



SIP Paging 25V/70V Amplifier Operations Guide

SIP Compliant Part #011579

Document Part #931996E for Firmware Version 21.0.0

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Operations Guide 931996E SIP Compliant 011579

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Technical Support

The fastest way to get technical support for your VoIP product is to submit a VoIP Technical Support form at the following website: https://support.cyberdata.net/

Phone: (831) 373-2601, Ext. 333 Email: support@cyberdata.net

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CyberData Corporation 931996E Operations Guide

Revision Information

Revision 931996E, which corresponds to firmware version 21.0.0, was released on August 6, 2024, and has the following changes:

• Adds the following note below Section 2.4.2, "Line In":

ESD Sensitivity: This equipment may be sensitive to ESD (electro-static discharge). It may cause the system to become unresponsive in some higher than normal ESD environments. As a precaution, during installation, it is best to make all external connections to the unit before powering on.

Pictorial Alert Icons



General Alert

This pictoral alert indicates a potentially hazardous situation. This alert will be followed by a hazard level heading and more specific information about the hazard.



Ground

This pictoral alert indicates the Earth grounding connection point.

Hazard Levels

Danger: Indicates an imminently hazardous situation which, if not avoided, will result in death or serious injury. This is limited to the most extreme situations.

Warning: Indicates a potentially hazardous situation which, if not avoided, could result in death or serious injury.

Caution: Indicates a potentially hazardous situation which, if not avoided, could result in minor or moderate injury. It may also alert users against unsafe practices.

Notice: Indicates a statement of company policy (that is, a safety policy or protection of property).

The safety guidelines for the equipment in this manual do not purport to address all the safety issues of the equipment. It is the responsibility of the user to establish appropriate safety, ergonomic, and health practices and determine the applicability of regulatory limitations prior to use. Potential safety hazards are identified in this manual through the use of words Danger, Warning, and Caution, the specific hazard type, and pictorial alert icons.

Important Safety Instructions

- 1. Read these instructions.
- 2. Keep these instructions.
- 3. Heed all warnings.
- 4. Follow all instructions.
- 5. Do not use this apparatus near water.
- 6. Clean only with dry cloth.
- 7. Do not block any ventilation openings. Install in accordance with the manufacturer's instructions.
- 8. Do not install near any heat sources such as radiators, heat registers, stoves, or other apparatus (including amplifiers) that produce heat.
- 9. Do not defeat the safety purpose of the polarized or grounding-type plug. A polarized plug has two blades with one wider than the other. A grounding type plug has two blades and a third grounding prong. The wide blade or the third prong are provided for your safety. If the provided plug does not fit into your outlet, consult an electrician for replacement of the obsolete outlet.
- 10. Protect the power cord from being walked on or pinched particularly at plugs, convenience receptacles, and the point where they exit from the apparatus.
- 11. Only use attachments/accessories specified by the manufacturer.
- 12. Refer all servicing to qualified service personnel. Servicing is required when the apparatus has been damaged in any way, such as power-supply cord or plug is damaged, liquid has been spilled or objects have fallen into the apparatus, the apparatus has been exposed to rain or moisture, does not operate normally, or has been dropped.
- 13. Prior to installation, consult local building and electrical code requirements.



Warning

Electrical Hazard: This product should be installed by a licensed electrician according to all local electrical and building codes.



Warning

Electrical Hazard: To prevent injury, this apparatus must be securely attached to the rack, wall or desktop in accordance with the installation instructions.

Abbreviations and Terms

Abbreviation or Term	Definition
A-law	A standard companding algorithm, used in European digital communications systems to optimize, i.e., modify, the dynamic range of an analog signal for digitizing.
AVP	Audio Video Profile
Cat 5	TIA/EIA-568-B Category 5
DHCP	Dynamic Host Configuration Protocol
LAN	Local Area Network
LED	Light Emitting Diode
Mbps	Megabits per second.
NTP	Network Time Protocol
PBX	Private Branch Exchange
PoE	Power over Ethernet (as per IEEE 802.3af standard)
RTFM	Reset Test Function Management
SIP	Session Initiated Protocol
SRTP	Secure Real Time Protocol
u-law	A companding algorithm, primarily used in the digital telecommunication
UC	Unified Communications
VoIP	Voice over Internet Protocol

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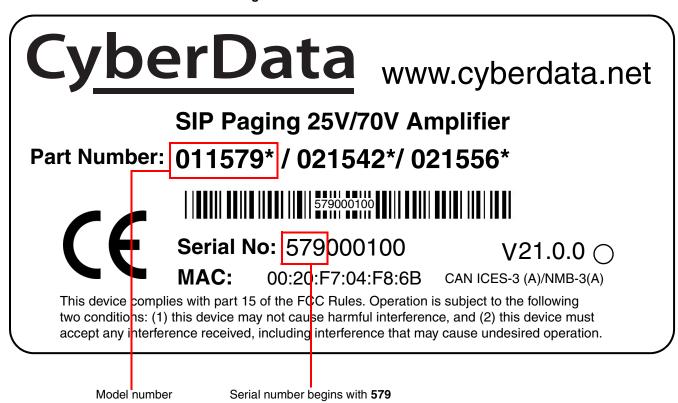
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1.1 How to Identify This Product

To identify the SIP Paging 25V/70V Amplifier, look for a model number label similar to the one shown in Figure 1-1. Confirm the following:

- The model number on the label should be 011579.
- The serial number on the label should begin with 579.

Figure 1-1. Model Number Label¹



^{1.} This figure is just an example. The information on your label may look different.

1.2 Features

- Supports one analog zone with hundreds¹ of 25V or 70V speakers/horns with a single home run cable to the Amplifier.
- · Calendar based bell scheduler
- Support for security code to prevent unwanted paging access
- Supports 100 DTMF controlled paging groups, that can transmit audio via multicast and the single analog channel
- Can page to unlimited endpoints that are enabled to receive multicast or Poly pages on any Poly channel
- Can receive pages directly from Poly phones as well as other devices that can send standard multicast
- Supports call-in stored message recording
- · Paging prioritization
- Loud/Night Ringer function second SIP extension
- · Supports delayed pages, i.e. call buffering
- Supports multiple user accounts
- Master volume adjustment via web interface or front panel LCD controls
- Fault sense input
- · Line-in for background music
- NTP-based internal clock with battery back-up
- LCD display for monitoring device status, configuration and diagnostics
- Configurable alerts for amplifier temperature threshold, protection, and recovery modes.
- 19-inch rack mountable with removable rack ears
- TLS 1.2 and SRTP enhanced security for IP Endpoints in a local or cloud-based environment
- HTTPS web based configuration
- Support for Cisco SRST resiliency
- · Autoprovisioning via HTTPS, HTTP or TFTP
- Configurable event generation for device health and status monitoring
- 802.11q VLAN tagging

^{1.*}Number of speakers supported will depend on the tap setting and the baseline voltage in your typical configuration. Continuous power output is up to 392W when in 70V mode and 160W when in 25V mode.

1.3 Specifications

Table 1-1. Product Specifications

Specifications	
Ethernet I/F	10/100 Mbps
Protocol	SIP RFC 3261 Compatible
Power Input	AC Voltage Input
	100-240VAC, 50/60 Hz
Analog Zone Output:	Continuous power output of up to 392W when driving a 70V distributed speakers, equivalent load > 12-Ohms
	Continuous power output of up to 160W when driving a 25V distributed speakers, equivalent load > 4-Ohms.
Line In:	
Input Signal Amplitudes	2.0 VPP maximum
Input Impedance	10k Ohm
Line Out:	
Output Signal Amplitudes	2.0 VPP maximum
Output Level	+2dBm nominal
Total Harmonic Distortion	0.5% maximum
Output Impedance	10k Ohm
Payload Types	G.711 a-law, G.711 μ-law, G.722, and G.729
Network Security	TLS 1.2, SRTP, HTTPS
Operating Range	Temperature: -40° C to 55° C (-40° F to 131° F)
	Humidity: 5-95%, non-condensing
Storage Temperature	-40° C to 70° C (-40° F to 158° F)
Storage Altitude	Up to 15,000 ft. (4573 m)
Dimensions ^a	19.0 inches [482.6 mm] Length (with rack ears)
	16.85 inches [428.0 mm] Length (w/o rack ears)
	10.83 inches [275.0 mm] Width
	2.25 inches [57.2 mm] Height
Weight	11.2 lbs [5.1 kg]
Boxed Weight	17.8 lbs [8.1 kg]
Compliance	CE: EMC Directive – Class A EN 55032 & EN 55024, LV Safety Directive – EN 62368-1; RoHS Compliant; FCC Part 15 Class A; Industry Canada ICES-3 Class A; IEEE 802.3 Compliant; TAA Compliant
Warranty	2 Years Limited
Part Number	011579

a. Dimensions are measured from the perspective of the product being upright with the front of the product facing you.

1.4 Compliance

1.4.1 CE Statement



As of the date of manufacture, the Paging Series has been tested and found to comply with the specifications for CE marking and standards per EMC and Radio communications Compliance. This applies to the following products: 011145, 011146, 011233, 011280, 011295, 011314, 011368, and 011372.

EMC Directive - Class A Emissions, Immunity, and LV Safety Directive, RoHS Compliant. Flammability rating on all components is 94V-0.

1.4.2 FCC Statement



This equipment has been tested and found to comply with the limits for a Class A digital device, pursuant to part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference when the equipment is operated in a commercial environment. This equipment generates, uses, and can radiate radio frequency energy and, if not installed and used in accordance with the instruction manual, may cause harmful interference to radio communications. Operation of this equipment in a residential area is likely to cause harmful interference in which case the user will be required to correct the interference at his own expense.

CAUTION: Changes or modifications not expressly approved by the manufacturer responsible for compliance could void the user's authority to operate the equipment.

1.4.3 Industry Canada (IC) Compliance Statement

Operation is subject to the following two conditions:

- 1. This device may not cause interference, and
- 2. This device must accept any interference, including interference that may cause undesired operations of the device.

ICES-3 Class A

2 Setting Up the SIP Paging 25V/70V Amplifier

The topics in this chapter provide information on setting up, configuring, and using the SIP Paging 25V/70V Amplifier.

2.1 Parts List

The packaging for the SIP Paging 25V/70V Amplifier includes the parts in Table 2-1.

Table 2-1. Parts List

Quantity	Part Name	Illustration
1	SIP Paging 25V/70V Amplifier	
1	Installation Quick Reference Guide	Checked and an analysis of the control of the contr
1	Power cord	
2	Accessory Kit Terminal Blocks	
6	Rubber Feet	000 000
2	Mounting Ears	
4	Mounting Screws	
1	Spare Screw Kit	

2.2 Typical Installation

Figure 2-1 illustrates how the SIP Paging 25V/70V Amplifier is normally installed as part of a paging system.

Music Source (Background Music) 011579 SIP Paging 25V/70V Amplifier Supports 10k Ohm Multicast VoIP Phone System or IP to Analog **On-Premise** Typical 25V or 70V Speaker Deployment PoE Switch Office Area SIP Phone Multicast **Customer Show Room** SIP CyberData SIP/Multicast Endpoints Parking Lot

Figure 2-1. Typical Hybrid Analog/IP Installation

Example Analog Speaker Matrix

Note: Configuration as pictured above			
	Qty	Power Tap	Total
Ceiling Speakers	6	3.75W	22.5W
Wallmount	4	10W	40W
Horns	3	15W	45W
			107.5W

2.3 General Wire Recommendation for 25V or 70V Speakers

For the majority of installations we recommend the following specifications for the speaker wiring.

16/2 (16AWG/2 Conductor) Gray Stranded In-Wall CL3R Speaker Wire (Figure 2-1)

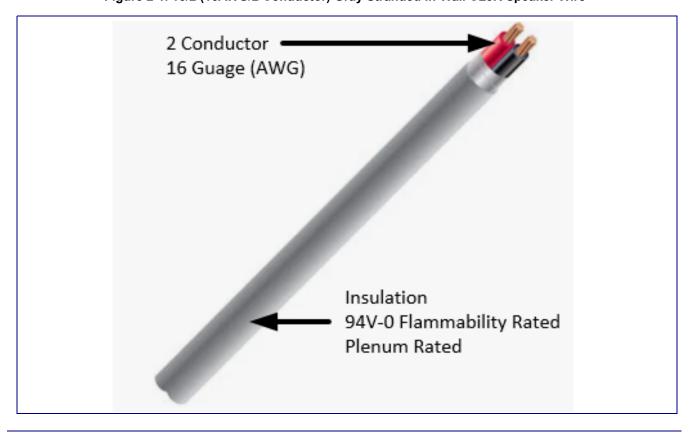
- · Use with speaker systems in an indoor audio system
- · Class 3, riser rated for in-wall installation in riser and non-riser spaces
- · PVC insulating jacket
- 2 fully annealed class B stranded bare copper conductors
- 16-Gauge
- UL Type CL3R
- Plenum rated

There are many brands that will work, but we have personally tested the Southwire company. The description below are the details on this wire in a 500 foot length.

P50002 SY 16/2 STR CU OAS CMP/CL3P CMP/CL3P/FPLP FT6 PLENUM SHIELDED STRANDED 185 25.07

Note Different gauge wiring can be used from our standard 16/2 recommendation if specific distances or power levels are trying to be maintained. Feel free to consult with our Design Services Group for additional assistance.

Figure 2-1. 16/2 (16AWG/2 Conductor) Gray Stranded In-Wall CL3R Speaker Wire

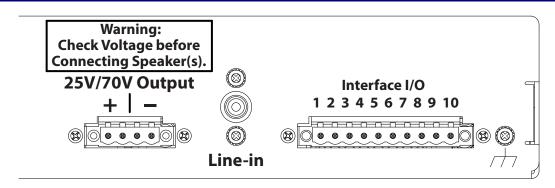


2.4 Connecting the SIP Paging 25V/70V Amplifier

Before you connect the SIP Paging 25V/70V Amplifier, be sure that you have received all of the parts described in Section 2.1, "Parts List".

See Figure 2-2 for the connection options that are available for the SIP Paging 25V/70V Amplifier.

Figure 2-2. Connection Options



25V/70V Output

There are two connections to hook analog speakers up to the amp. The connections accept up to a 12 AWG wire.

Line in

Line level input is for background music applications. The expected input level is 2Vpp (2 volts peak to peak voltage).

Interface I/O

- 1 Fault Sense Input (Common)
- 2 Fault Sense Input (Sense)
- 3-4 Reserved
- 5 Ground Reference
- 6 Relay Contact Common
- 7 Relay Contact Normally Open
- 8 Line Level Output (+)
- 9 Line Level Output (-)
- 10 Reserved

2.4.1 Ground Connection

This connection allows you to connect the device to an electrical ground.

2.4.2 Line In

This RCA 10K Ohm Hi-Z input connection allows you to connect an external music player to the internal amplifier.



ESD Sensitivity: This equipment may be sensitive to ESD (electro-static discharge). It may cause the system to become unresponsive in some higher than normal ESD environments. As a precaution, during installation, it is best to make all external connections to the unit before powering on.

2.4.3 Page Port Output Connections

Table 2-2. Page Port Output Connections

Description
Fault Sense Input (Common). See Section 2.4.3.1, "Pin 1 and 2—Fault Sense Input (Common/Sense)."
Fault Sense Input (Sense). See Section 2.4.3.1, "Pin 1 and 2—Fault Sense Input (Common/Sense)."
Reserved
Reserved
Ground Reference
Relay Contact - Common ^a . See Section 2.4.3.2, "Pin 6 and 7—Relay Contact (Common/Normally Open)".
Relay Contact - Normally Open ^a . See Section 2.4.3.2, "Pin 6 and 7—Relay Contact (Common/Normally Open)".
Line Level Output (+). See Section 2.4.3.3, "Pin 8 and 9 - Line Out."
Line Level Output (-). See Section 2.4.3.3, "Pin 8 and 9 - Line Out."
Reserved

a. 1 Amp at 30 VDC for continuous loads

2.4.3.1 Pin 1 and 2—Fault Sense Input (Common/Sense)

This input was designed as a method of monitoring an external amplifier that is equipped with a fault sense relay.

When enabled via the web interface (Section 2.8.13, "Configure the Fault Detection Parameters"), this input (when closed) will play a user uploadable audio file out of the line-out connection and/or place a SIP call to a pre-determined extension and play that file.

2.4.3.2 Pin 6 and 7—Relay Contact (Common/Normally Open)

When enabled on the web interface (Section 2.8.5, "Configure the Device Parameters"), every time an audio file is played out of the local line-out or 600 Ohm output, the relay will close, thereby enabling amplifiers with a remote turn-on capability to become active.

2.4.3.3 Pin 8 and 9 - Line Out

These RCA 10K Ohm Hi-Z output connections allow you to connect the device to The RCA line-in (10K Ohm Hi-Z) of an external audio amplifier.

2.4.4 Connect to the Power Source

To supply power, connect the SIP Paging 25V/70V Amplifier to a standard 100-240VAC 50/60Hz power supply. If required, connect the earth grounding wire to the chassis ground on the back of the unit. See Figure 2-3.

Figure 2-3. Connecting to the Power Source



To set up the device, connect the device to your network:

Power Supply

 Connect the Amplifier to a standard 100-240VAC, 50/60Hz external power supply

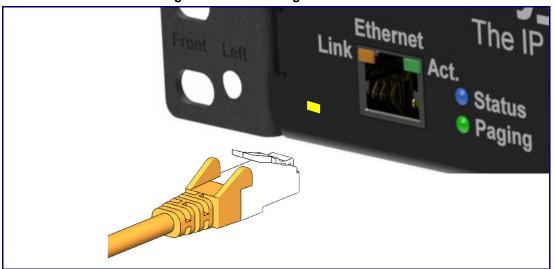
Chassis Ground

 Connect the earth grounding wire to the Chassis Ground. See the figure on the left.

2.4.5 Connect to the Network

Plug one end of a standard Ethernet cable into the device **Ethernet** port. Plug the other end into your network.

Figure 2-4. Connecting to the Network



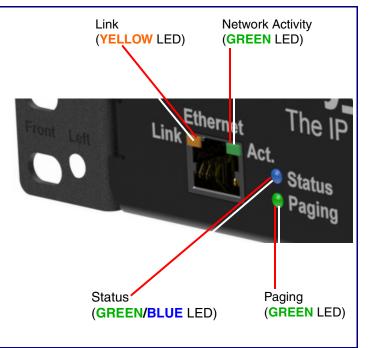
2.4.6 Confirm that the SIP Paging 25V/70V Amplifier is Up and Running

The LEDs on the front of the SIP Paging 25V/70V Amplifier verify the unit's operations.

