

Hidden Pages for Quadro Advanced Configuration



Revision: 1.6

Abstract: This document describes the frequently used hidden web pages available on Quadro units. It is intended to help administrators to identify and resolve the issues caused by inappropriate configuration of hidden settings.

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Document Revision History

Revision	Date	Description	Valid for models
1.0	28-Sep-07	Initial release	All models
1.1	02-Nov-07	Notes added to Introduction, Section 3.1.1.	All models
1.2	14-Nov-07	Edited	All models
1.3	02-Feb-09	Loadlogo.cgi added	All models
1.4	12-Jun-09	callcontrol.cgi and featuretrial.cgi pages added	All IP PBXs
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1 Introduction

Along with the basic web management pages available for the administrator, the Quadro IP-PBXs contain a number of hidden pages (CGIs) that allow for advanced configuration of the system. The default system settings will work for most applications but they can be changed if required. Some hidden pages also provide access to troubleshooting tools to collect additional information.

This document describes the frequently used hidden pages available on Quadro systems. It is intended to help administrators identify and resolve issues that cannot be resolved through the typical installation manuals. The options and parameters that are most commonly used will be explained. The screenshots in this document display the default values of hidden settings.

Please Note: The hidden pages contain low-level system settings and are mainly intended for Epygi's technical support engineers and advanced distributors for fine-tuning the system in some specific configuration scenarios. Wrong or inappropriate configuration of these settings may result in undesired system behavior. It is strongly recommended not to modify the default settings if their meanings are not fully clear.

2 The List of the Hidden Pages

Table 1 lists the frequently used hidden pages. More detailed descriptions of these pages are given in the sections below.

Table 1

	CGI	Description
1	advanced.cgi	Allows switching off the software watchdog and changing the default http/https server ports.
2	callcapture.cgi	Allows making call captures.
3	codecsconfig.cgi	Allows configuring voice codec settings.
4	extmanagement.cgi	Allows configuring certain advanced settings for extensions and attendants.
5	fxocfg.cgi	Allows configuring FXO lines.
6	fxoconfig.cgi	Allows configuring FXO lines.
7	fxodisconnect.cgi	Allows selecting one of the five basic scenarios for call disconnection.
8	generalconfig.cgi	Allows configuring the system general settings.
9	lantable.cgi	Allows packet traffic prioritization based on IP address.
10	linesconfig.cgi	Allows configuring FXS lines.
11	netcapture.cgi	Allows making packet captures from a specified network interface.
12	trunkecho.cgi	Allows configuring echo settings of ISDN, E1/T1 trunks.
13	tos.cgi	Allows specifying the value for the Type of Service setting (ToS) for SIP and RTP packets.

14	callcontrol.cgi	Allows making auto-generated call on the Quadro
15	featuretrial.cgi	Allows activating Quadro optional features for a specified trial period
16	kseswitch.cgi	Allows switching modes related to SLA

3 Accessing Hidden Pages

To access a hidden page, log into the system as the administrator and type the CGI name into the address field of the browser:

`http://Quadro_hostname_or_IP/hidden_page`

For example: `http://192.168.75.85/fxocfg.cgi`

3.1 advanced.cgi

The **Advanced System Setup** page allows switching off the software watchdog and specifying custom listening ports for HTTP and HTTPS.

The following settings are found on the **Advanced System Setup** page (see Figure 1):

- **Enables Software Watchdog:** The Quadro software watchdog is an application that checks the activity of sub-systems that are vital for the proper operation of the whole system and restarts the voice system or reboots the whole system if a sub-system has failed. The activity checking is executed every five minutes. The following sub-systems are monitored by the watchdog: *voice, vpnd, ipsecd, stats, tftpd, dhclient, dhcpd, named, ntpd, ntp, httpd*.
- **HTTP Server Port:** Listening port for HTTP.
- **HTTPS Server Port:** Listening port for HTTPS.

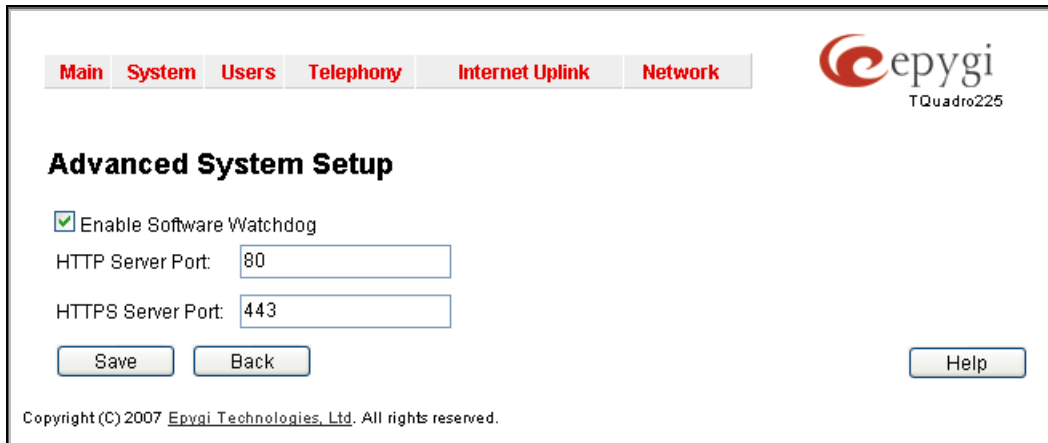


Figure 1 The Advanced System Setup page

3.2 codecsconfig.cgi

The **Voice Codecs Settings** page allows enabling/disabling the **CNG** and **Extrapolation** options for voice codecs and configuring the following settings (see Figure 2):

- **Select G729 Codec Type:** allows choosing between the G729 codec types.
- **Terminate stream if no packets received in:** Time (in seconds) after which the connection with the remote endpoint will be closed if no packet is received.

- **Short DTMF pulse suppression:** (not available in Quadro2x's) Enables short in-band DTMF suppression received from the remote endpoint when the out-of-band DTMF support is negotiated. Enabling this checkbox is recommended if the Quadro is used as a gateway between a SIP and non-SIP endpoints and the non-SIP endpoint detects short DTMFs (less than 20 ms) as valid DTMFs, resulting in, for example, double detection of DTMFs. In-band DTMFs shorter than 35-40 ms are normally ignored.
- **CNG (Comfort Noise Generation):** A technique used in codecs to generate an artificial noise if no voice packet or a CN packet is received from the remote party. With typical PSTN networks people are accustomed to hearing a very low level of background noise on a call. Without Comfort Noise packets being locally generated by the Quadro or far-end there would be complete silence on the call when a person is not speaking and the other party would think the call was disconnected.
- **Extrapolation:** A technique used in codecs to recover a lost or delayed RTP packet to provide a seamless voice experience to the user.
- **Telephone Event Draft Support:** Enables out-of-band DTMF transmission according to the draft-ietf-avt-rfc2833bis-04 (RFC 4733).

To disable the **CNG** or **Extrapolation** option for a codec, select the codec and click on **Edit** (see Figure 3). Clear the checkbox for the appropriate option and click on **Save**.

Voice Codecs Settings

[Edit](#)

Codecs	Extrapolation	CNG
<input type="checkbox"/> G.711u (PCM audio coding standard, 8 kHz sample rate, 8 bits, 64 kbit/s data rate)	Yes	Yes
<input type="checkbox"/> G.711a (PCM audio coding standard, 8 kHz sample rate, 8 bits, 64 kbit/s data rate)	Yes	Yes
<input type="checkbox"/> G.726-16 (ADPCM speech coding at 16 kbit/s rate)	Yes	Yes
<input type="checkbox"/> G.726-24 (ADPCM speech coding at 24 kbit/s rate)	Yes	Yes
<input type="checkbox"/> G.726-32 (ADPCM speech coding at 32 kbit/s rate)	Yes	Yes
<input type="checkbox"/> G.726-40 (ADPCM speech coding at 40 kbit/s rate)	Yes	Yes
<input type="checkbox"/> G.729a (CS-ACELP speech coding at 8 kbit/s rate)	Yes	Yes
<input type="checkbox"/> iLBC (Internet Low Bit Rate Codec at 13,33 kbit/s rate)	Yes	Yes
<input type="checkbox"/> G.722 (HD audio coding at 48-64 kbit/s data rate, 16 kHz sample rate)	No	No
<input type="checkbox"/> G.722.1 (HD audio coding at 24-32 kbit/s data rate, 16 kHz sample rate)	No	No

Select G729 Codec Type

☒ G729
☐ G729a

Terminate stream if no packets received in sec

☐ Short DTMF pulse suppression
☒ Telephone Event Draft Support

Please check your pending events!

