



SIP Paging Server Operations Guide

SIP Compliant Part #011146 Document Part #931073K for Firmware Version 12.2.0

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The IP Endpoint Company	The fastest way to get technical support for your VoIP product is to submit a VoIP Technical Support form at the following website: http://support.cyberdata.net/
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Revision Information

Revision 931073K, which corresponds to firmware version 12.2.0, was released on May 30, 2020, and has the following changes:

- Updates Figure 2-11, "Home Page"
- Updates Figure 2-12, "Device Page"
- Updates Figure 2-14, "Network Page"
- Updates Figure 2-15, "SIP Page"
- Updates Figure 2-16, "SIP Page Set to Point-to-Point Mode"
- Updates Figure 2-17, "PGROUPS Page"
- Updates Figure 2-18, "PGROUPS Page (continued)"
- Updates Figure 2-19, "PGROUPS Page (continued)"
- Updates Figure 2-28, "Fault Page"
- Updates Figure 2-29, "Audiofiles Page"
- Updates Figure 2-30, "Audiofiles Page"
- Updates Figure 2-35, "Events Page"
- Updates Figure 2-36, "Autoprovisioning Page"
- Updates Figure 2-38, "Firmware Page"
- Updates Figure 2-39, "Home Page"
- Updates Section 1.2, "Features" to add TLS 1.2 and SRTP enhanced security for IP endpoints in a local or cloud-based environment
- Updates Table 1-1, "Specifications"
- Updates Table 2-3, "Web Page Navigation"
- Updates Table 2-11, "SIP Configuration Parameters" to add the SRTP setting
- Adds Section 2.4.10, "Configure the SSL Parameters"

Browsers Supported

The following browsers have been tested against firmware version 12.2.0:

- Chrome (version 78.0.3904.70)
- Firefox (version 72.0.2)
- Microsoft Edge (80.0.361.50)
- Internet Explorer (version: 11)

Pictorial Alert Icons

GENERAL ALERT	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.

GENERAL ALERT	Warning <i>Electrical Hazard:</i> This product should be installed by a licensed electrician according to all local electrical and building codes.
GENERAL ALERT	Warning <i>Electrical Hazard:</i> To prevent injury, this apparatus must be securely attached to the floor/wall in accordance with the installation instructions.
GENERAL ALERT	Warning The PoE connector is intended for intra-building connections only and does not route to the outside plant.

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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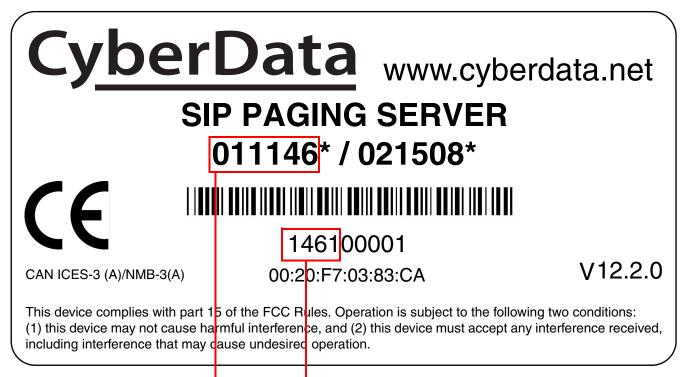
1 Product Overview

1.1 How to Identify This Product

To identify the SIP Paging Server, 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 011146.
- The serial number on the label should begin with 1461.

Figure 1-1. Model Number Label¹



Model number

Serial number begins with **1461**

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

1.2 Features

- Up to 25 unique bell files that can be used in up to 250 scheduled events
- Up to 25 stored messages that are utilized through paging groups
- Can page directly to Polycom phones
- Voice prompting and password controlled zones
- SIP RFC 3261 compatible
- NTP-based internal clock
- Multicast output
- Two SIP endpoints (one for Night Ringer)
- DTMF control of zone selection
- Delayed page support
- Line-In connection for background music multicasting
- Line-out connection to support analog amplifiers
- User uploadable audio files
- Web-based configuration and firmware upload
- PoE 802.3af enabled (Power-over-Ethernet)
- 19-inch Rack mount option
- TLS 1.2 and SRTP enhanced security for IP endpoints in a local or cloud-based environment

1.3 Specifications

Specifications	
Protocol	SIP RFC 3261 Compatible
Ethernet I/F	10/100 Mbps
Power Input	PoE 802.3af or 48VDC
Operating Temperature	-10° C to 50° C (14° F to 122° F)
Payload Types	G711, A-law and μ-law, G.722
Network Security	TLS/SSL 1.2 and SRTP
Page Port Output	Balanced 600 Ohm 5VPP
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
Dimensions ^a	6.11 inches [155.19 mm] Length
	4.05 inches [102.87 mm] Width
	1.15 inches [29.21 mm] Height
Boxed Weight	1.8 lbs.
Compliance	UL 62368-1, RoHS Compliant, FCC; Part 15 Class A, IEEE 802.3 Compliant; Reference Number for UL: E129569 Vol 4 Sec 1
Part Number	011146

Table 1-1. Specifications

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 Safety

This product is listed by UL. Representative samples of this product have been evaluated by UL and meet applicable safety standards. (Standard: UL 62368-1). This applies to the following products: 011145, 011146, 011233, 011280, 011295, 011368, 011372

Note You can download the Declaration of Conformity document from the **Downloads** tab of the product's webpage.

1.4.2 FCC Statement

This equipment has been tested and found to comply with the limits for a Class B 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.

2 Setting Up the SIP Paging Server

The topics in this chapter provide information on setting up, configuring, and using the SIP Paging Server.

2.1 Parts List

The packaging for the SIP Paging Server includes the parts in Table 2-1.

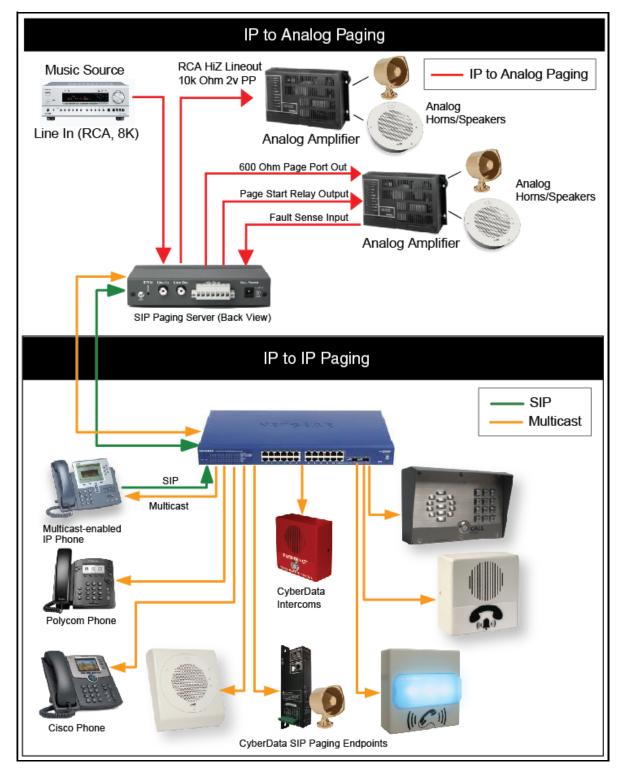
Quantity	Part Name	Illustration
1	SIP Paging Server	The second secon
1	Installation Quick Reference Guide	
1	Mounting Template (located on the last page of the <i>Installation Quick Reference</i>)	
1	Mounting Kit (part #070057A) which includes: (2) #4-6 x 7/8" Mounting Anchors (2) #4 x 1-1/4" Round Phillips Wood Screws	

Table 2-1. Parts List

2.2 Typical Installation

Figure 2-1 illustrates how the SIP Paging Server is normally installed as part of a paging system.





2.3 Connecting the SIP Paging Server

Before you connect the SIP Paging Server, 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 Server.

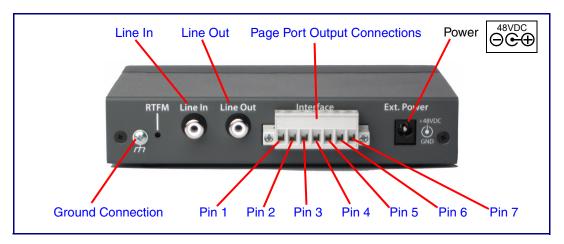


Figure 2-2. Connection Options

2.3.1 Ground Connection

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

2.3.2 Line In

This RCA 10K Ohm Hi-Z input connection allows you to connect the device to The RCA line-out (10K Ohm Hi-Z) of an external audio amplifier. The level of this input can be controlled by the potentiometer located on the front of the device (see Section 2.4.12, "Configure the Fault Detection Parameters").