Figure 2-5. LEDs

When you apply power, turn ON the Power switch and connect the Ethernet to a live network:

- The square, YELLOW Link LED above the **Ethernet** port indicates that the network connection has been established at 100Mbit speed.
- The square, **GREEN Network Activity** LED above the Ethernet port will blink to indicate network activity.
- The round GREEN/BLUE Status LED is BLUE, indicating power is ON. Once the device is initialized, the LED will blink GREEN at one second intervals.
- The round **GREEN Paging** LED comes on after the device is booted and initialized. This LED blinks when a page is in progress. You can disable Beep on Initialization on the Device Configuration page.



2.4.6.1 Verify Network Activity

The square, GREEN Network Activity LED blinks when there is network traffic.

2.5 LCD Display Explanation

The LCD Display can be interacted with via the Toggle and Enter buttons on the front panel of the device. See Figure 2-6.

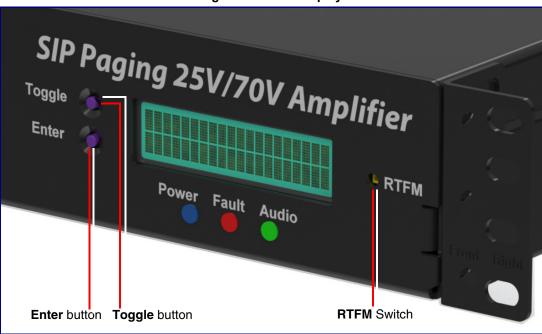


Figure 2-6. LCD Display

- The Toggle button is used to switch between menu pages and on specific pages to toggle between options.
- The **Enter** button is used on specific pages to confirm a setting.
- LCD Display menu pages:
 - Screen 1: CyberData Splash Screen
 - Screen 2: Serial Number and Mode (25v or 70v)
 - Screen 3: Temperature (Celsius /Fahrenheit) and fan status (on or off)
 - Screen 4: IP Address and MAC Address
 - Screen 5: Firmware Version and Part Number
 - Screen 6: Master Volume Level
 - Screen 7: Test Audio (Shows in Green)

2.6 Announcing the IP Address

To announce the IP address for the SIP Paging 25V/70V Amplifier, complete the following steps:

- 1. Use a paper clip to press the RTFM button, bringing up the RTFM screen. See Figure 2-7.
- 2. Use the toggle button to select **Speak IP Address** and then **Enter** to activate.
- 3. If a speaker is connected, the device will announce the IP address.



Figure 2-7. RTFM Switch

2.7 Restore the Factory Default Settings

The SIP Paging 25V/70V Amplifier is delivered with factory set default values for the parameters in Table 2-3. Use the RTFM switch (see Figure 2-8) on the front of the unit to restore these parameters to the factory default settings.

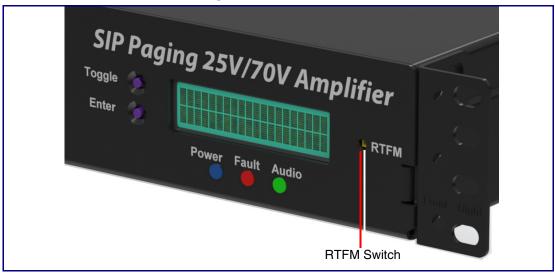


Figure 2-8. RTFM Switch

When you perform this procedure, the factory default settings are restored. The default Note parameters for access are shown in Table 2-3.

Factory Default Setting
DHCP
192.168.1.23
admin
admin
255.255.255.0
192.168.1.1

Table 2-3. Factory Default Settings

To restore these parameters to the factory default settings:

- 4. Use a paper clip to press the RTFM button, bringing up the RTFM screen.
- 5. Use the toggle button to select **Restore Defaults**, and then **Enter** to activate.
- 6. Selecting Restore Defaults will bring up the confirmations screen, where selecting Enter will restore defaults.
- 7. If a speaker is connected, the device will announce, "restoring default configuration" and "rebooting."

a. Default if there is not a DHCP server present.

2.8 Configuring the SIP Paging 25V/70V Amplifier

Use this section to configure the SIP Paging 25V/70V Amplifier.

2.8.1 Gather the Required Configuration Information

Have the following information available before you configure the SIP Paging 25V/70V Amplifier.

2.8.1.1 Static or DHCP Addressing?

Know whether your system uses static or dynamic (DHCP) IP addressing. If it uses static addressing, you also need to know the values to assign to the following SIP Paging 25V/70V Amplifier parameters:

- IP Address
- Subnet Mask
- **Default Gateway**

2.8.1.2 Username and Password for Configuration GUI

Determine the Username and Password that will replace the defaults after you initially log in to the configuration GUI.

- The Username is case-sensitive, and must be from four to 25 alphanumeric characters long.
- The Password is case-sensitive, and must be from four to 20 alphanumeric characters long.

2.8.1.3 SIP Settings

To configure the SIP parameters, determine whether you want to register with the server. If you do, determine the number of minutes the registration lease remains valid, and whether you want to automatically unregister when you reboot. To configure the SIP parameters, you also need to determine the values for these parameters:

- SIP Server IP Address
- Remote and Local SIP Port Numbers
- SIP User ID, and Authenticate ID and Password for this User ID

2.8.2 SIP Paging 25V/70V Amplifier Web Page Navigation

Table 2-4 shows the navigation buttons that you will see on every SIP Paging 25V/70V Amplifier web page.

Table 2-4. Web Page Navigation

Web Page Item	Description
Home	Link to the Home page.
Device	Link to the Device page.
Network	Link to the Network page.
SIP	Link to go to the SIP page.
PGROUPS	Link to the PGROUPS page.
SSL	Link to the SSL page.
Schedules	Link to the Schedules page.
Fault	Link to the Fault page.
WiFi-Buttons	Link to the WiFi-Buttons page.
Audiofiles	Link to the Audiofiles page.
Events	Link to the Events page.
Autoprov	Link to the Autoprovisioning page.
Firmware	Link to the Firmware page.
Events Autoprov	Link to the Autoprovisioning page.

2.8.3 Using the Toggle Help Button

The Toggle Help button allows you to see a short description of some of the settings on the webpage. To use the Toggle Help button, do the following:

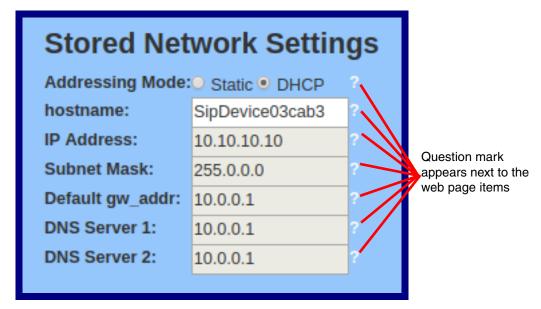
1. Click on the Toggle Help button that is on the UI webpage. See Figure 2-9 and Figure 2-10.

Figure 2-9. Toggle/Help Button

Toggle Help

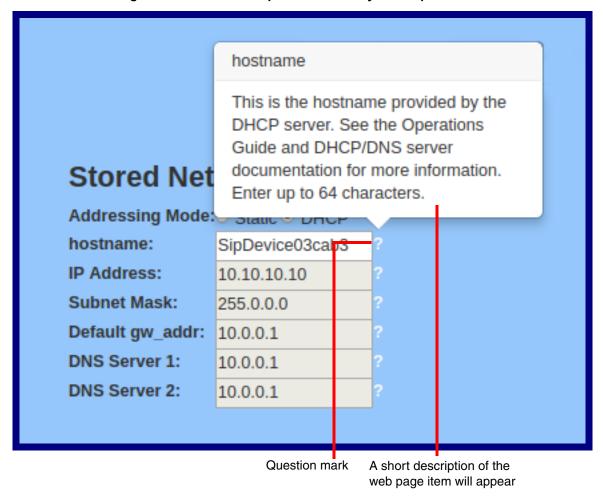
2. You will see a question mark (2) appear next to each web page item that has been provided with a short description by the Help feature. See Figure 2-10.

Figure 2-10. Toggle Help Button and Question Marks



3. Move the mouse pointer to hover over the question mark (?), and a short description of the web page item will appear. See Figure 2-11.

Figure 2-11. Short Description Provided by the Help Feature



2.8.4 Log in to the Configuration GUI

1. Open your browser to the SIP Paging 25V/70V Amplifier IP address.

Note If the network does not have access to a DHCP server, the device will default to an IP address of 192.168.1.23.

Note Make sure that the PC is on the same IP network as the SIP Paging 25V/70V Amplifier.

Note You may also download CyberData's VoIP Discovery Utility program which allows you to easily find and configure the default web address of the CyberData VoIP products.

CyberData's VoIP Discovery Utility program is available at the **Downloads** tab on the following webpage:

https://www.cyberdata.net/pages/011579

The unit ships in DHCP mode. To get to the user login page, use the discovery utility to scan for the device on the network and open your browser from there.

Note To work with the SIP Paging 25V/70V Amplifier configuration *after* the initial configuration, log in using the IP address you assign to the device. Section 2.8.6, "Configure the Network Parameters" provides instructions for entering the IP address.

2. Use the following default username and password on the **User Login** page. (Figure 2-12):

username: admin password: admin

Change the Default Username and Password

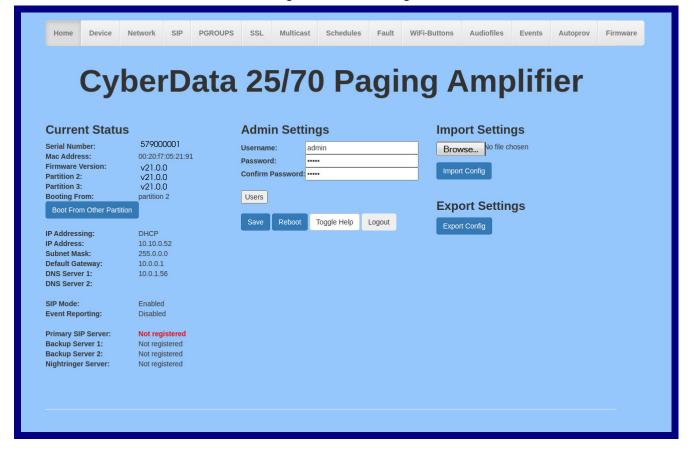
To change the default web access username and password from the **Home** page (Figure 2-13):

- Enter the new username from four to 25 alphanumeric characters in the Change Username field. The username is case-sensitive.
- 2. Enter the new Password from four to 20 alphanumeric characters in the **Change Password** field. The Password is case-sensitive.
- 3. Enter the new password again in the **Re-enter New Password** field.
- Click Save Settings.

Figure 2-12. User Login Page



Figure 2-13. Home Page



5. On the Home Page, review the setup details and navigation buttons described in Table 2-5

Note The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-5. Home Page Overview

Web Page Item	Description
Admin Settings	
Username ?	The username to access the web interface. Enter up to 25 characters.
Password ?	The password to access the web interface. Enter up to 25 characters.
Confirm Password ?	Confirm the web interface password.
Current Status	
Serial Number	Shows the device serial number.
Mac Address	Shows the device Mac address.
Firmware Version	Shows the current firmware version.
Partition 2	Contains a complete copy of bootable software.
Partition 3	Contains an alternate, complete copy of bootable software.
Booting From	Indicates the partition currently used for boot.
Boot From Other Partition	Allows the user to boot from the alternate partition.
IP Addressing	Shows the current IP addressing setting (DHCP or static).
IP Address	Shows the current IP address.
Subnet Mask	Shows the current subnet mask address.
Default Gateway	Shows the current default gateway address.
DNS Server 1	Shows the current DNS Server 1 address.
DNS Server 2	Shows the current DNS Server 2 address.
SIP Mode	Shows the current status of the SIP mode.
Event Reporting	Shows the current status of the Event Reporting mode.
Nightringer	Shows the current status of the Nightringer mode.
Primary SIP Server	Shows the current status of the Primary SIP Server.
Backup Server 1	Shows the current status of Backup Server 1.
Backup Server 2	Shows the current status of Backup Server 2.
Nightringer Server	Shows the current status of Nightringer Server.
Import Settings	
Browse	Use this button to select a configuration file to import.
Import Config	After selecting a configuration file, click Import to import the configuration from the selected file.
Export Settings	

Table 2-5. Home Page Overview (continued)

Web Page Item	Description
Export Config	Click Export to export the current configuration to a file.
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.
Logout	Logs out the current user and returns to the user log in page

Figure 2-14. Users List

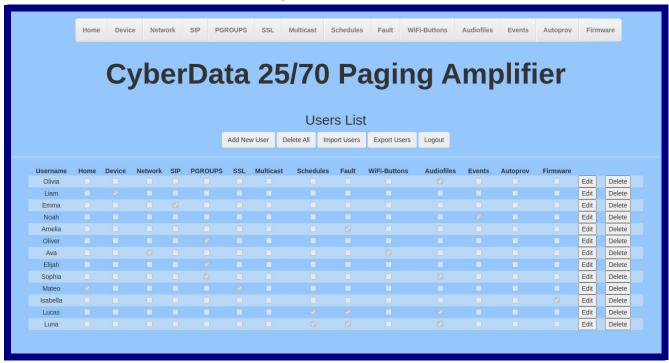


Table 2-6. Users List

Web Page Item	Description
Users List	
Add New User	Adds an authorized user to the user list
Delete All	Deletes all users from the current list
Import Users	Imports a previously exported user list
Export Users	Saves the current user list as a .json file. Please note that the data is encrypted and cannot be edited outside of the device Users dialog box.
Logout	Logs out the current user and returns to the user log in page
Edit	Allows administrator to modify the user's profile
Delete	Deletes the profile
Username	Name of the user
Home	Authorizes a user to view the Home page, allowing the user to see status, but does not grant administrative privileges.
Device	Authorizes a user to view and edit the Device web page.
Network	Authorizes a user to view and edit the Network web page.
SIP	Authorizes a user to view and edit the SIP web page.
PGROUPS	Authorizes a user to view and edit the PGROUPS web page.
SSL	Authorizes a user to view and edit the SSL web page.
Schedules	Authorizes a user to view and edit the Schedules web page.
Fault	Authorizes a user to view and edit the Fault web page.
WiFi-Button	Authorizes a user to view and edit the WiFi-Button web page.
Audiofiles	Authorizes a user to view and edit the Audiofiles web page.
Events	Authorizes a user to view and edit the Events web page.
Autoprov	Authorizes a user to view and edit the Autoprov web page.
Firmware	Authorizes a user to view and edit the Firmware web page.

Figure 2-15. Add New User

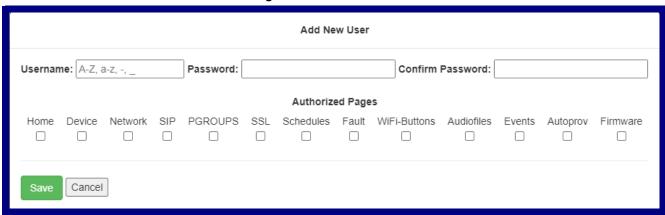


Table 2-7. Add New User

Web Page Item	Description	
Add New User		
Username ?	The username to access the web interface. Enter up to 25 characters.	
Password ?	The password to access the web interface. Enter up to 25 characters.	
Confirm Password ?	Confirm the web interface password.	
Authorized Pages		
Home	Authorizes a user to view the Home page, allowing the user to see status, but does not grant administrative privileges.	
Device	Authorizes a user to view and edit the Device web page.	
Network	Authorizes a user to view and edit the Network web page.	
SIP	Authorizes a user to view and edit the SIP web page.	
PGROUPS	Authorizes a user to view and edit the PGROUPS web page.	
SSL	Authorizes a user to view and edit the SSL web page.	
Schedules	Authorizes a user to view and edit the Schedules web page.	
Fault	Authorizes a user to view and edit the Fault web page.	
WiFi-Button	Authorizes a user to view and edit the WiFi-Button web page.	
Audiofiles	Authorizes a user to view and edit the Audiofiles web page.	
Events	Authorizes a user to view and edit the Events web page.	
Autoprov	Authorizes a user to view and edit the Autoprov web page.	
Firmware	Authorizes a user to view and edit the Firmware web page.	
Save	Click the Save button to save your configuration settings.	
Cancel	Closes the dialog box without saving changes	

Table 2-7. Add New User(continued)

Web Page Item	Description
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.

At this point you can:

- Review the SIP Paging 25V/70V Amplifier's Current Settings. Use the RTFM switch to restore the factory default settings. See Section 2.7, "Restore the Factory Default Settings".
- Configure the device parameters. Click on the **Device** button and see Section 2.8.5, "Configure the Device Parameters" for instructions.
- Configure the network parameters. Click on the **Network** button and see Section 2.8.6, "Configure the Network Parameters" for instructions.
- Configure the SIP parameters. Click on the SIP button and see Section 2.8.7, "Configure the SIP Parameters" for instructions.
- Configure the PGROUPS parameters. Click on the **PGROUPS** button and see Section 2.8.8, "Configure the Paging Groups (PGROUPS) Parameters" for instructions.
- Configure the fault detection parameters. Click on the Fault button and see Section 2.8.13, "Configure the Fault Detection Parameters" for instructions.
- Configure the audio parameters. Click on the Audiofiles button and see Section 2.8.15, "Configure the Audio Parameters" for instructions.
- Configure the event parameters. Click on the **Events** button and see Section 2.8.16, "Configure the Event Parameters" for instructions.
- Configure the autoprovisioning parameters. Click on the Autoprov button and see Section 2.8.17, "Configure the Autoprovisioning Parameters" for instructions.

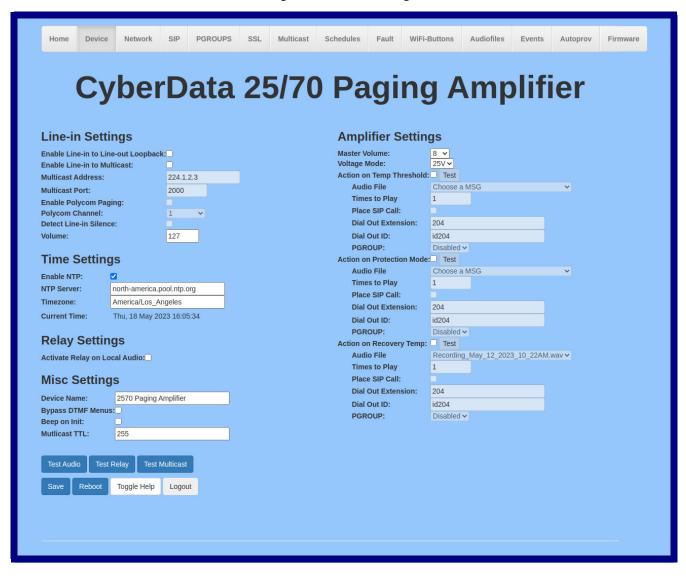
Note Click on the **Firmware** button any time you need to upload new versions of the firmware. See Section 2.9, "Upgrade the Firmware" for instructions.