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Figure 2 The Voice Codecs Settings page

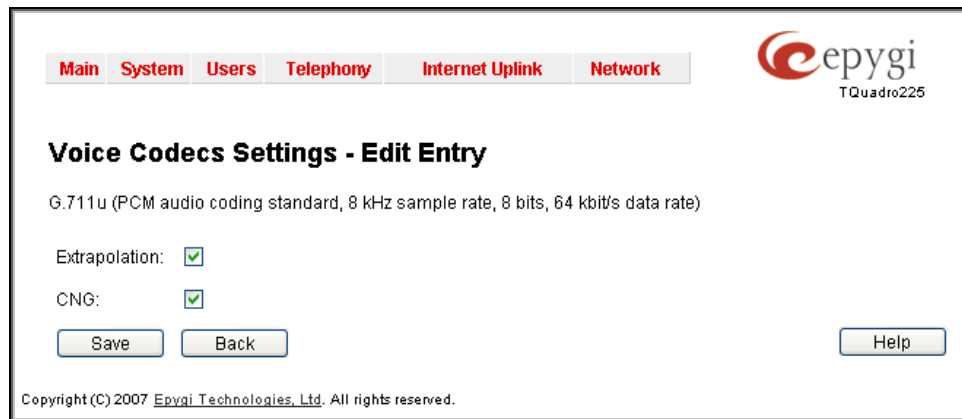


Figure 3 The Voice Codecs Settings – Edit Entry page

3.3 callcapture.cgi

The **Call Capture** page consists of two tabbed pages: **Active Calls** and **Interfaces** (see Figures 4 and 5 respectively). The **Active Calls** page allows for capturing calls currently in progress. The **Interfaces** page allows for capturing calls from a certain port (FXO, FXS, ISDN, etc.).

To capture an active call, select the call from the **Active Calls** table (see Figure 4). In the **Capture Timeout** text field, specify the length of the capture and click on the **Start** button. Once the capture is complete, the page will display the **Download Capture** and **Remove Captured Files** links. Click on the **Download Capture** link. This will download the resultant file which is a *.tar file containing archived binary files. Send the capture file to Epygi's Technical Support engineers for investigation or extract the file to a folder and use any audio editor (for example, Adobe Audition) to view its content.

Please Note: After downloading the capture, make sure to remove it from the Quadro to free the space.

To capture the call from a certain port, follow these steps:

1. On the **Interfaces** tabbed page, select the desired interface (FXO, FXS, ISDN or any other interface available on your Quadro model) and choose the desired port number from the appropriate drop-down list (see Figure 5).
2. In the **Capture Timeout** text field, specify the length of the capture.
3. Click on the **Start** button to start the capture.
4. Make or receive a call on the specified port.

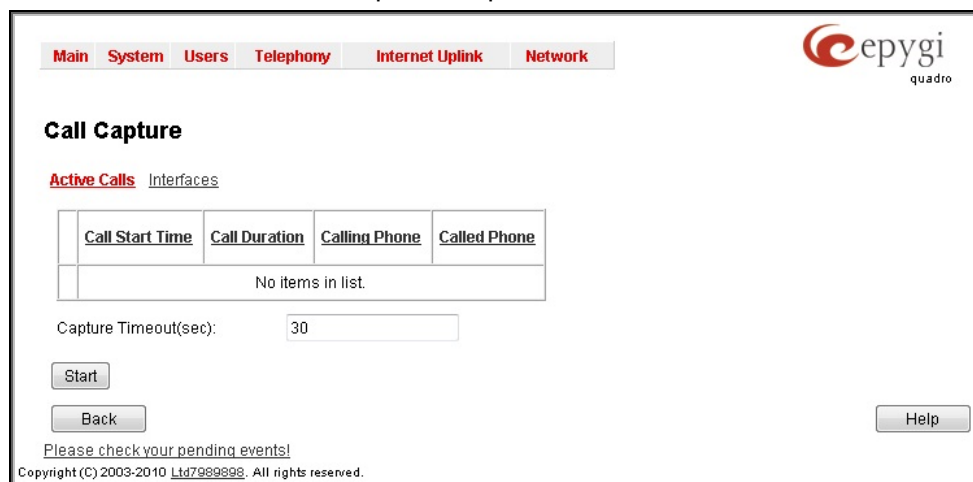
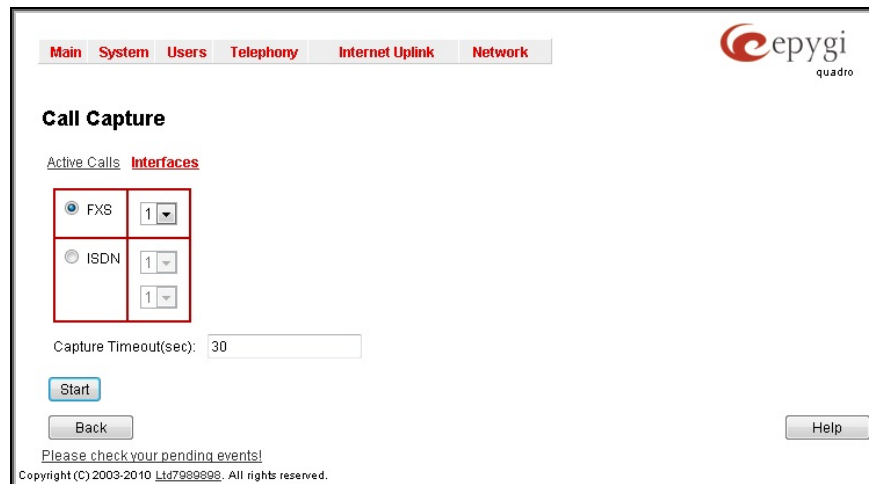


Figure 4 Call Capture – Active Calls page



Call Capture

Active Calls **Interfaces**

☒ FXS 1
☐ ISDN 1

Capture Timeout(sec): 30

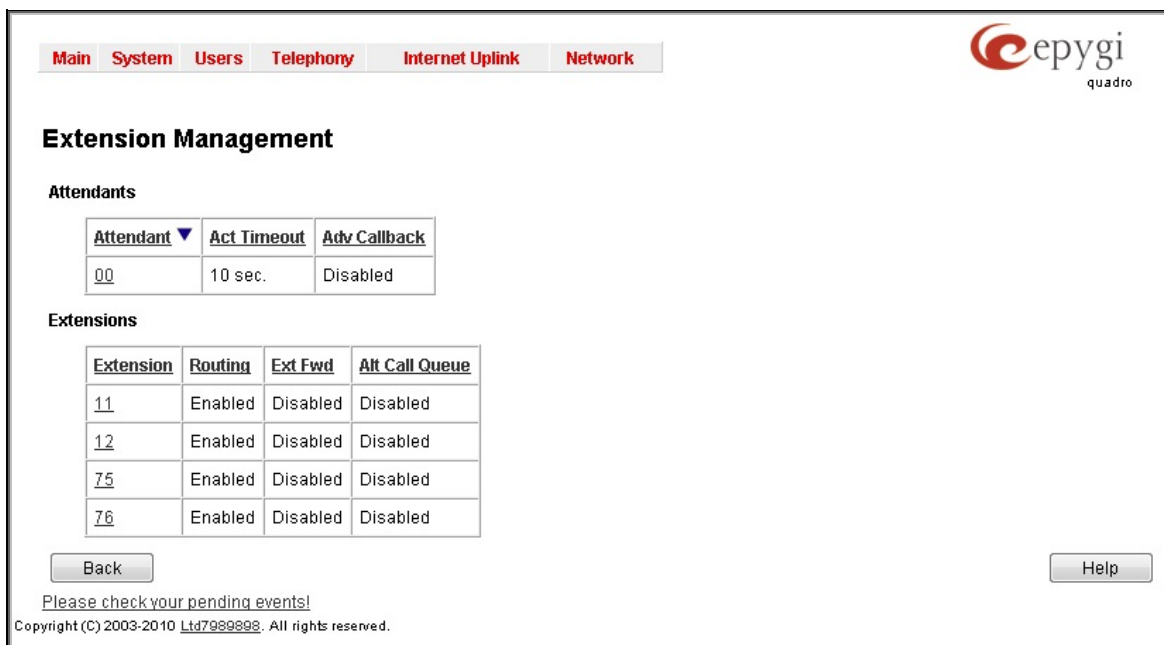
Start Back Help

Please check your pending events!
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Figure 5 Call Capture – Interfaces page

3.4 extmanagement.cgi

The **Extensions Management** page allows configuring certain advanced settings for extensions and attendants (see Figure 6).



Extension Management

Attendants

Attendant ▼	Act Timeout	Adv Callback
00	10 sec.	Disabled

Extensions

Extension	Routing	Ext Fwd	Alt Call Queue
11	Enabled	Disabled	Disabled
12	Enabled	Disabled	Disabled
75	Enabled	Disabled	Disabled
76	Enabled	Disabled	Disabled

Back Help

Please check your pending events!
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Figure 6 Extension Management page

The following setting can be edited through this hidden page for attendants (see Figure 7):

- **Callback Activation Timeout:** The time range within which the trusted users may interrupt the call to auto attendant in order to initiate the callback (18 sec is the default).
- **Enable Advanced Callback Capability (experimental):** If enabled, all incoming SIP calls received to the selected attendant will be required to pass authentication on the RADIUS server the Quadro is configured with. If authentication is successful, the system will disconnect the SIP call and call back the caller via ITSP using a corresponding call routing rule. The scenario that this setting uses assumes that incoming SIP calls are received from another Quadro (typically a Quadro gateway), which prior to making a SIP call, substitutes the originator's SIP username by the caller's PSTN number. If authentication fails, the system will

disconnect the call without calling back. If disabled, incoming calls to the attendant will be handled in the usual way according to the system configuration.

