	phone_setting.backgrounds =	
phones)	4.png,	Resource:1.png	
	5.png,	To configure a custom picture (e.g.,	
	6.png,	custom1.png) to be wallpaper, the	
	7.png,	value format is: Config:custom1.png	
	8.png or	The default value is	
	9.png) or	Resource:Default.jpg.	
	Config:wallp		
	aper name		
wallpaper_upl oad.url =			
(only	URL within	It configures the access URL of the	Settings->
applicable to	511	custom wallpaper image.	Preference->Uploa
SIP-T48G and	characters	The default value is blank.	d Wallpaper
SIPT46G IP			
phones)			
		It enables or disables always forward	
		feature.	Features->Forwar
forward.alwa	0 or 1	0 -Disabled	d &DND->Always
ys.enable =	0 01 1	1-Enabled	Forward->On/Off
		The default value is 0.	
forward.alwa	String within	It configures the target number the	Features->Forwar
ys.target =	32	phone forwards all incoming calls to.	d &DND->Always
, ortali got	characters	The default value is blank.	Forward->Target
forward.alwa	String within	It configures the always forward on	Features->Forwar
ys.on_code =	32	code.	d &DND->Always Forward->On
ys.on_code =	characters	The default value is blank.	Code
forward.alwa	String within	It configures the always forward off	Features->Forwar d &DND->Always
ys.off_code =	32	code.	Forward->Off
_	characters	The default value is blank.	Code
		It enables or disables busy forward	
		feature.	Features->Forwar
forward.busy.	0 or 1	0 -Disabled	d &DND->Busy
enable =		1-Enabled	Forward->On/Off
		The default value is 0.	·
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Parameter	Permitted Values	Descriptions	Web Setting Path
forward.busy.t arget =	String within 32 characters	It configures the target number the phone forwards incoming calls to when busy. The default value is blank.	Features->Forwar d &DND->Busy Forward->Target
forward.busy. on_code =	String within 32 characters	It configures the busy forward on code. The default value is blank.	Features->Forwar d &DND->Busy Forward->On Code
forward.busy. off_code =	String within 32 characters	It configures the busy forward off code. The default value is blank.	Features->Forwar d &DND->Busy Forward->Off Code
forward.no_a nswer.enable =	0 or 1	It enables or disables no answer forward feature. 0-Disabled 1-Enabled The default value is 0.	Features->Forwar d &DND->No Answer Forward->On/Off
forward.no_a nswer.target =	String within 32 characters	It configures the target number the phone forwards incoming calls to after a period of ring time. The default value is blank.	Features->Forwar d &DND->No Answer Forward->Target
forward.no_a nswer.timeout =	Integer from 0 to 20	It configures ring times (N) to wait before forwarding incoming calls. Incoming calls are forwarded when not answered after N*6 seconds. The default value is 2.	Features->Forwar d &DND->No Answer Forward->After Ring Time (0~120s)
forward.no_a nswer.on_cod e =	String within 32 characters	It configures the no answer forward on code. The default value is blank.	Features->Forwar d &DND->No Answer Forward->On Code
forward.no_a nswer.off_cod e =	String within 32 characters	It configures the no answer forward off code. The default value is blank.	Features->Forwar d &DND->No Answer Forward->Off Code
forward.intern	0 or 1	It enables or disables the phone to	Features->General

Parameter	Permitted Values	Descriptions	Web Setting Path
ational.enable =		forward incoming calls to international numbers. 0-Disabled 1-Enabled The default value is 1.	Information->Fwd International
acd.auto_avai lable = (not applicable to SIPT48G/T46G)	0 or 1	It enables or disables the phone to automatically change the status of the ACD agent to available. 0-Disabled 1-Enabled The default value is 0.	Features->ACD-> ACD Auto Available
acd.auto_avai lable_timer = (not applicable to SIPT48G/T46G)	Integer from 0 to 120	It configures the interval (in seconds) to automatically change the status of the ACD agent to available. The default value is 60.	Features->ACD-> ACD Auto Available Timer (0~120s)
action_url.setu p_completed =	URL within 511 characters	It configures the action URL the phone sends after startup. The value format is: http(s)://IP address of server/help.xml? variable name=variable value. Valid variable values are:	Features->Action URL->Setup Completed

Parameter	Permitted Values	Descriptions	Web Setting Path
		action_url. setup_completed = http://192.168.0.20/help.xml?IP=\$ip	
		The default value is blank.	
		It configures the action URL the phone sends after account is registered.	
		Example:	
action_url.regi stered =	URL within 511 characters	action_url.registered = http://192.168.0.20/help.xml?IP=\$ip	Features->Action URL->Registered
	characters	The default value is blank.	
		Note: The old parameter "action_url.log_on" is also applicable to IP phones.	
		It configures the action URL the phone sends after account is unregistered.	
		Example:	
action_url.unr	URL within 511	action_url.unregistered = http://192.168.0.20/help.xml?IP=\$ip	Features->Action
egistered =	characters	The default value is blank.	URL->Unregistered
		Note: The old parameter	
		"action_url.log_off" is also applicable to IP phones.	
		It configures the action URL the phone	
action_url.regi	URL within	sends after register failed. Example:	Features->Action
ster_failed =	511 characters	action_url.register_failed =	URL->Register Failed
	characters	http://192.168.0.20/help.xml?IP=\$ip	Tulled
		The default value is blank.	
		It configures the action URL the phone sends when off hook.	
action_url.off_	URL within	Example:	Features->Action
hook =	511 characters	action_url.off_hook = http://192.168.0.20/help.xml?IP=\$ip	URL->Off Hook
		The default value is blank.	
action_url.on_	URL within	It configures the action URL the phone	Features->Action
hook =	511 characters	sends when on hook. Example:	URL->On Hook

Parameter	Permitted Values	Descriptions	Web Setting Path
		action_url.on_hook = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	
action_url.inco ming_call =	URL within 511 characters	It configures the action URL the phone sends when receiving an incoming call. Example: action_url.incoming_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Incoming Call
action_url.out going_call =	URL within 511 characters	It configures the action URL the phone sends when placing a call. Example: action_url.outgoing_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Outgoing Call
action_url.call _established =	URL within 511 characters	It configures the action URL the phone sends when establishing a call. Example: action_url.call_established = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Established
action_url.call _terminated =	URL within 511 characters	It configures the action URL the phone sends when terminating a call. Example: action_url.call_terminated = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Terminated
action_url.dnd _on =	URL within 511 characters	It configures the action URL the phone sends when DND feature is enabled. Example: action_url.dnd_on = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Open DND
action_url.dnd _off =	URL within 511	It configures the action URL the phone sends when DND feature is disabled.	Features->Action URL->Close DND

Parameter	Permitted Values	Descriptions	Web Setting Path
	characters	Example: action_url.dnd_off = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	
action_url.alw ays_fwd_on =	URL within 511 characters	It configures the action URL the phone sends when always forward feature is enabled. Example: action_url.always_fwd_on = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Open Always Forward
action_url.alw ays_fwd_off =	URL within 511 characters	It configures the action URL the phone sends when always forward feature is disabled. Example: action_url.always_fwd_off = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Close Always Forward
action_url.bus y_fwd_on =	URL within 511 characters	It configures the action URL the phone sends when busy forward feature is enabled. Example: action_url.busy_fwd_on = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Open Busy Forward
action_url.bus y_fwd_off =	URL within 511 characters	It configures the action URL the phone sends when busy forward feature is disabled. Example: action_url.busy_fwd_off = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Close Busy Forward
action_url.no_ answer_fwd_o n =	URL within 511 characters	It configures the action URL the phone sends when no answer forward feature is enabled. Example:	Features->Action URL->Open No Answer Forward

Parameter	Permitted Values	Descriptions	Web Setting Path
		action_url.no_answer_fwd_on = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	
action_url.no_ answer_fwd_o ff =	URL within 511 characters	It configures the action URL the phone sends when no answer forward feature is disabled. Example: action_url.no_answer_fwd_off = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Close No Answer Forward
action_url.tran sfer_call =	URL within 511 characters	It configures the action URL the phone sends when performing a transfer. Example: action_url.transfer_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Transfer Call
action_url.blin d_transfer_call =	URL within 511 characters	It configures the action URL the phone sends when performing a blind transfer. Example: action_url.blind_transfer_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Blind Transfer
action_url.atte nded_transfer _call =	URL within 511 characters	It configures the action URL the phone sends when performing an attended or a semi-attended transfer. Example: action_url.attended_transfer_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Attended Transfer
action_url.hol d =	URL within 511 characters	It configures the action URL the phone sends when placing a call on hold. Example: action_url.hold = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Hold

Parameter	Permitted Values	Descriptions	Web Setting Path
action_url.unh old =	URL within 511 characters	It configures the action URL the phone sends when resuming a held call. Example: action_url.unhold = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->UnHold
action_url.mut e =	URL within 511 characters	It configures the action URL the phone sends when muting a call. Example: action_url.mute = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Mute
action_url.unm ute =	URL within 511 characters	It configures the action URL the phone sends when un-muting a call. Example: action_url.unmute = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->UnMute
action_url.miss ed_call =	URL within 511 characters	It configures the action URL the phone sends when missing a call. Example: action_url.missed_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Missed Call
action_url.bus y_to_idle =	URL within 511 characters	It configures the action URL the phone sends when changing the state of the phone from busy to idle. Example: action_url.busy_to_idle = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Busy To Idle
action_url.idle _to_busy =	URL within 511 characters	It configures the action URL the phone sends when changing the state of the phone from idle to busy. Example: action_url.idle_to_busy =	Features->Action URL->Idle To Busy

Parameter	Permitted Values	Descriptions	Web Setting Path
		http://192.168.0.20/help.xml?IP=\$ip	
		The default value is blank.	
action_url.ip_c hange =	URL within 511 characters	It configures the action URL the phone sends when changing the IP address of the phone. Example: action_url.ip_change = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->IP Changed
action_url.for ward_incomin g_call =	URL within 511 characters	It configures the action URL the phone sends when forwarding an incoming call. Example: action_url.forward_incoming_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Forward Incoming Call
action_url.reje ct_incoming_c all =	URL within 511 characters	It configures the action URL the phone sends when rejecting an incoming call. Example: action_url.reject_incoming_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Reject Incoming Call
action_url.ans wer_new_inco ming_call =	URL within 511 characters	It configures the action URL the phone sends when answering a new incoming call. Example: action_url.answer_new_incoming_call = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Answer New-In Call
action_url.tran sfer_finished =	URL within 511 characters	It configures the action URL the phone sends when completing a call transfer. Example: action_url.transfer_finished = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Transfer Finished

Parameter	Permitted Values	Descriptions	Web Setting Path
action_url.tran sfer_failed =	URL within 511 characters	It configures the action URL the phone sends when failing to transfer a call. Example: action_url.transfer_failed = http://192.168.0.20/help.xml?IP=\$ip The default value is blank.	Features->Action URL->Transfer Failed
lang.wui =	English, Chinese_S, Chinese_T, German, French, Italian, Spanish, Turkish or Portuguese	It configures the language of the web user interface. Chinese_S is only applicable to SIPT21P, SIPT19P, SIPT48G and SIP-T46G IP phones. Chinese_T is only applicable to SIPT48G and SIPT46G IP phones. French, Portuguese and Spanish are not applicable to SIP-T21P and SIP-T19P IP phones.	Settings->Preferen ce->Language
lang.gui = local_time.tim e_zone =	English, Chinese_S, Chinese_T, German, French, Turkish, Italian, Polish, Spanish or Portuguese	It configures the language of the phone user interface. Chinese_S and Chinese_T are only applicable to SIPT21P, SIPT19P, SIPT48G and SIPT46G IP phones. The default value is English. It configures the time zone. For more available time zones, refer to Time Zones on page 225.	Settings->Time & Date->Time Zone
local_time.tim e_zone_name =	String within 32 characters	The default value is +8. It configures the time zone name. For more available time zone names, refer to Time Zones on page 225. The default time zone name is China(Beijing).	Settings->Time & Date->Time Zone
local_time.ma	0 or 1	It configures the phone to obtain time from NTP server or manual settings.	Settings->Time & Date->Manual

Parameter	Permitted Values	Descriptions	Web Setting Path
able =		0-Manual 1-NTP The default value is 1.	Time
local_time.ma nual_ntp_srv_ prior =	0 or 1	It enables or disables the phone to use manually configured NTP server preferentially. 0-Disabled (use the NTP server obtained by DHCP preferentially) 1-Enabled The default value is 0.	Settings->Time & Date->NTP By DHCP Priority
local_time.ntp _server1 =	IP address or domain name	It configures the IP address or domain name of the NTP server 1. The default value is cn.pool.ntp.org.	Settings->Time & Date->Primary Server
local_time.ntp _server2 =	IP address or domain name	It configures the IP address or domain name of the NTP server 2. The default value is cn.pool.ntp.org.	Settings->Time & Date->Secondary Server
local_time.inte rval =	Integer from 15 to 86400	It configures the update interval (in seconds) when using the NTP server. The default value is 1000.	Settings->Time & Date->Synchronis m (15~86400s)
local_time.su mmer_time =	0, 1 or 2	It enables or disables daylight saving time (DST) feature. 0-Disabled 1-Enabled 2-Automatic The default value is 2.	Settings->Time & Date-> Daylight Saving Time
local_time.dst _time_type =	0 or 1	It configures the way DST works when DST feature is enabled. 0-DST By Date 1-DST By Week The default value is 0.	Settings->Time & Date->Fixed Type
local_time.star t_time =	Time	It configures the start time of the DST. Value formats are: Month/Day/Hour (for By Date) Month/ Day of Week/ Day of Week Last in Month/ Hour of Day	For DST By Date: Settings->Time & Date->Start Date For DST By Week: Settings->Time &

Parameter	Permitted Values	Descriptions	Web Setting Path
		(for By Week) The default value is 1/1/0.	Date->DST Start Month/DST Start Day of Week/DST Start Day of Week Last in Month/ Start Hour of Day
			For DST By Date:
local_time.en d_time =	Time	It configures the end time of the DST. Value formats are: Month/Day/Hour (for By Date) Month/ Day of Week/ Day of Week Last in Month/ Hour of Day (for By Week) The default value is 12/31/23.	Settings->Time & Date-> End Date For DST By Week: Settings ->Time & Date-> DST Stop Month/DST Stop Day of Week/DST Stop Day of Week Last in Month/Stop Hour of Day
local_time.dhc p_time =	0 or 1	It enables or disables the phone to update time with the offset time obtained from the DHCP server. It is only available to offset from GMT 0. 0-Disabled 1-Enabled The default value is 0.	Settings->Time & Date->DHCP Time
local_time.offs et_time =	Integer from -300 to 300	It configures the offset time (in minutes). The default value is blank.	Settings->Time & Date->Offset (minutes)
local_time.tim e_format =	0 or 1	It configures the time format. 0-12 Hour 1-24 Hour The default value is 1.	Settings->Time & Date->Time Format
local_time.dat e_format =	0, 1, 2, 3, 4, 5 or 6 (for SIPT19P/T21P /T22P/T26P/T 28P/T4X)	It configures the date format. For SIPT19P/SIPT21P/T22P/T26P/T28P/T4X IP phones: 0-WWW MMM DD	Settings->Time & Date->Date Format