2.8.5 Configure the Device Parameters

Miscellaneous device settings such as the page prompt and analog options are configured on this page. In addition, you may also enable Polycom Paging to page Polycom IP phones using their proprietary Polycom Paging protocol.

1. Click on the **Device** button to open the **Device** page. See Figure 2-16.

Figure 2-16. Device Page



2. On the **Device** page, you may enter values for the parameters indicated in Table 2-8.

Note The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-8. Device Configuration Parameters

Web Page Item	Description
Line-in Settings	
Enable Line-in to Line-out Loopback ?	Line-in audio will play back out the device's audio output ports. This is the lowest priority audio and will be preempted by any other audio stream.
Enable Line-in to Multicast ?	Line-in audio will be sent to the specified multicast address and port. Playback priority is determined by receiver(s)
	Cannot be combined with Play Line-in Audio via Multicast (Fault Detection)
Multicast Address ?	Address line-in audio will be sent to.
Multicast Port ?	Port line-in audio will be sent to (1-65535).
Enable Polycom Paging ?	Enabling Polycom Paging will result in a standard RTP multicast being sent to the specified address and port and a Polycom Group Paging multicast being sent to the specified address and port+1.
Polycom Channel ?	Destination channel for Polycom Group Paging multicast.
Detect Line-in Silence ?	If audio drops below a threshold on line-input, the fault line-in multicast stream will be stopped to reduce network traffic.
Volume	Volume setting for the line-in (0-127).
Time Settings	
Enable NTP ?	Sync device's local time with the specified NTP Server.
NTP Server ?	Use this field to set the address (in IPv4 dotted decimal notation or as a canonical name) for the NTP Server. This field can accept canonical names of up to 64 characters in length.
Timezone	Enter the tz database string of your timezone.
	Examples:
	America/Los_Angeles
	America/New_York
	Europe/London
	America/Toronto
	See https://en.wikipedia.org/wiki/List of tz database time zones for a full list of valid strings.
Current Time	Displays the current time.
Relay Settings	
Activate Relay on Local Audio ?	The relay will be activated (closed) when the device is playing audio. Use this to activate an external amplifier when the device is playing audio.
Misc Settings	
Device Name ?	Type the device name. Enter up to 25 characters.
Bypass DTMF Menus ?	Bypassing DTMF will result in all calls being relayed to PGROUP 0 Any security code entered for PGROUP 0 will be ignored if DTMF is bypassed
Beep on Init ?	Device will play the user defined "pagetone" audio file when it boots.

Table 2-8. Device Configuration Parameters (continued)

Multicast TTL The time to live (TTL) for multicast packets that aren't sent from PGROUPS. Amplifier Settings Voltage Mode Switch between 25v or 70v for the amplifier output voltage. Supplying a higher voltage Mode switch between 25v or 70v for the amplifier output voltage. Supplying a higher voltage than what your installed speakers are rated for could damage them! Action on Temp Threshold Sea	Web Page Item	Description
Switch between 25v or 70v for the amplifier output voltage. Supplying a higher voltage than what your installed speakers are rated for could damage them! Action on Temp Threshold	Multicast TTL ?	The time to live (TTL) for multicast packets that aren't sent from PGROUPS.
voltage than what your installed speakers are rated for could damage them! Action on Temp Threshold ? Enable an action when the 2570 hits a temperature threshold. Test button is for testing the action settings before saving them. Audio File ? Audio File of Audio File to be used with an action on temperature threshold. Custom stored message files can be uploaded to the board from the Audiofiles page. Number of times to play the selected audio file for a temperature threshold action. The default is set to 1. Place SIP Call ? Enable placing a SIP call on a temperature threshold action. Dial Out Extension ? The extension to dial on a temperature threshold action. Dial Out ID ? The ID to use with the dial out extension on a temperature threshold action. PGROUP ? The multicast PGROUP be use on a temperature threshold action. This uses the settings on the PGROUPS page. The default is set to disable. Action on Protection Mode ? Enable an action when the 2570 goes into protection mode. Test button is for testing the action settings before saving them. Test Audio File ? Audio File to be used with an action on protection mode. Custom stored message files can be uploaded to the board from the Audiofiles page. Number of times to play the selected audio file for a protection mode action. The default is set to 1. Place SIP Call ? Enable placing a SIP call on a protection mode action. Dial Out Extension ? The extension to dial on a protection mode action. The multicast PGROUP to use on a protection mode action. The multicast PGROUP to use on a protection mode action. The multicast PGROUP to use on a protection mode action. The multicast PGROUP to use on a protection mode action. The multicast PGROUP to use on a protection mode action. The multicast PGROUP to use on a protection mode action. The multicast PGROUP to use on a protection mode action. The settings on the PGROUPS page. The default is set to disabled. Action on Recovery Temp ? Enable an action when the 2570 hits a recovery temperatu	Amplifier Settings	
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	Dial Out ID ?	The ID to use with the dial out extension on a recovery temperature action.

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Table 2-8. Device Configuration Parameters (continued)

Web Page Item	Description
PGROUP ?	The multicast PGROUP to use on a recovery temperature action. This uses the settings on the PGROUPS page. The default is set to disabled.
Test Audio	Click on the Test Audio button to do an audio test. When the Test Audio button is pressed, you will hear a voice message for testing the device audio quality and volume.
Test Multicast	This button will cause the device to send a 5 second ULAW multicast stream to 234.2.1.200:2200.
Test Relay	Click on the Test Relay button to do a relay test.
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.
Logout	Logs out the current user and returns to the user log in page

2.8.6 Configure the Network Parameters

Configuring the network parameters enables your network to recognize the SIP Paging 25V/70V Amplifier and communicate with it. Click the **Network** button on the **Home** page to open the **Network** page.

CyberData 25/70 Paging Amplifier **Stored Network Settings VLAN Settings** Addressing Mode: O Static O DHCP VLAN ID (0-4095): SipDevice051c1a Hostname: VLAN Priority (0-7): IP Address: 10.10.10.10 255.0.0.0 Toggle Help Logout Default Gateway: 10.0.0.1 10.0.0.1 DNS Server 1: 10.0.0.1 DNS Server 2: DHCP Timeout in seconds: 60 **Current Network Settings** 10.10.1.51 IP Address: Subnet Mask: 255.0.0.0 Default Gateway: 10.0.0.1 DNS Server 1: 10.0.1.56 DNS Server 2:

Figure 2-17. Network Page

On the Network page, enter values for the parameters indicated in Table 2-9.

The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

Table 2-9. Network Configuration Parameters

Web Page Item	Description		
Stored Network Settings			
Addressing Mode ?	Select either DHCP IP Addressing or Static Addressing by marking the appropriate radio button. DHCP Addressing mode is enabled on default and the device will attempt to resolve network addressing with the local DHCP server upon boot. If DHCP Addressing fails, the device will revert to the last known IP address or the factory default address if no prior DHCP lease was established. See Section 2.7, "Restore the Factory Default Settings" for factory default settings. Be sure to click Save and Reboot to store changes when configuring a Static address.		
Hostname ?	This is the hostname provided by the DHCP server. See the DHCP/DNS server documentation for more information. Enter up to 64 characters.		
IP Address ?	Enter the Static IPv4 network address in dotted decimal notation.		
Subnet Mask ?	Enter the Subnet Mask in dotted decimal notation.		
Default Gateway ?	Enter the Default Gateway IPv4 address in dotted decimal notation.		
DNS Server 1 ?	Enter the primary DNS Server IPv4 address in dotted decimal notation.		
DNS Server 2 ?	Enter the secondary DNS Server IPv4 address in dotted decimal notation.		
DHCP Timeout in seconds ?	Specify the desired time-out duration (in seconds) that the device will wait for a response from the DHCP server before reverting back to the stored static IP address. The stored static IP address may be the last known IP address or the factory default address if no prior DHCP lease was established. Enter up to 8 characters. A value of -1 will retry forever.		
VLAN Settings			
VLAN ID (0-4095) 🛜	Specify the IEEE 802.1Q VLAN ID number. Enter up to 4 digits.		
	Note : The device supports 802.1Q VLAN tagging support. The switch port connected to the device will need to be in "trunking mode" for the VLAN tags to propagate.		
VLAN Priority (0-7)	Specify the IEEE 802.1p VLAN priority level. Enter 1 digit. A value of 0 may cause the VLAN ID tag to be ignored.		
Current Network Settings	Shows the current network settings.		
IP Address	Shows the current Static IP address.		
Subnet Mask	Shows the current Subnet Mask address.		
Default Gateway	Shows the current Default Gateway address.		
DNS Server 1	Shows the current DNS Server 1 address.		
DNS Server 2	Shows the current DNS Server 2 address.		

Table 2-9. Network Configuration Parameters (continued)

Web Page Item	Description	
Save	Click the Save button to save your configuration settings.	
Reboot	Click on the Reboot button to reboot the system.	
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark () appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.	
Logout	Logs out the current user and returns to the user log in page	

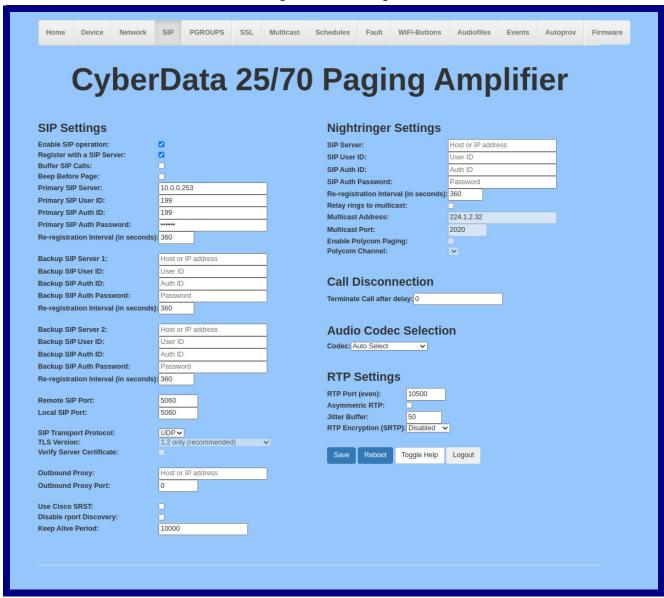
On this page:

- 1. Specify whether you use Static or DHCP IP Addressing by marking the appropriate radio button. If you select Static IP Addressing, go to Step 2.
- 2. For Static IP Addressing, also enter values for the following parameters:
 - The SIP Paging 25V/70V Amplifier's IP Address: The SIP Paging 25V/70V Amplifier is delivered with a factory default IP address. Change the default address to the correct IP address for your system.
 - The Subnet Mask.
 - The **Default Gateway.**

2.8.7 Configure the SIP Parameters

The SIP parameters enable the device to contact and register with the SIP server. On the Home page, click SIP Config to open the SIP page.

Figure 2-18. SIP Page



On the SIP page, enter values for the parameters indicated in Table 2-10.

The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-10. SIP Configuration Parameters

Web Page Item	Description
SIP Settings	
Enable SIP Operation ?	When enabled, the device will transmit, receive, and process SIP messages according to the configured SIP settings below.
Register with a SIP Server ?	When enabled, the device will attempt to register to the configured SIP Server(s) on this page. To configure the device to send and receive point-to-point SIP calls, enable SIP Operation and disable Register with a SIP Server (see Section 2.8.8, "Configure the Paging Groups (PGROUPS) Parameters").
Buffer SIP Calls ?	Also referred to as "delayed paging." Device will buffer up to four minutes of audio then play back the recording after hang up or after the buffer is full.
Beep Before Page ?	Device will play the user defined "pagetone" audio file before playing a SIP page.
Primary SIP Server ?	Enter the SIP server address as an IPv4 address in dotted decimal notation or a fully qualified domain name. This parameter also becomes the host portion of the SIP-URI for the device's extension on the primary SIP server. This field can accept entries of up to 255 characters in length.
Primary SIP User ID ?	Specify the SIP User ID for the Primary SIP Server. This parameter becomes the user portion of the SIP-URI for the device's extension on the primary SIP server. Enter up to 64 alphanumeric characters.
Primary SIP Auth ID ?	Specify the Authenticate ID for the Primary SIP Server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.
Primary SIP Auth Password ?	Specify the Authenticate Password for the Primary SIP Server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.
Re-registration Interval (in seconds)	The SIP Re-registration interval (in seconds) is the SIP Registration lease time, also known as the expiry. The supported range is 30-3600 seconds. Enter up to 4 digits.
Backup SIP Server 1 ?	Enter the backup SIP server address as an IPv4 address in dotted decimal notation or a fully qualified domain name. This parameter also becomes the host portion of the SIP-URI for the device's extension on the backup SIP server. This field can accept entries of up to 255 characters in length.
Backup SIP User ID ?	Specify the SIP User ID for the first backup SIP Server. This parameter becomes the user portion of the SIP-URI for the device's extension on the first backup SIP server. Enter up to 64 alphanumeric characters.
Backup SIP Auth ID ?	Specify the Authenticate ID for the first backup SIP server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.
Sackup SIP Auth Password Specify the Authenticate Password for the first backup SIP server. This parameter required for SIP registration authentication. Enter up to 64 alphanumeric chara	
Re-registration Interval (in seconds)	The SIP Re-registration interval (in seconds) is the SIP Registration lease time, also known as the expiry. The supported range is 30-3600 seconds. Enter up to 4 digits.
Backup SIP Server 2 ?	Enter a second backup SIP server address as an IPv4 address in dotted decimal notation or a fully qualified domain name. This parameter also becomes the host portion of the SIP-URI for the device's extension on the second backup SIP server. This field can accept entries of up to 255 characters in length.

Table 2-10. SIP Configuration Parameters (continued)

Web Page Item	Description	
Backup SIP User ID ?	Specify the SIP User ID for the second backup SIP Server. This parameter becomes the user portion of the SIP-URI for the device's extension on the second backup SIP server. Enter up to 64 alphanumeric characters.	
Backup SIP Auth ID ?	Specify the Authenticate ID for the second backup SIP server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.	
Backup SIP Auth Password ?	Specify the Authenticate Password for the second backup SIP server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters	
Re-registration Interval (in seconds) ?	The SIP Re-registration interval (in seconds) is the SIP Registration lease time, also known as the expiry. The supported range is 30-3600 seconds. Enter up to 4 digits.	
Remote SIP Port ?	The Remote SIP Port is the port number the device will use as the destination port when sending SIP messages. The default Remote SIP Port is 5060. The supported range is 0-65536. Enter up to 5 digits.	
Local SIP Port ?	The Local SIP Port is the port number the device will use to receive SIP messages. The default Local SIP Port is 5060. The supported range is 0-65536. Enter up to 5 digits.	
SIP Transport Protocol ?	Choose the transport protocol for SIP signaling. This will affect all extensions, including the Nightringer. Default is UDP.	
TLS Version ?	Choose the TLS version for SIP over TLS. Modern security standards strongly recommend using TLS 1.2.	
Verify Server Certificate ?	When enabled, the device will verify the authenticity of the server during the TLS handshake by its certificate and common name. The TLS handshake will be aborted if the server is deemed to be inauthentic and SIP registration will not proceed.	
Outbound Proxy ?	Enter the Outbound Proxy address as an IPv4 address in dotted decimal notation or a fully qualified domain name (FQDN). When an IP address is configured, the device will send all SIP messages to this IP address. When an FQDN is configured, the device will run DNS NAPTR, SRV, and A queries on the FQDN to resolve an IP address to which it will send all SIP messages. This field can accept entries of up to 255 characters in length.	
Outbound Proxy Port ?	The Outbound Proxy Port is port number used as the destination port when sending SIP messages to the outbound proxy. A value of 0 will default to 5060. The supported range is 0-65536. Enter up to 5 digits.	
Use Cisco SRST ?	When enabled, the backup servers are handled according to Cisco SRST (Survivable Remote Site Telephony). It is required for use in clustered Cisco Unified Communications Manager topologies.	
Disable rport Discovery ?	Disabling rport Discovery will prevent the device from including the public WAN IP address and port number in the contact information that is sent to the remote SIP servers. This will generally only need to be enabled when using an SBC or SIP ALG in conjunction with a remote SIP server.	
Unregister on Boot ?	When enabled, the device will send one registration with an expiry of 0 on boot.	
Keep Alive Period ?	The minimum time in milliseconds between keep-alive packets sent for nat traversal. A value of 0 will disable keep alive packets.	
Nightringer Settings		
SIP Server ?	Enter the SIP server address as an IPv4 address in dotted decimal notation or a fully qualified domain name. This parameter also becomes the host portion of the SIP-URI for the device's Nightringer extension on the SIP server. This field can accept entries of up to 255 characters in length.	

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Table 2-10. SIP Configuration Parameters (continued)

Web Page Item	Description	
SIP User ID ?	Specify the SIP User ID for the SIP server. This parameter becomes the user portion of the SIP-URI for the device's Nightringer extension. Enter up to 64 alphanumeric characters.	
SIP Auth ID 🛜	Specify the Authenticate ID for the SIP Server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.	
SIP Auth Password ?	Specify the Authenticate Password for the SIP Server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.	
Re-registration Interval (in seconds)	The SIP Re-registration Interval (in seconds) is the SIP Registration lease time, also known as the expiry. The supported range is 30-3600 seconds. Enter up to 4 digits.	
Relay rings to multicast ?	When selected, the device will play ring tones to the specified multicast address and port.	
Multicast Address ?	The multicast address used for nightring audio.	
Multicast Port ?	The multicast port used for nightring audio.	
Send Polycom Page	Enabling Polycom Paging will result in a standard RTP multicast being sent to the specified address and port and a Polycom Group Paging multicast being sent to the specified address and port+1. This will only enable polycom paging for nightringer.	
Polycom Transmit Channel	Destination channel for Polycom Group Paging multicast	
Call Disconnection		
Terminate Call After Delay ?	Automatically terminate an active call after a given delay in seconds. A value of 0 will disable this function. Enter up to 8 digits.	
Audio Codec Selection		
Codec ?	Select desired codec (only one may be chosen).	
RTP Settings		
RTP Port (even) ?	Specify the port number used for the RTP stream after establishing a SIP call. This port number must be an even number and defaults to 10500. The supported range is 0-65536. Enter up to 5 digits.	
Asymmetric RTP ?	Specify if the remote endpoint will send and receive RTP packets on different ports. If set to false, the device will track the address/port that is sending RTP packets during a SIP call. If the address/port changes mid-stream, the device will disregard the SDP and send all further RTP packets to this new address.	
	If set to true, this device will ignore the sending address/port and send RTP as specified in the SDP. Warning! Enabling asymmetric RTP can cause the RTP stream to be lost.	
	Most installations should not enable asymmetric RTP.	
Jitter Buffer ?	Specify the size of the jitter buffer (in milliseconds) used for SIP calls. Valid values are 50-1000.	
RTP Encryption (SRTP) ?	When enabled, a SIP call's audio streams are encrypted using SRTP.	
Save	Click the Save button to save your configuration settings.	
Reboot	Click on the Reboot button to reboot the system.	