Please Note: This setting is available in Quadro IP-PBXs and is used in a scenario that allows initiating a long-distance PSTN call from a remote Quadro IP-PBX by a local PSTN call to a QuadroGW.

Figure 7 Extension Management page for attendant 00

The following settings can be edited through the **Extensions Management** hidden page (see Figure 8):

- **Enable Routing Mode:** Indicates whether calls initiated from the selected extension should be made using the rules in the Call Routing table. If disabled, the extension will only be able to make PBX calls to two-digit extensions, PSTN calls and SIP calls through a server on which the extension is registered. PSTN calls need to be prefixed by 9 and SIP calls by 8. If enabled, calls initiated from the extension will be routed to the call routing table where the best matching pattern will be used to place the call.

This option should not be disabled so that the extension can use all the patterns in the call routing table.

- **Use extension settings in forwarding:** Indicates whether CallerIDs of SIP calls received on the selected extension and forwarded to another SIP destination will be modified to contain the SIP username of the extension that has forwarded the call. If enabled, the callerIDs of forwarded SIP calls will contain the call originator's display name and the SIP URL of the extension that has forwarded the call. If disabled, the originator's callerID will be sent to the forwarded destination without any modification.
- **Use alternative call queuing mode:** If this checkbox is enabled, the call queue message will be played simultaneously to all callers in the queue. If disabled, the call queue message will be played to callers at regular intervals.

Please Note: Do not enable this checkbox; it will break incoming calls' handling of the extension.

- **Fax message size (kb):** Size of system RAM to be allocated for temporary storage of an incoming fax. Once the fax receipt is complete, it will be stored in the system memory assigned to the extension as a tiff file. If an incoming fax requires larger storage space, the system will reject part of the fax that does not fit into the specified space.

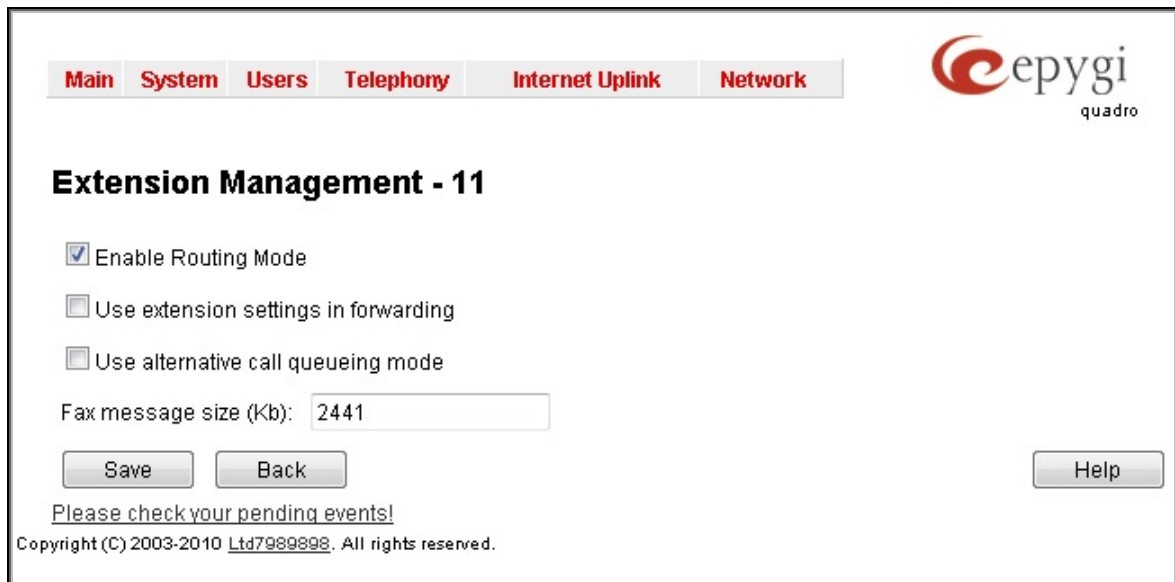


Figure 8 Extension Management page for extension 11

To edit any of the advanced settings using the **Extensions Management** page, click on the link associated with the corresponding extension or attendant.

3.5 fxocfg.cgi

The **Fxo Configuration** page allows configuring the FXO port settings to match the characteristics of PSTN lines (see Figure 9).

Please Note: All settings found on this page are automatically adjusted to the regional settings set in the System Configuration Wizard. It is strongly recommended to leave the settings with their default values unless the characteristics of a specific PSTN line differ from those predefined for the region.

The following settings are found on the **Fxo Configuration** page (see Figure 9):

- **Custom Fxo Disconnect Settings:** Enables the manual configuration of disconnect-call detection settings. Central offices (COs) typically use any or a mixture of the following three methods to notify call disconnection:
 1. **Disconnect notification by sending busy/disconnect tones.** The CO sends specific tones to notify that the call party has hung up. To configure the Quadro to detect disconnect tones and close the call upon detection, enable the **Busy Detection** checkbox and specify the following settings:
 - **Frequency 1:** Frequency of the disconnect tone used by CO. If the CO sends dual tone signals, enter the high frequency of the tone into the **Frequency 1** text field and the low frequency into the **Frequency 2** text field. In case of single tone signals, set the **Frequency 2** to 0.
 - **Duration:** Tone duration in ms.
 - **Duration Disp:** Amount of time (ms) that the tone signal may vary from its assigned value. It should be chosen reasonably large taking into account the value of **Duration**. By default 100 ms is recommended.
 - **Silence and Silence Disp:** Silence duration and the amount (in ms) that it may vary respectively.

- **Pattern Count:** Number of tone(s)-silence patterns the system should identify as a disconnect notification. Leave the factory default value if the exact number is unavailable.

If the Quadro detects signals with above-specified settings, it will close the call.

2. **Disconnect notification by power interruption.** The CO interrupts the power on the line to notify that the call party has hung up. To configure the Quadro to detect power interruption, enable the **Disconnect Detection** checkbox. Specify the **Disconnect Duration** setting, which is the minimum duration (ms) of power interruption to be identified as disconnect notification. If the Quadro does not detect the power for the time specified by the **Disconnect Duration** setting, it will close the call.
3. **Disconnect notification by putting silence on the line.** The CO puts silence on the line to notify that the call party has hung up. To configure the Quadro to close the call upon detecting silence, enable the **Silence Detection** checkbox and specify the following settings:

- **Min Duration:** Minimum duration of silence the system should identify as a disconnect notification.
- **Threshold:** Highest audio level that the system should identify as silence.

If the system detects audio lower than the specified threshold for more than the specified duration, it will close the call.

- **Custom Fxo Data.** Enables the manual configuration of different FXO related settings.
- **Voice Activity Level:** Lowest audio level (dial tone) the system should detect on an FXO line to know that the line is working. The activity detection takes place each time the system takes the FXO line off hook to make an outgoing call.
- **Voice Activity Timeout:** Maximum length of time that the system will wait for the voice activity on an FXO line. If the system does not detect the required voice activity level for the specified length of time, it assumes that that the PSTN line is not working (or no PSTN line is attached to the FXO port) and plays the notification message, "Check your external phone line."

Please Note: The **Voice Activity Level** and **Voice Activity Timeout** settings take effect only if the **Wait for dialtone** checkbox in the `fxoconfig.cgi` is enabled for the corresponding FXO line (see Section [3.6](#)).



Main System Users Telephony Internet Uplink Network

Fxo Configuration

☐ Custom Fxo Disconnect Settings.

☒ Busy Detection

Frequency1:

Frequency2:

Duration:

Duration Disp:

Silence:

Silence Disp:

Pattern Count:

☒ Disconnect Detection

Disconnect Duration:

☐ Silence Detection

Min Duration:

Threshold:

☐ Custom Fxo Data.

Voice Activity Level:

Voice Activity Timeout:

CallerID:

Current Detection Timeout:

Fxo Csrs:

Fxo Ring Stop Timeout:

☐ Enable On Hook Monitoring

Ring Confirmation Time: msec

Please check your pending events!

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Figure 9 The Network Capture page

- Caller ID:** Different standards of Caller ID transmissions. The standard appropriate to the local telephony company should be chosen from this list.

