



SIP Paging Adapter Operations Guide

SIP Compliant Part #011233 Document Part #931763E for Firmware Version 20.2.0

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	Phone: (831) 373-2601, Ext. 333 Email: support@cyberdata.net Fax: (831) 373-4193 Company and product information is at www.cyberdata.net .

Revision Information

Revision 931763E, which corresponds to firmware version 20.2.0, was released on May 7, 2022, and has the following changes:

• Updates Table 2-3, "Factory Default Settings" with the following change:

IP Address: 192.168.1.23

Subnet Mask: 255.255.255.0

Default Gateway: 192.168.1.1

• Updates Section 2.6.4, "Log in to the Configuration GUI" with the following change:

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

Microsoft Edge (version: 42.17134.1.0)Pictorial Alert Icons

GENERAL ALERT	General Alert This pictorial 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 pictorial 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 Adapter, 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 011233.
- The serial number on the label should begin with 2332.

Figure 1-1. Model Number Label



Model number Serial number begins with **2332**

1.2 Product Features

- Compatible with most analog amplifiers in the market
- Supports audio prioritization, including 10 multicast paging groups
- Loud/Night Ringer function second SIP extension
- 9 user-uploadable page messages
- Supports delayed pages with call buffering
- Support for security code to prevent unwanted SIP calls
- Can receive pages directly from Poly phones as well as other devices that can send standard multicast
- Sense input capable of generating events or SIP calls
- Built-in diagnostics
- Line-in for background music
- Line-out connector
- Remote amp fault sensor
- Audio controlled relay/remote amplifier enable
- DTMF entries for analog paging zones
- Rack mountable
- TLS 1.2 (including mutual authentication) and SRTP-enhanced security for IP Endpoints in a local or cloud-based environment
- · HTTPS or HTTP web-based configuration HTTPS is enabled by default
- Autoprovisioning via HTTPS, HTTP or TFTP
- Configurable event generation for device health and status monitoring
- 802.11q VLAN tagging
- HTTP command interface
- Support for Cisco SRST resiliency

1.3 Product Specifications

Specifications	
Ethernet I/F	10/100 Mbps
Protocol	SIP RFC 3261 Compatible
Power Input	PoE 802.3af or 48VDC
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
Page Port Output	Balanced 600 Ohm 5VPP
Payload Types	G.711 a-law, G.711 µ-law, G.722, and G.729
Network Security	TLS 1.2, SRTP, HTTPS
Operating Range	Temperature: -40° C to 55° C (-40° F to 131° F)
	Humidity: 5-95%, non-condensing
Storage Temperature	-40° C to 70° C (-40° F to 158° F)
Storage Altitude	Up to 15,000 ft. (4573 m)
Dimensions ^a	6.11 inches [155.19 mm] Length
	4.05 inches [102.87 mm] Width
	1.15 inches [29.21 mm] Height
Weight	1.2 lbs. [.54 kg]
Boxed Weight	1.8 lbs. [.82 kg]
Compliance	CE: EMC Directive - Class A EN 55032 & EN 55024, LV Safety Directive -
	EN 62368-1; RoHS Compliant; FCC Part 15 Class A; Industry Canada
	ICES-3 Class A; IEEE 802.3 Compliant; TAA Compliant
Warranty	2 Years Limited
Part Number	011233

Table 1-1. Product 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 CE Statement

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

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

1.4.2 FCC Statement

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

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

1.4.3 Industry Canada (IC) Compliance Statement

Operation is subject to the following two conditions:

1. This device may not cause interference, and

2. This device must accept any interference, including interference that may cause undesired operations of the device.

ICES-3 Class A

2 Setting Up the SIP Paging Adapter

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

2.1 Parts List

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

Quantity	Part Name	Illustration
1	SIP Paging Adapter	And a second sec
1	Installation Quick Reference Placemat	<section-header><section-header><section-header><section-header><section-header><section-header><section-header></section-header></section-header></section-header></section-header></section-header></section-header></section-header>
1	Mounting Template (located on the last page of the <i>Installation Quick</i> <i>Reference</i>)	
1	Mounting Kit (part #121007*) which includes: (2) Knotting Anchors (2) #6 x 1.25" Self-Tapping Screws	

Table 2-1. Parts List

2.2 Typical Installation

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





2.3 Connecting the SIP Paging Adapter

Before you connect the SIP Paging Adapter, 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 Adapter.