Table 2-10. SIP Configuration Parameters (continued)

Web Page Item	Description	
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.	
Logout	Logs out the current user and returns to the user log in page	

For specific server configurations, go to the following website address: Note

https://www.cyberdata.net/pages/connecting-to-ip-pbx-servers

- 1. Enter the IP address of the SIP Server.
- 2. Enter the port numbers used for SIP signaling:
 - a. Remote SIP Port
 - b. Local SIP Port
- 3. Enter the SIP registration parameters:
 - a. SIP User ID
 - b. Authenticate ID
 - c. Authenticate Password
- 4. For SIP Registration, designate whether you want the device to register with your SIP server.
- 5. At Unregister on Reboot:
 - a. Select Yes to automatically unregister the SIP Paging 25V/70V Amplifier when you reboot it.
 - b. Select No to keep the SIP Paging 25V/70V Amplifier registered when you reboot it.
- 6. In the Register Expiration field, enter the number of seconds the SIP Paging 25V/70V Amplifier registration lease remains valid with the SIP Server. The SIP Paging 25V/70V Amplifier automatically re-registers with the SIP server before the lease expiration timeout.

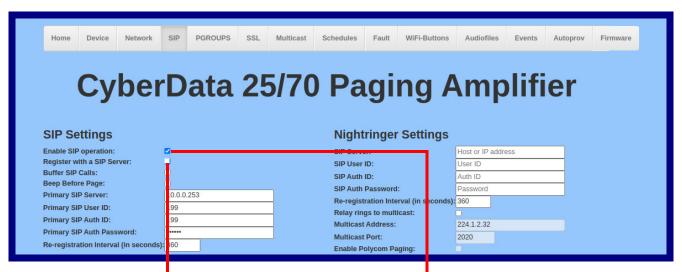
2.8.7.1 Point-to-Point Configuration

It is possible to use the device as a paging endpoint without registering it with a SIP server by configuring it for Point-to-Point paging. To do this, complete the following steps:

- 1. On the SIP page (Figure 2-19), make sure of the following:
 - The Register with a SIP Server parameter is not selected.
 - The Enable SIP Operation parameter is selected
- 2. Click on the Save button to save the changes.
- 3. Click on the **Reboot** button to reboot the device.
- 4. Enter the device's IP address as a "speed dial" (also called "auto-dial") key on the phone(s) from which you want to page.

Note Establishing point-to-point SIP calls may not work with all phones.

Figure 2-19. SIP Page



Register with a SIP Server is not selected

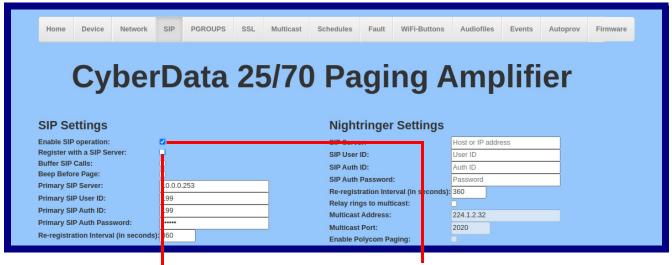
Enable SIP Operation is selected

2.8.7.2 Point-to-Point Fault Sense Reporting

It is possible to use the device to report faults detected at the device's Fault Sense Input without registering it with a SIP server by configuring it for Point-to-Point Fault Sense reporting. To do this, complete the following steps:

- 1. On the SIP page (Figure 2-20), make sure of the following:
 - The Register with a SIP Server parameter is not selected.
 - The Enable SIP Operation parameter is selected

Figure 2-20. SIP Page



Register with a SIP Server is not selected

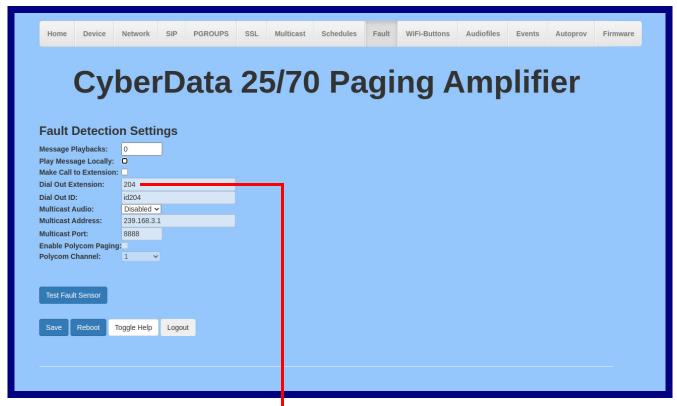
Enable SIP Operation is selected

- 2. Click on the **Save** button to save the changes.
- 3. Click on the Reboot button to reboot the device.

4. On the **Fault** page (Figure 2-21) in the **Dial Out Extension** field, enter the IP address of the phone that is to be called when a fault is detected at the Fault Sense Input.

Note Establishing point-to-point SIP calls may not work with all phones.

Figure 2-21. Fault Page



In the **Dial Out Extension** field, enter the IP address of the phone that is to be called when a fault is detected at the Fault Sense Input.

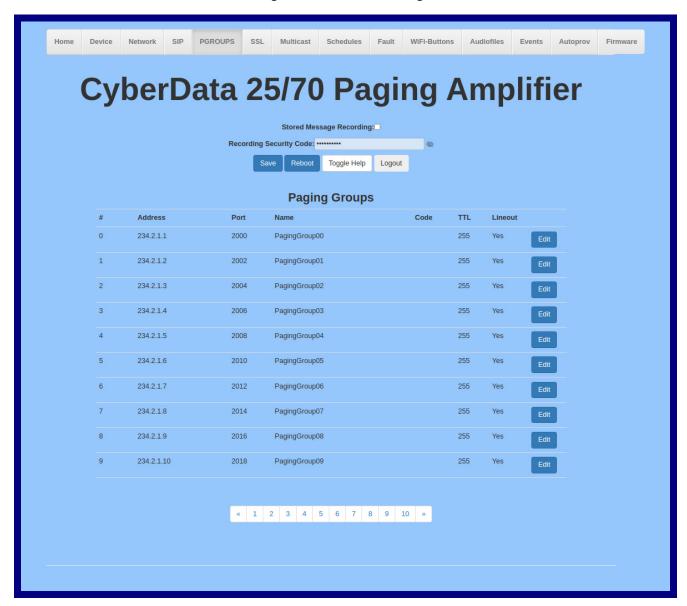
2.8.8 Configure the Paging Groups (PGROUPS) Parameters

Note A PGROUP is a way of assigning multicast addresses and port numbers when configuring multicast paging speakers.

To assign a multicast address, you must first configure the speakers that you want to put into a paging zone by entering a particular multicast address and port number combination in the web configuration for these speakers.

1. Click on the PGROUPS button to open the PGROUPS page. See Figure 2-22.

Figure 2-22. PGROUPS Page



2. On the PGROUPS page, enter values for the parameters indicated in Table 2-11.

Table 2-11. PGROUPS Parameters

Web Page Item	Description	
Stored message recording:	Enabling stored message recording allows a user to call into the device and record a stored message.	
	Note : When recording is enabled, a warning pops up (Figure 2-23): Remember to de-select stored message recording feature once finished recording.	
Recording Security Code: Set the recording security code value, which must ONLY use digit characters to code max length is 25 characters		
Save	Click the Save button to save your configuration settings.	
Click on the Reboot button to reboot the system.		
Click on the Toggle Help button to see a short description of some of the web pag click on the Toggle Help button, and you will see a question mark (?) appear nex the web page items. Move the mouse pointer to hover over a question mark to see description of a specific web page item.		
Logs out the current user and returns to the user log in page		
#	Shows the paging group number.	
Address	Shows the IP address of the PGROUP.	
Port	Shows the port number of the PGROUP.	
Name	Shows the name of the PGROUP.	
Code	Shows the security code of the PGROUP.	
TTL	Shows the "time to live" of the PGROUP.	
Lineout	Shows the Lineout setting of the PGROUP.	
	To make changes to the paging groups, complete the following steps:	
Edit	1. Click on the Edit button to open the Configure PGROUP window (Figure 2-24).	
_	2. In the Configure PGROUP window (Figure 2-24), edit the PGROUP (see Table 2-12, "Configure PGROUP Window") and click on the Save Changes button.	
	3. On the PGROUPS page (Figure 2-22), click on the Reboot button.	
Save	Click the Save button to save your configuration settings.	

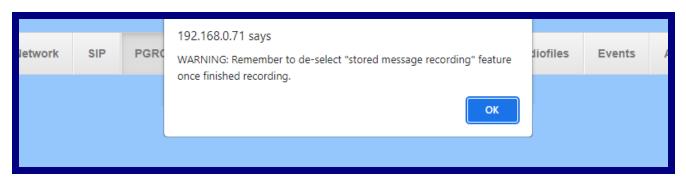
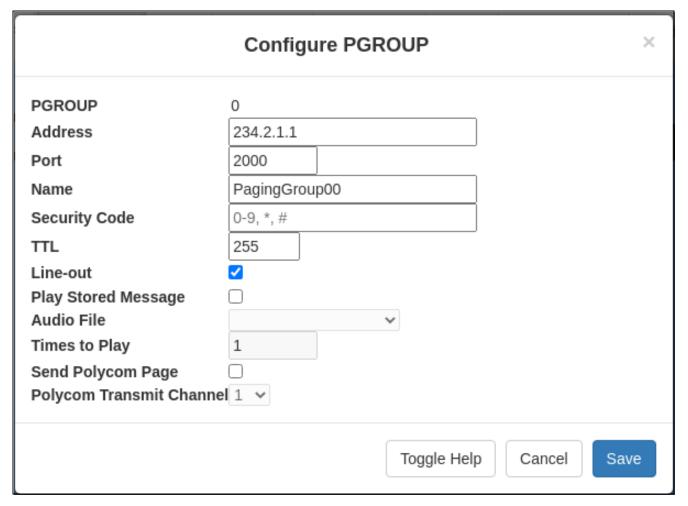


Figure 2-24. Configure PGROUP Window



The parameters for the **Configure PGROUP** window are shown in Table 2-12.

Table 2-12. Configure PGROUP Window

Web Page Item	Description		
PGROUP	Shows the paging group number.		
Address	Enter the IP address of the PGROUP.		
	Note: To disable a relay on a group, use an IP address of 0.0.0.0.		
Port	Enter the port number of the PGROUP.		
	Note : The port range can be from 2000 to 65534 and must be even. When configuring a Paging Group for Polycom Group Paging using an odd-numbered port, configure the next lower even port number. For example, when using the default Polycom paging port 5001 on Polycom phones, configure the next lower even port 5000 for the desired device's Paging Group port.		
Name	Enter a name for the PGROUP.		
Security Code	This field allows the user to add a security code to prevent unauthorized paging to the PGROUP. Code must be between two to five numeric digits (0 through 9). Leave the field empty for no security code. Any security code entered for PGROUP 0 will be ignored if DTMF is bypassed.		
TTL	The TTL field allows you to adjust the TTL. TTL is "time to live" and it describes how many networks (routers) a packet will go through before it is discarded.		
Line-out	The Lineout field determines whether or not the device will play audio out of the RCA output port and the 600 Ohm output port in addition to forwarding it to the PGROUP.		
Play Stored Message	When selected, entering this paging group will play the stored message selected in Audio File		
Audio File	Select a file from the drop down list of audio files previously saved in the Stored Messages portion of the Audiofiles Page .		
Times to Play	Enter the number of times the message will play (1-65535).		
Send Polycom Page	Enabling Polycom Paging will result in a standard RTP multicast being sent to the specified address and port and a Polycom Group Paging multicast being sent to the specified address and port+1. Tied to the specified paging group.		
Polycom Transmit Channel	Destination channel for Polycom Group Paging multicast		
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.		
Cancel	Click the Cancel button to close the Configure PGROUP window.		
Ok	Click the OK button to save your configuration settings.		

2.8.9 Operating the SIP Paging 25V/70V Amplifier

Call behavior changes based on the configuration of the **PGROUPs** page.

2.8.9.1 DTMF Bypassed

- When the SIP Paging 25V/70V Amplifier is called, it will send the "page tone" audio message to the caller.
- When the caller hears this message, the caller should begin speaking.

2.8.9.2 DTMF Not Bypassed

- When the SIP Paging 25V/70V Amplifier is called, it sends the "Enter PGROUP" audio message to the caller. By default, this message is "Enter the two digit zone number."
- When the caller hears this message, the caller should enter the two-digit code for the zone that the caller wants to page.
- If the zone is invalid or not configured, the SIP Paging 25V/70V Amplifier sends the "Invalid PGROUP" audio message to the caller. By default this message is "Invalid zone number. Enter the two digit zone number." The caller should repeat the previous step.
- If a security code is enabled on the zone, the SIP Paging 25V/70V Amplifier sends the "Enter Code" audio message to the caller. By default this message is "Enter the security code." When the caller hears this message, the caller should enter the security code for the selected zone. If no security code is enabled on the zone, the SIP Paging 25V/70V Amplifier will send the "page tone" audio message to the caller. The caller should begin speaking when this message is heard
- If the security code is invalid, the SIP Paging 25V/70V Amplifier will send the "Invalid Code" audio message to the caller. By default this message is "Invalid Security code. Enter the security code." The caller should repeat the previous step. When a valid security code is entered, the SIP Paging 25V/70V Amplifier will send the "page tone" audio message to the caller. The caller should begin speaking when this message is heard.
- For *page-all*, you simply configure *all* speakers with a particular multicast address and port number combination, which represents one of the 100 zones that the device will initially support. Each speaker can still be part of 100 other paging zones in addition to the one *page-all* zone.
- The SIP Paging 25V/70V Amplifier can negotiate the multicast stream via SIP regardless of the bypass state. However, if the SIP Paging 25V/70V Amplifier is not in bypass mode (or the multicast sender does not send any DTMF), the device will not play or relay any audio because the device will be waiting at the zone entry prompt. The DTMF from the sender would have to be sent as RFC2833 RTP events (i.e. "out of band").

2.8.10 Configure the SSL Parameters

1. Click SSL menu button to open the SSL page (Figure 2-25 and Figure 2-26).

Figure 2-25. SSL Configuration Page

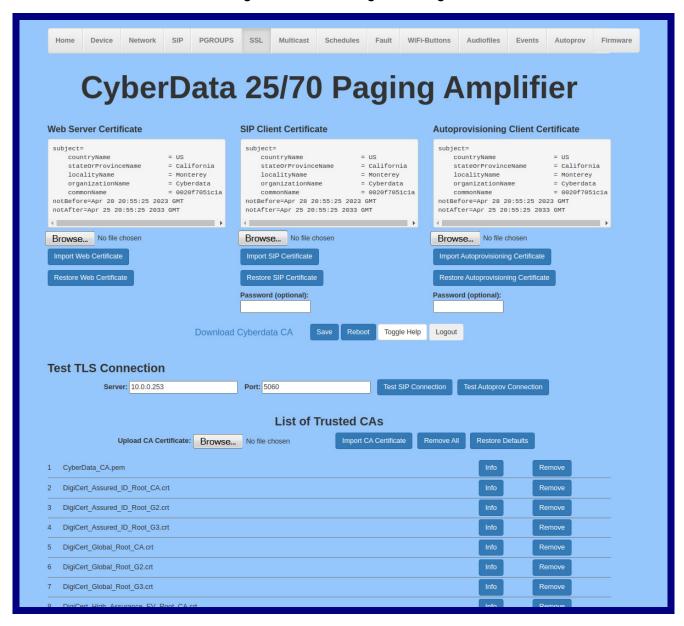


Figure 2-26. SSL Configuration Page

4 [igiCert_Assured_ID_Root_G3.crt	Info	Remove
5 [igiCert_Global_Root_CA.crt	Info	Remove
6 E	igiCert_Global_Root_G2.crt	Info	Remove
7 0	DigiCert_Global_Root_G3.crt	Info	Remove
8 0	pigiCert_High_Assurance_EV_Root_CA.crt	Info	Remove
9 [oigiCert_Trusted_Root_G4.crt	Info	Remove
10 0	GeoTrust_Global_CA.crt	Info	Remove
11 0	GeoTrust_Primary_Certification_Authority.crt	Info	Remove
12 0	GeoTrust_Primary_Certification_AuthorityG2.crt	Info	Remove
13 0	GeoTrust_Primary_Certification_AuthorityG3.crt	Info	Remove
14 0	GeoTrust_Universal_CA.crt	Info	Remove
15 0	GeoTrust_Universal_CA_2.crt	Info	Remove
16 0	o_Daddy_Class_2_CA.pem	Info	Remove
17 0	o_Daddy_Root_Certificate_AuthorityG2.pem	Info	Remove
18 V	eriSign_Class_3_Public_Primary_Certification_AuthorityG4.crt	Info	Remove
19 V	eriSign_Class_3_Public_Primary_Certification_AuthorityG5.crt	Info	Remove
20 V	eriSign_Universal_Root_Certification_Authority.crt	Info	Remove
21 V	erisign_Class_1_Public_Primary_Certification_Authority.crt	Info	Remove
22 V	erisign_Class_1_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
23 V	erisign_Class_2_Public_Primary_Certification_AuthorityG2.crt	Info	Remove
24 V	erisign_Class_2_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
25 V	erisign_Class_3_Public_Primary_Certification_Authority.crt	Info	Remove
26 V	erisign_Class_3_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
27 tl	nawte_Primary_Root_CA.crt	Info	Remove
28 tl	nawte_Primary_Root_CAG2.crt	Info	Remove
29 tl	nawte_Primary_Root_CAG3.crt	Info	Remove

2. On the SSL page, enter values for the parameters indicated in Table 2-13.

Note The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-13. SSL Configuration Parameters

Web Page Item	Description
Web Server Certificate	Certificate used by the web server.
Browse	Click Browse to select a certificate to import.
Import Web Certificate	After selecting a certificate, click Import Web Certificate to import it as the certificate used by this device's web server.
Restore Web Certificate	Restore the device's default web server certificate. This will remove the user-uploaded Web Server Certificate. (Server CAs and Trusted CAs are unaffected).
SIP Client Certificate	When doing mutual authentication this device will present a client certificate with these parameters.
Browse	Click Browse to select a certificate to import.
Import SIP Certificate	After selecting a certificate, click Import SIP Certificate to import it as the certificate used by the device during SIP transactions.
Restore SIP Certificate	Restore the device's default sip client certificate. This will remove any user-uploaded sip client certificates (Server CAs and Trusted CAs are unaffected).
Optional Password	Enter the optional password for the SIP certificate's private key.
	Note : When using a password, it must be entered and saved before importing the certificate.
Autoprovisioning Client Certificate	When doing mutual authentication this device will present a client certificate with these parameters.
Browse	Click Browse to select a certificate to import.
Import Autoprovisioning Certificate	After selecting a certificate, click Import Autoprovisioning Certificate to import it as this device's certificate. This certificate will be used when requesting files during autoprovisioning.
Restore Autoprovisioning Certificate	Restore the device's default autoprovisioning certificate. This will remove any user-uploaded autoprovisioning certificates. (Server CAs and Trusted CAs are unaffected).
Optional Password ?	Enter the optional password for the Autoprovisioning certificate's private key.
	Note : When using a password, it must be entered and saved before importing the certificate.
Cyberdata CA ?	Right click and Save Link As to get the Cyberdata CA used to sign this client certificate.