2.3.3 Line Out

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

2.3.4 Page Port Output Connections

Table 2-1. Page Port Output Connections

Pin	Description
Pin 1	Fault Sense Input (Common). See Section 2.3.4.1, "Pin 1 and 2-Fault Sense Input (Common/Sense)".
Pin 2	Fault Sense Input (Sense). See Section 2.3.4.1, "Pin 1 and 2-Fault Sense Input (Common/Sense)".
Pin 3	Positive 600-Ohm Audio Output ^a . See Section 2.3.4.2, "Pin 3, 4, and 5—Positive/Negative 600-Ohm Audio Output/Audio Ground Reference".
Pin 4	Negative 600-Ohm Audio Output. ^a . See Section 2.3.4.2, "Pin 3, 4, and 5—Positive/Negative 600-Ohm Audio Output/Audio Ground Reference".
Pin 5	Audio Ground Reference. See Section 2.3.4.2, "Pin 3, 4, and 5—Positive/Negative 600-Ohm Audio Output/Audio Ground Reference".
Pin 6	Relay Contact - Common ^b . See Section 2.3.4.3, "Pin 6 and 7—Relay Contact (Common/Normally Open)".
Pin 7	Relay Contact - Normally Open ^b . See Section 2.3.4.3, "Pin 6 and 7—Relay Contact (Common/Normally Open)".

a. The 600-Ohm audio output of the page port is also suited for interfaces with lower input impedances.

b. 1 Amp at 30 VDC for continuous loads

2.3.4.1 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 2.4.12, "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.

2.3.4.2 Pin 3, 4, and 5—Positive/Negative 600-Ohm Audio Output/Audio Ground Reference

This output allows direct connection to paging amplifiers requiring a "Page Port" type input that meets a balanced 600 Ohm 5VPP signal.

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

When enabled on the web interface (Section 2.4.5, "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.

2.3.5 Removable Interface Connector

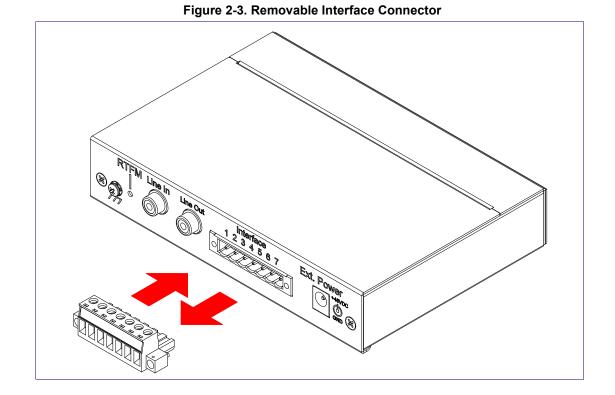


Figure 2-3 shows the interface connector that is removable on the SIP Paging Server.

2.3.6 Connect to the Power Source

To use PoE, plug a Cat 5 Ethernet cable from the SIP Paging Server **Ethernet** port to your network. As an alternative to PoE, you can plug one end of a +48V DC power supply into the Paging Server, and plug the other end into a receptacle. If required, connect the earth grounding wire to the chassis ground on the back of the unit. See Figure 2-4.

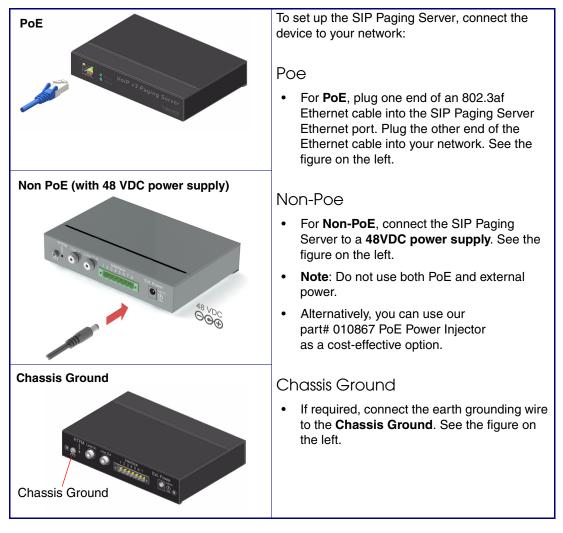


Figure 2-4. Connecting to the Power Source

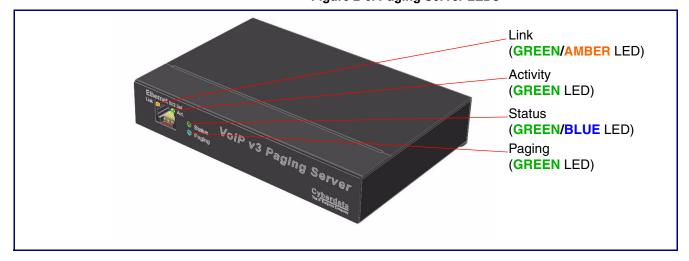
2.3.7 Connect to the Network

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





2.3.8 Confirm that the SIP Paging Server is Up and Running



The LEDs on the front of the SIP Paging Server verify the unit's operations.



2.3.8.1 Confirm Power on, Network Connectivity, and Connection Speed

When you plug in the Ethernet cable or power supply:

- The **GREEN/BLUE Status** LED and the **GREEN Paging** LED both blink at a rate of 10 times per second during the initial network setup.
- The round, GREEN/BLUE Status LED on the front of the SIP Paging Server comes on indicating that the power is on. Once the device has been initialized, this LED blinks at one second intervals.
- The square, **GREEN/AMBER Link** LED above the Ethernet port indicates that the network connection has been established. The Link LED changes color to confirm the auto-negotiated connection speed:
 - The Link LED is **GREEN** at 10 Mbps.
 - The Link LED is AMBER at 100 Mbps.
- The **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** page.

2.3.8.2 Verify Network Activity

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

2.3.9 Announcing the IP Address

To announce the IP address for the SIP Paging Server, briefly press and then quickly release the RTFM switch. See Figure 2-7.



2.3.10 Restore the Factory Default Settings

The SIP Paging Server is delivered with factory set default values for the parameters in Table 2-2. In addition, the settings for various UI web pages (such as the **Device Page**, **SIP Page**, etc.) are delivered with the factory default settings and can be restored to these default settings when you use the RTFM switch. However, uploaded audio files are not restored to the factory default settings when you use the RTFM switch.

Use the RTFM switch (see Figure 2-7) on the back of the unit to restore these parameters to the factory default settings.

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

Parameter	Factory Default Setting
IP Addressing	DHCP
IP Address ^a	10.10.10
Web Access Username	admin
Web Access Password	admin
Subnet Mask ^a	255.0.0.0
Default Gateway ^a	10.0.0.1

Table 2-2. Factory Default Settings

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

To restore these parameters to the factory default settings:

- 1. Press and hold the RTFM switch until the status and paging lights come on.
- 2. Continue to press the RTFM switch until after you see the indicator lights go off and you hear the "restoring defaults" announcement.
- 3. Release the RTFM switch.
- 4. The SIP Paging Server settings are restored to the factory defaults.

2.4 Configuring the SIP Paging Server

Use this section to configure the VoIP paging server.

2.4.1 Gather the Required Configuration Information

Have the following information available before you configure the SIP Paging Server.

2.4.1.1 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 Server parameters:

- IP Address
- Subnet Mask
- Default Gateway

2.4.1.2 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.

2.4.1.3 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

2.4.2 SIP Paging Server Web Page Navigation

Table 2-3 shows the navigation buttons that you will see on every SIP Paging Server web page.

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.
Audiofiles	Link to the Audiofiles page.
Events	Link to the Events page.
Autoprov	Link to the Autoprovisioning page.
Firmware	Link to the Firmware page.

Table 2-3. Web Page Navigation

2.4.3 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 2-8 and Figure 2-9.

Figure 2-8. Toggle/Help Button

2. You will see a question mark (?) appear next to each web page item that has been provided with a short description by the Help feature. See Figure 2-9.

Figure 2-9. Toggle Help Button and Question Marks

Stored Net			
Addressing Mode	?		
hostname:	SipDevice03cab3	?	
IP Address:	10.10.10.10	2	Questian merili
Subnet Mask:	255.0.0.0	?	Question mark appears next to the
Default gw_addr:	10.0.0.1	2	web page items
DNS Server 1:	10.0.0.1	?	
DNS Server 2:	10.0.0.1	?	

3. Move the mouse pointer to hover over the question mark (?), and a short description of the web page item will appear. See Figure 2-10.

	hostname				
Stored Net	This is the hostname provided by the DHCP server. See the Operations Guide and DHCP/DNS server documentation for more information.				
Addressing Mode:	Enter up to 64				
hostname:	SipDevice03cal				
IP Address:	10.10.10.10	?			
Subnet Mask:	255.0.0.0	?			
Default gw_addr:	10.0.0.1	?			
DNS Server 1:	10.0.0.1	?			
DNS Server 2:	10.0.0.1	?			
	Question ma	ark A short description of the			

Figure 2-10. Short Description Provided by the Help Feature

Question mark

A short description of the web page item will appear

2.4.4 Log in to the Configuration GUI

- 1. Open your browser to the SIP Paging Server IP address.
- **Note** If the network does not have access to a DHCP server, the device will default to an IP address of 10.10.10.10.
- Note Make sure that the PC is on the same IP network as the SIP Paging Server.
- **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/products/011146

The unit ships in DHCP mode. To get to the **Home** 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 Server configuration *after* the initial configuration, log in using the IP address you assign to the device. Section 2.4.6, "Configure the Network Parameters" provides instructions for entering the IP address.
- 2. When prompted, use the following default **Username** and **Password** to open the configuration Home page:

Username: admin

Password: admin

Change the **To change the default Web access Username and Password:** Default Username

- and Password
- 1. 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 2-11. Home Page

Home De	vice Network	SIP	PGROUPS	SSL Schedules	Fault	Audiofiles	Events	Autoprov	Firmware
	Cybe	erD	ata v	3.1 P	agi	ng	Ser	ver	
Current Sta	atus		Admin S	Settings		Impo	ort Settir	ngs	
Serial Number: Mac Address: Firmware Version:	146100001 00:20:f7:04:62:a9 v12.2.0		Username: Password: Confirm Pass	admin		Brow	<mark>/se</mark> No file (t Config		
IP Addressing: IP Address: Subnet Mask: Default Gateway: DNS Server 1: DNS Server 2:	DHCP 10.10.1.72 255.0.00 10.0.0.1 10.0.1.56		Save Re	boot Toggle Help		_	D rt Settir	ngs	
SIP Mode: Event Reporting: Nightringer:	Enabled Disabled Disabled								
Primary SIP Server Backup Server 1: Backup Server 2:	Not registered								

- 4. On the Home Page, review the setup details and navigation buttons described in Table 2-4
- **Note** The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

	-
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.
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. Then, click Save and Reboot to store changes.
Export Settings	
Export Config	Click Export to export the current configuration to a file.
	Click the Save button to save your configuration settings.
Save	Note: You need to reboot for changes to take effect.