Parameter	Permitted Values	Descriptions	Web Setting Path
	7, 8 or 9	1-DD-MMM-YY	
	(for SIP-T20P)	2-YYY-MM-DD	
		3-DD/MM/YYYY	
		4-MM/DD/YY	
		5-DD MMM YYYY	
		6-WWW DD MMM	
		The default value is 0.	
		For SIPT20P IP phones:	
		7-MM DD YY	
		8-DD MM YY	
		9-YY MM DD	
		The default value is 7.	
		"WWW" represents the abbreviation of	
		the week, "DD" represents a two-digit	
		day, "MMM" represents the first three	
		letters of the month, "YYYY" represents	
		a four-digit year, and "YY" represents a two-digit year which is not displayed	
		on the LCD screen of SIP-T20P IP	
		phones.	
		It enables or disables the phone to	
		provide input field of register name on	
hotdesking.st		the hot desking login wizard during	
artup_register	0 or 1	startup.	
_name_enabl e =		0 -Disabled	
		1-Enabled	
		The default value is 0.	
		It enables or disables the phone to	
		provide input field of user name on the	
hotdesking.st		hot desking login wizard during	
artup_userna	0 or 1	startup.	
me_enable =		0 -Disabled	
		1-Enabled	
		The default value is 1.	
hotdesking.st		It enables or disables the phone to	
artup_passwo	0 or 1	provide input field of password on the	
rd_enable =		hot desking login wizard during	

Parameter	Permitted Values	Descriptions	Web Setting Path
		startup. 0-Disabled 1-Enabled The default value is 1.	
hotdesking.st artup_sip_serv er_enable =	0 or 1	It enables or disables the phone to provide input field of SIP server on the hot desking login wizard during startup. 0-Disabled 1-Enabled The default value is 0.	
hotdesking.st artup_outboun d_enable =	0 or 1	It enables or disables the phone to provide input field of outbound server on the hot desking login wizard during startup. 0-Disabled 1-Enabled The default value is 0.	
hotdesking.ds skey_register_ name_enable =	0 or 1	It enables or disables the phone to provide input field of register name on the hot desking login wizard when pressing the Hot Desking key. 0-Disabled 1-Enabled The default value is 0.	
hotdesking.ds skey_usernam e_enable =	0 or 1	It enables or disables the phone to provide input field of user name on the hot desking login wizard when pressing the Hot Desking key. 0-Disabled 1-Enabled The default value is 1.	
hotdesking.ds skey_passwor d_enable =	0 or 1	It enables or disables the phone to provide input field of password on the hot desking login wizard when pressing the Hot Desking key.	

Parameter	Permitted Values	Descriptions	Web Setting Path
		0-Disabled	
		1-Enabled	
		The default value is 1.	
hotdesking.ds skey_sip_serv er_enable =	0 or 1	It enables or disables the phone to provide input field of SIP server on the hot desking login wizard when pressing the Hot Desking key. 0-Disabled 1-Enabled The default value is 0.	
hotdesking.ds skey_outboun d_enable =	0 or 1	It enables or disables the phone to provide input field of outbound server on the hot desking login wizard when pressing the Hot Desking key. 0-Disabled 1-Enabled The default value is 0.	
distinctive_rin g_tones.alert_i nfo.X.text = (X ranges from 1 to 10)	String within 32 characters	It configures the internal ringer text to the info texts contained in the Alert-Info header. Example: distinctive_ring_tones.alert_info.1.text = Family The default value is blank.	Settings->Ring-> Internal Ringer Text
distinctive_rin g_tones.alert_i nfo.X.ringer = (X ranges from 1 to 10)	Integer from 1 to 8	It configures the desired ring tones for each text. The value ranges from 1 to 8, the digit stands for the appropriate ring tone. Ring tones 6-8 are only applicable to SIPT46G and SIPT48G IP phones. The default value is 1.	Settings->Ring ->Internal Ringer File
auto_redial.en able =	0 or 1	It enables or disables the phone to automatically redial the called number when the called party is temporarily unavailable. 0-Disabled	Features->General Information->Auto Redial

Parameter	Permitted Values	Descriptions	Web Setting Path
		1-Enabled The default value is 0.	
auto_redial.int erval =	Integer from 1 to 300	It configures the interval (in seconds) for the phone to wait before redial. The default value is 10.	Features->General Information->Auto Redial Interval (1~300s)
auto_redial.ti mes =	Integer from 1 to 300	It configures the auto redial times when the called party is temporarily unavailable. The default value is 10.	Features->General Information->Auto Redial Times (1~300)
zero_touch.en able =	0 or 1	It enables or disables zero touch for the phone to perform provisioning during startup. O-Disabled 1-Enabled The default value is 0.	Settings->Auto Provision->Zero Active
zero_touch.wa it_time =	Integer from 1 to 100	It configures the duration time (in seconds) of the phone displaying the zero-sp-touch configuration screen when powered on. The default value is 5.	Settings->Auto Provision->Wait Time (1~100s)
push_xml.serv er =	IP address	It configures the IP address of the push XML server. The default value is blank.	Features->Remote Control->Push XML Server IP Address
push_xml.bloc k_in_calling =	0 or 1	It enables or disables the phone to block displaying the push XML screen when in calling status. O-Disabled 1-Enabled The default value is 0.	Features->Remote Control->Block XML In Calling
push_xml.sip_ notify =	0 or 1	It enables or disables the phone to use the push XML via SIP NOTIFY message. 0-Disabled 1-Enabled The default value is 0.	Features->Remote Control->SIP Notify

Parameter	Permitted Values	Descriptions	Web Setting Path
features.actio n_uri_limit_ip =	IP address or any	It configures the IP address of server from which the phone receives the action URI requests. Multiple IP addresses are separated by commas. If it is set to any, the phone will receive action URI requests from any server. If it is left blank, the phone will not receive action URI requests. The default value is blank.	Features->Remote Control->Action URI allow IP List
dialplan.area _code.code =	String within 16 characters	It configures the area code. The default value is blank.	Settings->Dial Plan->Area Code->Code
dialplan.area _code.min_len =	Integer from 1 to 15	It configures the minimum length of the number prefixed with the area code. The default value is 1.	Settings->Dial Plan->Area Code->Min Length (1-15)
dialplan.area _code.max_le n =	Integer from 1 to 15	It configures the maximum length of the number prefixed with the area code. The value must be larger than the minimum length. The default value is 15.	Settings->Dial Plan->Area Code->Max Length (1-15)
dialplan.area _code.line_id = (not applicable to SIPT19P IP phones)	Integer	It configures lines applying the area code. 0 to 6 (for SIP-T28P/T48G/T46G) 0 to 3 (for SIP-T26P/T22P/T42G/T41P) 0 to 2 (for SIP-T21P/T20P) Multiple line IDs are separated by commas. 0 stands for all lines. If it is left blank, the area code rule will apply to all accounts on the phone. The default value is blank.	Settings->Dial Plan->Area Code->Account
dialplan.block _out.number.X = (X ranges	String within 32 characters	It configures the block out string. The default value is blank.	Settings->Dial Plan->Block Out->BlockOut NumberX

Parameter	Permitted Values	Descriptions	Web Setting Path
from 1 to 10)			
dialplan.block _out.line_id.X = (X ranges from 1 to 10) (not applicable to SIPT19P IP phones)	Integer	It configures lines applying the block out rule. 0 to 6 (for SIP-T28P/T48G/T46G) 0 to 3 (for SIP-T26P/T22P/T42G/T41P) 0 to 2 (for SIP-T28P/T48G/T46G) Multiple line IDs are separated by commas. 0 stands for all lines. If it is left blank, the block out rule will apply to all accounts on the phone. The default value is blank.	Settings->Dial Plan->Block Out->Account
dialplan.dialn ow.rule.X = (X ranges from 1 to 100)	String within 511 characters	It configures the dial now rule. The default value is blank.	Settings->Dial Plan->Dial-now-> Rule
dialplan.dialn ow.line_id.X = (X ranges from 1 to 100) (not applicable to SIPT19P IP phones)	Integer	It configures lines applying the dial-now rule. 0 to 6 (for SIP-T28P/T48G/T46G) 0 to 3 (for SIP-T26P/T22P/T42G/T41P) 0 to 2 (for SIP-T21P/T20P) Multiple line IDs are separated by commas. 0 stands for all lines. If it is left blank, the dial-now rule will apply to all accounts on the phone. The default value is blank.	Settings->Dial Plan->Dial-now-> Account
dialplan_dialn ow.url =	URL within 511 characters	It configures the access URL of the dial-now rule file. The default value is blank.	
phone_setting .dialnow_dela y =	Integer from 1 to 14	It configures the delay time (in seconds) for the dial-now rule. The default value is 1.	Features->General Information->Time- Out for Dial-Now Rule
dialplan.repla ce.prefix.X = (X ranges from 1 to 100)	String within 32 characters	It configures the entered number to be replaced. The default value is blank.	Settings->Dial Plan->Replace Rule->Prefix

Parameter	Permitted Values	Descriptions	Web Setting Path
dialplan.repla ce.replace.X = (X ranges from 1 to 100)	String within 32 characters	It configures the alternate number to replace the entered number. The default value is blank.	Settings->Dial Plan->Replace Rule->Replace
dialplan.repla ce.line_id.X = (X ranges from 1 to 100) (not applicable to SIPT19P IP phones)	Integer	It configures lines applying the dial-now rule. 0 to 6 (for SIP-T28P/T48G/T46G) 0 to 3 (for SIP-T26P/T22P/T42G/T41P) 0 to 2 (for SIP-T21P/T20P) Multiple line IDs are separated by commas. 0 stands for all lines. The default value is blank.	Settings->Dial Plan->Replace Rule-> Account
dialplan_repl ace_rule.url =	URL within 511 characters	It configures the access URL of the replace rule file. The default value is blank.	
remote_phone book.data.X.u rl = (X ranges from 1 to 5) (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the remote phone book. The default value is blank.	Directory->Remote Phone Book-> Remote URL
remote_phone book.data.X.n ame = (X ranges from 1 to 5) (not applicable to SIPT20P IP phones)	String within 99 characters	It configures the display name of the remote phone book item. The default value is blank.	Directory->Remote Phone Book-> Display Name
remote_phone book.display_ name =	String within 99 characters	It configures the display name of the remote phone book. If it is left blank, Remote Phone Book is	

Parameter	Permitted Values	Descriptions	Web Setting Path
(not applicable to SIPT20P, SIPT42G and SIPT41P IP phones)		displayed on the LCD screen at the path Menu->Directory. The default value is blank.	
Idap.enable = (not applicable to SIPT20P and SIPT19P IP phones)	0 or 1	It enables or disables LDAP feature. 0 -Disabled 1 -Enabled The default value is 0.	Directory->LDAP-> Enable LDAP
Idap.name_filt er = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the criteria for searching the contact name attributes. Example: Idap.name_filter = ((cn=%)(sn=%)) The default value is blank.	Directory->LDAP-> LDAP Name Filter
Idap.number_f ilter = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the criteria for searching the contact number attributes. Example: Idap.number_filter = ((telephoneNumber=%)(mobile=%)(ipPhone=%)) The default value is blank.	Directory->LDAP-> LDAP Number Filter
Idap.host = (not applicable to SIPT20P and SIPT19P IP phones)	IP address or domain name	It configures the IP address or domain name of the LDAP server. The default value is blank.	Directory->LDAP-> Server Address
Idap.port = (not applicable to SIPT20P and SIPT19P IP phones)	Integer from 1 to 65535	It configures the port of the LDAP server. The default value is 389.	Directory->LDAP-> Port

Parameter	Permitted Values	Descriptions	Web Setting Path
Idap.base = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the LDAP search base which corresponds to the location of the LDAP phonebook. Example: Idap.base = dc=yealink,dc=cn The default value is blank.	Directory->LDAP-> Base
Idap.user = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the user name for accessing the LDAP server. The default value is blank.	Directory->LDAP-> Username
Idap.passwor d = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the password for accessing the LDAP server. The default value is blank.	Directory->LDAP-> Password
Idap.max_hits = (not applicable to SIPT20P and SIPT19P IP phones)	Integer from 1 to 32000	It configures the maximum of the search results returned by the LDAP server to be displayed. The default value is 50.	Directory->LDAP-> Max. Hits (1~32000)
Idap.name_at tr = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the name attributes of each record to be returned by the LDAP server. Multiple attributes are separated by spaces. Example: Idap.name_attr = sn cn The default value is blank.	Directory->LDAP-> LDAP Name Attributes
ldap.numb_at tr = (not	String within 99 characters	It configures the number attributes of each record to be returned by the LDAP server.	Directory->LDAP-> LDAP Number Attributes

Parameter	Permitted Values	Descriptions	Web Setting Path
applicable to SIPT20P and SIPT19P IP phones)		Multiple attributes are separated by spaces. Example: Idap.numb_attr = Mobile ipPhone The default value is blank.	
Idap.display_ name = (not applicable to SIPT20P and SIPT19P IP phones)	String within 99 characters	It configures the display name of the contact record displayed on the LCD screen. The value of this parameter must start with "%" symbol. Example: Idap.display_name = %cn The default value is blank.	Directory->LDAP-> LDAP Display Name
Idap.version = (not applicable to SIPT20P and SIPT19P IP phones)	2 or 3	It configures the LDAP version. The default value is 3.	Directory->LDAP-> Protocol
Idap.call_in_lo okup = (not applicable to SIPT20P and SIPT19P IP phones)	0 or 1	It enables or disables the phone to perform an LDAP search when receiving an incoming call. 0-Disabled 1-Enabled The default value is 0.	Directory->LDAP-> LDAP Lookup For Incoming Call
Idap.Idap_sort = (not applicable to SIPT20P and SIPT19P IP phones)	0 or 1	It enables or disables the phone to sort the search results in alphabetical order or numerical order. 0-Disabled 1-Enabled The default value is 0.	Directory->LDAP-> LDAP Sorting Results
features.call_n um_filter =	String within 99 characters	It configures the characters the phone filters when dialing. If the dialed number contains configured characters, the phone will automatically filter these characters	Features->General Information->Call Number Filter