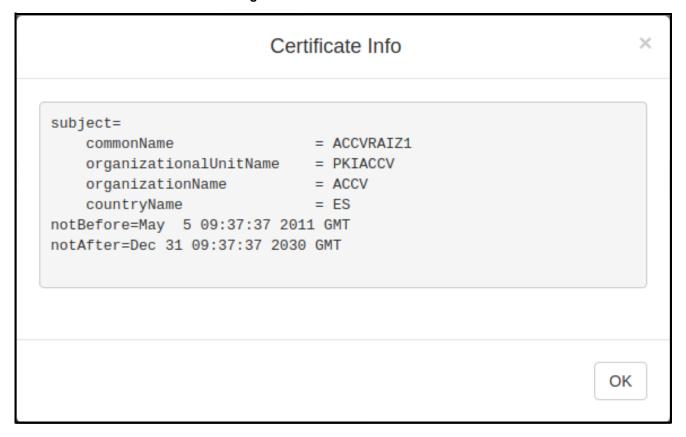
Web Page Item	Description
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.
Test TLS Connection	
Server ?	The ssl test server address as a fully qualified domain name or in IPv4 dotted decimal notation.
Port ?	The supported range is 0-65536. SIP connections over TLS to port 5060 are modified to connect to port 5061. This test button will do the same.
Test SIP Connection	Use this button to test a TLS connection to a remote server using the sip client key and password. This will attempt to make a socket connection to the configured test server and port and report the success or failure. This can be used to debug TLS connection issues separate from SIP registration issues.
Test Autoprov Connection	Use this button to test a TLS connection to a remote server using the autoprovisioning client key and password. This will attempt to make a socket connection to the configured test server and port and report the success or failure. This can be used to debug TLS connection issues with secure autoprovisioning.
List of Trusted CAs	
Browse	Use this button to select a configuration file to import.
Import CA Certificate	Click Browse to select a CA certificate to import. After selecting a server certificate authority (CA), click Import CA Certificate to import it to the list of trusted CAs. CAs are used to validate the certificate presented by the server when establishing a TLS connection.
Restore Defaults	Restore Defaults will restore the default list of registered CAs and Remove All will remove all registered CAs.
Remove All	Restore Defaults will restore the default list of registered CAs and Remove All will remove all registered CAs.
Info	Provides details of the certificate. After clicking on this button, the Certificate Info Window appears. See Section 2.8.10.1, "Certificate Info Window".

Web Page Item	Description
Remove	Removes this certificate from the list of trusted certificates. After clicking on this button, the Remove Server Certificate Window appears. See Section 2.8.10.2, "Remove Server Certificate Window".

2.8.10.1 Certificate Info Window

The **Certificate Info Window** provides details of the certificate. This window appears after clicking on the **Info** button. See Figure 2-27.

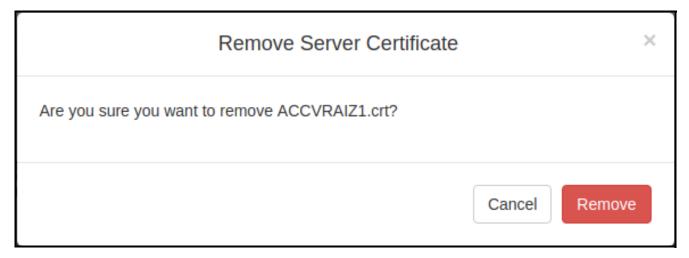
Figure 2-27. Certificate Info Window



2.8.10.2 Remove Server Certificate Window

The Remove Server Certificate Window will ask if the user wants to remove a certificate from the list of trusted certificates. This window appears after clicking on the Remove button. See Figure 2-28.

Figure 2-28. Remove Server Certificate Window



2.8.11 Configure the Multicast Parameters

Multicast groups use multicasting to create public address paging zones. Multicasting is based on the concept of a group. Multicast addresses specify an arbitrary group of IP hosts that have joined the group and want to receive traffic sent to the group. Group members send IGMP messages to their local multicast routers, allowing the group traffic traversal from the source.

The **Multicast Configuration** page allows the device to join up to 10 paging zones for receiving ulaw/alaw, G722, or G729 RTP audio streams. A paging zone can consist of one or many CyberData multicast group-enabled products. There is no limit to how many devices can be in a given paging zone. Each multicast group is defined by a multicast address and port number. Each multicast group is assigned a priority, allowing simultaneously arriving pages to be serviced based on importance. Multicast groups are compatible with IGMP through version three. The device supports simultaneous SIP and Multicast.

1. Click on the Multicast button to open the Multicast page. See Figure 2-29.

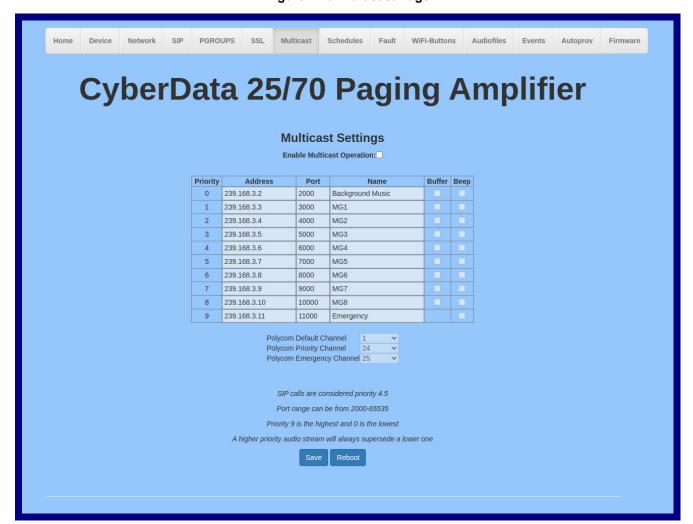


Figure 2-29. Multicast Page

2. On the Multicast page, enter values for the parameters indicated in Table 2-14.

Note The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-14. Multicast Configuration Parameters

Web Page Item	Description
Enable Multicast Operation	Enables or disables multicast operation.
Priority	Indicates the priority for the multicast group. Priority 9 is the highest (emergency streams). 0 is the lowest (background music). SIP calls are considered priority 4.5 . See Section 2.8.11.1, "Assigning Priority" for more details.
Address	Enter the multicast IP Address for this multicast group (15 character limit).
Port	Enter the port number for this multicast group (5 character limit [range can be from 2000 to 65535]).
	Note : The multicast ports have to be even values. The webpage will enforce this restriction.
Name	Assign a descriptive name for this multicast group (25 character limit).
Buffer	Device will buffer up to four minutes of audio and then play back the recording after the multicast stream finishes or after the buffer is full.
Веер	When selected, the device will play a beep before multicast audio is sent.
Polycom Default Channel	When a default Polycom channel/group number is selected, the device will subscribe to the default channel for one-way group pages. Group Numbers 1-25 are supported. Or, select Disabled to disable this channel.
Polycom Priority Channel	When a priority Polycom channel/group number is selected, the device will subscribe to the priority channel for one-way group pages. Group Numbers 1-25 are supported. Or, select Disabled to disable this channel.
Polycom Emergency Channel	When an emergency Polycom channel/group number is selected, the device will subscribe to the default channel for one-way group pages. Group Numbers 1-25 are supported. Or, select Disabled to disable this channel.
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.

You must click on the Save button for the changes to take effect. Note

2.8.11.1 Assigning Priority

When playing multicast streams, audio on different streams will preempt each other according to their priority in the list. An audio stream with a higher priority will interrupt a stream with a lower priority.

If both SIP and Multicast is enabled, SIP audio streams are considered priority 4.5. SIP audio will interrupt multicast streams with priority 0 through 4 and will be interrupted by multicast streams with priority 5 through 9.

Note SIP calls, multicast streams, ring tones, ringback tones, and nightring tones are all prioritized.

Ringtones and **Nightringtones** Ringtones all play at the same priority level. This means that it is possible to have a nightring tone and a normal ringtone playing at the same time.

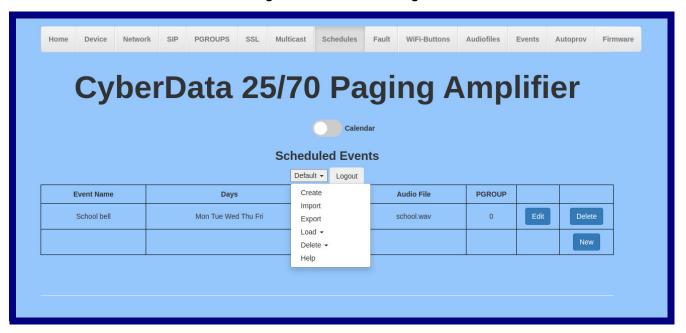
2.8.11.2 Polycom Paging

To use Polycom Group Paging, configure a multicast group with the IP address and port number of the Polycom phone. The default is 224.0.1.116, port 5001, but can be configured through the phone. Polycom defaults to channels 1, 24, and 25, but can also be configured. The payload should be 20 ms and the codec G711.Mu.

2.8.12 Configure the Schedules Parameters

1. Click on the **Schedules** button to open the **Schedules** page. See Figure 2-30.

Figure 2-30. Schedules Page



2. On the **Schedules** page, enter values for the parameters indicated in Table 2-15.

Table 2-15. Schedules Configuration Parameters

Web Page Item	Description
Calendar	Switches to Calendar mode, with augmented scheduling functionality. Note that scheduled events do not populate across the two modes. See Section 2.8.12.1, "Configure the Calendar"
Scheduled Events	
	Default: Default, unnamed schedule available for editing
Default ▼	Create: Used to create a new schedule. Upon saving, the new
Create	schedule will be loaded as the active schedule. Duplicate schedule names are not allowed.
	Import: Import a schedule previously exported. Imported
Import	schedules will need to be loaded to be set as the active schedule. Duplicate schedule names are not allowed.
Export	Export: Export currently active schedule.
Load →	Load : Load imported or created schedules. This makes the loaded schedule active.
Delete ▼	Delete : Delete imported or created schedules. Deleting currently active schedule is not allowed.
Help	Help: Launches the help text

Table 2-15. Schedules Configuration Parameters (continued)

Web Page Item	Description
Logout	Logs out the current user and returns to the user log in page
Event Name	Shows the name of the scheduled event.
Days	Shows the days of the scheduled event.
Time	Shows the time of the scheduled event.
Audio File	Shows the audio file that is associated with the scheduled event.
PGROUP	Select the paging group from the drop down menu.
Edit	Opens the Configure Scheduled Event window for the corresponding event. See Figure 2-31, "Configure Scheduled Event Window" and Table 2-16, "Configure Scheduled Event Window Parameters"
Delete	Removes the event from the schedule.
New	Opens a new Configure Scheduled Event window.

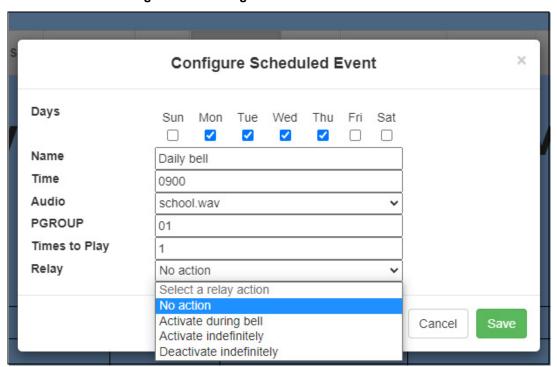


Figure 2-31. Configure Scheduled Event Window

The parameters for the **Configure Scheduled Event** window are shown in Table 2-16.

Table 2-16. Configure Scheduled Event Window Parameters

Web Page Item	Description
Days	Use the box beneath the day to select. Multiple days may be selected.
Name	Enter the name of the event, using a maximum of 16 characters.
Time	Enter time of the event, in the form HH:MM, using the 24 hour clock.
Audio	Choose the audio file to be played from the drop down list, which displays files previously uploaded in the Bells section of the Audio Files web page.
PGROUP	Select the paging group from the drop down menu.
Times to Play	Number of times the audio file will play
Relay	A relay can be activated or deactivated at the time of a scheduled bell.
	The relay options are:
	Activate during bell — The relay will be active while the bell is playing
	Activate indefinitely — The relay will be activated indefinitely.
	Deactivate indefinitely — The relay will be deactivated indefinitely.
Cancel	Click the Cancel button to close the Configure Scheduled Event window.
Save	Click the Save button to save your configuration settings.

Note www.cyberdata.net/bell-download/ has a set of sounds that can be downloaded.

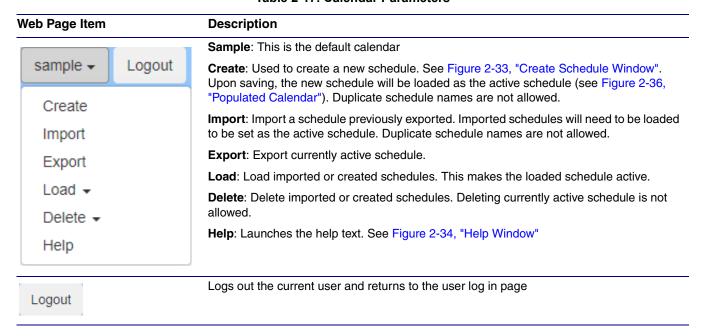
2.8.12.1 Configure the Calendar

1. Enable the Calendar setting to open its configuration options. See Figure 2-32.

Figure 2-32. Calendar Setting

The parameters for the Calendar are shown in Table 2-17.

Table 2-17. Calendar Parameters



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Table 2-17. Calendar Parameters

Web Page Item	Description
New Event	Launches the dialog box to create an event. See Section 2.8.12.2, "New Event Window"
<<<	Navigates back 1 year or to the first month of the calendar
<	Navigates to the previous month
Current Date	Navigates to the current date.
>	Navigates to the next month
>>>	Navigates to one year in the future
View Past	Displays the schedule for past days

Figure 2-33. Create Schedule Window



The parameters for the **Create Schedule** window are shown in Table 2-17.

Table 2-18. Create Schedule Parameters

Window Item	Description	
Schedule Name	Name of the schedule.	
Schedule Color	Color of the schedule.	
Save	Click the Save button to save your configuration settings.	
Cancel	Click the Cancel button to close the Create Schedule window.	

Help

Create: Used to create a new schedule. Upon saving, the new schedule will be loaded as the active schedule. Duplicate schedule names are not allowed.

Import: Import a schedule previously exported. Imported schedules will need to be loaded to be set as the active schedule. Duplicate schedule names are not allowed.

Export: Export currently active schedule.

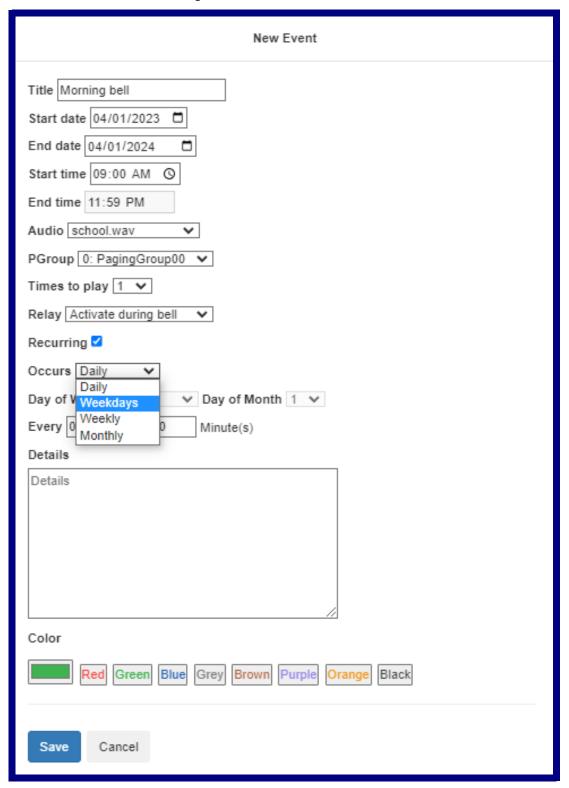
Load: Load imported or created schedules. This makes the loaded schedule active.

Delete: Delete imported or created schedules. Deleting currently active schedule is not allowed.

2.8.12.2 New Event Window

1. Click on the New Event button (Figure 2-32) to open the New Event window. See Figure 2-35.

Figure 2-35. New Event Window



The parameters for the **New Event** window are shown in Table 2-17.

Table 2-19. New Event Window Parameters

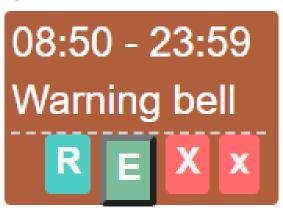
Window Item	Description
Title	The name of the event.
Start date	The start date of the event defaults to the current date; selecting the calendar allows the user to select a different date.
End Date	The end date defaults to the current date; selecting the calendar allows the user to select a different date.
Start time	Type a date or select the clock icon to indicate the start time.
End time	End time becomes enabled for a recurring event. Type a date or select the clock icon to indicate the start time.
Audio	Choose the audio file to be played from the drop down list, which displays files previously uploaded in the Bells section of the Audio Files web page.
PGROUP	Select the paging group from the drop down menu.
Times to Play	Number of times the audio file will play
Relay	A relay can be activated or deactivated at the time of a scheduled bell.
	The relay options are:
	Activate during bell — The relay will be active while the bell is playing
	Activate indefinitely — The relay will be activated indefinitely.
	Deactivate indefinitely — The relay will be deactivated indefinitely.
Recurring	Select this box to schedule a recurring event. Events may occur daily, only weekdays, weekly, monthly, or repeated after minutes or hours have elapsed.
Day of Week	From the drop down list, select the day of the week for a weekly recurring event.
Day of month	From the drop down list, select the day of the month for a recurring monthly event.
Every Hour(s)/Minute(s)	Type the number of hours or minutes to specify the interval of the recurring bell.
Details	This field is optional, and allows the user to provide details.
Color	Select the background color of the display of the bell.
Save	Click the Save button to save your configuration settings.
Cancel	Click the Cancel button to close the New Event window.