Table 2-4. Home Page Overview

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.

Note You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

At this point you can:

- Review the SIP Paging Server's **Current Settings**. Use the RTFM switch to restore the factory default settings. See Section 2.3.10, "Restore the Factory Default Settings".
- Configure the device parameters. Click on the **Device** button and see Section 2.4.5, "Configure the Device Parameters" for instructions.
- Configure the network parameters. Click on the **Network** button and see Section 2.4.6, "Configure the Network Parameters" for instructions.
- Configure the SIP parameters. Click on the **SIP** button and see Section 2.4.7, "Configure the SIP Parameters" for instructions.
- Configure the PGROUPS parameters. Click on the PGROUPS button and see Section 2.4.8, "Configure the Paging Groups (PGROUPS) Parameters" for instructions.
- Configure the fault detection parameters. Click on the **Fault** button and see Section 2.4.12, "Configure the Fault Detection Parameters" for instructions.
- Configure the audio parameters. Click on the **Audiofiles** button and see Section 2.4.13, "Configure the Audio Parameters" for instructions.
- Configure the event parameters. Click on the Events button and see Section 2.4.14, "Configure the Event Parameters" for instructions.
- Configure the autoprovisioning parameters. Click on the **Autoprov** button and see Section 2.4.15, "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 2.5, "Upgrading the Firmware" for instructions.

2.4.5 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 2-12.

Figure 2-12. Device Page

Line-in Settings		Relay Settings	
Enable Line-in to Line-out Loopback Enable Line-in to Multicast: Multicast Address: Multicast Port: Detect Line-in Silence:	224.1.2.3 2000	Activate Relay on Local Audio: 🗖	
Clock Settings		Misc Settings	
Set Time with NTP server on boot:		Device Name:	CyberData Paging Server
NTP Server:	north-america.pool.ntp.org	Bypass DTMF	
Posix Timezone String (see manual)		DTMF Duration: Beep on Init:	500
Periodically sync time with server: Time update period (in hours):	24	Beep on Page:	 ✓
Current Time:	18:23:55	Enable Polycom Paging on Multicast:	
Current finite.			
Current finie.		Polycom Transmit Channel: Disable HTTPS (NOT recommended):	1 •

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

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).
Detect Line-in Silence ?	If silence is detected on line-input, the multicast stream will be stopped to reduce network traffic.
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.
Clock Settings	
Set Time with NTP Server on boot ?	When selected, the time is set with an external NTP server when the device restarts.
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.
Posix Timezone String ?	See Section 2.4.5.2, "Time Zone Strings" for information about how to use the Posix Timezone String to specify time zone and daylight savings time where applicable. Enter up to 63 characters.
Periodically sync time with server ?	When selected, the time is periodically updated with the NTP server at the configured interval below.
Time update period (in hours) ?	The time interval after which the device will contact the NTP server to update the time. Enter up to 4 digits.
Current Time	Allows you to input the current time. (6 character limit)
Misc Settings	
Device Name ?	Type the device name. Enter up to 25 characters.
Bypass DTMF 🛜	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
DTMF Duration ?	The duration, in milliseconds, of DTMF tones played out of the device's analog audio ports (1-65535).

Table 2-5. Device Configuration Parameters

Web Page Item	Description
Beep on Init 🛜	Device will play the user defined "pagetone" audio file when it boots.
Beep on Page 🛜	Device will play the user defined "pagetone" audio file before playing a SIP page.
Enable Polycom Paging on Multicast ?	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 Transmit Channel ?	Destination channel for Polycom Group Paging multicast.
Disable HTTPS (NOT recommended) ?	Disables the encrypted connection to the webpage. We do not recommend disabling HTTPS for security reasons.
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.
	Click the Save button to save your configuration settings.
Save	Note: You need to reboot for changes to take effect.
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 2-5. Device Configuration	Parameters	(continued)
Tuble 2-0. Device Configuration	i urumeters	(continuca)

Note You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

2.4.5.1 Polycom Paging

The Polycom Paging feature is supported on Polycom IP phones using UC Software 4.0.0 and higher. The Polycom paging feature operates in two modes: Push-to-Talk (PTT) and Group Paging Only Group Paging mode pages are supported by the Paging Server.

Polycom phones use the same multicast IP address and port number for both PTT and Group Paging multicasts. Make sure to note the Polycom multicast IP address and port number before configuring the CyberData V3 Paging Server. Polycom phones use a default multicast IP address of 224.0.1.116 and odd-numbered port 5001.

While the same multicast IP address and port number is used for all Polycom pages in both modes, Polycom uses numbered "groups" or "channels" to differentiate between each paging group. Each "group" or "channel" is numbered 1 through 25.

The Paging Server can transmit to Group Paging groups 1 through 25 only for one-way audio pages. The transmit channel is configurable. The Polycom phones must subscribe to this channel in order to receive one-way audio pages from the Paging Server.

When configuring Polycom phones for their Group Paging feature, be sure the following settings are configured:

- Payload Size = 20 ms (milliseconds)
- Codec = G.711Mu

The Polycom Group Paging multicast transmitted by the Paging Server is G.711Mu encoded with a payload size of 20 ms.

It is imperative to note the Paging Server assumes the Polycom phones will use an odd-numbered port. Since it is not possible to configure the V3 Paging Server to transmit multicasts on odd-numbered ports (which maintains conformance with RFC 1889), it is necessary to use the next lower even port number when specifying the Polycom multicast IP address and port number on the **PGROUPS Page**. Using the Polycom default port 5001 will require you to configure the Paging Server to transmit on the next lower even port 5000.

Thus, configuring the Paging Server for Polycom Paging is a two-step process:

- 1. Enable Polycom Paging on the Paging Server by checking the box to **Enable Polycom Paging** on Multicast on the Device Page.
- 2. Specify the Polycom IP address and use the next lower even port number for the desired paging group on the **PGROUPS Page**.
- 3. Save and reboot to store changes.

2.4.5.2 Time Zone Strings

The posix time zone string tells the internal date and time utilities how to handle daylight savings time for different time zones. Table 2-6 shows some common strings.

Time Zone	Time Zone String		
US Pacific time	PST8PDT,M3.2.0/2:00:00,M11.1.0/2:00:00		
US Mountain time	MST7MDT,M3.2.0/2:00:00,M11.1.0/2:00:00		
US Eastern Time	EST5EDT,M3.2.0/2:00:00,M11.1.0/2:00:00		
Phoenix Arizona ^a	MST7		
US Central Time	CST6DST,M3.2.0/2:00:00,M11.1.0/2:00:00		

Table 2-6. Common Time Zone Strings

a.Phoenix, Arizona does not use daylight savings time.

Table 2-7 shows a breakdown of the parts that constitute the following time zone string:

• CST6DST,M3.2.0/2:00:00,M11.1.0/2:00:00

Time Zone String Part	Meaning
CST6CDT	The time zone offset from GMT and three character identifiers for the time zone.
CST	Central Standard Time
6	The (hour) offset from GMT/UTC
CDT	Central Daylight Time
M3.2.0/2:00:00	The date and time when daylight savings begins.
МЗ	The third month (March)
.2	The 2nd occurrence of the day (next item) in the month
.0	Sunday
/2:00:00	Time of day to change
M11.1.0/2:00:00	The date and time when daylight savings ends.
M11	The eleventh month (November)
.1	The 1st occurrence of the day (next item) in the month
.0	Sunday
/2:00:00	Time of day to change

Table 2-7. Time Zone String Parts

Time Zone String Examples

Table 2-8 has some more examples of time zone strings.

Time Zone	Time Zone String		
Tokyo ^a	IST-9		
Berlin ^b	CET-1MET,M3.5.0/1:00,M10.5.0/1:00		

Table 2-8. Time Zone String Examples

a.Tokyo does not use daylight savings time.

b.For Berlin, daylight savings time starts on the last Sunday in March at 01:00 UTC, and ends on the last Sunday in October at 01:00 UTC, and is one hour ahead of UTC.

Time Zone Identifier A user-definable three or four character time zone identifier (such as PST, EDT, IST, MUT, etc) is needed at the beginning of the posix time zone string to properly set the time. However, the specific letters or numbers used for the time zone identifier are not important and can be any three or four letter or number combination that is chosen by the user. However, the time zone identifier cannot be blank.