Please Note: The selected standard takes effect only if the **Caller ID** setting for the corresponding FXO line is set to **Default** in the fxoconfig.cgi. Otherwise, the standard selected in fxoconfig.cgi will be used.
- Current Detection Timeout:** Maximum length of time that the system will wait for the current on an FXO line to know that the line is working. On call origination, if the system does not detect the current within this timeout, it assumes that the PSTN line is not working (or no PSTN line is attached to the FXO port) and plays the notification message, "Check your external phone line."

Please Note: The **Current Detection Timeout** setting takes effect only if the **Current Detection** checkbox in the fxoconfig.cgi is enabled for the FXO line (see Section 3.6).
- Fxo Csrs:** Country-specific regional setting that should correspond to the line settings of the local telephony company. The value for this setting is set according to the regional settings selected in the System Configuration wizard. Please do not modify this setting unless instructed to do so by Epygi's technical support engineers.
- Fxo Ring Stop Timeout:** Period of time after which the system will validate ringing termination if no new ring pulse is received from CO. The value for this setting should be slightly greater than the time between two successive ringing signals sent by CO. Note that too small value for this setting will cause the system to drop the call before the next ringing signal arrives. Too large of a value will cause the system to ring the phone long after the

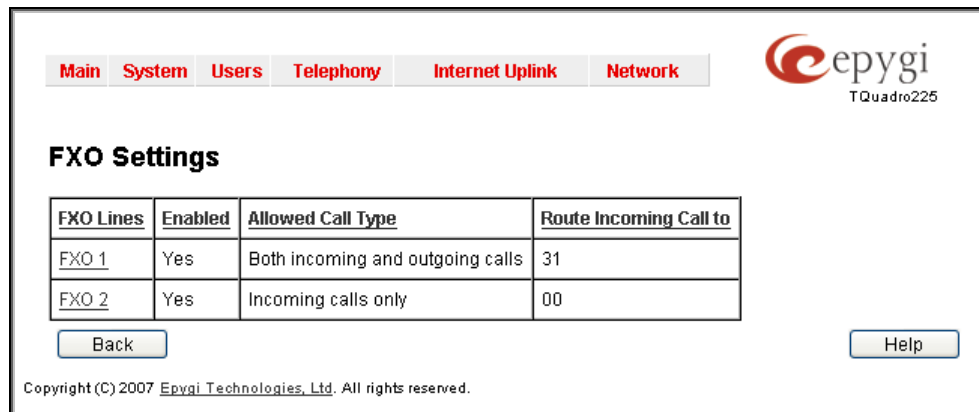
remote party has hung up, i.e. if the remote party hangs up before this timeout expires, the local user will hear busy tones when answering the ringing phone.

- **Enable On Hook Monitoring:** Enables on-hook monitoring of Caller ID signals. If selected, the system will continuously monitor FXO lines to detect Caller ID signals that are not preceded by a notification signal. If this checkbox is not selected, the system will detect only Caller IDs that are transmitted after a notification signal.
- **Ring Confirmation Time:** Minimum length of the ringing signal (in ms) required for the system to validate an incoming call. Ringing signals shorter than the specified time will be ignored by the system.

Please Note: For the instructions on how to determine the correct values for FXO settings appropriate to your CO, please see the troubleshooting guide *FXO Diagnostics*, which is available under the Downloads section of Epygi Technical Support Centre at www.epygi.com.

3.6 fxoconfig.cgi

The **FXO Settings** page allows configuring each FXO line (see Figure 10). As opposed to the `fxocfg.cgi` that contains common settings for all FXO lines, the `fxoconfig.cgi` allows for distinct configuration of each FXO line.



FXO Lines	Enabled	Allowed Call Type	Route Incoming Call to
FXO 1	Yes	Both incoming and outgoing calls	31
FXO 2	Yes	Incoming calls only	00

Back Help

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Figure 10 The FXO Settings page

To modify the settings of an FXO line, click on the link associated with the line on the **FXO Settings** page. This will open the **FXO Settings** for the selected line containing the following settings (see Figure 11):

- **Enable FXO:** Enables the corresponding FXO line.
- **Wait for dialtone:** Enables checking the availability of the connected PSTN line by detecting voice activity (dial tone) on the line. If enabled, the system will proceed with placing an outgoing PSTN call only if the required voice activity level is detected on the line. The voice activity and the timeout to wait for it are set in the `fxocfg.cgi`. If disabled, the system will send DTMFs regardless of whether or not the PSTN line is working.
- **DSP Echo Cancellation:** Enables the DSP echo canceller for the FXO line. The echo canceller uses adaptive linear filtering for linear processing of the echo that can be followed by non-linear processing depending on the residual signal level.
 - **Echo Tail:** Maximum anticipated length of echo signals. The setting value is expressed in samples with the sampling rate of 8000 samples/sec (8 samples = 1 ms) and is used in the adaptive liner filter.
 - **Echo NLP Diff:** Minimum value for the **Echo NLP Diff** setting, which is the difference in echo level before and after linear processing. If the difference in echo level is greater than the specified value, the echo signals will be further processed by the non-linear processor to remove the residual echo signal.

- **Echo NLP Shift:** Number of times to shift right the residual signal to decrease the echo signal level. This is equivalent to multiplying the specified number by 6 dB and deducting it from the residual signal strength. Note that this setting is used only if the difference in echo level is greater than the value specified by the **Echo NLP Diff** setting.

Please Note: For the instructions on how to determine the correct values for echo parameters appropriate to your Quadro setup, please see the troubleshooting guide *Echo Cancellation on Quadro*, which is available under the Downloads section of Epygi Technical Support Centre at www.epygi.com.

- **Current Detection:** Enables checking the availability of the connected PSTN line by detecting current on the line. If enabled, the system will proceed with placing an outgoing PSTN call only if the required current is detected on the line. The timeout to wait for the current is set in the `fxocfg.cgi`. If disabled, the system will send DTMFs regardless of whether or not the PSTN line is working.
- **Enable Parallel Handset Detection:** This setting is used when a PSTN line is shared between an FXO port and a parallel-connected phone(s). The checkbox allows the system to detect whether a parallel-connected phone is using the line and prevent placing an outgoing call if the line is in use. If enabled, the system will place an outgoing PSTN call only if the line is not in use. If disabled, the system will place an outgoing PSTN call if the level of the current on the line is sufficient for making a call.
- **Enable Pulse Dialing:** Enables pulse dialing on the line. If enabled, the dialed number will be sent to the PSTN using pulse signaling. If disabled, tone signaling will be used.
- **Wait For Next Ring Before Accept:**

Please Note: Regardless of the dialing method selected for the FXO line, numbers from devices connected to Quadro FXS ports (phones, modems or other terminal equipment) should be dialed using DTMF dialing.
- **Caller ID:** This drop-down list allows selecting the standard of Caller ID transmission used on the PSTN line. The *default* option for this setting configures the system to use the standard selected in the `fxocfg.cgi`.
- **Route Patterns:** This table is used to route incoming FXO calls whose ringing signals match a certain pattern to a specified call routing record. (The duet pattern is *ring-short pause-ring-long pause*; the trio pattern is *ring-short pause-ring-short pause-ring-long pause*). The capability of routing calls to different destinations based on the ringing pattern can be used, for example, to separate fax calls from ordinary phone calls.

FXO Settings - FXO 1

☒ Enable FXO

☒ Wait for dialtone

☒ DSP Echo Cancellation

Echo Tail:

Echo NLP Diff:

Echo NLP Shift:

☒ Current Detection

☐ Enable Parallel Handset Detection

☐ Enable Pulse Dialing

☐ Wait For Next Ring Before Accept

Caller ID:

Enabled	Pattern Name	Ring Duration	Pause Duration	Route Pattern
<input type="checkbox"/>	duet	Min 150 - 250 Max	Min 350 - 450 Max	<input type="text"/>
<input type="checkbox"/>	trio	Min 150 - 250 Max	Min 350 - 450 Max	<input type="text"/>

Please check your pending events!

1 - Any changes in the Ring Duration or Pause Duration settings take effect for all FXO lines

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Figure 11 The FXO Settings page for FXO 1

3.7 fxodisconnect.cgi

The **Fxo Disconnect Scenarios** page allows selecting one of the five basic scenarios for the disconnect-call notification methods. Each scenario is a set of prescribed values for busy tone parameters and duration of power interruption used by COs in specific countries (see Table 2).

The **Disconnect Scenario** drop-down list contains the five basic scenarios for call disconnection. The **Default** option corresponds to the scenario selected based on the system regional settings. Selecting a different disconnect scenario is useful if the Quadro does not recognize the disconnect signals sent from the CO resulting in calls not releasing.