On a populated calendar (Figure 2-36), hovering over an individual bell reveals the bell menu drop down menu. (Figure 2-37).

| CyberData 25/70 Paging Amplifier | Creek Autoprov | Premium | Cyber | CyberData 25/70 Paging Amplifier | Cyber | Cyb

Figure 2-36. Populated Calendar

Figure 2-37. Hovering over an individual bell reveals the bell menu drop down menu



The parameters for the bell menu drop down menu are shown in Table 2-20.

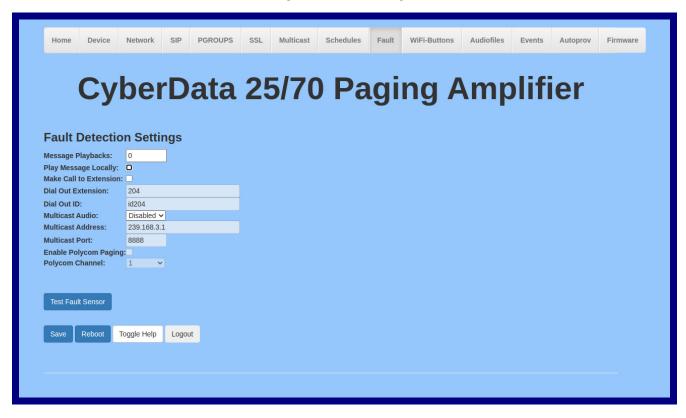
Table 2-20. Bell Menu Drop Down Menu Parameters

Menu Item	Description	
R	Indicates a recurring event	
E	Launches the Edit Event window	
X	Deletes all occurrences of this event.	
х	Deletes the current occurrence of this event.	

2.8.13 Configure the Fault Detection Parameters

1. Click on the Fault button to open the Fault page. See Figure 2-38.

Figure 2-38. Fault Page



2. On the Fault Detection page, enter values for the parameters indicated in Table 2-21.

Note The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

Table 2-21. Fault Detection Configuration Parameters

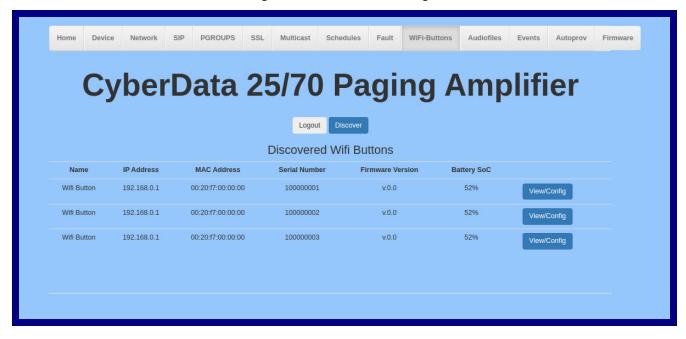
Web Page Item	Description
Triggered Settings	
Message Playbacks ?	The number of times to repeat the audio message through the local speaker or to the remote endpoint. A value of 0 will repeat forever. Enter a value from 0-65536.
Play Message Locally ?	When selected, the device will play the user defined "sensor triggered" audio file when the fault detection is triggered.
Make Call to Extension ?	When selected, the device will call an extension when fault detection is triggered. Use the Dial Out Extension field to specify the extension the device will call.
Dial Out Extension ?	Specify the extension the device will call when fault detection is triggered. Enter up to 64 alphanumeric characters.
Dial Out ID ?	An additional Caller identification string added to outbound calls. Enter up to 64 alphanumeric characters.
Multicast Audio ?	When enabled, the device will multicast audio to the specified multicast address and port while the fault sensor is activated.
Multicast Address ?	The multicast address used for fault detection audio.
Multicast Port ?	The multicast port used for fault detection audio.
Enable Polycom Paging ?	Enabling Polycom Paging will result in a standard RTP multicast being sent to the specified address and port and a Polycom Group Paging multicast being sent to the specified address and port+1.
Polycom Channel ?	Destination channel for Polycom Group Paging multicast.
Test Fault Sensor	Click on the Test Fault Sensor button to test the fault detection feature.
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (2) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.
Logout	Logs out the current user and returns to the user log in page

2.8.14 Configure the WiFi-Buttons Parameters

- 1. Click on the WiFi-Buttons button to open the WiFi-Buttons page. See Figure 2-39.
- Before WiFi Buttons become available to be associated with the device, WiFi network settings
 must be configured using the WiFi Utility. Please see the 011527 WiFi Alert Button Operations
 Guide at the Downloads tab at the following webpage:

https://www.cyberdata.net/products/011527/

Figure 2-39. WiFi-Buttons Page



2. On the WiFi-Buttons page, enter values for the parameters indicated in Table 2-22.

The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-22. WiFi-Buttons Configuration Parameters

Web Page Item	Description
Discovered WiFi Buttons	
Logout	Logs out the current user and returns to the user log in page
Discover	Discovers WiFI Buttons on the network
Name ?	Button's name (recommended to have a unique name for each)
IP Address?	Button's IP Address
MAC Address: ?	MAC address of the button
Serial Number 🛜	Serial number of the button
Firmware Version ?	Firmware version being used by the button
Battery SoC ?	Battery State of Charge (SoC) of the button
View/Config	Opens the Buttons Settings Page . See Figure 2-40 and Figure 2-41.
Disassociate	Severs the association with the SIP Paging 25V/70V Amplifier, erasing fields pertaining the button press and the heartbeat timeout.

Figure 2-40. Button Settings Page

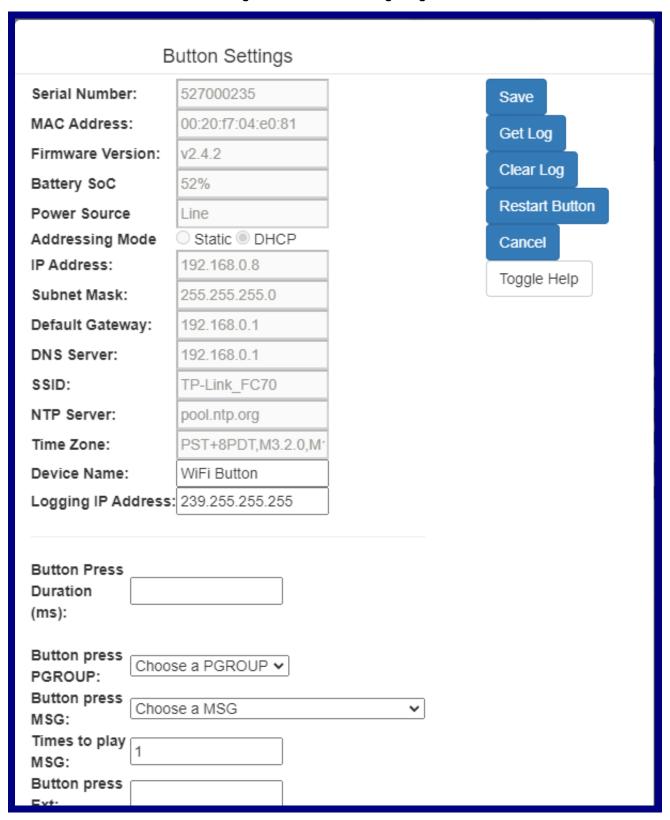


Figure 2-41. Button Settings Page

Heartbeat timeout (seconds): Heartbeat timeout PGROUP: Heartbeat timeout MSG: Times to play	Choose a PGROUP ▼ Choose a MSG ▼
MSG: Heartbeat timeout Ext:	
Buzzer Volume (Line 0-10): Buzzer Pattern (Line): Buzzer Volume (Batt 0-10): Buzzer Pattern (Batt):	
LED Brightness (Line 0-10): LED Pattern (Line): LED Brightness (Batt 0-10): LED Pattern (Batt):	

Table 2-23. Button Settings Page Parameters

Web Page Item	Description
Discovered WiFi Buttons	
Serial Number ?	Serial number of the button
MAC Address: ?	MAC address of the button
Firmware Version ?	Firmware version being used by the button
Battery SoC ?	Battery State of Charge (SoC) of the button
Power Source ?	Power source of the button
Addressing Mode ?	Indicates if the button is using DHCP or Static
IP Address ?	Button's IP Address
Subnet Mask 🛜	Subnet mask of the network
SSID ?	Service Set Identifier (SSID) of the network
NTP Server ?	Network Time Protocol (NTP) Server of the network
Time Zone ?	Time Zone
Device Name ?	Button's name (recommended to have a unique name for each)
Logging IP Address ?	Logging IP Address
ButtonPress Duration (ms) ?	The time in milliseconds the button must be pressed before an evenr is registered
Button press PGROUP ?	Paging Group to multicast an audio message on a button press
Button press MSG ?	Audio message to play on button press. To upload a message, please go the Stored Messages section of the Audiofiles page.
Times to play MSG 🔞	Times to play audio message on button press
Button press extension [7]	SIP extension to dial on button press
Heartbeat timeout ?	Buttons send a heartbeat message every 60 seconds. This is the timeout in seconds after which a failure to receive a heartbeat message will raise an event.
Heartbeat timeout PGROUP ?	Paging Group used to multicast an audio message on a heartbeat timeout
Heartbeat timeout MSG ?	Audio file to play on heartbeat timeout. TO upload a message, please go to the Stored Messages section of the Audiofiles page.
Times to play MSG ?	Times to play the audio message on heartbeat timeout
Heartbeat timeout extension ?	SIP extension to dial on heartbeat timeout.
Buzzer Volume ?	Set the buzzer power volume when the device is plugged in, where 0 is min power volume and 10 is the max power volume.
Buzzer Pattern ?	Set the buzzer pattern used when the device is plugged in.
Buzzer Volume ?	Set the buzzer power volume when the device is running on battery, where 0 is min power volume and 10 is the max power volume.
Buzzer Pattern ?	Set the buzzer pattern used when the device is running on battery.
LED Brightness ?	Set the LED brightness when the device is plugged in.
LED Pattern ?	Set the LED pattern used when the device is plugged in.
LED Brightness ?	Set the LED brightness when the device is running on battery.
Battery Power Pattern ?	Set the LED pattern used when the device is running on battery.

Table 2-23. Button Settings Page Parameters (continued)

Web Page Item	Description
Save	Click the Save button to save your configuration settings.
Get Log	Downloads the WiFi Button's log
Clear Log	Erases the Wifi Button's Log from the log partition
Restart Button	Restarts the WiFi Button
Cancel	Closes the Button Settings window without saving changes
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.

2.8.15 Configure the Audio Parameters

Page Tone:

Rebooting:

Click on the Audiofiles button to open the Audiofiles page. See Figure 2-42. The Audiofiles page is used to add custom audio to the board. User uploaded audio will take precedence over the audio files shipped with the device.

PGROUPS Multicast Schedules Fault WiFi-Buttons Audiofiles Autoprov Firmware CyberData 25/70 Paging Amplifier Logout Available Space: 1381MB **Audio Files** Browse... No file chosen Currently set to: default Browse... No file chosen 1: Currently set to: default Browse... No file chosen Delete Save 2: Currently set to: default Currently set to: default No file chosen Browse... 4: Currently set to: Browse... 5: Currently set to: default No file chosen 6: default Browse... No file chosen Delete Currently set to: 7: Currently set to: default No file chosen Browse... 8: Currently set to: default Browse... No file chosen Save Delete 9: Currently set to: default No file chosen Browse... Audio Test: Browse... No file chosen Currently set to: default Confused: Currently set to: No file chosen Browse... No file chosen Dot: Currently set to: Browse... Browse... No file chosen Enter Zone: Currently set to: default Browse... No file chosen Night Ring default Delete Save Currently set to:

Browse... No file chosen

Browse... No file chosen

Figure 2-42. Audiofiles Page

Currently set to:

Currently set to:

default

Figure 2-43. Audiofiles Page

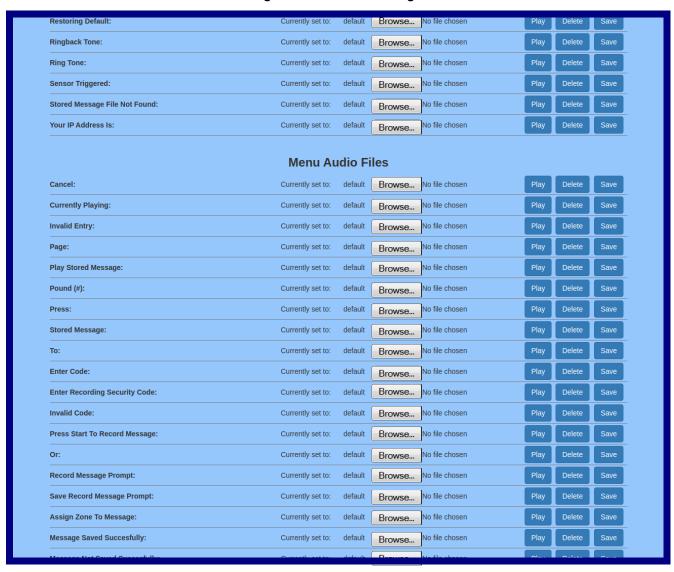


Figure 2-44. Audiofiles Page



On the Audiofiles page, enter values for the parameters indicated in Table 2-24.

Each entry on the Audiofiles page replaces one of the stock audio files on the board. When the input box displays the word default, the SIP Paging 25V/70V Amplifier is using the stock audio file. If that file is replaced with a user file, it will display the uploaded filename.

Note The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-24. Audiofiles Configuration Parameters

Web Page Item	Description
Audio Files	
Logout	Logs out the current user and returns to the user log in page
0-9	The name of the audio configuration option is the same as the spoken audio that plays on the board (24 character limit).
	'0' corresponds to the spoken word "zero."
	'1' corresponds to the spoken word "one."
	'2' corresponds to the spoken word "two."
	'3' corresponds to the spoken word "three."
	'4' corresponds to the spoken word "four."
	'5' corresponds to the spoken word "five."
	'6' corresponds to the spoken word "six."
	'7' corresponds to the spoken word "seven."
	'8' corresponds to the spoken word "eight."
	'9' corresponds to the spoken word "nine."
Dot	Corresponds to the spoken word "dot."
Audio Test	Corresponds to the message "This is the CyberData IP speaker test message"
Page Tone	Corresponds to a simple tone that is unused by default.
Your IP Address is	Corresponds to the message "Your IP address is"
Rebooting	Corresponds to the spoken word "Rebooting."
Restoring Default	Corresponds to the message "Restoring default."
Ringback Tone	Specifies the Ringback Tone.
Ring Tone	Specifies the Ring Tone.
Night Ring	Specifies the ringtone for nightring. By default this parameter uses the same audio file that is selected for the Ring Tone parameter.
Sensor Triggered	Corresponds to the message "Sensor Triggered."
Stored Message File Not Found	Corresponds to the message "Stored Message File Not Found."
Enter Zone	Corresponds to the words "Enter Zone" used in the audio menu played to the caller.
Confused	Corresponds to the message "Confused."
Browse	The Browse button will allow you to navigate to and select an audio file.

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Table 2-24. Audiofiles Configuration Parameters (continued)

Web Page Item	Description
Play	The Play button will play that audio file.
Delete	The Delete button will delete any user uploaded audio and restore the stock audio file.
Save	The Save button will download a new user audio file to the board once you've selected the file by using the Browse button. The Save button will delete any pre-existing user-uploaded audio files.
Menu Audio Files	Menu Audio Files are user-uploadable messages that create the audio menu played to the caller.
Cancel	Corresponds to the word "Cancel" used in the audio menu played to the caller.
Currently Playing	Corresponds to the words "Currently Playing" used in the audio menu played to the caller.
Invalid Entry	Corresponds to the words "Invalid Entry" used in the audio menu played to the caller.
Page	Corresponds to the word "Page" used in the audio menu played to the caller.
Play Stored Message	Corresponds to the words "Play Stored Message" used in the audio menu played to the caller.
Pound (#)	Corresponds to whatever word or phrase the user wishes to call the pound key in the audio menu played to the caller.
Press	Corresponds to the word "Press" used in the audio menu played to the caller.
Stored Message	Corresponds to the words "Stored Message" used in the audio menu played to the caller.
То	Corresponds to the word "To" used in the audio menu played to the caller.
Enter Code	Corresponds to the message "Enter Code."
Invalid Code	Corresponds to the message "Invalid Code."
Press Start to Record Message	Corresponds to the message "Press Start to Record Message."
Or	Corresponds to the message "Or."
Record Message Prompt	Corresponds to the message "Record Message Prompt."
Save Record Message Prompt	Corresponds to the message "Save Record Message Prompt."
Assign Zone to Message	Corresponds to the message "Assign Zone to Message."
Message Saved Successfully	Corresponds to the message "Message Saved Successfully."
Message Not Saved Successfully	Corresponds to the message "Message Not Saved Successfully."
You Recorded	Corresponds to the message "You Recorded."
Browse	The Browse button will allow you to navigate to and select an audio file.
Play	The Play button will play that audio file.

Table 2-24. Audiofiles Configuration Parameters (continued)

Web Page Item	Description
Delete	The Delete button will delete any user uploaded audio and restore the stock audio file.
Save	The Save button will download a new user audio file to the board once you've selected the file by using the Browse button. The Save button will delete any pre-existing user-uploaded audio files.
Stored Messages	Upload files here to be used as stored messages.
	Note: The user may upload a maximum of 25 files.
Browse	The Browse button will allow you to navigate to and select an audio file.
Upload Message	The Upload Message button will upload the stored message.
Delete All Messages	The Delete All Messages button will delete all of the stored messages.
Bells	Upload files here to be used in scheduled events.
	Note: The user may upload a maximum of 25 files.
Browse	The Browse button will allow you to navigate to and select an audio file.
Upload Bell	The Upload Bell button will upload the bell.
Delete All Bells	The Delete All Bells button will delete all of the bells.

2.8.15.1 User-created Audio Files

User-created audio files must be saved in one of the following formats:

- RIFF (little-endian) data,
- WAVE audio, Microsoft PCM
- 16 bit, mono 8000 Hz

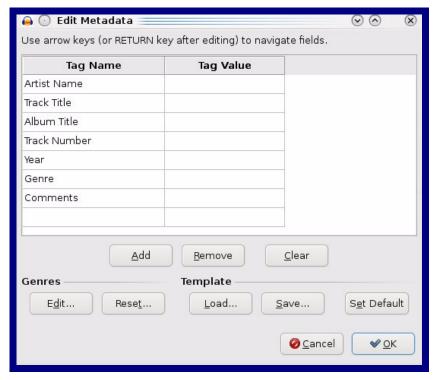
These audio format restrictions are enforced by the webpage.