Figure 2-13	. Three or	Four Character	Time Zone	Identifier
-------------	------------	----------------	-----------	------------

PST8PDT,M3.2.0/2:00:00,M11.1.0/2:00:00

Three or four character time zone identifier at the beginning of the time zone string. The identifier can be any three or four letter or number combination chosen by the user.

You can also use the following URL when a certain time zone applies daylight savings time:

http://www.timeanddate.com/time/dst/2011.html

World GMT Table Table 2-9 has information about the GMT time in various time zones.

Table 2-9. World GMT Table

Time Zone	City or Area Zone Crosses		
GMT-12	Eniwetok		
GMT-11	Samoa		
GMT-10	Hawaii		
GMT-9	Alaska		
GMT-8	PST, Pacific US		
GMT-7	MST, Mountain US		
GMT-6	CST, Central US		
GMT-5	EST, Eastern US		
GMT-4	Atlantic, Canada		
GMT-3	Brazilia, Buenos Aries		
GMT-2	Mid-Atlantic		
GMT-1	Cape Verdes		
GMT	Greenwich Mean Time, Dublin		

Table 2-9.	World	GMT	Table	(continued)	١
	V UIIU		Table	Continueu	,

City or Area Zone Crosses		
Berlin, Rome		
Israel, Cairo		
Moscow, Kuwait		
Abu Dhabi, Muscat		
Islamabad, Karachi		
Almaty, Dhaka		
Bangkok, Jakarta		
Hong Kong, Beijing		
Tokyo, Osaka		
Sydney, Melbourne, Guam		
Magadan, Soloman Is.		
Fiji, Wellington, Auckland		
	Berlin, Rome Israel, Cairo Moscow, Kuwait Abu Dhabi, Muscat Islamabad, Karachi Almaty, Dhaka Bangkok, Jakarta Hong Kong, Beijing Tokyo, Osaka Sydney, Melbourne, Guam Magadan, Soloman Is.	

2.4.6 Configure the Network Parameters

Configuring the network parameters enables your network to recognize the SIP Paging Server and communicate with it. Click the **Network** button on the **Home** page to open the **Network** page.

		PGROUPS SSL Schedules Fault Audiofiles	Events Autoprov Firmware
			_
C	vberĽ	ata v3.1 Paging	Server
	,		
Stored Network	k Settings	VLAN Settings	
ddressing Mode:	Static • DHCP	VLAN ID (0-4095): 0	
lostname:	SipDevice0462a9	VLAN Priority (0-7): 0	
P Address:	10.10.10		
ubnet Mask:	255.0.0.0		
efault Gateway:	10.0.0.1		
NS Server 1:	10.0.0.1		
NS Server 2:	10.0.0.1		
HCP Timeout in seconds			
A value of -1 will retry fore	ever		
		Save Reboot Toggle Help	
Current Networ	rk Settings	our report roggie ricip	
P Address: 10.10.1.7	72		
ubnet Mask: 255.0.0.0			
efault Gateway: 10.0.0.1			
NS Server 1: 10.0.1.56 NS Server 2:	0		

Figure 2-14. Network Page

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

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

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 2.3.10, "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 2-10. Network Configuration Parameters

Web Page Item	Description
	Click the Save button to save your configuration settings.
Save	Note: You need to reboot for changes to take effect.
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 2-10. Network Configuration Parameters (continued)

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 Server's **IP Address**: The SIP Paging Server 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.
- **Note** You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

2.4.7 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.

Home Device Network	SIP PGROUPS SSL	Schedules Fault Audiofiles	Events Autoprov Firmware
_		_	
Cyb	erData v?	.1 Paging	Server
Cyb	ci Data vo	. T i aging	
SIP Settings		Nightringer Settings	
Enable SIP operation:	₹	Enable Nightringer:	
SIP Transport Protocol:	UDP T	SIP Server:	10.0.253
TLS Version:	1.2 only (recommended)	Remote SIP Port:	5060
Verify Server Certificate: Register with a SIP Server:	 ✓ 	Local SIP Port:	5061
		Outbound Proxy:	
	10.0.0.253	Outbound Proxy Port:	0
Primary SIP User ID:	199	User ID:	241
	199	Authenticate ID:	241
Primary SIP Auth Password:		Authenticate Password:	
		Re-registration Interval (in seconds):	: 360
Backup SIP Server 1:		Relay rings to multicast:	
Backup SIP User ID 1:		Multicast Address:	224.1.2.32
Backup SIP Auth ID 1:		Multicast Port:	2020
Backup SIP Auth Password 1:			
		Call Disconnection	
Backup SIP Server 2:			
Backup SIP User ID 2:		Terminate Call after delay: 0	
Backup SIP Auth ID 2:			
Backup SIP Auth Password 2:		Codec Selection	
		Force Selected Codec:	
	5060	Codec: PCMU (G.711	1, u-law) 🔻
	5060		
Outbound Proxy:		DTD Sottings	
Outbound Proxy Port:	0	RTP Settings	
Disable rport Discovery:		RTP Port (even): 10500	
		Jitter Buffer: 50	
Re-registration Interval (in seconds):		SRTP: Disabled •	
		Save Reboot Toggle Help	

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

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

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.
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.
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 2.4.7.1, "Point-to-Point Configuration").
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.
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.
Backup SIP Server 1 <mark>?</mark>	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 1 ?	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 1 ?	Specify the Authenticate ID for the first backup SIP server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.

Table 2-11. SIP Configuration Parameters

Web Page Item	Description
Backup SIP Auth Password 1 🛜	Specify the Authenticate Password for the first backup SIP server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.
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.
Backup SIP User ID 2 🛜	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 2 🛜	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 2 🛜	Specify the Authenticate Password for the second backup SIP server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.
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.
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.
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.
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.
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.
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.
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.

Table 2-11. SIP Configuration Parameters (continued)

Web Page Item	Description
Jitter Buffer <mark>?</mark>	Specify the size of the jitter buffer (in milliseconds) used for SIP calls. Valid values are 50-1000.
SRTP ?	When enabled, a SIP call's audio streams are encrypted using SRTP.
Nightringer Settings	
Enable Nightringer 🛜	When Nightringer is enabled, the device will attempt to register a second extension with the SIP server. Any calls made to this extension will play a ringtone (corresponds to Night Ring on the Audiofiles page). By design, it is not possible to answer a call to the Nightringer extension.
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.
Remote SIP Port 🛜	The Remote SIP Port is the port number the device will use as the destination port when sending SIP messages for the Nightringer extension. 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 for the Nightringer extension. This value cannot be the same as the Local SIP Port for the primary extension. The default Local SIP Port is 5061. The supported range is 0-65536. Enter up to 5 digits.
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 for the Nightringer extension. 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 for the Nightringer extension. 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 for the Nightringer extension. A value of 0 will default to 5060. The supported range is 0-65536. Enter up to 5 digits.
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.
Authenticate ID ?	Specify the Authenticate ID for the SIP Server. This parameter is required for SIP registration authentication. Enter up to 64 alphanumeric characters.
Authenticate 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.

Web Page Item	Description
Terminate Call After De	Iay ?Automatically terminate an active call after a given delay in seconds. A value of 0 will disable this function. Enter up to 8 digits.
Codec Selection	
Force Selected Codec	When configured, this option will allow you to force the device to negotiate for the selected codec [PCMU(G.711, u-law), PCMA(G.711, a-law), or G.722]. Otherwise, the device will perform codec negotiation using the default list of supported codecs.
Codec ?	Select desired codec (only one may be chosen).
	Click the Save button to save your configuration settings.
Save	Note: You need to reboot for changes to take effect.
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.
Note	You must click on the Save button and then the Reboot button for the changes to take effect.
Note	• For specific server configurations, go to the following website address:
	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
	For SIP Registration , designate whether you want the VoIP Paging Server to register with yo SIP server.
5. /	At Unregister on Reboot:
	a. Select Yes to automatically unregister the SIP Paging Server when you reboot it.
	b. Select No to keep the SIP Paging Server registered when you reboot it.
C	In the Desister Evaluation field, enter the number of seconds the SID Desing Conver

Table 2-11. SIP Configuration Parameters (continued)

6. In the **Register Expiration** field, enter the number of seconds the SIP Paging Server registration lease remains valid with the SIP Server. The SIP Paging Server automatically reregisters with the SIP server before the lease expiration timeout.

Note You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

2.4.7.1 Point-to-Point Configuration

When the board is set to not register with a SIP server, it's possible to set the device to dial out to a single endpoint. To do this, do the following:

- 1. On the SIP page (Figure 2-16), make sure that the **Register with a SIP Server** parameter is not selected.
- 2. Type the IP address of the remote device that you want to contact into the **Dial out Extension** field

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

Figure 2-16. SIP Page Set to Point-to-Point Mode

Home Device Netw	ork SIP PGROUPS SSL	Schedules Fault Audiofiles	Events Autoprov Firmware
Cv	perData v	3.1 Paging	Server
. .			
SIP Settings		Nightringer Setting	ie -
Enable SIP operation: SIP Transport Protocol:	UDP V	Enable Nightringer:	
TLS Version:	1.2 only (recommended)	SIP Server:	10.0.253
Verify Server Certificate:		Remote SIP Port:	5060
Register with a SIP Server:		Local SIP Port:	5061
Use Cisco SRST:		Outbound Proxy:	
Primary SIP Server:	10.0.0253	Outbound Proxy Port:	0
Primary SIP User ID:	199	User ID:	241
Primary SIP Auth ID:	199	Authenticate ID:	241
Primary SIP Auth Password:		Authenticate Password:	•••••
		Re-registration Interval (in seco	nds): 360
Backup SIP Server 1:		Relay rings to multicast:	
Backup SIP User ID 1:		Multicast Address:	224.1.2.32
Backup SIP Auth ID 1:		Multicast Port:	2020
Backup SIP Auth Password 1:			
Rackun SID Server 2:		Call Disconnection	

Device is set to **NOT** register with a SIP server

2.4.8 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 2-17.