Parameter	Permitted Values	Descriptions	Web Setting Path
		when dialing. The default value is "-,".	
features.dnd_r efuse_code =	404, 480 or 486	It configures a return code and reason of SIP response messages when rejecting an incoming call by DND. 404-No Found 480-Temporarily not available 486-Busy here The default value is 480.	Features->General Information-> Return Code When DND
features.norm al_refuse_cod e =	404, 480 or 486	It configures a return code and reason of SIP response messages when rejecting an incoming call. 404-No Found 480-Temporarily not available 486-Busy here The default value is 486.	Features->General Information-> Return Code When Refuse
features.call_c ompletion_en able =	0 or 1	It enables or disables call completion feature. 0-Disabled 1-Enabled The default value is 0.	Features->General Information->Call Completion
features.fwd_ mode = (not applicable to SIPT19P IP phones)	0 or 1	It configures the call forward mode. 0-Phone 1-Custom The default value is 0.	Features-> Forward&DND->Fo rward->Mode
features.dnd_ mode = (not applicable to SIPT19P IP phones)	0 or 1	It configures the DND mode. 0 -Phone 1 -Custom The default value is 0.	Features-> Forward&DND-> DND->Mode
features.dnd.e nable =	0 or 1	It enables or disables DND feature. 0 -Disabled 1 -Enabled	Features-> Forward&DND-> DND->DND Status

Parameter	Permitted Values	Descriptions	Web Setting Path
		The default value is 0.	
features.dnd.o n_code =	String within 32 characters	It configures the DND on code when the DND mode is configured as Phone. The default value is Blank.	Features-> Forward&DND-> DND->DND On Code
features.dnd.o ff_code =	String within 32 characters	It configures the DND off code when the DND mode is configured as Phone. The default value is Blank.	Features-> Forward&DND-> DND->DND Off Code
features.dnd.e mergency_en able =	0 or 1	It enables or disables the phone to receive incoming calls from authorized numbers when DND feature is enabled. 0-Disabled 1-Enabled The default value is 0.	Features-> Forward&DND-> DND Emergency
features.dnd.e mergency_aut horized_numb er =	String within 511 characters	It configures the numbers the phone will receive incoming calls from when DND feature is enabled. Multiple numbers are separated by commas. The default value is blank.	Features-> Forward&DND-> DND Authorized Numbers
features.fwd_ diversion_ena ble =	0 or 1	It enables or disables forward diversion feature. 0- Disabled 1-Enabled The default value is 1.	Features->General Information-> Diversion/History-In fo
call_waiting.e nable =	0 or 1	It enables or disables call waiting feature. 0-Disabled 1-Enabled The default value is 1.	Features->General Information->Call Waiting
call_waiting.to ne =	0 or 1	It enables or disables the phone to play the call waiting tone. 0-Disabled 1-Enabled	Features->Audio- > Call Waiting Tone

Parameter	Permitted Values	Descriptions	Web Setting Path
		The default value is 1.	
call_waiting.o n_code =	String within 32 characters	It configures the call waiting on code. The default value is blank.	Features->General Information->Call Waiting On Code
call_waiting.of f_code =	String within 32 characters	It configures the call waiting off code. The default value is blank.	Features->General Information->Call Waiting Off Code
features.interc om.allow =	0 or 1	It enables or disables the phone to automatically answer an incoming intercom call. O-Disabled 1-Enabled The default value is 1.	Features->Interco m ->Accept Intercom
features.interc om.mute =	0 or 1	It enables or disables the phone to mute the speaker when answering an intercom call. 0-Disabled 1-Enabled The default value is 0.	Features->Interco m ->Intercom Mute
features.interc om.tone =	0 or 1	It enables or disables the phone to play a warning tone when answering an intercom call. 0-Disabled 1-Enabled The default value is 1.	Features->Interco m ->Intercom Tone
features.interc om.barge =	0 or 1	It enables or disables the phone to barge in an intercom call. O-Disabled 1-Enabled The default value is 0.	Features->Interco m ->Intercom Barge
features.remot e_phonebook. enable = (not applicable to SIPT20P IP	0 or 1	It enables or disables the phone to perform a remote phone book search when receiving an incoming call. O-Disabled 1-Enabled The default value is 0.	Directory->Remote Phone Book-> Search Remote Phonebook Name

Parameter	Permitted Values	Descriptions	Web Setting Path
phones)			
features.remot e_phonebook. flash_time = (not applicable to SIPT20P IP phones)	Integer from 3600 to 2592000	It configures the interval (in seconds) for the phone to update the data of the remote phone book from the remote phone book server. The default value is 21600.	Directory->Remote Phone Book-> Search Flash Time (Seconds)
features.hotlin e_number =	String within 32 characters	It configures the hotline number. The default value is blank.	Features->General Information-> Hotline Number
features.hotlin e_delay =	Integer from 0 to 10	It configures the delay time (in seconds) for the phone to dial out the hotline number automatically. The default value of delay time is 4.	Features->General Information-> Hotline Delay (0~10s)
features.dtmf. hide =	0 or 1	It enables or disables the phone to suppress the display of DTMF digits. 0-Disabled 1-Enabled The default value is 0.	Features->General Information->Supp ress DTMF Display
features.dtmf. hide_delay = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the DTMF digits for a short period before displaying as asterisks when the parameter "features.dtmf.hide" is set to 1 (Enabled). 0-Disabled 1-Enabled The default value is 0.	Features->General Information->Supp ress DTMF Display Delay
features.dtmf.r epetition =	1, 2 or 3	It configures the repetition times for sending the DTMF packets. The default value is 3.	Features->General Information->DTM F Repetition
features.dtmf.r eplace_tran =	0 or 1	It enables or disables the phone to send DTMF sequences for transfer function when pressing the transfer soft key or the TRAN key. 0-Disabled	Features->General Information->DTM F Replace Tran

Parameter	Permitted Values	Descriptions	Web Setting Path
		1-Enabled The default value is 0.	
features.dtmf.t ransfer =	String within 32 characters	It configures DTMF sequences for transfer key function to be sent. It consists of 0-9, A-D, * and #. The default value is blank.	Features->General Information->Tran Send DTMF
features.head set_prior =	0 or 1	It enables or disables headset prior feature. 0-Disabled 1-Enabled The default value is 0.	Features->General Information-> Headset Prior
features.head set_training = (not applicable to SIPT21P and SIPT19P IP phones)	0 or 1	It enables or disables dual headset feature. 0-Disabled 1-Enabled The default value is 0.	Features->General Information->Dual- Headset
features.play_ local_dtmf_ton e_enable=	0 or 1	It enables or disables the phone to play a local DTMF tone. 0-Disabled 1-Enabled The default value is 1.	Features->General Information->Play Local DTMF Tone
features.busy_ tone_delay =	0, 3 or 5	It configures the duration time (in seconds) for the busy tone. The default value is 0.	Features->General Information->Busy Tone Delay (Seconds)
features.send _pound_key =	0 or 1	It configures whether to send one pound key by pressing the pound key twice when the pound key is configured as a send key. O-Send one pound key 1-Do not send any pound key The default value is 0.	Features->General Information->Send Pound Key
features.key_a s_send =	0, 1 or 2	It configures the "#" or "*" key as a send key.	Features->General Information->Key

Parameter	Permitted Values	Descriptions	Web Setting Path
		O-Disabled 1-# key 2-* key The default value is 1. Note: The old parameter "features.pound_key.mode" is also applicable to IP phones.	As Send
features.send _key_tone =	0 or 1	It enables or disables the phone to play key tone when pressing the send key. O-Disabled 1-Enabled The default value is 1.	Features->Audio- > Send Sound
features.key_t one =	0 or 1	It enables or disables the phone to play key tone when pressing any key. 0-Disabled 1-Enabled The default value is 1.	Features->Audio- > Key Tone
features.play_ hold_tone.ena ble =	0 or 1	It enables or disables the phone to play a warning tone when there is a call on hold. O-Disabled 1-Enabled The default value is 1.	Features->General Information->Play Hold Tone
features.play_ hold_tone.del ay =	Integer from 3 to 3600	It configures the interval (in seconds) for playing a hold warning tone. The default value is 30.	Features->General Information->Play Hold Tone Delay
features.redial _tone =	Integer within 6 digits	It configures the phone to continue to play the dial tone after inputting the preset numbers in the dialing screen. Example: features.redial_tone = 125 The phone will continue to play the dial tone after inputting "125" in the dialing screen. If it is left blank, the phone will not play	Features->Audio- >Redial Tone

Parameter	Permitted Values	Descriptions	Web Setting Path
		the dial tone after inputting numbers in the dialing screen. The default value is blank.	
features.partiti on_tone =	0 or 1	It enables or disables the phone with active accounts to play tones in the dialing screen differently from the phone with no active accounts. O-Disabled 1-Enbaled The default value is 0.	
features.pass word_dial.ena ble =	0 or 1	It enables or disables password dial feature for the phone. 0-Disabled 1-Enabled The default value is 0.	Features->General Information->PswD ial
features.pass word_dial.pref ix =	String within 32 characters	It configures the prefix of the password-dial number. For example, set the prefix to 12 and the length to 3, when you want to dial the number 123456, the entered number is displayed as 12***6 on the LCD screen. The default value is blank.	Features->General Information-> PswPrefix
features.pass word_dial.len gth =	Integer from 0 to 99	It configures the number of digits to be hidden. The hidden digits are displayed as asterisks on the LCD screen. The default value is blank.	Features->General Information-> PswLength
features.save_ call_history =	0 or 1	It enables or disables the phone to save the call history. O-Disabled 1-Enabled The default value is 1.	Features->General Information->Save Call Log
phone_setting .common_po wer_led_enab	0 or 1	It enables or disables the power indicator LED to be turned on. For SIPT2xP/T19P IP phones:	Features->Power LED->Common Power Light On

Parameter	Permitted Values	Descriptions	Web Setting Path
le =		0-Disabled (power indicator LED is off)	
		1-Enabled (power indicator LED is solid green)	
		The default value is 1.	
		For SIPT4X IP phones:	
		0 -Disabled (power indicator LED is off)	
		1-Enabled (power indicator LED is solid red)	
		The default value is 0.	
		Note: The old parameter "features.power_led_on" is also applicable to IP phones and "features.idle_talk_power_led_flash_e nable" is also applicable to SIP-T21P/T19P/T4X IP phones.	
		It enables or disables the power indicator LED to flash when the phone receives an incoming call.	
		For SIPT2xP/T19P IP phones:	
		0 -Disabled (power indicator LED does not flash)	
		1-Enabled (power indicator LED fast flashes (300ms) green)	
phone_setting		The default value is 1.	Factures > Dower
.ring_power_l	0 or 1	For SIPT4X IP phones:	Features->Power LED->Ring Power
ed_flash_ena ble =		0 -Disabled (power indicator LED does not flash)	Light Flash
		1-Enabled (power indicator LED fast flashes (300ms) red)	
		If it is set to 0, the status of the power	
		indicator LED is determined by the	
		value of the parameter "phone_setting.common_power_led_e nable".	
		The default value is 1.	
phone_setting		It enables or disables the power	Features->Power
.mail_power_l	0 or 1	indicator LED to flash when the phone	LED->Voice/Text
ed_flash_ena		receives a voice mail or a text	Mail Power Light

Parameter	Permitted Values	Descriptions	Web Setting Path
ble =		message.	Flash
		For SIPT2xP/T19P IP phones:	
		0 -Disabled (power indicator LED does not flash)	
		1-Enabled (power indicator LED slow flashes (1000ms) green)	
		The default value is 0.	
		For SIPT4X IP phones:	
		0 -Disabled (power indicator LED does not flash)	
		1-Enabled (power indicator LED slow flashes (1000ms) red)	
		The default value is 1.	
		If it is set to 0, the status of the power indicator LED is determined by the value of the parameter "phone_setting.common_power_led_e nable".	
		It enables or disables the power indicator LED to flash when a call is mute.	
		For SIPT2xP/T19P IP phones:	
		0 -Disabled (power indicator LED does not flash)	
		1-Enabled (power indicator LED fast flashes (300ms) green)	
phone_setting		The default value is 1.	Facetures & Davis
.mute_power_I	0 or 1	For SIPT4X IP phones:	Features->Power LED->Mute Power
ed_flash_ena ble =	0 01 1	0 -Disabled (power indicator LED does not flash)	Light On
		1-Enabled (power indicator LED fast flashes (300ms) red)	
		The default value is 0.	
		If it is set to 0, the status of the power indicator LED is determined by the value of the parameter "phone_setting.common_power_led_e nable".	

Parameter	Permitted Values	Descriptions	Web Setting Path
		It enables or disables the power indicator LED to flash when a call is placed on hold or is held.	
		For SIPT2xP/T19P IP phones:	
		0 -Disabled (power indicator LED does not flash)	
		1-Enabled (power indicator LED fast flashes (500ms) green)	
phone_setting		The default value is 0.	Features->Power
.hold_and_hel d_power_led_	0 or 1	For SIPT4X IP phones:	LED->Hold/Held
flash_enable		0 -Disabled (power indicator LED does not flash)	Power Light On
		1-Enabled (power indicator LED fast flashes (500ms) red)	
		The default value is 0.	
		If it is set to 0, the status of the power indicator LED is determined by the value of the parameter	
		"phone_setting.common_power_led_e nable".	
		It enables or disables the power indicator LED to be turned on when the	
		phone is busy.	
		For SIPT2xP/T19P IP phones:	
		0 -Disabled (power indicator LED is off)	
phone_setting		1-Enabled (power indicator LED is solid green)	
.talk_and_dial	0 or 1	The default value is 1.	Features->Power
_power_led_e	0 01 1	For SIPT4X IP phones:	LED->Talk/Dial Power Light On
nable =		0 -Disabled (power indicator LED is off)	G
		1-Enabled (power indicator LED is solid red)	
		The default value is 0.	
		If it is set to 0, the status of the power	
		indicator LED is determined by the value of the parameter	
		"phone_setting.common_power_led_e	

Parameter	Permitted Values	Descriptions	Web Setting Path
		nable".	
features.relog _offtime =	Integer from 1 to 1000	It configures the web access timeout (in minutes). The default value is 5. It takes effect after a reboot.	Features->General Information->Auto- Logout Time (1~1000min)
features.direct _ip_call_enabl e =	0 or 1	It enables or disables the phone to make an IP call directly. O-Disabled 1-Enabled The default value is 1.	Features->General Information-> Allow IP Call
features.allow _mute =	0 or 1	It enables or disables the phone to mute an active call. O-Disabled 1-Enabled The default value is 1.	Features->General Information->Allo w Mute
features.grou p_listen_in_tal king_enable =	0 or 1	It enables or disables the phone to enter into the group listening mode by pressing the speakerphone key when it is in talking using the handset. 0-Disabled 1-Enabled The default value is 1.	
features.ringer _device.is_use _headset =	0, 1 or 2	It configures the ringer device for the phone in the headset mode. 0-Use Speaker 1-Use Headset 2-Use Headset & Speaker The default value is 0.	Features->Audio- > Ringer Device for Headset
features.factor y_pwd_enabl e =	0 or 1	It enables or disables the phone to prompt for the administrator password when you long press the OK key to perform factory reset. 0-Disabled 1-Enabled The default value is 0.	