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 2.6.10, "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 2-2. 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 2.6.10, "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 2.6.5, "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 Adapter.

2.3.6 Connect to the Power Source

To use PoE, plug a Cat 5 Ethernet cable from the SIP Paging Adapter **Ethernet** port to your network. As an alternative to PoE, you can plug one end of a +48V DC power supply into the SIP Paging Adapter, 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 SIP Paging Adapter **Ethernet** port. Plug the other end into your network.





2.3.8 Confirm that the SIP Paging Adapter is Up and Running

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



Figure 2-6. SIP Paging Adapter LEDs

2.3.8.1 Verify Network Activity

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

2.4 Announcing the IP Address

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

Note The IP address announcement can be heard if a speaker or amplified speaker is connected to the unit.



Figure 2-7. RTFM Switch

2.5 Restore the Factory Default Settings

The SIP Paging Adapter is delivered with factory set default values for the parameters in Table 2-3. Use the **RTFM** switch (see Figure 2-8) on the back of the unit to restore these parameters to the factory default settings.



Figure 2-8. RTFM Switch

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

Parameter	Factory Default Setting
IP Addressing	DHCP
IP Address ^a	192.168.1.23
Web Access Username	admin
Web Access Password	admin
Subnet Mask ^a	255.255.255.0
Default Gateway ^a	192.168.1.1

Table 2-3. 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 switch until after the indicator lights go off, and then release it.
- **Note** The "Restoring Defaults" announcement can be heard if a speaker or amplified speaker is connected to the unit.
- 3. The SIP Paging Adapter settings are restored to the factory defaults.

2.6 Configuring the SIP Paging Adapter

Use this section to configure the SIP Paging Adapter.

2.6.1 Gather the Required Configuration Information

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

2.6.1.1 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 Adapter parameters:

- IP Address
- Subnet Mask
- Default Gateway

2.6.1.2 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 25 alphanumeric characters long.

2.6.1.3 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

2.6.2 SIP Paging Adapter Web Page Navigation

Table 2-4 shows the navigation buttons that you will see on every SIP Paging Adapter 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.
SSL	Link to the SSL page.
Multicast	Link to the Multicast 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 2-4. Web Page Navigation

2.6.3 Using the Toggle Help Button

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

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

Figure 2-9. Toggle/Help Button

Toggle Help

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

Figure	2-10.	Toggle	Help	Button	and	Question	Marks
iguio		109910	11010	Datton	ana	Quootion	manco

Stored Net	igs						
Addressing Mode	Addressing Mode: O Static O DHCP						
hostname:	SipDevice03cab3	?					
IP Address:	10.10.10.10		Quality				
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 2-11.

	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	nark A short description of the				

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

Question mark

A short description of the web page item will appear

2.6.4 Log in to the Configuration GUI

- 1. Open your browser to the SIP Paging Adapter IP address.
- **Note** If the network does not have access to a DHCP server, the device will default to an IP address of 192.168.1.23.
- Note Make sure that the PC is on the same IP network as the SIP Paging Adapter.
- **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 following website address:

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

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 Adapter configuration *after* the initial configuration, log in using the IP address you assign to the device. Section 2.6.6, "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 Default Username and Password: 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 25 alphanumeric characters in the **Change Password**

3. Enter the new password again in the **Re-enter New Password** field.

Click Save Settings.

field. The Password is case-sensitive.