Figure 2-17. PGROUPS Page

Home Device Network SIP	PGROUPS Schedules Fault Audiofiles Events Autoprov Firmware
CyberDa	ata v3.1 Paging Server
	Paging Groups
	# Address Port Name Code TTL Lineout
	0 234.2.1.1 2000 PagingGroup00 255 Yes Edit
	1 234.2.1.2 2002 PagingGroup01 255 Yes Edit
	2 234.2.1.3 2004 PagingGroup02 255 Yes Edit
	3 234.2.1.4 2006 PagingGroup03 255 Yes Edit
	4 234.2.1.5 2008 PagingGroup04 255 Yes Edit
	5 234.2.1.6 2010 PagingGroup05 255 Yes Edit
	6 234.2.1.7 2012 PagingGroup06 255 Yes Edit
	7 234.2.1.8 2014 PagingGroup07 255 Yes Edit
	8 234.2.1.9 2016 PagingGroup08 255 Yes Edit
	9 234.2.1.10 2018 PagingGroup09 255 Yes Edit
	10/234.2.1.11 2020 PagingGroup10 255 Yes Edit
	11/234.2.1.12 2022 PagingGroup11 255 Yes Edit
	12 234.2.1.13 2024 PagingGroup12 255 Yes Edit
	13 234.2.1.14 2026 PagingGroup13 255 Yes Edit
	14/234.2.1.15 2028 PagingGroup14 255 Yes Edit
	15/234.2.1.16 2030 PagingGroup15 255 Yes Edit
	16 (234.2.1.17 2032 PagingGroup16 255 Yes Edit
	17/234.2.1.18 2034 PagingGroup17 255 Yes Edit
	18/234.2.1.19 2036 PagingGroup18 255 Yes Edit
	19/234.2.1.20 2038 PagingGroup19 255 Yes Edit
	20 234.2.1.21 2040 PagingGroup20 255 Yes Edit
	21/234.2.1.22 2042 PagingGroup21 255 Yes Edit
	22/234.2.1.23 2044 PagingGroup22 255 Yes Edit
	23/234.2.1.24 2046 PagingGroup23 255 Yes Edit
	24/234.2.1.25 2048 PagingGroup24 255 Yes Edit
	25/234.2.1.26 2050 PagingGroup25 255 Yes Edit
	26 234.2.1.27 2052 PagingGroup26 255 Yes Edit
	07/024/03/09/00/00/00/07/00/07/00/07/07/07/07/07/07/

Figure 2-18. PGROUPS Page (continued)

Figure 2-19. PGROUPS Page (continued)

		PagingGroup77	255		Edit	
78 234.	2.1.79 2156	PagingGroup78	255	Yes	Edit	
79 234.	2.1.80 2158	PagingGroup79	255	Yes	Edit	
80 234.	2.1.81 2160	PagingGroup80	255	Yes	Edit	
81 234.	2.1.82 2162	PagingGroup81	255	Yes	Edit	
82 234.	2.1.83 2164	PagingGroup82	255	Yes	Edit	
83 234.	2.1.84 2166	PagingGroup83	255	Yes	Edit	
84 234.	2.1.85 2168	PagingGroup84	255	Yes	Edit	
85 234.	2.1.86 2170	PagingGroup85	255	Yes	Edit	
86 234.	2.1.87 2172	PagingGroup86	255	Yes	Edit	
87 234.	2.1.88 2174	PagingGroup87	255	Yes	Edit	
88 234.	2.1.89 2176	PagingGroup88	255	Yes	Edit	
89 234.	2.1.90 2178	PagingGroup89	255	Yes	Edit	
90 234.	2.1.91 2180	PagingGroup90	255	Yes	Edit	
91 234.	2.1.92 2182	PagingGroup91	255	Yes	Edit	
92 234.	2.1.93 2184	PagingGroup92	255	Yes	Edit	
93 234.	2.1.94 2186	PagingGroup93	255	Yes	Edit	
94 234.	2.1.95 2188	PagingGroup94	255	Yes	Edit	
95 234.	2.1.96 2190	PagingGroup95	255	Yes	Edit	
96 234.	2.1.97 2192	PagingGroup96	255	Yes	Edit	
97 234.	2.1.98 2194	PagingGroup97	255	Yes	Edit	
98 234.	2.1.99 2196	PagingGroup98	255	Yes	Edit	
99 234.	2.1.100 2198	PagingGroup99	255	Yes	Edit	
		Reboot				

Figure 2-20. PGROUPS Page (continued)

2. On the **PGROUPS** page, enter values for the parameters indicated in Table 2-12.

Web Page Item	Description
#	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.
Reboot	Click on the Reboot button to reboot the system.
	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 2-21).
	2. In the Configure PGROUP window (Figure 2-21), edit the PGROUP (see Table 2-13, "Configure PGROUP Window") and click on the Save Changes button.
	3. On the PGROUPS page (Figure 2-19), click on the Reboot button.

Table 2-12. PGROUPS Parameters

Figure 2-21. Configure PGR	OUP Window
----------------------------	------------

GROUP	0	
ddress	234.2.1.1]
Port	2000]
Name	PagingGroup00]
Security Code]
TTL	255]
Line-out	•	
Play Stored Message	0	
Audio File	•	
Times to Play	1]

The parameters for the **Configure PGROUP** window are shown in Table 2-13.

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 V3 Paging Server'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.
Lineout	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).
	Click the Save Changes button to save your configuration settings.
Save Changes	Note: You need to reboot for changes to take effect.
Cancel	Click the Cancel button to close the Configure PGROUP window.

Table 2-13. Configure PGROUP Window

2.4.9 Operating the Paging Server

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

2.4.9.1 DTMF Bypassed

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

2.4.9.2 DTMF Not Bypassed

- When the SIP Paging Server 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 Server 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 Server 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 Server 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 Server 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 Server 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 paging server 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 Server can negotiate the multicast stream via SIP regardless of the bypass state. However, if the SIP Paging Server 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").

2.4.10 Configure the SSL Parameters

1. Click **SSL** menu button to open the **SSL** page (Figure 2-22).

Figure 2-22. SSL Configuration Page

н	lome Device Network	SIP PGROUPS SSL Schedules	Fault Audiofiles Events Autoprov Firmward
	Cyber	Data v3.1 Pa	ging Server
Sei	rver CAs	Client Certificate	Test SSL Connection
Imp	OWSE No file chosen	commonName = CyberData SIP Device validFrom = Jul 10 17:56:03 2018 (validTo = Jul 7 17:56:03 2028 GMI	1011. 0000
	store Defaults Remove All	Client CA	
		List of Trusted CA	S
L	DST_ACES_CA_X6.crt		Info Remove
2	DST_Root_CA_X3.crt		Info Remove
3	Deutsche_Telekom_Root_CA_2.crt		Info Remove
ļ	DigiCert_Assured_ID_Root_CA.crt		Info Remove
j	DigiCert_Assured_ID_Root_G2.crt		Info Remove
6	DigiCert_Assured_ID_Root_G3.crt		Info Remove
	DigiCert_Global_Root_CA.crt		Info Remove
3	DigiCert_Global_Root_G2.crt		Info Remove
)	DigiCert_Global_Root_G3.crt		Info

12	Equifax_Secure_CA.crt	Info	Remove
13	Equifax_Secure_Global_eBusiness_CA.crt	Info	Remove
14	Equifax_Secure_eBusiness_CA_1.crt	Info	Remove
15	GeoTrust_Global_CA.crt	Info	Remove
16	GeoTrust_Global_CA_2.crt	Info	Remove
17	GeoTrust_Primary_Certification_Authority.crt	Info	Remove
18	GeoTrust_Primary_Certification_Authority_+_G2.crt	Info	Remove
19	GeoTrust_Primary_Certification_AuthorityG3.crt	Info	Remove
20	GeoTrust_Universal_CA.crt	Info	Remove
21	GeoTrust_Universal_CA_2.crt	Info	Remove
22	ISRG_Root_X1.crt	Info	Remove
23	VeriSign_Class_3_Public_Primary_Certification_AuthorityG4.crt	Info	Remove
24	VeriSign_Class_3_Public_Primary_Certification_AuthorityG5.crt	Info	Remove
25	VeriSign_Universal_Root_Certification_Authority.crt	Info	Remove
26	Verisign_Class_1_Public_Primary_Certification_Authority.crt	Info	Remove
27	Verisign_Class_1_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
28	Verisign_Class_2_Public_Primary_Certification_AuthorityG2.crt	Info	Remove
29	Verisign_Class_2_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
30	Verisign_Class_3_Public_Primary_Certification_Authority.crt	Info	Remove
31	Verisign_Class_3_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
32	thavte_Primary_Root_CA.crt	Info	Remove
33	thavte_Primary_Root_CAG2.crt	Info	Remove
34	thavte_Primary_Root_CA_*_G3.crt	Info	Remove

Figure 2-23. SSL Configuration Page

- 2. On the SSL 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.