You can use the free utility Audacity to convert audio files into this format. See Figure 2-45 through Figure 2-47.

Figure 2-45. Audacity 1



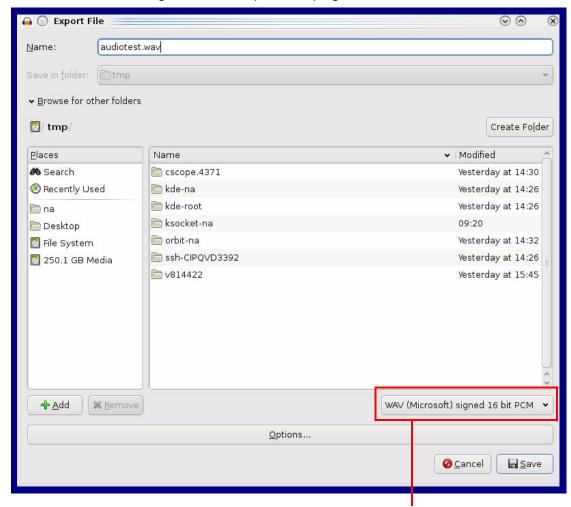
Figure 2-46. Audacity 2



When you export an audio file with Audacity, save the output as:

WAV (Microsoft) signed 16 bit PCM.

Figure 2-47. WAV (Microsoft) signed 16 bit PCM



WAV (Microsoft) signed 16 bit PCM

2.8.16 Configure the Event Parameters

Click on the **Events** button to open the **Events** page (Figure 2-48). The **Events** page specifies a remote server that can be used to receive HTTP POST events when actions take place on the board.

Figure 2-48. Events Page

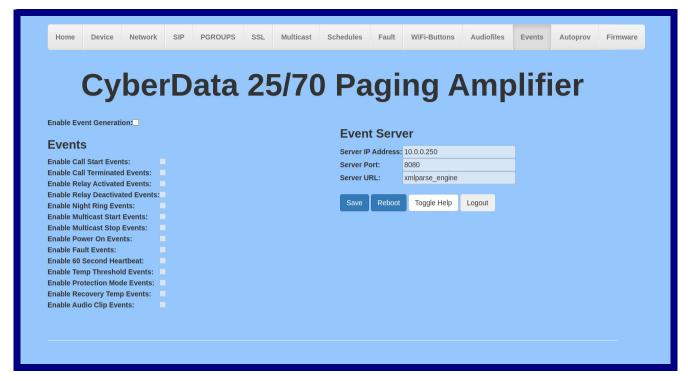


Table 2-25 shows the web page items on the **Events** page.

Note The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

Table 2-25. Events Configuration

Web Page Item	Description
Enable Event Generation ?	The device will send HTTP POST events to the specified remote server and port number whenever a certain action takes place. Select an event type below to generate an HTTP POST event. See Section 2.8.16.1, "Example Packets for Events" for sample packets.
Events	
Enable Call Start Events ?	When selected, the device will report the start of a SIP call.
Enable Call Terminated Events ?	When selected, the device will report the end of a SIP call.
Enable Relay Activated Events ?	When selected, the device will report relay activation.
Enable Relay Deactivated Events ?	When selected, the device will report relay deactivation.
Enable Night Ring Events ?	When selected, the device will report when it starts ringing upon an incoming SIP call to the Nightringer extension. As a reminder, the Nightringer extension always rings upon an incoming SIP call and it is not possible to alter this behavior.
Enable Multicast Start Events ?	When selected, the device will report when the device starts playing a multicast audio stream.
Enable Multicast Stop Events ?	When selected, the device will report when the device stops playing a multicast audio stream.
Enable Power On Events ?	When selected, the device will report when it boots.
Enable Fault Events ?	When selected, the device will report when the on-board fault detection is activated.
Enable 60 Second Heartbeat ?	When enabled, the device will report a Heartbeat event every 60 seconds. SIP registration is not required to generate Heartbeat events.
Enable Temp Threshold Events ?	When selected, the device will report a temperature threshold event.
Enable Protection Mode Events ?	When selected, the device will report a protection mode event.
Enable Recovery Temp Events ?	When selected, the device will report a recovery temperature event.
Enable Audio Clip Events ?	When selected, the device will report an event when a waveform that is playing is distorted, possibly indicating that the amplifier is overdriven.
Event Server	
Server IP Address ?	The IPv4 address of the event server in dotted decimal notation.
Server Port ?	Specify the event server port number. The supported range is 0-65536. Enter up to 5 digits.
Server URL ?	Generally, the destination URL is the name of the application that receives the events and the string in the HTTP POST command. It can be a script used to parse and process the HTTP POST events. Enter up to 127 characters.
Check All	Click on Check All to select all of the events on the page.
Uncheck All	Click on Uncheck All to de-select all of the events on the page.
Save	Click the Save button to save your configuration settings.

Table 2-25. Events Configuration (continued)

Web Page Item	Description
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.
Logout	Logs out the current user and returns to the user log in page

2.8.16.1 Example Packets for Events

The server and port are used to point to the listening server and the 'Remote Event Server URL' is the destination URL (typically the script running on the remote server that's used to parse and process the POST events).

Note The XML is URL-encoded before transmission so the following examples are not completely accurate.

Here are example packets for every event:

```
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 197
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>POWERON</event>
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 199
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>HEARTBEAT</event>
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 196
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>BUTTON</event>
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 201
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>CALL ACTIVE
</cyberdata>
POST xmlparse engine HTTP/1.1
```

```
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 205
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>CALL TERMINATED
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 197
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>RINGING
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 234
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>MULTICAST START
<index>8</index>
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 233
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>MULTICAST STOP</event>
<index>8</index>
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 234
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>RELAY ACTIVATED
</cyberdata>
POST xmlparse engine HTTP/1.1
```

```
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 234
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>RELAY DEACTIVATED</event>
</cyberdata>
POST xmlparse engine HTTP/1.1
Host: 10.0.3.79
User-Agent: CyberData/1.0.0
Content-Length: 234
Content-Type: application/x-www-form-urlencoded
<?xml version="1.0" encoding="ISO-8859-1"?>
<cyberdata NAME='CyberData VoIP Device' MAC='0020f70015b6'>
<event>NIGHTRINGING</event>
</cyberdata>
```

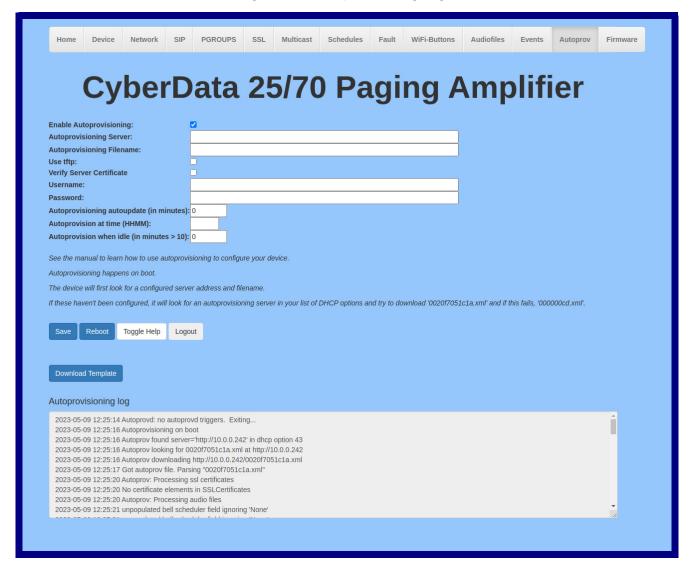
2.8.17 Configure the Autoprovisioning Parameters

Autoprovisioning can be used to automatically configure your device. The autoprovisioning file is an xml file with the device configuration. Values found in this file will override values stored in on-board memory.

Note By default, the device will try to set up its configuration with autoprovisioning.

1. Click the Autoprov menu button to open the Autoprovisioning page. See Figure 2-49.

Figure 2-49. Autoprovisioning Page



2. On the Autoprovisioning page, you may enter values for the parameters indicated in Table 2-26.

Note The question mark icon (?) in the following table shows which web page items will be defined after the Toggle Help button is pressed.

Table 2-26. Autoprovisioning Configuration Parameters

Web Page Item	Description		
Disable Autoprovisioning ?	Prevent the device from automatically trying to download a configuration file. See Section 2.8.17.1, "Autoprovisioning" for more information.		
Autoprovisioning Server ?	Enter the IPv4 address of the provisioning server in dotted decimal notation.		
Autoprovisioning Filename ?	The autoprovisioning filename is the configuration filename. The default autoprovisioning filename is in the format of <mac address="">.xml.</mac>		
	Supported filename extensions are ".txt", and ".xml." The current filename is denoted by an asterisk at the bottom of the Autoprovisioning Page . Enter up to 256 characters.		
	A file may have any name with an xml extension. If a file name is entered, the device will look for the specified file name, and only that file.		
Use tftp ?	The device will use TFTP (instead of http) to download autoprovisioning files.		
Username ?	The username used to authenticate with an autoprovisioning server. Leave this field blank to disable authentication.		
Password ?	The password used to authenticate with an autoprovisioning server. Leave this field blank to disable authentication.		
Autoprovisioning autoupdate (in minutes)	The reoccurring time (in minutes) the device will wait before checking for new autoprovisioning files. Enter up to 6 digits. A value of 0 will disable this option.		
	Note : To use the auto update options, make sure that the Enable NTP setting on the Device Page page is selected (see Table 2-8).		
Autoprovision at time (HHMMSS) ?	The time of day the device will check for a new autoprovisioning file. The time must be 6 characters in length and in HHMMSS format. An empty value will disable this option.		
	Note : To use the auto update options, make sure that the Enable NTP setting on the Device Page page is selected (see Table 2-8).		
Autoprovision when idle (in minutes > 10)	The idle time (in minutes greater than 10) after which the device will check for a new autoprovisioning file. Enter up to 6 digits. A value of 0 will disable this option.		
	Note : To use the auto update options, make sure that the Enable NTP setting on the Device Page page is selected (see Table 2-8).		
Save	Click the Save button to save your configuration settings.		
Reboot	Click on the Reboot button to reboot the system.		
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.		

Table 2-26. Autoprovisioning Configuration Parameters (continued)

Web Page Item	Description	
Download Template	Press the Download Template button to create an autoprovisioning file for the device. See Section 2.8.17.3, "Get Autoprovisioning Template Button"	
Autoprovisioning log	The autoprovisioning log provides information about the latest autoprovisioning attempt (i.e. dhcp options and server accessed and files parsed or not found).	

2.8.17.1 Autoprovisioning

On boot, the device will look for an autoprovisioning server configured on the Autoprovisioning Page or specified as a DHCP option. When it finds a server, it will try to download the following (in order of preference):

- 1. The file configured on the autoprovisioning page.
- 2. A file named according to it's mac address (for example: 0020f7350058.xml).
- 3. The file 00000cd.xml

The file can be hosted using a standard web server (like apache, IIS, or nginx), and the device can download over SSL. The file server can be an ipv4 address in dotted decimal notation or a fully qualified domain name.

By default, the device will get its autoprovisioning server from the DHCP options. See Section 2.8.17.2, "Sample dhcpd.conf" for an example of how to configure dhcpd to offer autoprovisioning server addresses. If multiple options are set, the device will attempt to download autoprovisioning files from every server.

The DHCP option determines the protocol used to download the autoprovisioning file. The device looks for DHCP options in the following order:

- 1. Option 43 a FQDN or an IP address to an http server
- 2. Option 72 an IP address to an http server
- 3. Option 150 an IP address to a tftp server
- 4. Option 66 an IP address to a tftp server or if the entry starts with 'http', a FQDN to a http server.

You can download an autoprovisioning template file from the Autoprovisioning Page using the Download Template button (see Table 2-26). This file contains every configuration option that can be set on the board.

Autoprovisioning files can contain the whole configuration or a subset of this file. The first autoprovisioning file can also contain links to other autoprovisioning files.

The <MiscSettings> section contains some examples of additional autoprovisioning files:

```
<MiscSettings>
       <DeviceName>CyberData VoIP Intercom
       <AutoprovFile>common.xml</AutoprovFile>-->
<!--
--!>
       <AutoprovFile>sip reg[macaddress].xml</AutoprovFile>-->
<!--
       <AutoprovFile>audio[macaddress]</AutoprovFile>-->
       <AutoprovFile>device[macaddress].xml</AutoprovFile>-->
   </MiscSettings>
```

After downloading the first autoprovisioning file, the device will step through up to twenty additional <AutoprovFile> entries and try to download these files from the same server.

When the device finds a filename with the string [macaddress], it will replace this string with the mac address.

As an example, the user has configured option 43 on their DHCP server to "http://example.com," and on their server, they have a file named 0020f7123456.xml (the same as the mac address of the device).

The file 0020f7123456.xml contains:

```
<?xml version="1.0" encoding="utf-8" ?>
<specific>
         <MiscSettings>
                <DeviceName>Newname/DeviceName>
                <AutoprovFile>common.xml</AutoprovFile>
                <AutoprovFile>sip_reg[macaddress].xml</AutoprovFile>
                <AutoprovFile>audio[macaddress]</AutoprovFile>
                <AutoprovFile>device.xml</AutoprovFile>
         </MiscSettings>
</specific>
```

- 1. The device will first set it's name to 'Newname'.
- 2. It will try to download http://example.com/common.xml.
- 3. It will try to download http://example.com/sip_reg0020f7123456.xml.
- 4. It will try to download http://example.com/audio0020f7123456.
- 5. It will try to download http://example.com/device.xml.

The device is reconfigured every time it downloads a new file so if two files configure the same option the last one will be the one that is saved.

It is possible to autoprovision autoprovisioning values (for example, to disable autoprovisioning or to configure a time to check for new files).

Checking for New **Autoprovisioning** Files after Boot

The device will always check for an autoprovisioning files on boot but it can be configured to also check after a periodic delay, when idle, or at a specified time. When one of these options is set, the device will download its autoprovisioning files again, and if it finds any differences from the files it downloaded on boot, it will force a reboot and reconfigure.

The **Autoprovisioning** Filename

The autoprovisioning filename can contain a file, a file path, or a directory.

Table 2-27. Autoprovisioning File Name

Autoprovisioning Filename	Autoprovisioning Server	File Downloaded
config.xml	10.0.1.3	10.0.1.3/config.xml
/path/to/config.xml	10.0.1.3	10.0.1.3/path/to/config.xml
subdirectory/path/	10.0.1.3	10.0.1.3/subdirectory/path/0020f7020002.xml

TFTP options may not support subdirectories. If a directory is set in the filename field, firmware and audio files will also be downloaded from this subdirectory.

If the filename ends with a forward slash "/," the device will treat it as a subdirectory.

For example:

The autoprovisioning server is set to "https://www.example.com"

The autoprovisioning filename is set to "cyberdata/"

On boot, the device will try to download:

https://www.example.com/cyberdata/0020f7123456.xml

...and if this fails:

https://www.example.com/cyberdata/00000cd.xml

Audio files and firmware files will also add "cyberdata" to the URL before downloading.

Autoprovisioning Firmware Updates

```
<FirmwareSettings>
  <FirmwareFile>505-uImage-ceilingspeaker/FirmwareFile>
  <FirmwareServer>10.0.1.3/FirmwareServer>
  <OutdoorIntercom30>firmware file v9.3.0</OutdoorIntercom30>
  <OutdoorIntercom31>firmware file v10.3.0</OutdoorIntercom31>
  <CallButton31>firmware_file_v10.3.0</CallButton31>
</FirmwareSettings>
```

In the <FirmwareSettings> section, the <FirmwareServer> element can be used to specify a different server for hosting firmware files. When this element is not available, the device will try to download the file from the autoprovisioning server.

The device will use the filename to determine when to autoprovision firmware updates. The default configuration is blank, so the first time you set a value in your autoprovisioning file, it may force a firmware update even if the firmware version has not changed.

The <FirmwareFile> name can contain path elements (i.e. /path/to/firmware/10.3.0-ulmage-[device_file_name]).

The device also supports product strings for downloading firmware. If the <FirmwareFile> option is not set, the device will look for its particular product string for a firmware filename. In this way, a generic autoprovisioning file can specify unique firmware for a range of products.

The list of valid product strings:

```
<ProductString>CallButton31/ProductString>
<ProductString>EmergencyIntercom31</ProductString>
<ProductString>EmergencyIntercom31SW</ProductString>
<ProductString>IndoorIntercom31</ProductString>
<ProductString>IndoorIntercom31SW</ProductString>
<ProductString>IndoorKeypad31</productString>
<ProductString>IndoorKeypad31SW</ProductString>
<ProductString>OfficeRinger31</productString>
<ProductString>OfficeRinger31SW</ProductString>
<ProductString>OutdoorIntercom31</ProductString>
<ProductString>OutdoorIntercom31SW</ProductString>
<ProductString>OutdoorKeypad31/ProductString>
<ProductString>OutdoorKeypad31SW</ProductString>
<ProductString>Strobe31</ProductString>
<ProductString>Strobe31SW</ProductString>
```

Autoprovisioning Example 1

Here's a simple example using four autoprovisioning files to configure two devices:

We boot up two devices with mac addresses 00:20:f7:02:00:01 and 00:20:f7:02:00:02 (Device1 and Device2).

The devices are set to use DHCP and that server provides an autoprovisioning server address with option 43. The address is "https://autoprovtest.server.net." The files on this server are as follows:

```
00000cd.xml
<MiscSettings>
<DeviceName>CyberData Autoprovisioned/DeviceName>
<AutoprovFile>sip common.xml</AutoprovFile>
<AutoprovFile>sip_[macaddress].xml</AutoprovFile>
</MiscSettings>
sip common.xml
<SIPSettings>
<SIPServer>10.0.0.253</SIPServer>
<RemoteSIPPort>5060</RemoteSIPPort>
</SIPSettings>
sip 0020f7020001.xml
<SIPSettings>
<SIPUserID>198</SIPUserID>
<SIPAuthPassword>ext198</SIPAuthPassword>
<DialoutExtension0>204</DialoutExtension0>
</SIPSettings>
sip 0020f7020002.xml
<SIPSettings>
<SIPUserID>500</SIPUserID>
<SIPAuthPassword>ext500</SIPAuthPassword>
<DialoutExtension0>555</DialoutExtension0>
```

On boot, Device1 tries to fetch the file 0020f7023614.xml from "https://autoprovtest.server.net". This file is not available, so device1 then tries to fetch the file 000000cd.xml. This file exists, and Device1 parses the three elements.