Fxo Disconnect Scenarios

Disconnect Scenario:

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Figure 12 The FXO Disconnect Scenarios page

Table 2

Settings	Scenario A	Scenario B	Scenario C	Scenario D	Scenario E
Frequency 1	425	450	425	400	480
Frequency 2	0	400	0	0	625
Duration	300	300	250	500	300
Duration Disp	100	100	100	100	100
Silence	300	300	250	500	300
Silence Disp	100	100	100	100	100
Pattern	3	3	3	3	3
Disconnect Duration	200	200	200	200	200

3.8 generalconfig.cgi

The **General Configuration** page allows configuring the system general settings (see Figure 13). The following settings are found on this page:

- **Max Number of Records in DB cache:** Maximum number of records in the external SIP callers' cache. The setting was used with a feature available in older software versions.
- **DNS cache MAX size:** Maximum number of DNS records (hostnames to IP address resolutions) stored for Call Manager's reference.
- **DNS cache cleanup timeout:** Time after which the system will refresh all records stored in the DNS cache.
- **Flash timeout:** Time that the system will wait for the input of a next digit(s) to handle the keys as a key combination (for example 'Flash'+ '3'). If no digit is pressed within the specified timeout, the system will assume that the Flash key has been followed by the key 0. Digits pressed after the timeout expiration will be ignored.
- **Call progress notification timeout:** Period of time after which the system will play the message "Trying to connect, please wait" if no response has been received from the remote endpoint when establishing a SIP call.
- **IP line registration timeout:** Registration timeout to be included in configuration files of supported IP phones generated by the Quadro. A registration timeout is the time after which an IP phone will try to re-register on the Quadro.
- **IP line registration timeout minimum:** It is the minimum registration timeout which is supported by the Quadro. If some device tries to register on the Quadro with the less registration timeout, Quadro will reject that registration with "Too Brief" SIP error message.
- **Play user friendly voice messages instead of tones:** Response type (system voice messages or busy tones) to receive from the system if it encounters a problem while executing user-initiated operations (e.g., the user tries to dial out when all FXO lines are in use). If the *default* or *always* options are selected, the system will play a voice message to notify the user that a problem occurred. When the *never* option is selected, the system will send busy tones.
- **Templates for CallerID:** The text boxes allow specifying which component(s) of the received caller ID the system should send to the called extension's phone. The following options for forming a caller ID pattern are available:
 - %a (automatic) – send the caller ID as provided by the calling party's service provider.
 - %d (display name) – send the calling party's display name.
 - %u (username) – send the calling party's SIP or PSTN number.
 - %h (hostname, available for SIP calls only) – send the hostname of the calling party's SIP URL.

Please Note: These templates for CallerID are effective for FXS lines only.

The **IP call**, **PBX call**, **PSTN call** text boxes are used to assign different caller ID patterns to different call types. For example, to configure the system to send the display name and host name for incoming SIP calls, enter `%d%h` into the **IP call** text box.

- **Presence:** This section allows to specify the count of subscriptions which can be supported by the Quadro.
- **Subscription limitation:** This text box allows to specify the number of subscriptions which can be supported by the Quadro. Increasing this value can bring instability of Quadro.
- **Accept stray SIP requests:** Indicates whether the system should accept SIP requests without checking the presence of the destination URI in the sip user agent.
- **Change SIP Error Code to Busy Here:** Indicates whether the system should substitute SIP error code 480 (Temporary Unavailable) with code 486 (Busy Here). As responses to SIP error 480 may be implemented differently depending on the SIP device, error code substitution is useful to ensure identical responses from different SIP devices.
- **Ignore To header in incoming SIP INVITE requests:** When this feature is enabled and in the incoming SIP "INVITE" message "To" header and "Request Line" contains different information, Quadro will ignore information provided in the "To" header of SIP "INVITE" message and will use the information which is provided in the "Request Line" of the same SIP "INVITE" message.
- **Use Rport:** Indicates whether the system should add an empty **Rport** to SIP packets. The **Rport** parameter allows the sender (the Quadro) to request proxy servers to send the responses to a particular address and port (the originating address and port) if it supports the feature.
- **Enable IP Loop:** Indicates whether the system should make a SIP call when extension-to-extension calls are made using extensions' SIP usernames. If disabled, the system will make a PBX call in the mentioned scenario. This checkbox is designed for testing purposes only. Please do not enable this checkbox unless instructed to do so by Epygi's technical support engineers.
- **Force Hold Music:** Forces Quadro's music on hold (MOH) to be sent to the remote party when the Quadro is behind an IP/PSTN provider that does not properly support remote MOH. If the checkbox is disabled, the Quadro will send an SIP re-INVITE (HOLD) message with the `c=IN IP4 0.0.0.0, a=sendonly` media attributes to notify the server that the local party has placed the remote party on hold. In this case, some servers completely close the receive audio stream resulting in the remote party not hearing the Quadro's MOH. If the checkbox is enabled, the Quadro will send an SIP re-INVITE message with active media attributes, leaving local IP (`c=`) address and media stream direction (`a=`) unchanged, forcing the server to send Quadro's MOH to the remote party.
- **Do Not Send External RE-INVITE:** When this feature is enabled Quadro will not send external "RE-INVITE" message to the remote party. This feature can be used for solving Hold/Un-Hold, Transfer and similar problems.
- **Do Not Send REFER:** When this feature is enabled Quadro will not send external "REFER" message to the remote party. This feature can be used when the remote party doesn't support SIP "REFER" message.
- **Enable SIP IDS:** When this feature is enabled Quadro will automatically detect the SIP attacks and block that remote party by adding appropriate rule in the Firewall.
- **Security Level:** This feature specifies the security password level which will be used for local IP phones registration. Also, depending on the Security Level, during the System Diagnostic, Quadro can provide additional information about its security protection.
- **Callback through Routing:** This feature allows to make a Call Back (*2 feature code) through the Call Routing table. This feature is relevant to use, when you want the system depending on the caller ID to decide how to make the Call Back. If you do not have proper routing for call back destination, the Call Back feature will not work.

- **Enable VM silence disconnect:** This feature allows to stop the Voice Mail recording when Quadro receives silence from remote party and close that call.
- **Disconnect timeout:** This drop-down list allows to specify the silence timeout, after which Voice Mail recording will be automatically stopped and that call will be closed.

The screenshot displays the 'General Configuration' page of the Quadro Advanced Configuration interface. The page is divided into several sections:

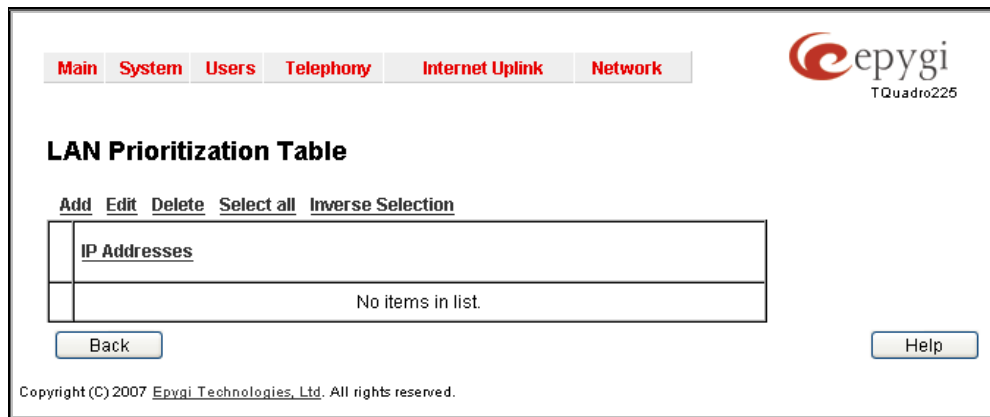
- Navigation Tabs:** Main, System, Users, Telephony, Internet Uplink, Network.
- General Configuration Section:**
 - Max Number of Records in DB cache:** 32 recs
 - DNS cache MAX size:** 32 recs
 - DNS cache cleanup timeout:** 6 hours
 - Flash timeout:** 2 sec
 - Call progress notification timeout:** 10 sec
 - IP line registration timeout:** 3600 sec
 - IP line registration timeout minimum:** 120 sec
 - Play user friendly voice messages instead of tones:** default
 - Templates for Caller ID:**
 - IP call: %a (%a%d%u%h)
 - PBX call: %a (%a%d%u)
 - PSTN call: %a (%a%d%u)
 - Presence:**
 - Subscription limitation: 180
 - Enable VM silence disconnect:** ☒ (checked)
 - Disconnect timeout:** 10
 - Accept stray SIP requests:** ☐
 - Change SIP Error Code to Busy Here:** ☐
 - Ignore To header in incoming SIP INVITE requests:** ☐
 - Use Rport:** ☒ (checked)
 - Enable IP Loop:** ☐
 - Force Hold Music:** ☐
 - Do Not Send External RE-INVITE:** ☒ (checked)
 - Do Not Send REFER:** ☐
 - Callback through Routing:** ☐
 - Enable SIP IDS:** ☐
 - Security Level:** High
- Buttons:** Save, Back, Help.
- Footnote:** Please check your pending events!
- Footnote:** 1 - %a - automatically, %d - display name, %u - user name or phone number, %h - host name or IP address
- Footnote:** 2 - Warning: The device should be rebooted manually to apply changed value of this field. Increasing this value can bring instability of this device.
- Footnote:** 3 - Warning: For SIP IDS to take effect the firewall should be enabled.
- Copyright:** Copyright (C) 2003-2010 Epygi Technologies, Ltd. All rights reserved.