Parameter	Permitted Values	Descriptions	Web Setting Path
features.expor t_cfg_erase_p wd =	0 or 1	It configures the phone to export the configuration file with what type of password. 0-Encrypted 1-Blank 2-Plaintext The default value is 1.	
features.picku p.group_picku p_enable = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the GPickup soft key when the phone is in the pre-dialing screen. 0-Disabled 1-Enabled The default value is 0.	Features->Call Pickup->Group Call Pickup
features.picku p.group_picku p_code =	String within 32 characters	It configures the group call pickup code. The default value is blank.	Features->Call Pickup->Group Call Pickup Code
features.picku p.direct_picku p_enable = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the DPickup soft key when the phone is in the pre-dialing screen. 0-Disabled 1-Enabled The default value is 0.	Features->Call Pickup->Directed Call Pickup
features.picku p.direct_picku p_code =	String within 32 characters	It configures the directed call pickup code. The default value is blank.	Features->Call Pickup->Directed Call Pickup Code
features.picku p.blf_visual_e nable = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display a visual alert when the monitored user receives an incoming call. 0-Disabled 1-Enabled The default value is 0.	Features->Call Pickup->Visual Alert for BLF Pickup
features.picku p.blf_audio_e nable =	0 or 1	It enables or disables the phone to play an audio alert when the monitored user receives an incoming	Features->Call Pickup->Audio Alert for BLF Pickup

Parameter	Permitted Values	Descriptions	Web Setting Path
		call.	
		0 -Disabled	
		1-Enabled	
		The default value is 0.	
features.blf_le d_mode =		It configures BLF LED mode and provides four kinds of definition for the BLF/BLF list key LED status.	
(not		For more information, refer to BLF LED	Features->General
applicable to	0, 1, 2 or 3	Mode on page 228.	Information->BLF
SIP-T19P and		The default value is 0.	LED Mode
SIPT48G IP phones)		Note: The old parameter	
phonocy		"features.blf_and_callpark_idle_led_e nable" is also applicable to IP phones.	
		It enables or disables the phone to	
		deal with the Version header in the BLF	
features.blf_lis		NOTIFY message sent by the server.	
t_version =	0 or 1	0 -Disabled	
		1-Enabled	
		The default value is 0.	
		It takes effect after a reboot.	
		It enables or disables the phone to	
features.voice		play the warning tone when receiving a voice mail.	
_mail_tone_en	0 or 1	0 -Disabled	
able =		1-Enabled	
		The default value is 1.	
	PCMU		
	PCMA		
multicast.code	G729	It configures the codec of multicast	
c =	G727	paging.	
(not applicable to SIPT19P IP	G726-16	Codecs G726-16, G726-24 and	Features->General Information->
	G726-10	G726-40 are not applicable to SIP-T21P	Multicast Codec
	G726-32	IP phones.	
phones)	G726-40	The default value is G722.	
	G723_53		
	G/25_55		

Parameter	Permitted Values	Descriptions	Web Setting Path
multicast.recei ve_priority.en able =	0 or 1	It enables or disables the phone to handle the incoming multicast paging calls when there is a multicast paging call on the phone. 0-Disabled 1-Enabled The default value is 1.	Directory-> Multicast IP-> Paging Priority Active
multicast.recei ve_priority.pri ority =	Integer from 0 to 10	It configures the priority of multicast paging calls. 1 is the highest priority, 10 is the lowest priority. The default value is 10.	Directory-> Multicast IP->Paging Barge
multicast.listen _address.X.ip_ address = (X ranges from 1 to 10)	IP address: port	It configures the listening multicast IP address and port number for the phone. Example: multicast.listen_address.1.ip_address = 224.5.6.20:10008 The default value is blank.	Directory-> Multicast IP-> Listening Address
multicast.listen _address.X.la bel = (X ranges from 1 to 10)	String within 99 characters	It configures the label displayed on the LCD screen when receiving the multicast paging. The default value is blank.	Directory-> Multicast IP->Label
phone_setting .predial_auto dial =	0 or 1	It enables or disables the phone to automatically dial out the entered digits in the pre-dialing screen. O-Disabled 1-Enabled The default value is 0.	Settings->Preferen ce->Live Dialpad
phone_setting .inter_digit_tim e =	Integer from 1 to 14	It configures the time (in seconds) for the phone to automatically dial out the entered digits without pressing a send key. The default value is 4.	Settings-> Preference->Inter Digit Time (1~14s)

Parameter	Permitted Values	Descriptions	Web Setting Path
phone_setting .lock = (only applicable to SIPT2xP and SIPT19P IP phones)	0, 1, 2 or 3	It configures the keypad lock type. 0-Disabled 1-Menu Key 2-Function Keys 3-All Keys The default value is 0.	Features->Phone Lock->Keypad Lock Type
phone_setting .phone_lock.e nable = (only applicable to SIP- T46G, T42G and T41P IP phones)	0 or 1	It enables or disables keypad lock feature. 0 -Disabled 1 -Enabled The default value is 0.	Features->Phone Lock->Keypad Lock Enable
phone_setting .phone_lock.lo ck_key_type = (only applicable to SIP-T46G, T42G and T41P IP phones)	0, 1 or 2	It configures the keypad lock type. 0-All Keys 1-Function Keys 2-Menu Key The default value is 0.	Features->Phone Lock->Keypad Lock Type
phone_setting .phone_lock.u nlock_pin = (not applicable to SIPT48G IP phones)	characters within 15 digits	It configures the password for unlocking the keypad. The default value is 123.	Features->Phone Lock->Phone Unlock PIN (0~15 Digit)
phone_setting .phone_lock.lo ck_time_out = (not applicable to SIPT48G IP phones)	Integer from 0 to 3600	It configures the interval (in seconds) to automatically lock the keypad. The default value is 0 (the keypad is locked only by long pressing the pound key or pressing the keypad lock key)	Features->Phone Lock->Phone Lock Time Out (0~3600s)

Parameter	Permitted Values	Descriptions	Web Setting Path
features.bluet ooth_enable = (only applicable to SIP-48G and SIPT46G IP phones)	0 or 1	It enables or disables Bluetooth feature. 0 -Disabled 1 -Enabled The default value is 0.	Features-> Bluetooth-> Bluetooth Active
phone_setting .ring_type =	Ring1.wav, Ring2.wav, Ring3.wav, Ring4.wav, Ring5.wav Ring6.wav, Ring7.wav, Ring8.wav	It configures the ring tone for the phone. Example: phone_setting.ring_type = Ring1.wav The default value is Ring1.wav. Note: Ring tones 6-8 are only applicable to SIP-T48G and SIP-T46G IP phones.	Settings->Preferen ce->Ring Type
phone_setting .contrast = (only applicable to SIPT28P, SIPT21P, SIP-19P IP phones, EXP39 connected to SIPT28P and SIPT26P IP phones, and EXP40 connected to SIPT48G/T46G IP phones)	Integer from 1 to 10	It configures the contrast of the LCD screen. For SIPT21P and SIPT19P IP phones, it configures the LCD's contrast of the phone only. For SIPT26P IP phones, it configures the LCD's contrast of the connected EXP39 only. For SIPT28P IP phones, it configures the LCD's contrast of the IP phone and the connected EXP39. For SIPT48G/T46G IP phones, it configures the LCD's contrast of the connected EXP40 only. The default value is 6.	Settings->Preferen ce->Contrast
phone_setting .lcd_logo.mod e = (not applicable to SIPT48G/T46G	0, 1 or 2	It configures the logo mode of the LCD screen (except for SIP-T20P IP phones). For SIP-T26P/T22P/T21P/T19P/ T42G/T41P IP phones: 0-Disabled 1-System logo	Features->General Information->Use Logo

Parameter	Permitted Values	Descriptions	Web Setting Path
IP phones)	Values	2-Custom logo The default value is 0. For SIP-T28P IP phones: 1-System logo 2-Custom logo The default value is 1. It enables or disables a text logo (for SIP-T20P IP phones). 0-Disabled	
phone_setting .lcd_logo.text = (only	String within	1-Enabled The default value is 0. It configures a text logo.	Features->General
applicable to SIPT20P IP phones) Icd_logo.url =	characters	The default value is Yealink.	Logo
(not applicable to SIP-T20P and SIP-T48G/T46G IP phones)	URL within 511 characters	It configures the access URL of logo file. The default value is blank.	Features->General Information-> Upload Logo
Icd_logo.delet e = (not applicable to SIPT20P and SIPT48G/T46G IP phones)	URL within 511 characters	It deletes all custom logo files. The valid value is: http://localhost/all The default value is blank.	
phone_setting .active_backli ght_level = (only applicable to SIPT28P IP	Integer from 1 to 3 for SIPT28P and the connected EXP39 IP	It configures the level of the active backlight intensity. For SIP-T28P IP phones and the connected EXP39: The default value is 2. For SIP-T48G IP phones and the	Settings->Preferen ce->Backlight Active Level

Parameter	Permitted Values	Descriptions	Web Setting Path
phones and the connected EXP39, and SIPT48G/T46G IP phones and the connected EXP40)	phones. Integer from 1 to 10 for SIP-T48G/ T46G IP phones and the connected EXP40	connected EXP40: The default value is 10. For SIP-T46G IP phones and the connected EXP40: The default value is 8.	
phone_setting .inactive_back light_level = (only applicable to SIPT48G/T46G IP phones)	0 or 1	It configures the phone to go out or reduce intensity of the backlight on the LCD screen after a period of inactivity. 0-Off 1-Low The default value is 1.	Settings-> Preference-> Backlight Inactive Level
phone_setting .backlight_tim e = (not applicable to SIPT21P, SIPT20P and SIPT19P IP phones)	0, 1, 15, 30, 60, 120, 300, 600 or 1800	It configures the backlight time (in seconds). For SIP-T28P/T26P/T22P IP phones: 0-Always off 1-Always on 15-15s 30-30s 60-60s 120-120s 300-300s 600-600s The default value is 30. For SIP-T4X IP phones: 0-Always on 15-15s 30-30s 60-60s 120-120s 300-300s 600-600s	Settings->Preferen ce->Backlight Time (seconds)

Parameter	Permitted Values	Descriptions	Web Setting Path
		1800 -1800s	
		The default value is 0.	
phone_setting .ring_for_tranf ailed =	Ring1.wav Ring2.wav Ring3.wav Ring4.wav Ring5.wav Ring6.wav Ring7.wav Ring8.wav	It configures the ring tone when the phone fails to transfer a call. The default value is Ring1.wav. Note: Ring tones 6-8 are applicable to SIP-T48G/T46G IP phones only.	
phone_setting .logon_wizard =	0 or 1	It enables or disables the phone to provide the logon wizard during startup. O-Disabled 1-Enabled The default value is 0.	Features->General Information-> Logon Wizard
phone_setting .is_deal180 =	0 or 1	It enables or disables the phone to deal with the 180 SIP message received after the 183 SIP message. O-Disabled 1-Enabled The default value is 1.	Features->General Information->180 Ring Workaround
phone_setting .headsetkey_ mode =	0 or 1	It configures headset mode precedence during a call. O-Always use (pressing the Speakerphone key and picking up the handset are not effective when the headset mode is activated) 1-Use as normal The default value is 1.	
phone_setting .emergency.n umber = (not applicable to SIPT48G IP	String within 99 characters	It configures emergency numbers. Multiple emergency numbers are separated by commas. The default value is 110,911,120.	Features->Phone Lock->Emergency

Parameter	Permitted Values	Descriptions	Web Setting Path
phones)			
phone_setting .show_code40 3 =	String within 99 characters	It configures the display message on the LCD screen when receiving a 403 message. If it is left blank, the phone will display the value sent from the server when receiving the 403 message. The default value is blank. It takes effect after a reboot.	
super_search. recent_call = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables recent call in dialing feature. If it is enabled, you can see the placed calls list when the phone is in the pre-dialing screen. 0-Disabled 1-Enabled The default value is 0.	Directory-> Setting->Recent Call In Dialing
directory_setti ng.url = (not applicable to SIP-T20P IP phones)	URL within 511 characters	It configures the access URL of the custom directory list file. The default value is blank.	Directory-> Setting->Directory
super_search. url = (not applicable to SIPT20P IP phones)	URL within 511 characters	It configures the access URL of the custom search source list in dialing file. The default value is blank.	Directory-> Setting->Search Source List In Dialing
firmware.url =	URL within 511 characters	It configures the access URL of the firmware file. The default value is blank. It takes effect after a reboot.	Settings-> Upgrade->Select and Upgrade Firmware
ringtone.url =	URL within 511 characters	It configures the access URL of the custom ring tone file. The default value is blank.	Settings-> Preference-> Upload Ringtone
ringtone.delet	URL within 511	It deletes all custom ring tone files.	

Parameter	Permitted Values	Descriptions	Web Setting Path
e =	characters	The valid value is: http://localhost/all	
		The default value is blank.	
gui_lang.url =	URL within 511 characters	It configures the access URL of the custom language file. The default value is blank.	
gui_lang.delet e =	URL within 511 characters	It deletes all custom language files. The valid value is: http://localhost/all The default value is blank.	
trusted_certific ates.url =	URL within 511 characters	It configures the access URL of the custom trusted certificate file. The default value is blank.	Security->Trusted Certificates->Load trusted certificates file
trusted_certific ates.delete =	URL within 511 characters	It deletes all uploaded trusted certificate files. The valid value is: http://localhost/all The default value is blank.	
server_certific ates.url =	URL within 511 characters	It configures the access URL of the custom server certificate file. The default value is blank.	Security->Server Certificates->Load server cer file
server_certific ates.delete =	URL within 511 characters	It deletes the uploaded server certificate file. The valid value is: http://localhost/all The default value is blank.	
local_contact. data.url =	URL within 511 characters	It configures the access URL of the local contact file. The default value is blank.	Directory->Local Directory->Import Local Directory File
local_contact. data_photo_t ar.url = (only applicable to SIPT48G/T46G IP phones)	URL within 511 characters	It configures the access URL of the TAR file compressed from custom contact file and avatar TAR file. All avatars needed for contacts should be tarred in advance. The default value is blank.	
auto_dst.url =	URL within 511	It configures the access URL of the DST Time file.	