Web Page Item	Description
Server 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.
Apply/Reboot	Reboots the device and applies settings and activates imported certificates.
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.
Client Certificate	When doing mutual authentication this device will present a client certificate with these parameters.
Client CA ?	Right click and Save Link As to get the Cyberdata CA used to sign this client certificate.
Test SSL Connection	
Server ?	The ssl test server address as a fully qualified domain name or in IPv4 dotted decimal notation.
Port ?	The ssl test server port. The supported range is 0-65536. SIP connections over TLS to port 5060 will do the same.
Test TLS connection	Use this button to test a TLS connection to a remote server. 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.
List of Trusted CAs	
Info	Provides details of the certificate. After clicking on this button, the Certificate Info Window appears. See Section 2.4.10.1, "Certificate Info Window".

Table 2-14. SSL Configuration Parameters

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 2.4.10.2, "Remove Server Certificate Window".

Table 2-14. SSL Configuration Parameters (continued)

2.4.10.1 Certificate Info Window

The **Certificate Info Window** provides details of the certificate. This window appears after clicking on the **Info** button. See Figure 2-24.

Figure 2-24. Certificate Info Window

Cer	rtificate Info	
organizationalUnitName	= ACCV = ES 11 GMT	
		ОК

2.4.10.2 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 2-25.

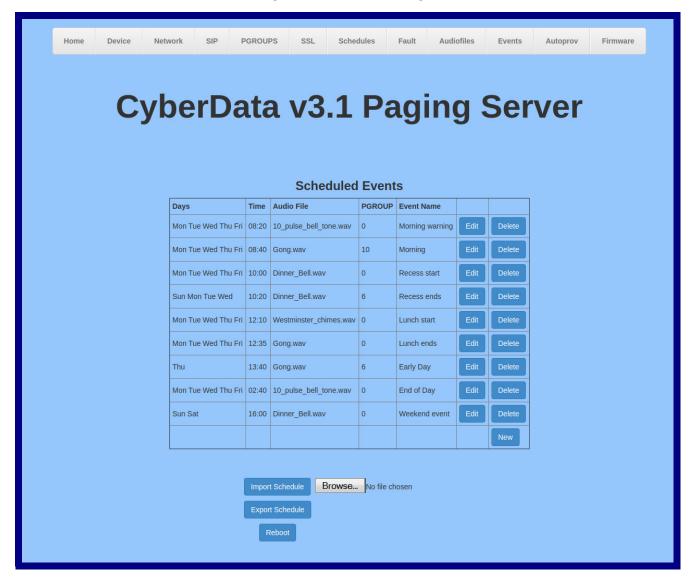
Figure 2-25. Remove Server Certificate Window

Remove Server Certificate		×
Are you sure you want to remove ACCVRAIZ1.crt?		
	Cancel	Remove

2.4.11 Configure the Schedules Parameters

1. Click on the **Schedules** button to open the **Schedules** page. See Figure 2-26.

Figure 2-26. Schedules Page



2. On the **Schedules** page, enter values for the parameters indicated in Table 2-17.

Web Page Item	Description
Scheduled Events	
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.
Event Name	Shows the name of the scheduled event.
Edit	Opens the Configure Scheduled Event window for the corresponding event. See Figure 2-27, "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.
Import Schedule	After selecting a file, imports a schedule.
Export Schedule	Exports the current schedule.
Reboot	Reboots the device, which must be rebooted for changes to the schedule to take effect.

Table 2-15. Schedules Configuration Parameters

Note The schedule is not affected by restoring the factory defaults.

Configure Scheduled Event					
Event # 1					
Days Sun M	Mon Tue Wed Thu Fri Sat				
	~ ~ ~ ~ ~				
Time	07:55				
Audio	Short_school_bell.wav •				
Times to Play	1				
PGROUP	00 •				
Name	warning bell				
Relay	No action No action				
	Activate during bell Activate indefinitely Deactivate indefinitely				
Save Change	-				

Figure 2-27. Configure Scheduled Event Window

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

Web Page Item	Description
Days	Use the box beneath the day to select. Multiple days may be selected.
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.
Name	Enter the name of the event, using a maximum of 16 characters.
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.
	Click the Save Changes button to save your configuration settings.
Save Changes	Note: You need to reboot for changes to take effect.
Cancel	Click the Cancel button to close the Configure Scheduled Event window.
	 In addition to the bells, the user can have 25 stored messages that are utilized through paging groups.
	 <u>www.cyberdata.net/bell-download/</u> has a set of sounds that can be downloaded.

Table 2-16. Configure Scheduled Event Window Parameters

• The user can upload 25 unique bell files that can be used in up to 250 scheduled events.

2.4.12 Configure the Fault Detection Parameters

1. Click on the Fault button to open the Fault page. See Figure 2-28.

Figure 2-28. Fault Page

U		ulu	v3.1 P	ugn	900	
Fault Detection	Settings					
Play Stored Audio Locally:						
Make call to extension:						
Dial Out Extension:	204					
Dial Out ID:	id204					
Repeat Message:	0					
Play Stored Audio via Multi	cast: 🗌					
Play Line-in Audio via Multi	cast: 🔲					
Multicast Address:						
Multicast Port:	0					
Save Reboot Togg	e Help					

- 2. On the **Fault Detection** page, enter values for the parameters indicated in Table 2-17.
- **Note** The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

Web Page Item	Description
Triggered Settings	
Play Stored Audio 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.
Repeat Message ?	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 Stored Audio via Multicast 🛜	When selected, the device will send the user-defined "sensor triggered" audio file to the specified multicast address and port. Cannot be combined with Play Line-in Audio via Multicast .
Play Line-in Audio via Multicast ?	When selected, the device will send line-in audio to the specified multicast address and port. Cannot be combined with Play Stored Audio via Multicast or Enable Line-in to Line-out Loopback .
Multicast Address ?	The multicast address used for fault detection audio.
Multicast Port ?	The multicast port used for fault detection audio.
Test Fault Detection	Click on the Test Fault Detection button to test the fault detection feature.
	Click the Save button to save your configuration settings.
Save	Note: You need to reboot for changes to take effect.
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.

Note You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

2.4.13 Configure the Audio Parameters

Click on the **Audiofiles** button to open the **Audiofiles** page. See Figure 2-29. 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.

Home Device Network SIP PGROUPS SSL Schedules Fault Audiofiles Events Autoprov Firmware				
CyberData v3.1 Paging Server				
CyberDala VJ.I Faying Server				
Available Space: 35.22MB				
Built-in Audio Files				
0: Currently set to default BrowseNo file chosen Play Delete Save				
1: Currently set to default BrowseNo file chosen Play Delete Save				
2: Currently set to default BrowseNo file chosen Play Delete Save				
3: Currently set to default Browse No file chosen Play Delete Save				
4: Currently set to default Browse No file chosen Play Delete Save				
5: Currently set to default Browse No file chosen Play Delete Save 6: Currently set to default				
7: Currently set to default				
BrowseNo file chosen Play Delete Save 8: Currently set to default				
Browse No file chosen Play Delete Save				
Browse				

Figure 2-29. Audiofiles Page

9:	Currently set to default	Browse No file chosen	Play Delete Save	
Dot:	Currently set to default			
		Browse No file chosen	Play Delete Save	
Audio Test:	Currently set to default	Browse No file chosen		n
Enter PGROUP:	Currently set to default	Browse No file chosen	Play Delete Save	
	and a second	Browse No file chosen	Play Delete Save	
Invalid PGROUP:	Currently set to default			
		Browse No file chosen	Play Delete Save	
Enter Code:	Currently set to default	Browse No file chosen	Play Delete Save	
Invalid Code:	Currently set to default			
		Browse No file chosen	Play Delete Save	
Page Tone:	Currently set to default	Browse No file chosen		
Your IP Address Is:	Currently set to default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
Rebooting:	Currently set to default			
Protection Profession		Browse No file chosen	Play Delete Save	
Restoring Default:	Currently set to default	Browse No file chosen	Play Delete Save	1
Sensor Triggered:	Currently set to default			
		Browse No file chosen	Play Delete Save	
Night Ring:	Currently set to default			
		Browse No file chosen	Play Delete Save	
Bells			Stored Message	es

Figure 2-30. Audiofiles Page

Enter Code	e: Currently set to default			
		Browse No file chosen	Play Delete Save	
Invalid Coo	de: Currently set to default	Personal and the second		
		Browse No file chosen	Play Delete Save	
Page Tone	: Currently set to default	Browse No file chosen		
Your ID Ad	dress Is: Currently set to default	Drowse No file chosen	Play Delete Save	
Tour IP Au	uress is. Currently set to default	Browse No file chosen	Play Delete Save	
Rebooting	: Currently set to default	No hie chosen		
	uoidait	Browse No file chosen	Play Delete Save	
Restoring	Default: Currently set to default	,,		
		Browse No file chosen	Play Delete Save	
Sensor Trig	ggered: Currently set to default			
		Browse No file chosen	Play Delete Save	
Night Ring	: Currently set to default			
		Browse No file chosen	Play Delete Save	
E	Bells		Stored Messages	
	_		_	
Gong.wav	Play	elete Tsunami_Warning	.wav Play	Delete
10_pulse_bell_tone.wav	Play D	elete Browse No	o file chosen Upload Stored Mes	sage
Dinner_Bell.wav	Play	elete		
Westminster_chimes.wav	Play	elete		
Browse No file chosen	Upload New Bell			

Figure 2-31. Audiofiles Page

Browse...