Figure 13 The General Configuration page


3.9 lantable.cgi

By default, all traffic from the LAN to WAN interfaces has ordinary priority. The **LAN Prioritization Table** allows changing the precedence of the traffic created by a particular IP device residing on the LAN from ordinary to high. Prioritizing the outbound traffic of a device can be useful, for example, in network configurations where one Quadro resides on the LAN of another Quadro. In this case the Quadro residing on the LAN should have higher traffic priority on the other Quadro to ensure better call quality.

To change the priority of a device from ordinary to high, add its IP address to the **LAN Prioritization** table (see Figure 14). Note that IP phones should not be added to the table as they are added implicitly once registered on the Quadro. To add an entry to the table, click the **Add** button and specify the IP address of the device (see Figure 15).



[Main](#) [System](#) [Users](#) [Telephony](#) [Internet Uplink](#) [Network](#)

 TQuadro225

LAN Prioritization Table

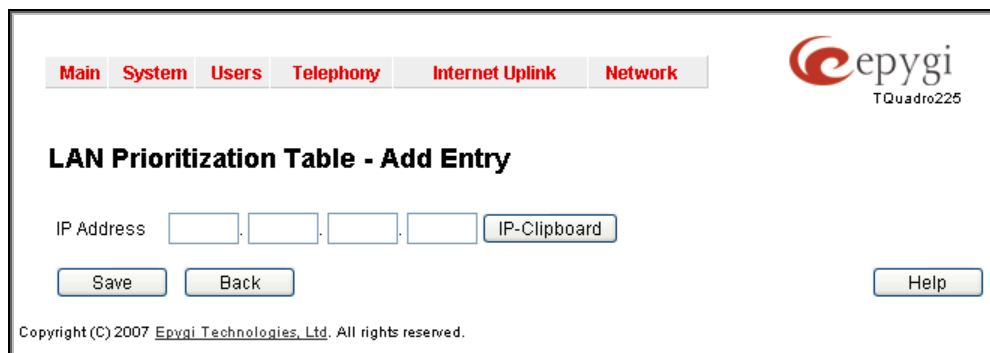
[Add](#) [Edit](#) [Delete](#) [Select all](#) [Inverse Selection](#)

IP Addresses
No items in list.


[Back](#) [Help](#)

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Figure 14 The LAN Prioritization Table



[Main](#) [System](#) [Users](#) [Telephony](#) [Internet Uplink](#) [Network](#)

 TQuadro225

LAN Prioritization Table - Add Entry

IP Address [IP-Clipboard](#)

[Save](#) [Back](#) [Help](#)

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Figure 15 Adding an Entry to LAN Prioritization Table

3.10 linesconfig.cgi

The **Lines Configuration** page allows configuring FXS lines (see Figure 16). Each FXS line has an associated tab on the page that should be clicked to access the line settings.

The following settings can be configured for a selected FXS line:

- **Prering Caller ID System:** The standard of caller ID transmissions to be used on the line when sending the caller ID prior the first ring.
- **Postring Caller ID System:** The standard of caller ID transmissions to be used on the line when sending the caller ID between the first and second rings.
- **Send off-hook Caller ID:** Enables caller ID transmission to the FXS line when the line is in a call (provided that the call waiting feature is enabled on the extension).
- **DT-AS Level:** The level of alert signals preceding off-hook caller ID signals.
- **Send Only Mandatory Caller ID:** If selected, only certain fields of caller ID information will be sent to the called party when using the FSK system. These fields include *caller ID*, *caller name* and *time* as defined by the ETSI standard. Selecting this checkbox is useful if the called party's phone does not recognize all FSK parameters and doesn't show Caller ID on incoming calls.
- **Flash MIN Duration and Flash MAX Duration:** The Flash button causes the line to go on-hook for a short time then return back to the off-hook state again. The **Flash MIN Duration** and **Flash MAX Duration** specify the minimum and maximum durations of keeping the line on-hook for the system to assume the Flash button is pressed (i.e. if the line goes on hook for longer than the specified **Flash MIN Duration** period of time, but shorter than the specified **Flash MAX Duration**, then the system will assume that the Flash button has been pressed). On-hook intervals shorter than the **Flash MIN Duration** will be ignored by the system. On-

hook intervals longer than the **Flash MAX Duration** will be interpreted as hanging up the handset.

Figure 16 The Lines Configuration page

- **DSP Echo Cancellation:** This checkbox enables the DSP echo canceller for the FXS line. For the meaning of the **DSP Echo Tail**, **DSP Echo NLP Diff**, and **DSP Echo NLP Shift**, see Section 3.6.
- **Codec Echo Cancellation:** This checkbox enables the echo canceller running on the hardware codec (ADC/DAC).

Please Note: It is strongly recommended that only one echo canceller run on the line. If the DSP echo canceller is enabled, leave the codec echo canceller disabled.

Please Note: The DSP echo canceller provides a better suppression of echo compared with the Codec echo canceller.

3.11 netcapture.cgi

The **Network Capture** page allows network packets captures from a specified network interface (see Figure 17). This page contains the following settings:

- **Capture on interface:** Network interface from where the capture should be made. The **Local Loopback Interface** option is available for QuadroM units only and is used to capture the traffic within the unit.
- **Stop after receiving count packet:** Number of packets to be captured.

Please Note: After the capture is initiated, the system will create a file and log all the packets meeting the specified criteria into it. Specifying a large number for this setting or leaving it unspecified and making a long capture may consume system resources and cause a system reboot.

- **Restrict to host:** Packets can be captured from the specified IP address.
- The **Capture All Packets** and **Capture Specific Protocol Packets** radio buttons are used to choose between capturing all the packets on the interface and restricting the capture to specific packets only (ARP, SIP, DNS, and RTP).