1. Device1 changes its device name to CyberData Autoprovisioned.

</SIPSettings>

- 2. Device1 finds an AutoprovFile element containing the filename sip common.xml. The device downloads sip common.xml from "https://autoprovtest.server.net," and imports this configuration, setting the sip server to 10.0.0.253 and the remote port to 5060.3.
- Device1 finds another AutoprovFile element containing the filename sip [macaddress].xml. The device replaces the [macaddress] with its own mac address value creating sip_0020f7020001.xml, downloads this file from "https://autoprovtest.server.net," and imports this configuration. This sets the user ID to 198, the password to ext198, and the dialout extension to 204. Device1 is now finished with autoprovisioning.

Device2 goes through the same steps by setting its device name to **CyberData Autoprovisioned**, its SIP server to **10.0.0.253**, and its port to **5060**. When Device2 "sees" **sip_[macaddress].xml**, Device2 replaces it with its own mac address and downloads **sip_0020f7020002.xml** from "https://autoprovtest.server.net." Device2 sets the SIP User ID to **500**, the password to **ext500**, and the dialout extension to **555**.

Autoprovisioning Example 2

Here is another example of setting up your autoprovisioning files:

We boot up two devices with mac addresses **00:20:f7:02:00:01** and **00:20:f7:02:00:02** (Device1 and Device2) and boot them on a network with a DHCP server configured with an autoprovisioning server at **10.0.1.3** on option **150**. Our TFTP server has three files:

```
0020f7020001.xml
<MiscSettings>
<AutoprovFile>common_settings.xml</AutoprovFile>
</MiscSettings>
<SIPSettings>
<SIPUserID>198</SIPUserID>
<SIPAuthPassword>ext198</SIPAuthPassword>
<DialoutExtension0>204</DialoutExtension0>
</SIPSettings>
0020f7020002.xml
<MiscSettings>
<AutoprovFile>common_settings.xml</AutoprovFile>
</MiscSettings>
<SIPSettings>
<SIPUserID>500</SIPUserID>
<SIPAuthPassword>ext500</SIPAuthPassword>
<DialoutExtension0>555</DialoutExtension0>
</SIPSettings>
common settings.xml
<MiscSettings>
<DeviceName>CyberData Autoprovisioned/DeviceName>
</MiscSettings>
<SIPSettings> <SIPServer>10.0.0.253</SIPServer>
<RemoteSIPPort>5060/RemoteSIPPort>
```

</SIPSettings>

- 1. On boot, Device1 downloads **0020f7020001.xml** from **10.0.1.3** and imports these values. The SIP User ID is **198**, the password is **ext198**, and the dialout extension is **204**.
- 2. Device1 then gets the filename **common_settings.xml** from the AutoprovFile element and downloads this file from the TFTP server at **10.0.1.3**. and imports these settings. The device name is set to **CyberData Autoprovisioned**, the SIP server is set to **10.0.0.253**, and the port is set to **5060**.

Device2 does the same except it downloads **0020f7020002.xml** on boot and imports these values instead. The Sip User ID is **500**, password is **ext500**, and dialout extension is **555**. Device2 then downloads the **common_settings.xml** file and imports those values. The device name is set to **CyberData Autoprovisioned**, the SIP server is set to **10.0.0.253**, and the port is set to **5060**.

XMI Files

XML files can contain <AutoprovFile> elements. If multiple DHCP options are specified, the device will try to download autoprovisioning files from each in turn. The device will only look for <AutoprovFile> elements in the first file downloaded from each server. You can specify up to 20 <AutoprovFile> elements in the first autoprovisioning file.

There are numerous ways to change an element of the configuration(xml) file. Using sip ext as an example, the extension can be changed:

Within the device-specific xml, i.e. [macaddress].xml, via the AutoprovFile element:<SIPSettings>/<SIPExt>

From the device specific xml, a pointer to a sip_common file

From the device specific xml, a pointer to the device specific sip [macaddress].xml

From the common file, a pointer to sip_common.xml

From the common file, a pointer to the device specific (sip_[macaddress].xml)

Autoprovisioned Audio Files

Audio files are stored in non-volatile memory and an autoprovisioned audio file will only have to be downloaded once for each device. Loading many audio files to the device from the web page could cause it to appear unresponsive. If this happens, wait until the transfer is complete and then refresh the page.

The device uses the file name to determine when to download a new audio file. This means that if you used autoprovisioning to upload a file and then changed the contents of this file at the TFTP server, the device will not recognize that the file has changed (because the file name is the same).

Since audio files are stored in non-volatile memory, if autoprovisioning is disabled after they have been loaded to the board, the audio file settings will not change. You can force a change to the audio files on the board by clicking Restore Default on the Audio Configuration page or by changing the autoprovisioning file with "default" set as the file name.

2.8.17.2 Sample dhcpd.conf

```
# Sample configuration file for ISC dhcpd for Debian
ddns-update-style none;
option domain-name "voiplab";
option domain-name-servers 10.0.0.252;
option option-150 code 150 = ip-address;
option ntp-servers north-america.pool.ntp.org;
option space VendorInfo;
option VendorInfo.text code 10 = { text };
authoritative;
log-facility local7;
subnet 10.0.0.0 netmask 255.0.0.0 {
   max-lease-time 3600;
   default-lease-time 3600;
   option routers
                                  10.0.0.1;
   option subnet-mask
                                  255.0.0.0;
                                  "voiplab";
   option domain-name
   option domain-name-servers
                                  10.0.0.252;
   option time-offset
                                                   # Pacific Standard Time
                                                                     # OPTION 72
     option www-server
                                   99.99.99.99;
                                      "10.0.1.52";
                                                                     # OPTION 66
     option tftp-server-name
     option tftp-server-name
                                     "http://test.cyberdata.net";
                                                                    # OPTION 66
                                                                     # OPTION 150
     option option-150
                                     10.0.0.252;
# These two lines are needed for option 43
     vendor-option-space VendorInfo;
                                                                     # OPTION 43
     option VendorInfo.text "http://test.cyberdata.net";
                                                                    # OPTION 43
    range 10.10.0.1 10.10.2.1; }
```

2.8.17.3 Get Autoprovisioning Template Button

The Get Autoprovisioning Template button allows the user to generate, download, edit, and then store an autoprovisioning template on the server that serves the autoprovisioning files for devices.

To generate an autoprovisioning template directly from the device, complete the following steps:

- 1. On the **Autoprovisioning** page, click on the **Get Autoprovisioning Template** button.
- 2. You will see a window prompting you to save a configuration file (.xml) to a location on your computer (Figure 2-50). The configuration file is the basis for the default configuration settings for your unit).
- 3. Choose a location to save the configuration file and click on **OK**. See Figure 2-50.

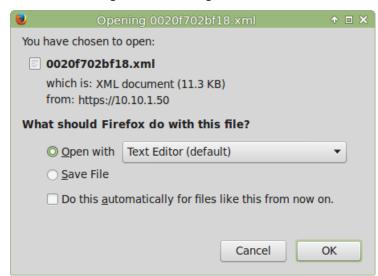


Figure 2-50. Configuration File

- 4. At this point, you can open and edit the autoprovisioning template to change the configuration settings in the template for the unit.
- 5. You can then upload the autoprovisioning file to a TFTP or HTTP server where the file can be loaded onto other devices.

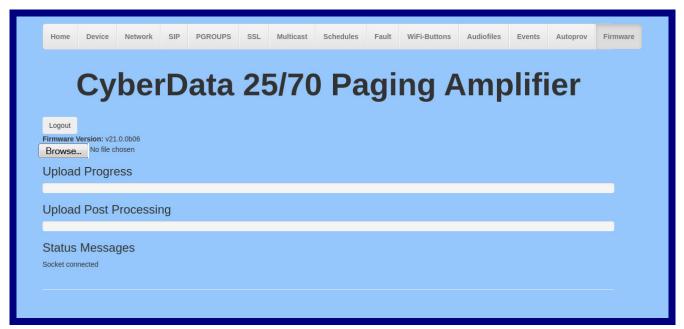
2.9 Upgrade the Firmware

Note CyberData strongly recommends that you do not upgrade the firmware when the device is likely to be in use.

To upgrade the firmware of your device:

- 1. Download the latest firmware file from the **Downloads** tab at the following webpage: https://www.cyberdata.net/pages/011579
- 2. Unzip the firmware version file. This file may contain the following:
- Firmware file
- Release notes
- Autoprovisioning template
- 3. Log in to the **Home** page as instructed in 2.8.4 "Log in to the Configuration GUI".
- 4. Click on the Firmware menu button to open the Firmware page (Figure 2-51).

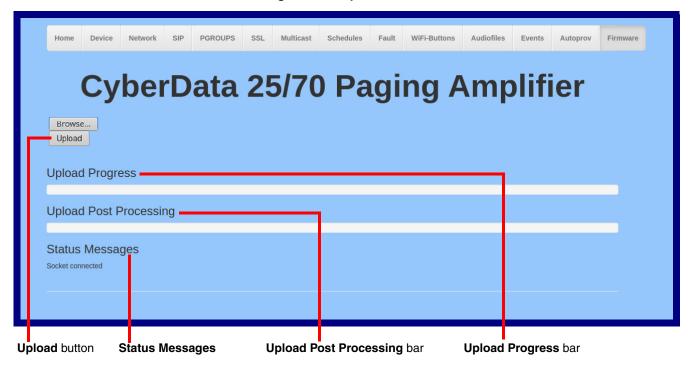
Figure 2-51. Firmware Page



5. Click on the **Browse** button, and then navigate to the location of the firmware file.

6. Select the firmware file. This reveals the Upload button (Figure 2-52).

Figure 2-52. Upload Button



- 7. Click on the Upload button. After selecting the Upload button, you will see the progress of the upload in the Upload Progress bar.
- 8. When the upload is complete, you will see the words Upload finished under Status Messages.
- 9. At this point, you will see the progress of the upload's post processing in the Upload Post Processing bar.

Do not reboot the device before the upgrading process is complete.

- 10. When the process is complete, you will see the words SWUPDATE Successful under Status Messages.
- 11. The device will reboot automatically.
- 12. The Home page will display the version number of the firmware and indicate which boot partition is active.

Table 2-28 shows the web page items on the **Firmware** page.

Table 2-28. Firmware Page Parameters

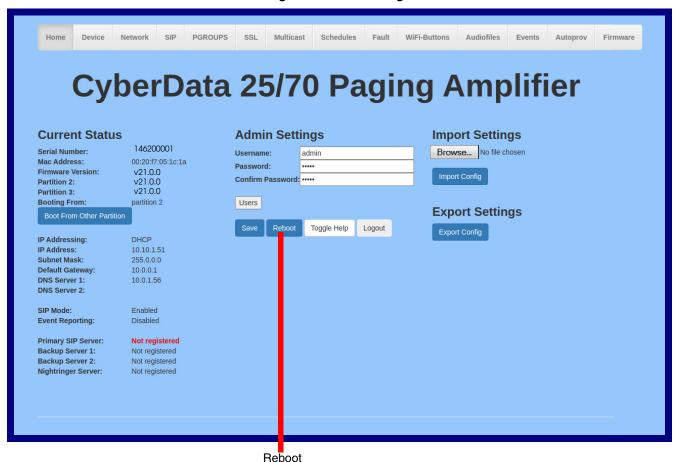
Web Page Item	Description	
Logout	Logs out the current user and returns to the user log in page	
Browse	Use the Browse button to navigate to the location of the firmware file that you want to upload.	
Upload	Click on the Upload button to automatically upload the selected firmware and reboot the system.	
	Note: This button only appears after the user has selected a firmware file.	
Upload progress	Status bar indicates the progress in uploading the file.	
Upload Post Processing	Processing Status bar indicates the progress of the software installation.	
Status Messages	Messages relevant to the firmware update process appear here.	

2.9.1 Reboot the SIP Paging 25V/70V Amplifier

To reboot a SIP Paging 25V/70V Amplifier, log in to the web page as instructed in Section 2.8.4, "Log in to the Configuration GUI".

1. Click **Reboot** (Figure 2-53). A normal restart will occur.

Figure 2-53. Home Page



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2.10 Command Interface

Some functions on the device can be activated using simple Post commands to the web interface. The examples in Table 2-29 use the free unix utility, wget commands. However, any program that can send HTTPS Post commands to the device should work.

2.10.1 Command Interface Post Commands

The commands in Table 2-29 require an authenticated session (a valid username and password to work).

Table 2-29. Command Interface Post Commands

Device Action	Post Command ^a wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=reboot"		
Reboot			
Place call to extension (example: extension 600)	wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command"post-data "request=call&extension=600"		
Terminate a call	wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=terminate"		
Test Relay	wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command"post-data "request=test_relay"		
Activate Relay	wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=activate_relay"		
eactivate Relay wgetuser adminpassword adminauth-no-challenge /dev/nullno-check-certificate "https://10.10.1.247/comm post-data "request=deactivate_relay"			
Speak IP Address wgetuser adminpassword adminauth-no-ch/dev/nullno-check-certificate "https://10.10.1.24 post-data "request=speak_ip_address"			
est Audio wgetuser adminpassword adminauth-no-challenge /dev/nullno-check-certificate "https://10.10.1.247/comma post-data "request=test_audio"			
Swap Boot partitions	wgetuser adminpassword adminauth-no-challengeno-check-certificatequiet -O /dev/null "https://10.10.1.81/command"post-data "request=swap_boot_partition"		

a. Type and enter all of each Post command on one line.

Appendix A: Mounting the SIP Paging 25V/70V Amplifier

A.1 Mounting

The SIP Paging 25V/70V Amplifier has several mounting options:

A.1.1 Wall Mount

Included with each unit are mounting ears which can be attached to the SIP Paging 25V/70V Amplifier unit to mount it to a wall (Figure A-1). Please note the hardware to mount the unit to the wall is not included.

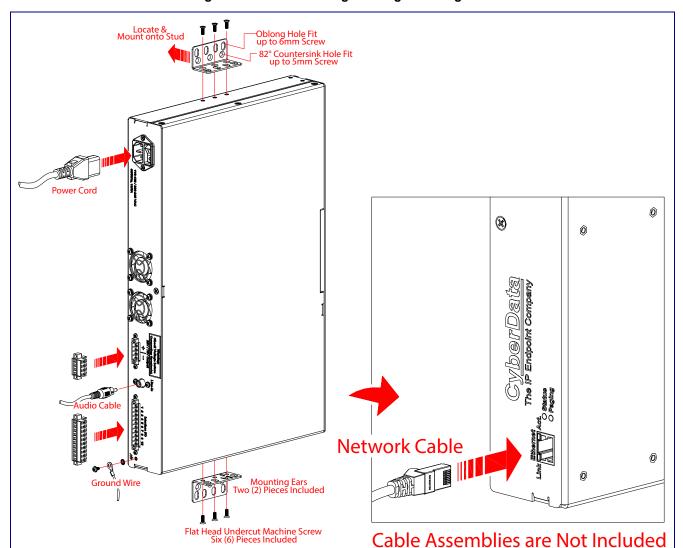
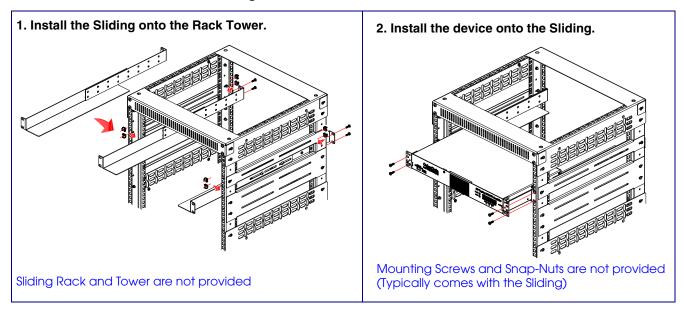


Figure A-1. Wall Mounting Utilizing Mounting Ears

A.1.2 Rackmount

Included with each unit are mounting ears to mount the SIP Paging 25V/70V Amplifier unit (Figure A-2). The mounting hardware should be compatible with most standard rackmount systems. Please note the hardware to mount to the rackmount system is unique to the rackmount enclosure and this hardware is not included. Each of the units also has multiple mounting holes on the side to mount standard rail systems.

Figure A-2. Rack Tower Installation



A.1.3 Desktop

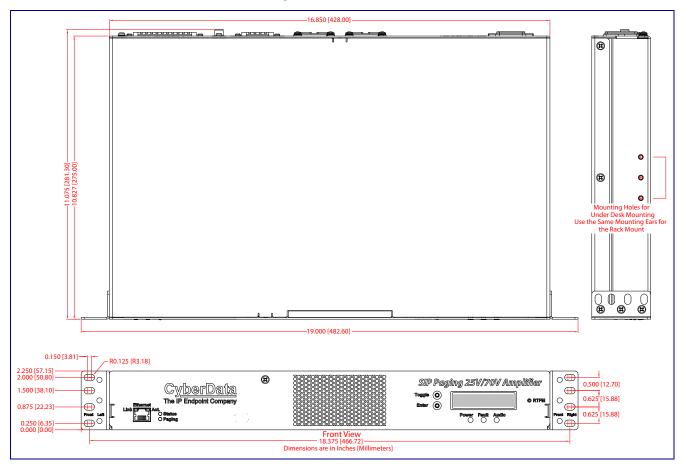
Included with the unit are rubber mounting feet that can be placed on the bottom of the SIP Paging 25V/70V Amplifier unit (Figure A-3). These rubber feet are meant for use when the unit is placed on a desktop or shelf.

Figure A-3. Desktop Version with Rubber Feet

A.2 Dimensions

Figure A-4 shows the dimensions for the SIP Paging 25V/70V Amplifier.

Figure A-4. Dimensions



Appendix A: Troubleshooting/Technical Support

A.1 Frequently Asked Questions (FAQ)

To see a list of frequently asked questions for your product, click on the **FAQs** tab at the following webpage:

https://www.cyberdata.net/pages/011579

A.2 Documentation

The documentation for this product is released in an English language version only.

To download PDF copies of CyberData product documentation, click on the **Downloads** tab at the following webpage:

https://www.cyberdata.net/pages/011579

A.3 Contact Information

Contact CyberData Corporation

3 Justin Court

Monterey, CA 93940 USA www.CyberData.net Phone: 831-373-2601 Fax: 831-373-4193

Sales Sales 831-373-2601, Extension 334

Technical Support The fastest way to get technical support for your VoIP product is to submit a VoIP Technical Support form at the following website:

https://support.cyberdata.net/

The Support Form initiates a ticket which CyberData uses for tracking customer requests. Most importantly, the Support Form tells us which PBX system and software version that you are using, the make and model of the switch, and other important information. This information is essential for troubleshooting. Please also include as much detail as possible in the **Comments** section of the Support Form.

Phone: (831) 373-2601, Extension 333

A.4 Warranty and RMA Information

The most recent warranty and RMA information is available at the following website address:

https://support.cyberdata.net/

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