Parameter	Permitted Values	Descriptions	Web Setting Path
	characters	The default value is blank.	
custom_factor y_configuratio n.url =	URL within 511 characters	It configures the access URL of the custom factory configuration files. The default value is blank. It takes effect after a reboot.	
features.custo m_factory_con fig.enable =	0 or 1	It enables or disables Import Factory Configuration feature. 0-Disabled 1-Enabled The default value is 0.	
configuration. url =	URL within 511 characters	It configures the access URL for the custom configuration files. The default value is blank. It takes effect after a reboot.	Settings-> Configuration-> Export or Import Configuration
custom_mac_ cfg.url =	URL within 511 characters	It configures the access URL of the custom MAC-Oriented CFG file. The default value is blank.	
account.X.out _dialog_blf_en able = (SIP-T19P: X=1 SIP-T21P/T20P: X ranges from 1 to 2. SIP-T26P/T22P/T 42G/T41P: X ranges from 1 to 3. SIP-T28P/T48G/ T46G: X ranges from 1 to 6)	0 or 1	It enables or disables the phone to handle NOTIFY messages out of the BLF dialog for account X. 0-Disabled 1-Enabled The default value is 0.	
account.X.ena ble = (SIP-T28P/T48G /T46G: X ranges from 1	0 or 1	It enables or disables the account X. 0 -Disabled 1 -Enabled The default value is 0.	Account->Register ->Line Active

to 6. SIRT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIRT21P/T20P: X ranges from 1 to 6. SIRT21P/T20P: X ranges from 1 to 6. SIRT26P/T22P/ T42G/T41P: X ranges from 1 to 6. SIRT21P/T20P: X ranges from 1 to 6. SIRT21P/T20P: X ranges from 1 to 3. SIRT21P/T20P: X ranges from 1 to 6. SIRT26P/T22P/ T42G/T41P: X ranges from 1 to 2. SIRT19P: X=1) account.X.dis play_name = (SIRT28P/T48G T44G: X ranges from 1 to 6. String within 99 characters the default value is blank. Account.>Register ->Label Account.>Register ->Label It configures the label displayed on the LCD screen for account X. The default value is blank. Account.>Register ->Label Account.>Register ->Label Account.>Register ->Label Account.>Register ->Label T46G: X ranges from 1 to 6. SIRT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIRT21P/T20P: X ranges from 1 to 3. SIRT21P/T20P: X ranges from 1 to 3. SIRT21P/T20P: X ranges from 1	Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.dis play_name = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from	SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1) account.X.lab el = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from	String within 99	LCD screen for account X.	_
1 to 2. SIPT19P: X=1) account.X.aut String within It configures the user name for register Account->Register	account.X.dis play_name = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	99 characters	account X. The default value is blank.	->Display Name

Parameter	Permitted Values	Descriptions	Web Setting Path
h_name = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	99 characters	authentication for account X. The default value is blank.	->Register Name
account.X.use r_name = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	String within 32 characters	It configures the register user name for account X. The default value is blank.	Account->Register ->User Name
account.X.pas sword = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from	String within 99 characters	It configures the password for register authentication for account X. The default value is blank.	Account->Register ->Password

Parameter	Permitted Values	Descriptions	Web Setting Path
1 to 2. SIPT19P: X=1) account.X.tran sport = (SIPT28P/T48G /T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1)	Integer	It configures the transport type for account X. 0-UDP 1-TCP 2-TLS 3-DNS-NAPTR The default value is 0.	Account->Register ->Transport
account.X.rere gister_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	0 or 1	It configures whether the phone needs to re-register the account when encountering an INVITE failover, if the SIP server is configured with a domain name for account X. 0-Do not need to re-register 1-Need to re-register The default value is 0.	
account.X.nap tr_build = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/	0 or 1	It configures the way of SRV query when there is no result from the NAPTR query for account X. 0-SRV query using UDP only 1-SRV query using UDP, TCP or TLS. The default value is 0.	

Parameter	Permitted Values	Descriptions	Web Setting Path
T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1) account.X.fall back.redunda ncy_type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	0 or 1	It configures the registration mode for account X. O-Concurrent registration 1-Successive registration The default value is 0.	
account.X.fall back.timeout = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	Integer from 10 to 2147483647	It configures the time interval (in seconds) for the phone to detect whether the working server is available by sending the registration request for account X. It is only applicable to successive registration mode. The default value is 120.	

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.sip_server.Y.address = (SIP-T28P/T48G/T46G: X) ranges from 1 to 6. SIP-T26P/T22P/T42G/T41P: X) ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1 Y ranges from 1 to 2.)	IP address or domain name	It configures the IP address or domain name of the SIP server Y for account X. Example: account.1.sip_server.1.address = 10.2.1.128 The default value is blank. Note: The old parameter "account.X.sip_server_host" is also applicable to IP phones.	Account->Register ->SIP Server Y-> Server Host
account.X.sip_server.Y.port = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1 Y ranges from 1 to 2)	Integer from 0 to 65535	It configures the port of SIP server Y for account X. The default value is 5060. Note: The old parameter "account.X.sip_server_port" is also applicable to IP phones.	Account->Register ->SIP Server Y-> Port
account.X.sip_ server.Y.expire s = (SIP-T19P: X=1 SIP-T20P/T21P:	Integer from 30 to 2147483647	It configures the registration expiration time (in seconds) to SIP server Y for account X. The default value is 3600.	Account->Register ->SIP Server Y-> Server Expires

Parameter	Permitted Values	Descriptions	Web Setting Path
X ranges from 1 to 2. SIPT22P/T26P/T 41P/T42G: X ranges from 1 to 3. SIPT28P/T46G/ T48G: X ranges from 1 to 6. Y ranges from 1 to 2) account.X.sip_ server.Y.retry_			
server.Y.retry_counts = (SIP-T28P/T48G)/T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1 Y ranges from 1 to 2)	Integer from 0 to 20	It configures the times for the phone to retransmit the request when the SIP server Y is unavailable or there is no response from the SIP server Y for account X. The default value is 3.	Account->Register ->SIP Server Y ->Server Retry Counts
account.X.sip_server.Y.failback_mode = (SIP-T28P/T48G/T46G: X ranges from 1 to 6. SIP-T26P/T22P/T42G/T41P: X ranges from 1	0, 1, 2 or 3	It configures the mode for the phone to retry the primary server in failover mode for account X. O-newRequests: all requests are forwarded to the primary server first, regardless of the last used server. 1-DNSTTL: the phone retries to use the primary server after the timeout of the DNSTTL configured for the SIP server. 2-Registration: the phone retries to use	

Parameter	Permitted Values	Descriptions	Web Setting Path
to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1. Y ranges from 1 to 2) account.X.sip_		the primary server when the SIP server's registration requires renewal. 3-duration: the phone retries to use the primary server after the timeout defined by the parameter "account.X.failback_timeout". The default value is 0.	
server.Y.failba ck_timeout = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Integer 0, from 60 to 65535	It configures the timeout (in seconds) for the phone to retry to use the primary server after failing over to the current working server for account X when the parameter "account.X.sip_server.Y.failback_mode" is set to 3 (duration). If you set the parameter between 1 and 59, the timeout will be 60 seconds. The default value is 3600.	
account.X.sip_server.Y.regist er_on_enable = (SIP-T19P: X=1 SIP-T20P/T21P: X ranges from 1 to 2. SIP-T22P/T26P/T41P/T42G: X ranges from 1 to 3. SIP-T28P/T46G/T48G: X ranges from 1	0 or 1	It enables or disables the phone to send registration requests to the secondary server for account X when encountering a failover. 0-Disabled 1-Enabled The default value is 0.	

Parameter	Permitted Values	Descriptions	Web Setting Path
to 6. Y ranges from 1 to 2.)			
account.X.stat ic_cache_pri = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It configures whether preferentially to use the DNS static cache for domain name resolution of the SIP server for account X. O-Use domain name server preferentially 1-Use DNS static cache preferentially The default value is 0.	
account.X.dns _cache_type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0, 1 or 2	It configures the content that the DNS static cache records for account X. O-Do not use DNS static cache. 1-Use DNS static cache, but do not record the additional records. 2-Use DNS static cache and record the additional records. The default value is 1.	
account.X.dns _cache_a.Y.na me = (SIP-T28P/T48G /T46G: X	Domain name	It configures the domain name of A record Y in the DNS static cache for account X. The default value is blank. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1. Y ranges from 1 to 2.) account.X.dns			
_cache_a.Y.ip = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	IP address	It configures the IP address that the domain name of A record Y maps to in the DNS static cache for account X. The default value is blank. It takes effect after a reboot.	
account.X.dns _cache_a.Y.ttl = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1	Integer from 30 to 2147483647	It configures the time interval that A record Y may be cached before the record should be consulted again for account X. The default value is 300. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1. Y ranges from 1 to 2.) account.X.dns _cache_srv.Y.n ame = (SIPT28P/T48G /T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1. Y ranges from	Domain name	It configures the domain name of SRV record Y in the DNS static cache for account X. It takes effect after a reboot.	
1 to 2.) account.X.dns _cache_srv.Y.p ort = ((SIPT28P/T48 G/T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2.	Integer from 0 to 65535	It configures the port to be used in SRV record Y for account X. The default value is 0. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T19P: X=1.			
Y ranges from 1 to 2.)			
account.X.dns _cache_srv.Y.p riority = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Integer from 0 to 65535	It configures the priority for the specific host entry in SRV record Y for account X. Lower priority is more preferred. The default value is 0. It takes effect after a reboot.	
account.X.dns _cache_srv.Y.t arget = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Domain name	It configures the actual host for an A query for account X. The default value is blank. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.dns _cache_srv.Y. weight = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Integer from 0 to 65535	It configures the weight of SRV record Y for account X. When priorities are equal, weight is used to differentiate the preference. Higher weight is more preferred. The default value is 0. It takes effect after a reboot.	
account.X.dns _cache_srv.Y.tt I = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Integer from 30 to 2147483647	It configures the time interval that SRV record Y may be cached before the record should be consulted again for account X. The default value is 300. It takes effect after a reboot.	
account.X.dns _cache_naptr. Y.name = (SIP-T28P/T48G /T46G: X	Domain name	It configures the domain name to which NAPTR record Y refers in the DNS static cache for account X. The default value is blank.	

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)		It takes effect after a reboot.	
account.X.dns _cache_naptr. Y.flags = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	S, A, U or P	It configures the flag of NAPTR record Y in the DNS static cache for account X. (Always "s" for SIP, which means to do an SRV lookup on whatever is in the replacement field) S-Do an SRV lookup next. A-Do an A lookup next. U-No need to do a DNS query next. P-Service customized by the user The default value is blank. It takes effect after a reboot.	
account.X.dns _cache_naptr. Y.order = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1	Integer from 0 to 65535	It configures the order of NAPTR record Y for account X. NAPTR record with lower order is more preferred. The default value is 0. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1. Y ranges from 1 to 2.) account.X.dns _cache_naptr. Y.preference = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Integer from 0 to 65535	It configures the preference of NAPTR record Y for account X. NAPTR record with lower preference is more preferred. The default value is 0. It takes effect after a reboot.	
account.X.dns _cache_naptr. Y.replace = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.	Domain name	It configures a domain name to be used for the next SRV query in NAPTR record Y for account X. The default value is blank. It takes effect after a reboot.	

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT19P: X=1. Y ranges from 1 to 2.)			
account.X.dns _cache_naptr. Y.service = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	String within 32 characters	It configures the transport protocol available for SIP in NAPTR record Y for account X. The default value is blank. It takes effect after a reboot.	
account.X.dns _cache_naptr. Y.ttl = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 2.)	Integer from 30 to 2147483647	It configures the time interval that NAPTR record Y may be cached before the record should be consulted again for account X. The default value is 300. It takes effect after a reboot.	
account.X.srv_	0 or 1	It enables or disables the phone to	

Parameter	Permitted Values	Descriptions	Web Setting Path
ttl_timer_enab le = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)		refresh the DNS-SRV query record at the regular time for account X. 0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	
account.X.out bound_proxy_ enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to use the outbound proxy server for account X. O-Disabled 1-Enabled The default value is 0.	Account->Register ->Enable Outbound Proxy Server
account.X.out bound_host = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3.	IP address or domain name	It configures the IP address or domain name of the outbound proxy server for account X. The default value is blank.	Account->Register ->Outbound Proxy Server

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.out bound_port = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 0 to 65535	It configures the port of the outbound proxy server for account X. The default value is 5060.	Account->Register ->Outbound Proxy Server->Port
voice_mail.nu mber.X = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	String within 99 characters	It configures the voice mail number for account X. The default value is blank.	Account-> Advanced->Voice Mail
account.X.pro xy_require = (SIP-T28P/T48G /T46G: X ranges from 1 to 6.	String within 256 characters	It configures the proxy server for account X. The default value is blank.	Account->Basic-> Proxy Require

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.) account.X.sip_ trust_ctrl = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.	0 or 1	It enables or disables the phone to only accept the message from the trusted server for account X. 0-Disabled 1-Enabled The default value is 0.	
SIPT19P: X=1.) account.X.ano nymous_call = (SIPT28P/T48G /T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)	0 or 1	It enables or disables anonymous call feature for account X. 0-Disabled 1-Enabled The default value is 0.	Account->Basic-> Local Anonymous
account.X.sen d_anonymous	0 or 1	It configures the phone to send anonymous on/off code to	Account->Basic-> Send Anonymous

Parameter	Permitted Values	Descriptions	Web Setting Path
_code = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)		activate/deactivate the server-side anonymous call feature for account X. 0-Off Code 1-On Code The default value is 0.	Code
account.X.ano nymous_call_o ncode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	String within 32 characters	It configures the code for activating the server-side anonymous call feature for account X when the parameter "account.X.send_anonymous_code" is set to 1 (On Code). The default value is blank.	Account->Basic-> Anonymous Call-> On Code
account.X.ano nymous_call_o ffcode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3.	String within 32 characters	It configures the code for deactivating the server-side anonymous call feature for account X when the parameter "account.X.send_anonymous_code" is set to 0 (Off Code). The default value is blank.	Account->Basic-> Anonymous Call-> Off Code

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.reje ct_anonymous _call = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables anonymous call rejection feature for account X. 0-Disabled 1-Enabled The default value is 0.	Account->Basic-> Anonymous Call Rejection
account.X.ano nymous_reject _oncode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	String within 32 characters	It configures the code for activating the server-side anonymous call rejection feature for account X. The default value is blank.	Account->Basic-> Anonymous Call Rejection->On Code
account.X.ano nymous_reject _offcode = (SIP-T28P/T48G	String within 32 characters	It configures the code for deactivating the server-side anonymous call rejection feature for account X. The default value is blank.	Account->Basic-> Anonymous Call Rejection->Off Code