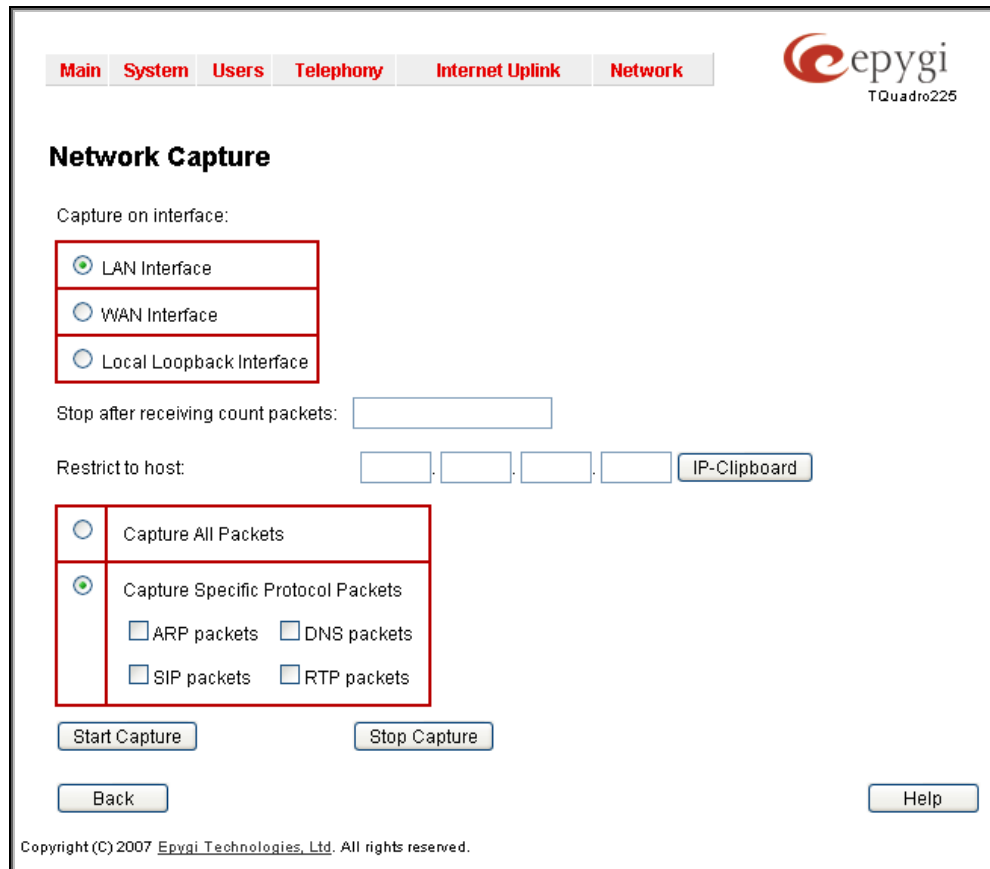


Figure 17 The Network Capture page

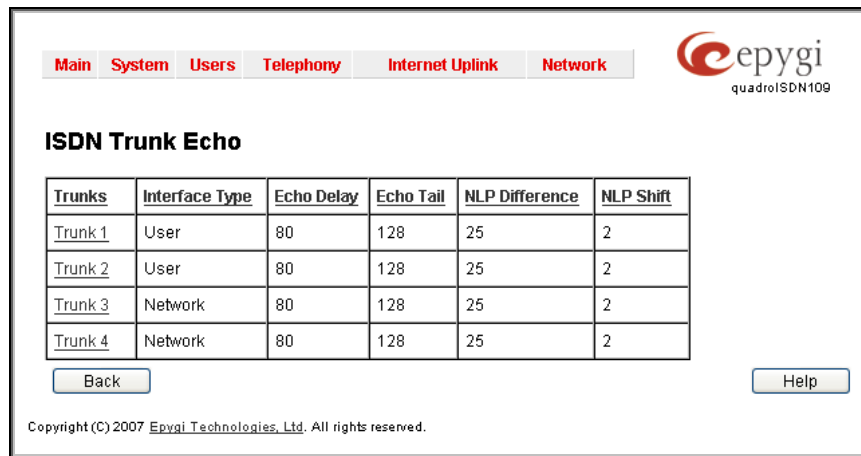
3.12 trunkecho.cgi

The **ISDN Trunk Echo** page allows configuring echo settings of ISDN and E1/T1 trunks (see Figure 18). To modify the settings of an ISDN (or E1/T1) trunk, click on the link associated with the trunk. This will open the **Trunk – Echo** page for the corresponding trunk (see Figure 19):

The following settings are found on the **Trunk – Echo** page:

- **Echo Delay:** Delay between the original voice and the corresponding echo signal. For the meaning of the **DSP Echo Tail**, **DSP Echo NLP Diff**, and **DSP Echo NLP Shift**, see Section [3.6](#).
- **Advanced Echo Settings:** Indicates whether the system should automatically switch off the DSP echo canceller if no echo is detected on the call, i.e. if the call path does not contain any echo source, such as FXO or FXS interfaces.
 - If the **Disable** option is selected, the echo canceller will be on regardless of whether or not echo occurs on the call.
 - If the **Enable** option is selected, the system will automatically switch off the canceller if the transmitted signal level is higher than the specified threshold (**Transmit Energy**) and the received signal level is lower than the specified threshold (**Receive Energy**) for the specified length of time (**Process Saturation**) and these conditions hold true for the specified number of times (**Pattern Count**). Note that the higher the **Transmit Energy**

level and the lower the **Receive Energy** level, the stricter the conditions for switching the canceller off.

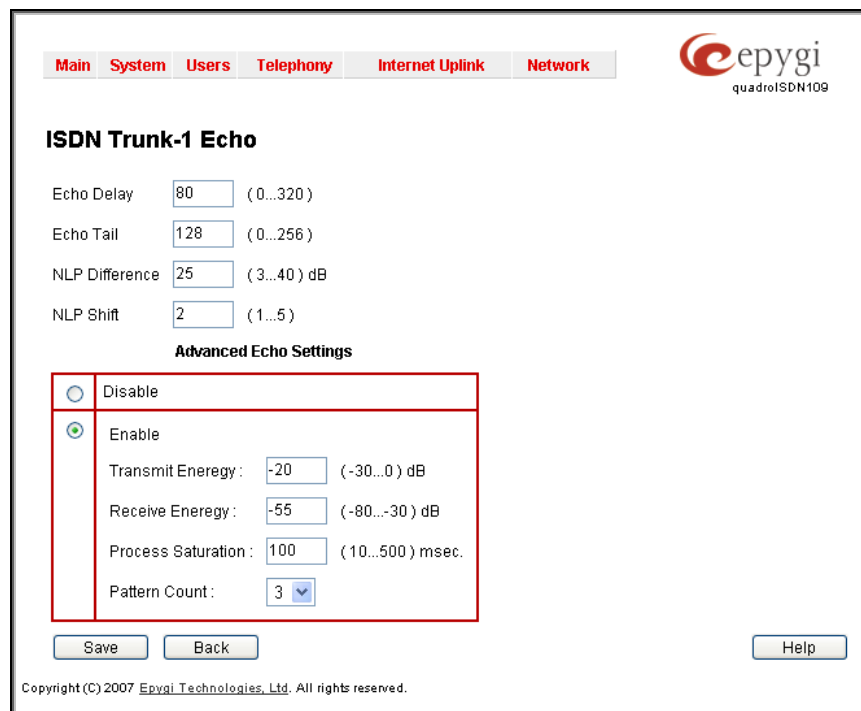


Trunks	Interface Type	Echo Delay	Echo Tail	NLP Difference	NLP Shift
Trunk 1	User	80	128	25	2
Trunk 2	User	80	128	25	2
Trunk 3	Network	80	128	25	2
Trunk 4	Network	80	128	25	2

Back Help

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Figure 18 The ISDN Trunk Echo table



Echo Delay: 80 (0...320)

Echo Tail: 128 (0...256)

NLP Difference: 25 (3...40) dB

NLP Shift: 2 (1...5)

Advanced Echo Settings

☐ Disable

☒ Enable

Transmit Energy: -20 (-30...0) dB

Receive Energy: -55 (-80...-30) dB

Process Saturation: 100 (10...500) msec.

Pattern Count: 3

Save Back Help

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Figure 19 Configuring Echo Settings of ISDN Trunk-1

3.13 tos.cgi

The **TOS Settings** page allows specifying the value for the **Type of Service** setting (ToS) for SIP and RTP packets (see Figure 20). The ToS setting is used in IP packets to notify routers how to handle the packet (precedence, delay, reliability, etc.).

- **low, medium, high:** Values for the ToS setting for low, medium and high priority RTP packets respectively. The priority level of RTP is chosen from the Users→Extension Management→Edit Entry page for the selected extension (see Figure 21).
- **SIP tos:** The ToS value for SIP packets.

The screenshot shows the 'TOS Settings' page. At the top, there is a navigation bar with tabs: Main, System, Users, Telephony, Internet Uplink, and Network. The 'Telephony' tab is selected. The Epygi logo and 'TQuadro225' are in the top right. The page title is 'TOS Settings'. Under the 'RTP' section, there are three input fields: 'low' with value '0x80', 'medium' with value '0xB8', and 'high' with value '0xF8'. Under the 'SIP' section, there is an input field for 'SIP tos' with value '0x0'. At the bottom, there are 'Save', 'Back', and 'Help' buttons. A copyright notice at the bottom reads: 'Copyright (C) 2007 Epygi Technologies, Ltd. All rights reserved.'

Figure 20 The TOS Settings page

The screenshot shows the 'Extensions Management - Edit Entry' page. At the top, there is a navigation bar with tabs: Main, System, Users, Telephony, Internet Uplink, and Network. The 'Telephony' tab is selected. The Epygi logo and 'quadro' are in the top right. The page title is 'Extensions Management - Edit Entry'. On the left, there is a sidebar with links: General Settings, SIP Settings, SIP Advanced Settings (selected), Remote Settings, Call Queue Settings, and Voice Mailbox Settings. Below these are links: Go To User Settings, Go To Line Settings, and Go To Codec Settings. The main content area is titled 'SIP Advanced Settings - 13'. It contains several sections: 'Advanced Settings' with fields for 'Authentication User Name', a checkbox for 'Send Keep-alive Messages to Proxy', a 'Timeout (sec)' field with value '60', a circled 'RTP priority level' dropdown menu with 'medium' selected, and a checkbox for 'Do Not Use SIP Old Hold Method'; 'Outbound Proxy' with fields for 'Host address' and 'Port'; 'Secondary SIP Server' with fields for 'Host address', 'Port' (value '5060'), and 'Outbound Proxy for Secondary SIP Server' with fields for 'Host address' and 'Port'. At the bottom, there are 'Save', 'Back', and 'Help' buttons. A note at the bottom reads: 'Please check your pending events!'. A copyright notice at the bottom reads: 'Copyright (C) 2003-2010 Ltd7989898. All rights reserved.'

Figure 21 Choosing RTP priority level

3.14 loadlogo.cgi

The **Upload Company Details** page allows changing the default company logos and the link to the company's website.

Restore Default Logo restores the default company logo. The Company Logo is displayed in the right top corner. If this checkbox is selected, the text field **Upload New Logo** will be disabled.

Upload New Logo text field can be used to type the path where new company logo image file is located. If the image file is browsed with the help of file-chooser, this field displays the path of the browsed file. **Browse** button is used to browse for the image file. The requirements for the company logo are the following:

- File type - *.gif
- Image Sizes - 120x43 pixels

The **Download Logo** link appears only if a file has been uploaded recently. It downloads the image file to the PC and opens a window where the saving location can be specified.

Restore Default Running Logo restores the default company running logo. The Company Running Logo is displayed in the right top corner, during saving or other operations. If this checkbox is selected, the text field **Upload New Running Logo** will be disabled.