Browse...

On the Audiofiles page, enter values for the parameters indicated in Table 2-18.

- **Note** 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 Server 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.

Web Page Item	Description			
Audio Files				
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." (24 character limit).			
Audio Test	Corresponds to the message "This is the CyberData IP speaker test message" (24 character limit).			
Enter PGROUP	Corresponds to the message "Enter PGROUP" (24 character limit).			
Invalid PGROUP	Corresponds to the message "Invalid PGROUP" (24 character limit).			
Enter Code	Corresponds to the message "Enter Code" (24 character limit).			
Invalid Code	Corresponds to the message "Invalid Code" (24 character limit).			
Page Tone	Corresponds to a simple tone that is unused by default (24 character limit).			
Your IP Address is	Corresponds to the message "Your IP address is" (24 character limit).			
Rebooting	Corresponds to the spoken word "Rebooting" (24 character limit).			
Restoring Default	Corresponds to the message "Restoring default" (24 character limit).			
Sensor Triggered	Corresponds to the message "Sensor Triggered" (24 character limit).			
Night Ring	Specifies the ringtone for nightring. By default this parameter uses the same audio file that is selected for the Ring Tone parameter.			
Bells	Upload files here to be used in scheduled events.			
	Note: The user may upload a maximum of 25 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.			

Table 2-18. Audiofiles Configuration Parameters

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.

Table 2-18. Audiofiles Configuration Parameters (continued)

2.4.13.1 User-created Audio Files

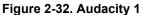
User-created audio files must be saved in one of the following formats:

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

Note 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 2-32 through Figure 2-34.





Tag Name	Tag Value	
Artist Name		
Track Title		
Album Title		
Track Number		
Year		
Genre		
Comments		
<u>A</u> dd	<u>R</u> emove <u>G</u> Template	<u>C</u> lear
E <u>d</u> it Rese <u>t</u>	Load Sa	set Default

Figure 2-33. Audacity 2

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

• WAV (Microsoft) signed 16 bit PCM.

🔒 💽 Export File			\odot	×			
Name: audiotest.	wav			٦			
Save in <u>f</u> older: Etmp							
<u></u>							
✓ Browse for other folders							
🛃/ tmp/			Create Fo <u>l</u> der	-)			
Places	Name		✓ Modified	1			
🍂 Search	🛅 cscope.4371		Yesterday at 14:30				
🛞 Recently Used	🛅 kde-na		Yesterday at 14:26				
🛅 na	🛅 kde-root		Yesterday at 14:26 09:20 Yesterday at 14:32				
🛅 Desktop	🛅 ksocket-na						
👩 File System	🛅 orbit-na						
👩 250.1 GB Media	ssh-CIPQVD3392		Yesterday at 14:26				
	► v814422		Yesterday at 15:45				
₽ Add ≈ <u>B</u> emove			WAV (Microsoft) signed 16 bit PCM 🗸)			
	<u>(</u>	<u>p</u> tions		1			
			⊘ Cancel Save				

Figure 2-34. WAV (Microsoft) signed 16 bit PCM

WAV (Microsoft) signed 16 bit PCM

2.4.14 Configure the Event Parameters

Click on the **Events** button to open the **Events** page (Figure 2-35). The **Events** page specifies a remote server that can be used to receive HTTP POST events when actions take place on the board.

Home	Device	Network	SIP	PGROUPS	SSL	Schedules	Fault	Audiofiles	Events	Autoprov	Firmware
	C	ybe	rD	ata	v3	.1 P	agi	ing	Ser	ver	
nable Event	Generation:										
						Ever	t Serve	er			
Events						Server I	P Address:	10.0.0.250			
nable Call St	art Events:					Server I	Port:	8080			
nable Call Te						Server l	JRL:	xmlparse_engine			
nable Relay											
nable Relay											
nable Night I	•										
nable Power											
nable Fault E nable 60 Sec											
heck All		al. Unche	ck All								
		Cheric									
Save Re	boot Tog	gle Help									
Care Like	iogę	jie neip									

Figure 2-35. Events Page

Table 2-19 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.

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 2.4.14.1, "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 SIF 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 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.					
Event Server						
Server IP Address ?	The IPv4 address of the event server in dotted decimal notation.					
Server Port <mark>?</mark>	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.					
Jncheck All	Click on Uncheck All to de-select all of the events on the page.					
	Click the Save button to save your configuration settings.					
Save	Note: You need to reboot for changes to take effect.					
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 2-19. Events Configuration

Note You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

2.4.14.1 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</event>
</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</event>
</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</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>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>
```

2.4.15 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 2-36.

Figure 2-36. Autoprovisioning Page

Home	Device	Network	SIP	PGROUPS	SSL	Schedules	Fault	Audiofiles	Events	Autoprov	Firmware
	C	ybe	rD	ata	v 3	.1 P	agi	ing S	Ser	ver	
							Ŭ	Ŭ			
Disable Auto	provisioning	:									
Autoprovisio	oning Server:										
	oning Filenam	ie:									
Use tftp: Verify Serve	Certificate										
Username:	Certificate										
Password:											
Autoprovisio	oning autoupo	date (in minutes): 0								
Autoprovisio	on at time (HH	IMMSS):									
Autoprovisio	on when idle (in minutes > 10): 0								
See the manı	ual to learn how	v to use autoprov	/isionina to	configure your de	evice.						
	ing happens o			j							
		a configured ser	ver addres	s and filename.							
					r in your list	of DHCP options a	and try to dow	nload '0020f70462	2a9.xml' and if	this fails, '000000	cd.xml'.
Save R	eboot Tog	igle Help									
Download 1	Template										
Autoprovic	ioning log										
Autoprovis											
	provisioning De prov found opt	evice ion 43 in DHCP s	server="htt	p://10.0.0.242"							
21:01 Auto	prov looking fo	r 0020f70462a9.	xml at http								
		arsing "0020f704 ion 72 in DHCP s		0.1.118"							
		r 0020f70462a9.									

- 2. On the **Autoprovisioning** page, you may enter values for the parameters indicated in Table 2-20.
- **Note** The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

Web Page Item	Description
Disable Autoprovisioning ?	Prevent the device from automatically trying to download a configuration file. See Section 2.4.15.1, "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, enable the Set Time with NTP Server on boot setting on the Device Page page (see Table 2-5).
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, enable the Set Time with NTP Server on boot setting on the Device Page page (see Table 2-5).
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, enable the Set Time with NTP Server on boot setting on the Device Page page (see Table 2-5).
	Click the Save button to save your configuration settings.
Save	Note: You need to reboot for changes to take effect.

Table 2-20. Autoprovisioning Configuration Parameters

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.				
Download Template	Press the Download Template button to create an autoprovisioning file for the device. See Section 2.4.15.3, "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).				

Table 2-20. Autoprovisioning Configuration Parameters (continued)

Note You must click on the **Save** button and then the **Reboot** button for the changes to take effect.

2.4.15.1 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 000000cd.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 2.4.15.2, "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 2-20). 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:

<mi< th=""><th>scSettings></th></mi<>	scSettings>
	<devicename>CyberData VoIP Intercom</devicename>
</td <td><autoprovfile>common.xml</autoprovfile>></td>	<autoprovfile>common.xml</autoprovfile> >
</td <td><autoprovfile>sip_reg[macaddress].xml</autoprovfile>></td>	<autoprovfile>sip_reg[macaddress].xml</autoprovfile> >
</td <td><autoprovfile>audio[macaddress]</autoprovfile>></td>	<autoprovfile>audio[macaddress]</autoprovfile> >
</td <td><autoprovfile>device[macaddress].xml</autoprovfile>></td>	<autoprovfile>device[macaddress].xml</autoprovfile> >
<td>liscSettings></td>	liscSettings>

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:

- 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 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.

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

Table 2-21. Autoprovisioning File Name

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/000000cd.xml

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

Autoprovisioning <FirmwareSettings>

```
Firmware Updates <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-uImage-[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>IndoorIntercom31SW</ProductString> <ProductString>IndoorIntercom31SW</ProductString> <ProductString>IndoorKeypad31</ProductString> <ProductString>OfficeRinger31</ProductString> <ProductString>OfficeRinger31SW</ProductString> <ProductString>OfficeRinger31SW</ProductString> <ProductString>OutdoorIntercom31SW</ProductString> <ProductString>OutdoorIntercom31</ProductString> <ProductString>OutdoorIntercom31</ProductString> <ProductString>OutdoorIntercom31SW</ProductString> <ProductString>OutdoorKeypad31</ProductString> <ProductString>OutdoorKeypad31SW</ProductString> <ProductString>Strobe31</ProductString> <ProductString>Strobe31</ProductString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString></productString> Autoprovisioning H Example 1