Parameter	Permitted Values	Descriptions	Web Setting Path
/T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.dnd .enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	0 or 1	It enables or disables DND feature for account X when the DND mode is configured as Custom. 0-Disabled 1-Enabled The default value is 0.	Features->Forwar d& DND->DND ->DND Status
account.X.dnd .on_code = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P:	String within 32 characters	It configures the DND on code for account X when the DND mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->DND On Code

Parameter	Permitted Values	Descriptions	Web Setting Path
X ranges from 1 to 2.) (not applicable to SIPT19P IP phones)			
.off_code = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	String within 32 characters	It configures the DND off code for account X when the DND mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->DND Off Code
account.X.alw ays_fwd.enab le = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to	0 or 1	It enables or disables always forward feature for account X when the call forward mode is configured as Custom. 0-Disabled 1-Enabled The default value is 0.	Features->Forwar d& DND->Always Forward->On/Off

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT19P IP phones)			
account.X.alw ays_fwd.targe t = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	String within 32 characters	It configures the target number the phone forwards all incoming calls to for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->Always Forward->Target
account.X.bus y_fwd.enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	0 or 1	It enables or disables busy forward feature for account X when the call forward mode is configured as Custom. 0-Disabled 1-Enabled The default value is 0.	Features->Forwar d& DND->Busy Forward->On/Off

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.bus y_fwd.target = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	String within 32 characters	It configures the target number the phone forwards incoming calls to when busy for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->Busy Forward->Target
account.X.tim eout_fwd.ena ble = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	0 or 1	It enables or disables no answer forward feature for account X when the call forward mode is configured as Custom. 0-Disabled 1-Enabled The default value is 0.	Features->Forwar d& DND->No Answer Forward-> On/Off
account.X.tim eout_fwd.targ et =	String within 32 characters	It configures the target number the phone forwards incoming calls to after a period of ring time for account X	Features->Forwar d& DND->No Answer Forward->

Parameter	Permitted Values	Descriptions	Web Setting Path
(SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)		when the call forward mode is configured as Custom. The default value is blank.	Target
account.X.tim eout_fwd.time out = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	Integer from 0 to 20	It configures ring times (N) to wait before forwarding incoming calls for account X when the call forward mode is configured as Custom. Incoming calls are forwarded when not answered after N*6 seconds. The default value is 2.	Features->Forwar d& DND->No Answer Forward-> After Ring Time (0~120s)
account.X.alw ays_fwd.off_c ode = (SIP-T28P/T48G /T46G: X ranges from 1	String within 32 characters	It configures the always forward off code for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->Always Forward ->Off Code

Parameter	Permitted Values	Descriptions	Web Setting Path
to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)			
account.X.alw ays_fwd.on_c ode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	String within 32 characters	It configures the always forward on code for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->Always Forward->On Code
account.X.bus y_fwd.off_cod e = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X	String within 32 characters	It configures the busy forward off code for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->Busy Forward ->Off Code

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones) account.X.bus y_fwd.on_cod e = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3.	String within 32 characters	It configures the busy forward on code for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->Busy Forward->On Code
SIPT21P/T20P: X ranges from 1 to 2.) (not applicable to SIPT19P IP phones)			
account.X.tim eout_fwd.off_c ode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P:	String within 32 characters	It configures the no answer forward off code for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->No Answer Forward ->Off Code

Parameter	Permitted Values	Descriptions	Web Setting Path
X ranges from 1 to 2.) (not applicable to SIPT19P IP phones)			
account.X.tim eout_fwd.on_c ode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	String within 32 characters	It configures the no answer forward on code for account X when the call forward mode is configured as Custom. The default value is blank.	Features->Forwar d& DND->No Answer Forward ->On Code
account.X.sip_ listen_port = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 1024 to 65535	It configures the local SIP port for account X. The default value is 5060.	Account-> Advanced->Local SIP Port

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.100 rel_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the 100 reliable retransmission feature for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced-> Retransmission
account.X.sub scribe_mwi = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to subscribe the message waiting indicator for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced-> Subscribe for MWI
account.X.sub scribe_mwi_e xpires = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3.	Integer from 0 to 84600	It configures the interval (in seconds) of MWI subscription for account X. The default value is 3600.	Account-> Advanced->MWI Subscription Period (Seconds)

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.cid_source = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0, 1, 2, 3, 4 or 5	It configures the source caller identity for presentation when receiving an incoming call for account X. 0-FROM 1-PAI 2-PAI-FROM 3-PRID-PAI-FROM 4-PAI-RPID-FROM, 5-RPID-FROM The default value is 0.	Account-> Advanced->Caller ID Source
account.X.cid_source_privac y = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to deal with PRIVACY header field in the 180 or 200 OK message for account X. 0-Disabled 1-Enabled The default value is 1.	
account.X.cid_ source_ppi = (SIP-T28P/T48G /T46G: X ranges from 1	0 or 1	It enables or disables the phone to process the P-Preferred-Identity header for caller identity presentation when receiving an incoming call for account X.	

Parameter	Permitted Values	Descriptions	Web Setting Path
to 6.		0 -Disabled	
SIP-T26P/T22P/		1-Enabled	
T42G/T41P: X		The default value is 1.	
ranges from 1			
to 3.			
SIP-T21P/T20P: X ranges from			
1 to 2.			
SIP-T19P: X=1.)			
account.X.cp_			
source =			
(SIP-T28P/T48G			
/T46G: X		It configures the source callee identity	
ranges from 1 to 6.		for presentation for account X.	
SIP-T26P/T22P/		0 -PAI-RPID	
T42G/T41P: X	0, 1 or 2	1-Dialed Digits	
ranges from 1		2 -RFC4916	
to 3.		The default value is 0.	
SIP-T21P/T20P:			
X ranges from			
1 to 2.			
SIP-T19P: X=1.)			
account.X.ses			
sion_timer.ena ble =			
(SIP-T28P/T48G			
/T46G: X			
ranges from 1		It enables or disables the session timer	
to 6.		for account X.	Account->
SIP-T26P/T22P/	0 or 1	0 -Disabled	Advanced->
T42G/T41P: X		1-Enabled	Session Timer
ranges from 1 to 3.		The default value is 0.	
SIP-T21P/T20P:			
X ranges from			
1 to 2.			
SIP-T19P: X=1.)			

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.ses sion_timer.exp ires = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 30 to 7200	It configures the interval (in seconds) for refreshing the SIP session for account X. The default value is 1800.	Account-> Advanced-> Session Expires (30~7200s)
account.X.ses sion_timer.refr esher = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It configures the refresher of the session timer for account X. 0-Uac 1-Uas The default value is 0.	Account-> Advanced-> Session Refresher
account.X.ena ble_user_equ al_phone = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X	0 or 1	It enables or disables the "user=phone" carried in the INVITE message for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced->Send user=phone

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)			
account.X.srtp _encryption = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0, 1 or 2	It configures whether to use voice encryption service for account X. 0-Disabled 1-Optional 2-Compulsory The default value is 0.	Account-> Advanced->RTP Encryption (SRTP)
account.X.pti me = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 (Disabled), 10, 20, 30, 40, 50 or 60	It configures the RTP packet time for account X. The default value is 20.	Account-> Advanced->PTime (ms)
account.X.bla _number = (SIP-T28P/T48G /T46G: X	String within 99 characters	It configures the BLA number for account X. The default value is blank.	Account-> Advanced->BLA Number

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)			
account.X.bla _subscribe_pe riod = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 60 to 7200	It configures the period (in seconds) of BLA subscription for account X. The default value is 300.	Account-> Advanced->BLA Subscription Period
account.X.regi ster_mac = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to carry the MAC address in the REGISTER message for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced->SIP Send MAC

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.regi ster_line = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to carry the line number in the REGISTER message for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced->SIP Send Line
account.X.dial oginfo_callpic kup = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	0 or 1	It enables or disables the phone to pick up a call according to the SIP header of dialog-info for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced-> Dialog Info Call Pickup
account.X.gro up_pickup_co de = (SIP-T28P/T48G /T46G: X ranges from 1 to 6.	String within 32 characters	It configures the group pickup code for account X. The default value is blank.	Account-> Advanced->Group Call Pickup Code

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.dire ct_pickup_cod e = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	String within 32 characters	It configures the directed pickup code for account X. The default value is blank.	Account-> Advanced-> Directed Call Pickup Code
account.X.aut o_answer = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables auto answer feature for account X. 0-Disabled 1-Enabled The default value is 0.	Account->Basic-> Auto Answer

Parameter	Permitted Values	Descriptions	Web Setting Path
features.auto_ answer_delay =	Integer from 1 to 4	It configures the delay time (in seconds) before the phone automatically answers an incoming call. The default value is 1.	
account.X.mis sed_calllog = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to record the missed call of account X. O-Disabled 1-Enabled The default value is 1.	Account->Basic-> Missed Call Log
account.X.sub scribe_mwi_to _vm = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to subscribe to the voice mail number for the message waiting indicator for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced-> Subscribe MWI To Voice Mail
account.X.reg _fail_retry_inte rval = (SIP-T28P/T48G	Integer from 0 to 1800	It configures the interval (in seconds) for the phone to retry to register account X when registration fails. The default value is 30.	Account-> Advanced->SIP Registration Retry Timer (0~1800s)

Parameter	Permitted Values	Descriptions	Web Setting Path
/T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)			
account.X.con f_type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 2	It configures the conference type for account X. 0-Local Conference 2-Network Conference The default value is 0.	Account-> Advanced-> Conference Type
account.X.con f_uri = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	SIP URI within 511 characters	It configures the network conference URI for account X. The default value is blank.	Account-> Advanced-> Conference URI

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.blf.s ubscribe_peri od = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 30 to 2147483647	It configures the period (in seconds) of the BLF subscription for account X. The default value is 1800.	
account.X.blf.s ubscribe_even t = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It configures the event of the BLF subscription for account X. O-Dialog 1-Presence The default value is 0.	
account.X.sip_server_type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1	0, 2, 4 or 6	It configures the SIP server type for account X. 0-Default 2-BroadSoft 4-Cosmocom 6-UCAP The default value is 0.	Account-> Advanced->SIP Server Type

Parameter	Permitted Values	Descriptions	Web Setting Path
to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.mus ic_server_uri = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	SIP URI within 256 characters	It configures the URI of the Music On Hold server for account X. The default value is blank.	Account-> Advanced->Music Server URI
account.X.dtm f.type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0, 1, 2 or 3	It configures the DTMF type for account X. 0-INBAND 1-RFC2833 2-SIP INFO 3-AUTO or SIP INFO The default value is 1.	Account-> Advanced->DTMF Type
account.X.dtm f.dtmf_payloa d = (SIP-T28P/T48G	Integer from 96 to 127	It configures the RFC2833 payload for account X. The default value is 101.	Account-> Advanced->DTMF Payload Type (96~127)

Parameter	Permitted Values	Descriptions	Web Setting Path
/T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.dtm f.info_type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	1, 2 or 3	It configures the DTMF info type when the DTMF type is configured as "SIP INFO", "AUTO or SIP INFO" for account X. 0-Disabled 1-DTMF-Relay 2-DTMF 3-Telephone-Event The default value is 0.	Account-> Advanced->DTMF Info Type
account.X.nat. nat_traversal = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.	0 or 1	It enables or disables the NAT traversal for account X. 0-Disabled 1-Enabled The default value is 0.	Account->Register ->NAT

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T19P: X=1.)			
account.X.nat. stun_server = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	IP address or domain name	It configures the IP address or domain name of the STUN server for account X. The default value is blank.	Account->Register ->STUN Server
account.X.nat. stun_port = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 1024 to 65000	It configures the port of the STUN server for account X. The default value is 3478.	Account->Register ->STUN Server ->Port
account.X.nat. udp_update_e nable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X	0, 1, 2 or 3	It configures the type of keep-alive packets sent by the phone to the NAT device to keep the communication port open so that NAT can continue to function for account X. 0-Dsiabled 1-Default: the phone sends UDP packets to the server. 2-Option: the phone sends SIP OPTION	Account-> Advanced->Keep Alive Type

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.) account.X.nat. udp_update_ti me = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 15 to 2147483647	packets to the server. 3-Notify: the phone sends SIP NOTIFY packets to the server. The default value is 1. It configures the keep-alive interval (in seconds) for account X. The default value is 30.	Account-> Advanced->Keep Alive Interval (Seconds)
account.X.nat. rport = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables NAT Rport feature for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced->RPort
account.X.adv anced.timer_t 1 =	Float from 0.5~10s	It configures the session timer T1 (in seconds) for account X. The default value is 0.5.	Account-> Advanced->SIP Session Timer T1

Parameter	Permitted Values	Descriptions	Web Setting Path
(SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)			(0.5~10s)
account.X.adv anced.timer_t 2 = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Float from 2~40s	It configures the session timer T2 (in seconds) for account X. The default value is 4.	Account-> Advanced->SIP Session Timer T2 (2~40s)
account.X.adv anced.timer_t 4 = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P:	Float from 2.5~60s	It configures the session timer T4 (in seconds) for account X. The default value is 5.	Account-> Advanced->SIP Session Timer T4 (2.5~60s)

Parameter	Permitted Values	Descriptions	Web Setting Path
X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.aler t_info_url_ena ble = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the IP phone to download the ring tone from the URL contained in the Alert-Info header for account X. 0-Disabled 1-Enabled The default value is 0.	Account-> Advanced-> Distinctive Ring Tones
features.alert_ info_tone =	0 or 1	It enables and disables the phone to map the keywords in the Alert-Info header to the specified Bellcore ring tones. 0-Disabled 1-Enabled The default value is 0.	
account.X.ring tone.ring_type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from	Common, Ring1.wav, Ring2.wav, Ring3.wav Ring4.wav Ring5.wav Ring6.wav, Ring6.wav, Ring7.wav,	It configures a ring tone for account X. Example: account.1.ringtone.ring_type = Ring3.wav means configuring Ring3.wav for account1. account.1.ringtone.ring_type = Common means account1 will use the ring tone selected for the phone. The default value is Common. Note: Ring tones 6-8 are only applicable to SIP-T48G/T46G IP phones.	Account->Basic-> Ring Type