Upload New Running Logo text field can be used to type the path where new company running logo image file is located. If the image file is browsed with the help of file-chooser, this field displays the path of the browsed file. **Browse** button is used to browse for the image file. The requirements for the company running logo are the following:

- File type - *.gif
- Image Sizes - 120x43 pixels

The **Download Running Logo** link appears only if a file has been uploaded recently. It downloads the image file to the PC and opens a window where the saving location can be specified.

Figure 22 Upload Company Details page

Restore Default Icon restores the default icon. The Icon is displayed in the browser address bar, favorites, tabs. If this checkbox is selected, the text field **Upload New Icon** will be disabled.

Upload New Icon text field can be used to type the path where new company logo image file is located. If the image file is browsed with the help of file-chooser, this field displays the path of the

browsed file. **Browse** button is used to browse for the image file. The requirements for the company logo are the following:

- File type - *.ico
- Image Sizes – 12x12 pixels

The **Download Icon** link appears only if a file has been uploaded recently. It downloads the audio file to the PC and opens a window where the saving location can be specified.

Restore Default Link restores the default link. The Link is the redirecting address to the Company website for the Logo and Copyright link in the bottom of all pages. If this checkbox is selected, the following text fields will be disabled.

Copyright Link Text text field requires the copyright text visible on the bottom of all pages.

Logo Link Text text field requires the text displayed when mouse over the company logo in the right top corner.

Company Website text field requires the URL where the click on the copyright link or the company logo will take.

3.15 callcontrol.cgi

The **callcontrol.cgi** page allows to activate a call from a WEB, connecting two parties together via auto-generated call. This feature is useful for remote diagnostic purposes.

First Party text field requires the address for the first party of the auto-generated call.

Second Party text field requires the address for the second party of the auto-generated call.

The syntax of the **call type** drop down list for both first and second parties is the following:

Tel: is used for the calls via the call routing table,

Sip: is used when called destination reached through a SIP server,

PSTN: when called destination dialed to PSTN.

Examples:

pbx: 11 – call to extension 11

pstn: 0123456789 – call to 0123456789 PSTN number via FXO, ISDN or E1/T1

sip:20225@sip.epygi.com – call to 20225@sip.epygi.com via SIP

tel:820225 – call to 20225 number via call routing.

First Party Answering Timeout drop down list requires the time period during which the first party has to answer the call in order to transfer it to the second party.

Start Processing button is used to initiate the auto-generated call.

How it works:

The user enters the addresses for the first and second parties and presses the Start Processing button.

The call setup takes the following steps: Quadro makes a call to the “First Party” (Quadro extension or a remote destination). As soon as the extension answers the call, Quadro plays a message, then transfers the call to the second party (remote user) and connects the call between Quadro extension and remote user.

In case if the first party doesn’t answer within the specified timeout, Quadro shows an error in the progress text box and stops processing the call.

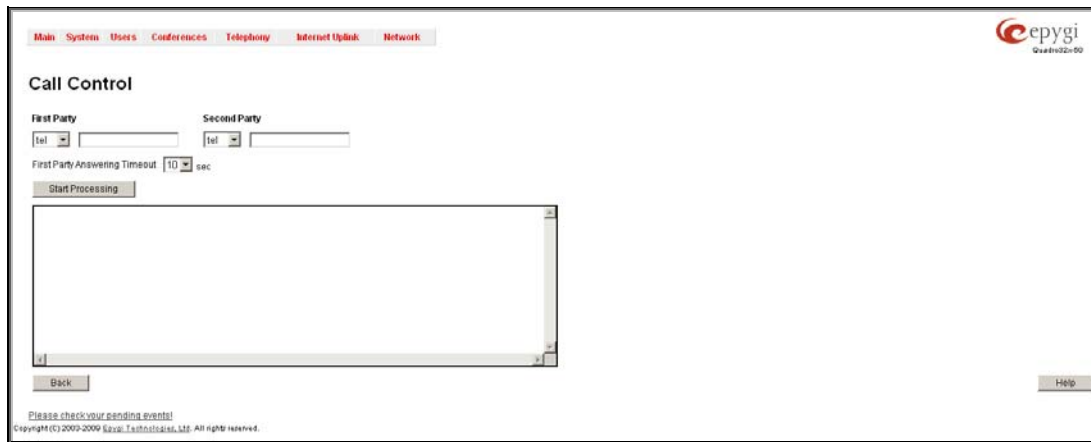


Figure 233 Call Control page

3.16 featuretrial.cgi

The **featuretrial.cgi** page allows activating the Quadro optional features for a trial. This page lists all Quadro features that may be activated for a trial, characterized by a Feature Description and provided with its Status.

Expiration Date/Time- used to specify the trial period. Upon expiring the specified period the Quadro will reboot and trial features will disable.

How it works

User has to select the appropriate checkboxes under **Activate** column, specify the needed count under **Count** Column and save. The Quadro will reboot and trial features activate.

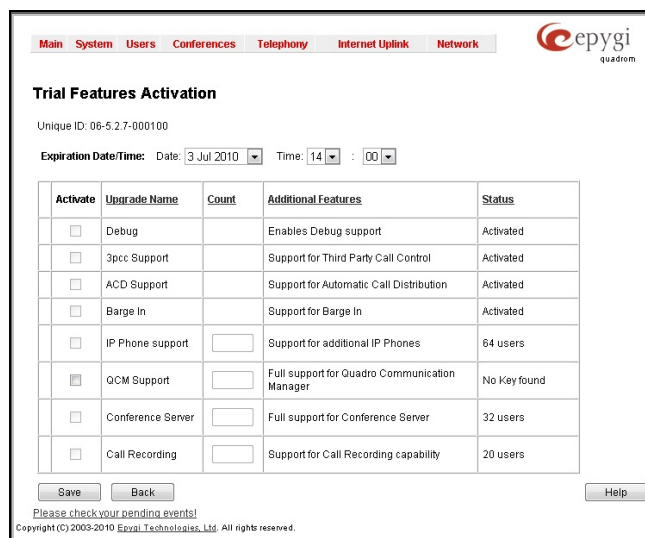
The syntax for values under **Count** is the following:

For G.729 – it is the number for additional 729 calls

For IP Phone support – it is the number for additional IP lines

For QCM Support – the number for licensed QCM's

Call recording- the number for simultaneous call recordings



Activate	Upgrade Name	Count	Additional Features	Status
<input type="checkbox"/>	Debug		Enables Debug support	Activated
<input type="checkbox"/>	3pcc Support		Support for Third Party Call Control	Activated
<input type="checkbox"/>	ACD Support		Support for Automatic Call Distribution	Activated
<input type="checkbox"/>	Barge In		Support for Barge In	Activated
<input type="checkbox"/>	IP Phone support		Support for additional IP Phones	64 users
<input checked="" type="checkbox"/>	QCM Support		Full support for Quadro Communication Manager	No Key found
<input type="checkbox"/>	Conference Server		Full support for Conference Server	32 users
<input type="checkbox"/>	Call Recording		Support for Call Recording capability	20 users

Figure 244 Trial Features Activation page

3.17 kseswitch.cgi

On the Quadro6L and Quadro4Li the **kseswitch.cgi** page allows switching between **KSE Mode** and **PBX Mode** (see Figure 25).

KSE Switch

☒ **KSE Mode** Key System Emulation mode. SLA lines are configured by default.

☐ **PBX Mode** Regular PBX working mode. SLA lines are not configured, but can be set up manually, if needed.

Please check your pending events!
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Figure 255 KSE Switch page

KSE Mode – is a **Key System Emulation mode**. When your Quadro is in **KSE Mode** (first radio button is selected), SLA lines are configured by default. This means that each FXO line attaches to one SLA correspondingly and by default configuration files for IP Phones are generated with **SLA Watching** (see Figure 26).

Key System Emulation

Line	External Trunk	Name	Advanced Settings
SLA1	FXO 1	FXO1	Forwarding for SLA1
SLA2	FXO 2	FXO2	Forwarding for SLA2
SLA3	FXO 3	FXO3	Forwarding for SLA3
SLA4	FXO 4	FXO4	Forwarding for SLA4
SLA5	FXO 5	FXO5	Forwarding for SLA5
SLA6	FXO 6	FXO6	Forwarding for SLA6
SLA7	Inactive		Forwarding for SLA7
SLA8	Inactive		Forwarding for SLA8

[Advanced Configuration](#)

Please check your pending events!
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Figure 266 Key System Emulation page

PBX Mode – is a **Regular PBX working mode**. When your Quadro is in **PBX Mode** (second radio button is selected), SLA lines aren't configured by default, but can be set up manually, if needed. This means that by default no FXO line is attached to SLA and by default configuration files for IP Phones are generated without **SLA Watching**.

Making a selection

If your Quadro is in one of these modes and you want to change it, just select the one you want and press the **save** button, you will be asked to confirm, also you will be informed that the Quadro is going to be rebooted and all your configuration changes will be lost, after pressing the **Yes** button, all configurations will be resetted and Quadro will be rebooted.