Dning 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>
</SIPSettings>
```

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.
- 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.
- 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

ng 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 **10.0.253**, and the port is set to **5060**.

XML 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)

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.

2.4.15.2 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
                                  10.0.0.252;
   option domain-name-servers
   option time-offset
                                   -8;
                                                   # 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; }

2.4.15.3 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 2-37). 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 2-37.

🕘 Openi	ng 0020f702bf18.xml	↑ □ ×
You have chosen to c	ppen:	
0020f702bf18 which is: XML do from: https://10.	ocument (11.3 KB)	
What should Firef	ox do with this file?	
Open with	Text Editor (default)	•
○ <u>S</u> ave File		
Do this <u>a</u> utor	natically for files like this from now or	ı.
	Cancel	Ж

Figure 2-37. 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.

2.5 Upgrading the Firmware



Caution

Equipment Hazard: Devices with a serial number that begins with 1461xxxxx can only run firmware versions 11.0.0 or later.

2.5.1 Upgrade the Firmware

To upload the firmware from your computer:

- 1. Retrieve the latest SIP Paging Server firmware from the SIP Paging Server **Downloads** page at: <u>https://www.cyberdata.net/products/011146</u>
- 2. Unzip the firmware version file. This file may contain the following:
- Firmware file
- Release notes
- 3. Log in to the SIP Paging Server home page as instructed in 2.4.4 "Log in to the Configuration GUI".

4. Click on the **Firmware** menu button to open the **Firmware** page. See Figure 2-38.



Caution

Equipment Hazard: CyberData strongly recommends that you first reboot the device before attempting to upgrade the firmware of the device. See Section 2.5.2, "Reboot the SIP Paging Server".

Figure 2-38. Firmware Page

Home	Device	Network	SIP	PGROUPS	SSL	Schedules	Fault	Audiofiles	Events	Autoprov	Firmware
	C	vbe	rD	ata	v3	.1 P	adi	ina s	Ser	ver	
		<i>y</i> c		utu			ag.			••••	
									_		
Current Firm	vare Version: \	v12.2.0			ecify a file: No file	chosen		Uploa	1		

- 5. Click on the Browse button, and then navigate to the location of the firmware file.
- 6. Select the firmware file.
- 7. Click on the Upload button.
- **Note** Do not reboot the device after clicking on the **Upload** button.
- **Note** This starts the upgrade process. Once the SIP Paging Server has uploaded the file, the **Uploading Firmware** countdown page appears, indicating that the firmware is being written to flash. The SIP Paging Server will automatically reboot when the upload is complete. When the countdown finishes, the **Firmware** page will refresh. The uploaded firmware filename should be displayed in the system configuration (indicating a successful upload and reboot).
- 8. Table 2-22 shows the web page items on the **Firmware** page.

Table	2-22.	Firmware	Parameters	

Web Page Item	Description				
Current Firmware Version	Shows the current firmware version.				
Browse	Use the Browse button to navigate to the location of the Intercom firmware file that you want to upload.				
Upload	Click on the Upload button to automatically upload the selected firmware and reboot the system.				

2.5.2 Reboot the SIP Paging Server

To reboot a SIP Paging Server, log in to the web page as instructed in Section 2.4.4, "Log in to the Configuration GUI".

1. Click Reboot (Figure 2-39). A normal restart will occur.

Figure 2-39. Home Page

	Cybe	rD	ata v	3.1 P	ani	na	Sor	vor	
	Cybe			J.T.F	ayı	ing .	JCI	VCI	
						_			
Current Sta	atus		Admin S	Settings		Impo	rt Settin	gs	
Serial Number:	146100001		Username:	admin		Brow	se No file d	hosen	
Mac Address:	00:20:f7:04:62:a9		Password:						
Firmware Version:	v12.2.0					Import	Config		
			Confirm Pass	word:			Connig		
P Addressing:	DHCP					Evne	rt Cottin	ac	
IP Address:	10.10.1.72					Expo	rt Settin	igs	
Subnet Mask:	255.0.0.0		Save Re	boot Toggle Help					
Default Gateway:						Export	Config		
DNS Server 1:	10.0.1.56								
DNS Server 2:									
SIP Mode:	Enabled								
Event Reporting:	Disabled								
Nightringer:	Disabled								
Primary SIP Server	: Not registered								
Backup Server 1:									
Backup Server 2:									
Nightringer Server	: Not registered								

Reboot

2.6 Command Interface

Some functions on the device can be activated using simple POST commands to the web interface. The examples in Table 2-23 use the free unix utility, **wget commands**. However, any program that can send HTTP POST commands to the device should work.

2.6.1 Command Interface Post Commands

These commands require an authenticated session (a valid username and password to work).

Device Action	HTTP Post Command wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi- bin/command.cgi"post-data "test_relay=yes" wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi- bin/command.cgi"post-data "terminate=yes"			
Trigger relay (fixed at 5 seconds)				
Terminate active call				
Force reboot	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi- bin/command.cgi"post-data "reboot=yes"			
Play "audio test message"	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi- bin/command.cgi"post-data "test_audio=yes"			
Announce IP address	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi- bin/command.cgi"post-data "speak_ip_address=yes"			
Play the "0" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_0=y			
Play the "1" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_1=yes"			
Play the "2" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_2=yes"			
Play the "3" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet - O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_3=yes"			
Play the "4" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_4=yes"			
Play the "5" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_5=yes"			

Table 2-23. Command Interface Post Commands^a

Device Action HTTP Post Command				
Play the "6" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_6=yes"			
Play the "7" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_7=yes"			
Play the "8" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_8=yes"			
Play the "9" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_9=yes"			
Play the "Dot" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_d=yes"			
Play the "Page Tone" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_pagetone=yes"			
Play the "Your IP Address Is" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_youripaddressis=yes"			
Play the "Rebooting" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_rebooting=yes"			
Play the "Restoring Default" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_restoringdefault=yes"			
Play the "Sensor Triggered" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_sensortriggered=yes"			
Play the "Night Ring" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_nightring=yes"			
Play the "Enter PGROUP" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_enterpgroup=yes"			
Play the "Invalid PGROUP" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_invalidpgroup=yes"			
Play the "Enter Code" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_entercode=yes"			

Table 2-23. Command Interface Post Commands^a (continued)

Device Action	HTTP Post Command			
Play the "Invalid Code" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "play_invalidcode=yes"			
Delete the "0" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_0=yes"			
Delete the "1" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_1=yes"			
Delete the "2" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_2=yes"			
Delete the "3" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_3=yes"			
Delete the "4" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certifica quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_4=yes"			
Delete the "5" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_5=yes"			
Delete the "6" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certific quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_6=yes"			
Delete the "7" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_7=yes"			
Delete the "8" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_8=yes"			
Delete the "9" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_9=yes"			
Delete the "Audio Test" audio file	file wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_audiotest=yes"			
Delete the "Page Tone" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_pagetone=yes"			

Table 2-23. Command Interface Post Commands^a (continued)

Device Action	HTTP Post Command		
Delete the "Your IP Address Is" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_youripaddressis=yes"		
Delete the "Rebooting" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_rebooting=yes"		
Delete the "Restoring Default" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate - quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_restoringdefault=yes"		
Delete the "Sensor Triggered" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_sensortriggered=yes"		
Delete the "Night Ring" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_nightring=yes"		
Delete the "Enter PGROUP" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_enterpgroupl=yes"		
Delete the "Invalid PGROUP" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_invalidpgroup=yes"		
Delete the "Enter Code" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate - quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_entercode=yes"		
Delete the "Invalid Code" audio file	wgetuser adminpassword adminauth-no-challengeno-check-certificate quiet -O /dev/null "https://10.0.3.71/cgi-bin/audiofiles.cgi"post-data "delete_invalidcode=yes"		

Table 2-23. Command Interface Post Commands^a (continued)

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

Appendix A: Setting Up a TFTP Server

A.1 Set up a TFTP Server

Autoprovisioning requires a TFTP server for hosting the configuration file.

A.1.1 In a LINUX Environment

To set up a TFTP server on LINUX:

- 1. Create a directory dedicated to the TFTP server, and move the files to be uploaded to that directory.
- Run the following command where /tftpboot/ is the path to the directory you created in Step 1: the directory that contains the files to be uploaded. For example:

in.tftpd -l -s /tftpboot/your_directory_name

A.1.2 In a Windows Environment

You can find several options online for setting up a Windows TFTP server. This example explains how to use the Solar Winds freeware TFTP server, which you can download at:

http://www.cyberdata.net/support/voip/solarwinds.html

To set up a TFTP server on Windows:

- 1. Install and start the software.
- 2. Select File/Configure/Security tab/Transmit Only.

Make a note of the default directory name, and then move the firmware files to be uploaded to that directory.

Appendix B: Troubleshooting/Technical Support

B.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/products/011146

B.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/products/011146

B.3 Contact Information

Contact CyberData Corporation 3 Justin Court Monterey, CA 93940 USA <u>www.CyberData.net</u> Phone: 800-CYBERDATA (800-292-3732) Fax: 831-373-4193

Sales Sales 831-373-2601, Extension 334

TechnicalThe fastest way to get technical support for your VoIP product is to submit a VoIP TechnicalSupportSupport form at the following website:

http://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

B.4 Warranty and RMA Information

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

http://support.cyberdata.net/

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