Parameter	Permitted Values	Descriptions	Web Setting Path
1 to 2.			
SIP-T19P: X=1.)			
account.X.cod ec.Y.payload_ type = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. SIP-T28P/T26P/T 22P/T20P: Y ranges from 1 to 11. SIP-T21P/T19P: Y ranges from 1 to 8. SIP-T4X: Y ranges from 1 to 12)	PCMU PCMA G729 G722 G723_53 G723_63 G726-16 G726-24 G726-32 G726-40 iLBC GSM	It configures the codec for account X. Codecs G726-16, G726-24 and G726-40 are not applicable to SIP-T21P and SIP-T19P IP phones. Codec GSM is only applicable to SIPT4X IP phones. For SIPT28P/T26P/T22P/T20P/T4X IP phones: When Y=1, the default value is PCMU; When Y=2, the default value is PCMA; When Y=3, the default value is G723_53; When Y=4, the default value is G729; When Y=5, the default value is G722; When Y=7, the default value is G726-16; When Y=9, the default value is G726-24; When Y=10, the default value is G726-32; When Y=11, the default value is G726-40. When Y=12, the default value is GSM. For SIPT21P/T19P IP phones: When Y=3, the default value is PCMU; When Y=3, the default value is PCMU; When Y=3, the default value is PCMU; When Y=3, the default value is PCMA; When Y=3, the default value is PCMA; When Y=3, the default value is G723_53; When Y=4, the default value is G723_63; When Y=5, the default value is G729; When Y=5, the default value is G729; When Y=5, the default value is G722;	Account->Codec

When Y=7, the default value is iLBC; When Y=8, the default value is G726-32. It enables or disables the specified codec for account X.
G726-32. It enables or disables the specified
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O-Disabled 1-Enabled Example:

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.cod ec.Y.priority = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. SIP-T28P/T26P/T 22P/T20P: Y ranges from 1 to 11. SIP-T21P/T19P: Y ranges from 1 to 8. SIP-T4X: Y ranges from 1 to 12)	Integer from 0 to 11 for SIPT4X IP phones Integer from 0 to 10 for SIPT20P/T22P /T26P/T28P IP phones Integer from 0 to 7 for SIPT19P/T21P IP phones	It configures the priority of the enabled codec for account X. Example: account.1.codec.1.priority = 1 For SIPT28P/T26P/T22P/T20P/T4X IP phones: When Y=1, the default value is 1; When Y=3, the default value is 0; When Y=4, the default value is 0; When Y=5, the default value is 3; When Y=6, the default value is 0; When Y=7, the default value is 0; When Y=8, the default value is 0; When Y=9, the default value is 0; When Y=10, the default value is 0; When Y=11, the default value is 0. (only applicable to SIPT4X IP phones) For SIPT21P/T19P IP phones: When Y=1, the default value is 2; When Y=3, the default value is 0; When Y=3, the default value is 0; When Y=4, the default value is 0; When Y=5, the default value is 0; When Y=6, the default value is 3; When Y=6, the default value is 4; When Y=7, the default value is 0; When Y=8, the default value is 0;	Account->Codec
account.X.cod ec.Y.rtpmap = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X	Integer from 0 to 127	It configures rtpmap of the audio codec for account X. Example: account.1.codec.1.rtpmap = 0 For SIPT28P/T26P/T22P/T20P/T4X IP phones: When Y=1, the default value is 0;	

	leb Setting Path
ranges from 1 When Y=2, the default value is 8;	
to 3. When Y=3, the default value is 4;	
SIPT21P/T20P: When Y=4, the default value is 4;	
X ranges from When Y=5, the default value is 18;	
1 to 2. When Y=6, the default value is 9;	
SIPT19P: X=1. When Y=7, the default value is 106;	
SIPT28P/T26P/T When Y=8, the default value is 103:	
22P/12UP: Y	
to 11. When Y=10, the default value is 102;	
SIPT21P/T19P: When Y=11, the default value is 105.	
Y ranges from When Y=12, the default value is 97.	
1 to 8. (only applicable to SIP-T4X IP phones)	
SIPT4X: Y For SIPT21P/T19P IP phones:	
ranges from 1 When Y=1, the default value is 0;	
to 12) When Y=2, the default value is 8;	
When Y=3, the default value is 4;	
When Y=4, the default value is 4;	
When Y=5, the default value is 18;	
When Y=6, the default value is 9;	
When Y=7, the default value is 106;	
When Y=8, the default value is 102.	
account.X.unr egister_on_re	
boot =	
(SIP-T28P/T48G	
/T46G: X	
ranges from 1 It enables or disables the phone to	count->
10 6. On-register account x before a repoot.	dvanced->
SIPT26P/T22P/	nregister When
ranges from 1	boot
ranges from 1 The default value is 0.	
SIP-T21P/T20P:	
X ranges from	
1 to 2.	
SIPT19P: X=1.)	

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.pict ure_info_enab le = (X ranges from 1 to 6) (only applicable to SIPT48G/T46G IP phones)	0 or 1	It enables or disables the phone to download the picture information for account X when receiving an incoming call or during a call. 0-Disabled 1-Enabled The default value is 0.	
account.X.co mpact_heade r_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to support compact SIP header for account X. O-Disabled 1-Enabled The default value is 0.	
account.X.mus ic_on_hold_ty pe = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It configures the way on how the phone processes Music On Hold when placing an active call on hold for account X. 0-Calling the music server before holding 1-Calling the music server after holding The default value is 0.	

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.acd .enable = (SIP-T28P: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables ACD feature for account X. 0-Disabled 1-Enabled The default value is 0.	
(not applicable to SIPT48G/T46G) account.X.acd .available = (SIPT28P: X ranges from 1 to 6. SIPT26P/T22P/T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.) (not	0 or 1	It enables or disables the phone to display the available and unavailable soft keys for account X after the phone logs into the ACD system. 0-Disabled 1-Enabled The default value is 0.	
applicable to SIPT48G/T46G) account.X.sub scribe_acd_ex pires = (SIPT28P: X ranges from 1	Integer from 120 to 3600	It configures the period (in seconds) of ACD subscription for account X. The default value is 1800.	Account-> Advanced->ACD Subscrip Period (120~3600s)

Parameter	Permitted Values	Descriptions	Web Setting Path
to 6.			
SIP-T26P/T22P/			
T42G/T41P: X			
ranges from 1			
to 3.			
SIP-T21P/T20P:			
X ranges from			
1 to 2.			
SIP-T19P: X=1.)			
(not			
applicable to			
SIP-T48G/T46G			
)			

The following table lists configuration parameters that are integrated with BroadSoft platform. For more information on BroadSoft features, refer to Yealink_IP_Phones_Deployment_Guide_for_BroadSoft_UC-One_Environment.

Parameter	Permitted Values	Descriptions	Web Setting Path
bw.enable =	0 or 1	It enables or disables BroadSoft features for IP phones. 0-Disabled 1-Enabled The default value is 0. It takes effect after a reboot.	
features.uc_en able = (only applicable to SIPT48G/ T46G IP phones)	0 or 1	It enables or disables UC feature. 0-Disabled 1-Enabled The default value is 1. It takes effect after a reboot.	
features.uc_us ername = (only applicable to SIPT48G/ T46G IP phones)	String within 99 characters	Configures the user name for UC authentication. The default value is blank. It takes effect after a reboot.	Directory-> Network Directory->UC Username
features.uc_p assword = (only applicable to SIPT48G/ T46G IP phones)	String within 32 characters	It configures the password for UC authentication. The default value is blank. It takes effect after a reboot.	Directory-> Network Directory->UC Password
features.confi g_dsskey_len gth = (only applicable to SIPT46G IP phones)	0 or 1	It enables or disables extended length of the label displayed on the idle LCD screen for the line key. 0-Disabled 1-Enabled The default value is 0.	

Parameter	Permitted Values	Descriptions	Web Setting Path
phone_setting .dsskey_direct ory_auto.ena ble = (only applicable to SIPT48G/ T46G IP phones)	0 or 1	It enables or disables Auto Favorite feature. If enabled, the IP phone will download information of favorites from the cloud server and automatically configure UC Favorite keys from the first unused line key (the line key is configured as N/A). If a line key is used, the IP phone will skip to the next unused line key. 0-Disabled 1-Enabled The default value is 1.	Features->General Information->Auto Favorite
account.X.xsi. host = (X=1)	IP address or domain name	It configures the IP address or domain name of the Xtended Services Platform server for account X. Example: account.1.xsi.host = xsp1.iop1.broadworks.net The default value is blank.	Directory-> Network Directory->Host Server
account.X.xsi. port = (X=1)	Integer from 1 to 65535	It configures the port of the Xtended Services Platform server for account X. Example: account.1.xsi.port = 80 The default value is 80.	Directory-> Network Directory->Port
account.X.xsi. server_type = (X=1)	"http" or "https"	It configures the access protocol of the Xtended Services Platform server for account X. Example: account.1.xsi.server_type = http The default value is http.	Directory-> Network Directory->XSI Server Type
account.X.xsi. user = (X=1)	String within 99 characters	It configures the user name for XSI authentication for account X. Example: account.1.xsi.user = 3502@as.iop1.broadworks.net The default value is blank.	Directory-> Network Directory->User ID

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.xsi. password = (X=1)	String within 99 characters	It configures the password for XSI authentication for account X. Example: account.1.xsi.password = 123456 The default value is blank.	Directory-> Network Directory-> Password
bw_phoneboo k.group_enabl e = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the group directory. O-Disabled 1-Enabled The default value is 1.	Directory-> Network Directory->Group
bw_phoneboo k.group_displ ayname = (not applicable to SIPT20P IP phones)	String within 99 characters	It configures the display name on the LCD screen for the group directory. The default value is Group.	Directory-> Network Directory->Group
bw_phoneboo k.enterprise_e nable = (not applicable to SIP-T20P IP phones)	0 or 1	It enables or disables the phone to display the enterprise directory. O-Disabled 1-Enabled The default value is 1.	Directory-> Network Directory->Enterpri
bw_phoneboo k.enterprise_d isplayname = (not applicable to SIP-T20P IP phones)	String within 99 characters	It configures the display name on the LCD screen for the enterprise directory. The default value is Enterprise.	Directory-> Network Directory->Enterpri
bw_phoneboo k.group_com mon_enable = (not	0 or 1	It enables or disables the phone to display the group common directory. 0 -Disabled 1 -Enabled	Directory-> Network Directory->Group Common

Parameter	Permitted Values	Descriptions	Web Setting Path
applicable to SIPT20P IP phones)		The default value is 1.	
bw_phonebook.group_common_displayname = (notapplicable to SIPT20P IP phones)	String within 99 characters	It configures the display name on the LCD screen for the group common directory. The default value is GroupCommon.	Directory-> Network Directory->Group Common
bw_phoneboo k.enterprise_c ommon_enabl e = (not applicable to SIP-T20P IP phones)	0 or 1	It enables or disables the phone to display the enterprise common directory. O-Disabled 1-Enabled The default value is 1.	Directory-> Network Directory->Enterpri se Common
bw_phonebook.enterprise_common_displayname = (not applicable to SIP-T20P IP phones)	String within 99 characters	It configures the display name on the LCD screen for the enterprise common directory. The default value is EnterpriseCommon.	Directory-> Network Directory->Enterpri se Common
bw_phonebook.personal_enable= (notapplicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the personal directory. 0-Disabled 1-Enabled The default value is 1.	Directory-> Network Directory->Person
bw_phoneboo k.personal_dis playname= (not applicable to	String within 99 characters	It configures the display name on the LCD screen for the personal directory. The default value is Personal.	Directory-> Network Directory->Person al

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT20P IP phones)			
bw_phoneboo k.custom = (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables custom directory feature. 0-Disabled 1-Enabled The default value is 0.	Directory-> Network Directory->Enable Custom Directory
directory.upd ate_time_inter val = (not applicable to SIP.T20P IP phones)	Integer from 2 to 43200	It configures the interval (in minutes) for the phone to update the data of the BroadSoft directory from the BroadSoft server. The default value is 60.	
bw_phonebook.call_log_enable = (notapplicable to SIPT20P IP phones)	0 or 1	It enables or disables BroadSoft call log feature. 0 -Disabled 1 -Enabled The default value is 0.	
call_park.ena ble= (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the Park soft key during a call. 0-Disabled 1-Enabled The default value is 0.	Features->Call Pickup->Call Park
call_park.grou p_enable= (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to display the GPark soft key during a call. O-Disabled 1-Enabled The default value is 0.	Features->Call Pickup->Group Call Park
call_park.park _visual_notify_ enable = (not	0 or 1	It enables or disables the phone to display a parked indicator when a call is parked against its line.	Features->Call Pickup->Visual Alert for Parked Call

Parameter	Permitted Values	Descriptions	Web Setting Path
applicable to SIPT20P IP phones)		O-Disabled 1-Enabled The default value is 0.	
call_park.park _ring= (not applicable to SIPT20P IP phones)	0 or 1	It enables or disables the phone to play a warning tone when a call is parked against its line. 0-Disabled 1-Enabled The default value is 0.	Features->Call Pickup->Audio Alert for Parked Call
bw.feature_ke y_sync =	0 or 1	It enables or disables feature key synchronization. O-Disabled 1-Enabled The default value is 0.	Features->General Information-> Feature Key Synchronization
account.X.blf. blf_list_uri = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	SIP URI within 256 characters	It configures the BLF List URI to monitor a list of users for account X. The default value is blank.	Account-> Advanced->BLF List URI
account.X.blf_I ist_code = (SIP-T28P/T48G /T46G: X ranges from 1 to 6.	String within 32 characters	It configures the feature access code for directed call pickup (default: *97) for account X. The default value is blank.	Account-> Advanced->BLF List Code

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP			
phones) account.X.blf_I ist_barge_in_c ode = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.) (not applicable to SIP-T19P IP phones)	String within 32 characters	It configures the feature access code for directed call pickup with barge-in (default: *33) for account X. The default value is blank.	Account-> Advanced->BLF List Barge In Code
phone_setting .auto_blf_list_ enable = (not applicable to SIPT19P IP phones)	0 or 1	It enables or disables the phone to automatically configure the BLF list keys in order. 0-Disabled 1-Enabled The default value is 1.	
phone_setting .blf_list_seque nce_type =	0 or 1	It configures the order of BLF list keys assigned automatically. 0 -Line Keys->Memory	

Parameter	Permitted Values	Descriptions	Web Setting Path
(only applicable to SIPT28P, SIPT26P, SIPT48G and SIPT46G IP phones) account.X.sha red_line = (SIPT28P/T48G /T46G: X ranges from 1 to 6. SIPT26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1)	0 or 1	Keys->Extension Keys 1-Extension Keys->Memory Keys->Line Keys Memory keys are not applicable to SIPT46G/SIPT48G IP phones. The default value is 0. It enables or disables Broadsoft SCA feature for account X. 0-Disabled 1-Broadsoft SCA The default value is 0.	Account-> Advanced->Share d Line
account.X.acd .initial_state = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)	1 or 2	It configures the initial agent state for account X. 1-Avaliable 2-Unavailable The default value is 1.	
account.X.acd .unavailable_r eason_enable =	0 or 1	It enables or disables unavailable reason code feature for account X. 0 -Disabled	

Parameter	Permitted Values	Descriptions	Web Setting Path
(SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1)		1-Enabled The default value is 0.	
account.X.rea son_code.Y = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 100)	Integer from 1 to 2147483647	It configures the unavailable code which must match one of the codes configured on BroadWorks for account X. The value Y must be continuous. The default value is blank.	
account.X.rea son_code_na me.Y = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3.	String within 99 characters	It configures the unavailable reason which must match one of the reasons configured on BroadWorks for account X. The value Y must be continuous. The default value is blank.	

Parameter	Permitted Values	Descriptions	Web Setting Path
SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1. Y ranges from 1 to 100)			
account.X.call _center.call_in fo_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables call center call information feature for account X. 0-Disabled 1-Enabled The default value is 0.	
account.X.call _center.show_ call_info_time = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	Integer from 1 to 86400	It configures the interval (in seconds) to specify how long the call center call information displays for account X. The default value is 30 seconds.	

Parameter	Permitted Values	Descriptions	Web Setting Path
account.X.call _center.disp_c ode_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables disposition code feature for account X. 0-Disabled 1-Enabled The default value is 0.	
account.X.bw_disp_code.Y = (SIP-T28P/T48G/T46G: X) ranges from 1 to 6. SIP-T26P/T22P/T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 100)	Integer from 1 to 2147483647	It configures the disposition code which must match one of the codes configured on BroadWorks for account X. The value Y must be continuous. The default value is blank.	
account.X.bw_disp_code_name.Y = (SIP-T28P/T48G/T46G: X) ranges from 1 to 6. SIP-T26P/T22P/	String within 99 characters	It configures the disposition code name which must match one of the names configured on BroadWorks for account X. The value Y must be continuous. The default value is blank.	

Parameter	Permitted Values	Descriptions	Web Setting Path
T42G/T41P: X ranges from 1 to 3.			
SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.			
Y ranges from 1 to 100)			
account.X.call _center.trace_ enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables customer originated trace feature for account X. O-Disabled 1-Enabled The default value is 0.	
account.X.call _center.emerg ency_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2.	0 or 1	It enables or disables the emergency escalation feature for account X. 0-Disabled 1-Enabled The default value is 0.	

Parameter	Permitted Values	Descriptions	Web Setting Path
SIP-T19P: X=1.			
account.X.sup ervisor_info_c ode.Y = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 100)	Integer from 1 to 2147483647	It configures the supervisor number for account X. The value Y must be continuous. The default value is blank.	
account.X.sup ervisor_info_c ode_name.Y = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1. Y ranges from 1 to 100)	String within 99 characters	It configures the supervisor name for account X. The value Y must be continuous. The default value is blank.	
account.X.call _center.queue	0 or 1	It enables or disables the queue status notification feature for account X.	

Parameter	Permitted Values	Descriptions	Web Setting Path
_status_enabl		0-Disabled	
e =		1-Enabled	
(SIP-T28P/T48G		The default value is 0.	
/T46G: X			
ranges from 1			
to 6.			
SIP-T26P/T22P/			
T42G/T41P: X ranges from 1			
to 3.			
SIP-T21P/T20P:			
X ranges from			
1 to 2.			
SIP-T19P: X=1.)			
account.X.call			
_center.queue		It enables or disables the power	
_status_light_e		indicator LED to flash when the ACD	
nable =		call queue has reached the maximum	
(SIP-T28P/T48G		number of calls for account X.	
/T46G: X		0 -Disabled (power indicator LED does	
ranges from 1		not flash)	
to 6.	0 or 1	1-Enabled (power indicator LED fast	
SIP-T26P/T22P/	3 31 1	flashes (300ms) green)	
T42G/T41P: X		If it is set to 0, the status of the power	
ranges from 1 to 3.		indicator LED is determined by the	
		value of the parameter	
SIP-T21P/T20P:		"phone_setting.common_power_led_e nable".	
X ranges from 1 to 2.			
SIP-T19P: X=1.)		The default value is 0.	
account.X.hot			
eling.enable		It enables or disables hoteling feature	
=		for account X.	
(SIP-T28P/T48G		0 -Disabled	
/T46G: X	0 or 1	1-Enabled	
ranges from 1			
to 6.		The default value is 0.	
SIP-T26P/T22P/		It takes effect after a reboot.	
T42G/T41P: X			

Parameter	Permitted Values	Descriptions	Web Setting Path
ranges from 1 to 3. SIPT21P/T20P: X ranges from 1 to 2. SIPT19P: X=1.)			
account.X.hot eling.auto_log in_enable = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	0 or 1	It enables or disables the phone to save login credentials automatically when logging into the guest profile for account X. 0-Disabled 1-Enabled The default value is 0.	
account.X.hot eling.user_id = (SIP-T28P/T48G /T46G: X ranges from 1 to 6. SIP-T26P/T22P/ T42G/T41P: X ranges from 1 to 3. SIP-T21P/T20P: X ranges from 1 to 2. SIP-T19P: X=1.)	String within 99 characters	It configures the user ID used to log into the guest profile for account X. The default value is blank.	
account.X.hot eling.passwor	String within	It configures the password used to log into the guest profile for account X.	

Parameter	Permitted Values	Descriptions	Web Setting Path
d =	characters	The default value is blank.	
(SIP-T28P/T48G			
/T46G: X			
ranges from 1			
to 6.			
SIP-T26P/T22P/			
T42G/T41P: X			
ranges from 1			
to 3.			
SIP-T21P/T20P:			
X ranges from			
1 to 2.			
SIP-T19P: X=1.)			

The following parameters are not applicable to IP phones running firmware version 72 or later, so the description of these parameters has been deleted in this guide.

```
account.X.bw_acd_reason_code.Y =
account.X.dns_query_timeout=
account.X.failback_mode =
account.X.failback_timeout =
account.X.retry_counts =
acd.bw =
bw.calllog_and_dir =
bw_phonebook.server_search_enable =
directory.incoming_call_match_enable =
features.action_uri_reboot_now =
features.auto_release_bla_line =
features.dsskey_blind_tran =
features.hold_trans_delay =
features.ldap.input_type =
|ldap.dial_lookup =
network.snmp.enable =
network.snmp.port =
network.snmp.trust_ip =
phone_setting.blf_list_enable =
phone_setting.sms_popup_enable =
voice.call_preview_mode=
voice.tone.record =
features.hoteling.enable =
action_url.call_interrupt =
action_url.setup_autop_finish =
features.blf_pickup_only_send_code =
```

Time Zones

Time Zone	Time Zone Name
-11:00	Samoa
-10:00	United States-Hawaii-Aleutian
-10:00	United States-Alaska-Aleutian
-09:00	United States-Alaska Time
-08:00	Canada(Vancouver, Whitehorse)
-08:00	Mexico(Tijuana, Mexicali)
-08:00	United States-Pacific Time
-07:00	Canada(Edmonton, Calgary)
-07:00	Mexico(Mazatlan, Chihuahua)
-07:00	United States-Mountain Time
-07:00	United States-MST no DST
-06:00	Canada-Manitoba(Winnipeg)
-06:00	Chile(Easter Islands)
-06:00	Mexico(Mexico City, Acapulco)
-06:00	United States-Central Time
-05:00	Bahamas(Nassau)
-05:00	Canada(Montreal, Ottawa, Quebec)
-05:00	Cuba(Havana)
-05:00	United States-Eastern Time
-04:30	Venezuela(Caracas)
-04:00	Canada(Halifax, Saint John)
-04:00	Chile(Santiago)
-04:00	Paraguay(Asuncion)
-04:00	United Kingdom-Bermuda(Bermuda)
-04:00	United Kingdom(Falkland Islands)
-04:00	Trinidad&Tobago
-03:30	Canada-New Foundland(St.Johns)
-03:00	Denmark-Greenland(Nuuk)
-03:00	Argentina(Buenos Aires)
-03:00	Brazil(no DST)
-03:00	Brazil(DST)
-02:00	Brazil(no DST)
-01:00	Portugal(Azores)
0	GMT
0	Greenland
0	Denmark-Faroe Islands(Torshavn)
0	Ireland(Dublin)
0	Portugal(Lisboa, Porto, Funchal)
0	Spain-Canary Islands(Las Palmas)

Time Zone	Time Zone Name
0	United Kingdom(London)
0	Morocco
+01:00	Albania(Tirane)
+01:00	Austria(Vienna)
+01:00	Belgium(Brussels)
+01:00	Caicos
+01:00	Chad
+01:00	Spain(Madrid)
+01:00	Croatia(Zagreb)
+01:00	Czech Republic(Prague)
+01:00	Denmark(Kopenhagen)
+01:00	France(Paris)
+01:00	Germany(Berlin)
+01:00	Hungary(Budapest)
+01:00	Italy(Rome)
+01:00	Luxembourg(Luxembourg)
+01:00	Macedonia(Skopje)
+01:00	Netherlands(Amsterdam)
+01:00	Namibia(Windhoek)
+02:00	Estonia(Tallinn)
+02:00	Finland(Helsinki)
+02:00	Gaza Strip(Gaza)
+02:00	Greece(Athens)
+02:00	Israel(Tel Aviv)
+02:00	Jordan(Amman)
+02:00	Latvia(Riga)
+02:00	Lebanon(Beirut)
+02:00	Moldova(Kishinev)
+02:00	Russia(Kaliningrad)
+02:00	Romania(Bucharest)
+02:00	Syria(Damascus)
+02:00	Turkey(Ankara)
+02:00	Ukraine(Kyiv, Odessa)
+03:00	East Africa Time
+03:00	Iraq(Baghdad)
+03:00	Russia(Moscow)
+03:30	Iran(Teheran)
+04:00	Armenia(Yerevan)
+04:00	Azerbaijan(Baku)
+04:00	Georgia(Tbilisi)
+04:00	Kazakhstan(Aktau)
+04:00	Russia(Samara)

Time Zone	Time Zone Name
+04:30	Afghanistan
+05:00	Kazakhstan(Aqtobe)
+05:00	Kyrgyzstan(Bishkek)
+05:00	Pakistan(Islamabad)
+05:00	Russia(Chelyabinsk)
+05:30	India(Calcutta)
+06:00	Kazakhstan(Astana, Almaty)
+06:00	Russia(Novosibirsk, Omsk)
+07:00	Russia(Krasnoyarsk)
+07:00	Thailand(Bangkok)
+08:00	China(Beijing)
+08:00	Singapore(Singapore)
+08:00	Australia(Perth)
+09:00	Korea(Seoul)
+09:00	Japan(Tokyo)
+09:30	Australia(Adelaide)
+09:30	Australia(Darwin)
+10:00	Australia(Sydney, Melbourne, Canberra)
+10:00	Australia(Brisbane)
+10:00	Australia(Hobart)
+10:00	Russia(Vladivostok)
+10:30	Australia(Lord Howe Islands)
+11:00	New Caledonia(Noumea)
+12:00	New Zealand(Wellington, Auckland)
+12:45	New Zealand(Chatham Islands)
+13:00	Tonga(Nukualofa)

BLF LED Mode

BLF LED Mode provides four kinds of definition for the BLF/BLF list key LED status. The following tables list the LED statuses of the BLF/BLF list key when BLF LED Mode is set to 0, 1, 2 or 3 respectively. The default value of the BL LED Mode is 0. BLF LED Mode is not applicable to SIP-T48G and SIP-T19PIP phones. BLF/BLF list key LED status for line keys on T46G/T42G/T41P IP phones are the same as that for memory keys.

Line key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 0)

LED Status	Description
Solid green	The monitored user is idle.
Fast flashing green (200ms)	The monitored user receives an incoming call.
Slow flashing green (500ms)	The monitored user is dialing. The monitored user is talking. The monitored user's conversation is placed on hold.
Slow flashing green (1s)	The call is parked against the monitored user's phone number.
Off	The monitored user does not exist.

Memory key/Expansion Module key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 0)

LED Status	Description
Solid green	The monitored user is idle.
Fast flashing red (200ms)	The monitored user receives an incoming call.
Solid red	The monitored user is dialing. The monitored user is talking.
Slow flashing red (1s)	The call is parked against the monitored user's phone number. The monitored user's conversation is placed on hold.
Off	The monitored user does not exist.

Line key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 1)

LED Status	Description
Fast flashing green (200ms)	The monitored user receives an incoming call.
Solid green	The monitored user is dialing. The monitored user is talking.
Slow flashing green (500ms)	The monitored user's conversation is placed on hold.

Slow flashing green (1s)	The call is parked against the monitored user's phone number.
Off	The monitored user is idle.
	The monitored user does not exist.

Memory key/Expansion Module key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 1)

LED Status	Description
Fast flashing red (200ms)	The monitored user receives an incoming call.
Solid red	The monitored user is dialing. The monitored user is talking.
Slow flashing red (1s)	The call is parked against the monitored user's phone number. The monitored user's conversation is placed on hold.
Off	The monitored user is idle. The monitored user does not exist.

Line key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 2)

LED Status	Description
Fast flashing green (200ms)	The monitored user receives an incoming call.
Slow flashing green (500ms)	The monitored user is dialing.
	The monitored user is talking.
	The monitored user's conversation is placed on hold.
Slow flashing green (1s)	The call is parked against the monitored user's phone
	number.
Off	The monitored user is idle.
	The monitored user does not exist.

Memory key/Expansion Module key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 2)

LED Status	Description
Fast flashing red (200ms)	The monitored user receives an incoming call.
Solid red	The monitored user is dialing. The monitored user is talking.
Slow flashing red (1s)	The call is parked against the monitored user's phone number. The monitored user's conversation is placed on hold.
Off	The monitored user is idle. The monitored user does not exist.

Line key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 3)

LED Status	Description
Fast flashing green (200ms)	The monitored user receives an incoming call.
Solid green	The monitored user is dialing.
	The monitored user is talking. The monitored user's conversation is placed on hold.
Slow flashing green (1s)	The call is parked against the monitored user's phone number.
Off	The monitored user is idle.
	The monitored user does not exist.

Memory key/Expansion Module key LED (configured as a BLF/BLF list key and BLF LED Mode is set to 3)

LED Status	Description
Fast flashing red (200ms)	The monitored user receives an incoming call.
Solid red	The monitored user is dialing. The monitored user is talking. The monitored user's conversation is placed on hold.
Slow flashing red (1s)	The call is parked against the monitored user's phone number.
Off	The monitored user is idle. The monitored user does not exist.

Customer Feedback

We are striving to improve our documentation quality and we appreciate your feedback. Email your opinions and comments to DocsFeedback@yealink.com.