Figure 2-12. Home Page

Home	Device	Network	SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
	(Cvb	erD)ata	Pa	ain	g Ad	lant	er	
		- j				3	9			
Current	Status			Admin	Settings		lr	nport Set	tings	
Serial Number	•	233100001		Username:	admin			Browse No f	ile chosen	
Mac Address: Firmware Vers		0:20:f7:04:68:2b		Password:	••••					
Partition 2:		v20.2.0 v20.2.0		Confirm Pas	sword: •••••			Import Config		
Partition 3: Booting From:		v20.2.0 Dartition 2								
Boot From Ot				Save F	Reboot Toggle	Help	E	xport Set	tings	
IP Addressing:	г	ОНСР						Export Config		
IP Address:		10.10.1.81								
Subnet Mask:		255.0.0.0								
Default Gatewa DNS Server 1:		L0.0.0.1 L0.0.1.56								
DNS Server 2:	Ĩ									
SIP Mode:	E	Enabled								
Multicast Mode		Disabled								
Event Reportin Nightringer:	-	Disabled Disabled								
Primary SIP Se	erver: N	Not registered								
Backup Server		Not registered								
Backup Server Nightringer Se		Not registered Not registered								
Nightinger Se	IVCI.	vorregistered								

- 4. On the Home Page, review the setup details and navigation buttons described in Table 2-5
- **Note** The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

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).
P 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.
Multicast Mode	Shows the current status of the Multicast 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.
mport 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. Click Save.
Export Settings	
Export Config	Click Export to export the current configuration to a file.
Save	Click the Save button to save your configuration settings.

Table 2-5. Home Page Overview

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 for the changes to take effect.
	At this point you can:
	• Review the SIP Paging Adapter's Current Settings . Use the RTFM switch to restore the factory default settings. See Section 2.5, "Restore the Factory Default Settings".
	• Configure the device parameters. Click on the Device button and see Section 2.6.5, "Configure the Device Parameters".
	 Configure the network parameters. Click on the Network button and refer to Section 2.6.6, "Configure the Network Parameters" for instructions.
	 Configure the SIP parameters. Click on the SIP button and see Section 2.6.7, "Configure the SIP Parameters".
	 Configure the multicast parameters. Click on the Multicast button and see Section 2.6.9, "Configure the Multicast Parameters" for instructions.
	 Configure the fault detection parameters. Click on the Fault button and see Section 2.6.10, "Configure the Fault Detection Parameters".
	 Configure the audio parameters. Click on the Audiofiles button and see Section 2.6.11, "Configure the Audio Parameters" for instructions.
	• Configure the event parameters. Click on the Events button and see Section 2.6.12, "Configure the Event Parameters" for instructions.
	 Configure the autoprovisioning parameters. Click on the Autoprov button and see Section 2.6.13, "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 2.7, "Upgrade the Firmware" for instructions.

Table 2-5. Home Page Overview (continued)

2.6.5 Configure the Device Parameters

1. Click on the **Device** button to open the **Device** page. See Figure 2-13.

Figure	2-13.	Device	Page
--------	-------	--------	------

Home Device	Network	SIP SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
C	ybe	rData	a Pa	ain	a Aa	lan	ter	
	- j 10 01			9	9			
Line-in Settings				Relay S	ettings			
Enable Line-in to Line-out Loo				Activate Rela	y on Local Audio:			
Volume:	127							
Clock Settings				DTMF S	ettings			
Enable NTP: 🗹				DTMF Duratio			00	
NTP Server: north-america.po Timezone: America/Los Ang					Menus (Go straig figured DTMF for J			
Current Time:Mon, 17 Jan 2022	-			Analog Zone:			37456	
					Entry for Analog	Zone:		
				Require Secu Security Code				
Misc Settings								
Device Name:	strobe b2		1					
Beep on Init:								
Beep Before Page: Disable HTTPS (NOT recomme								
Disable HTTPS (NOT recomme	inucuj.							
Test Audio Test Relay								
Save Reboot Toggle H	Help							

- 2. On the **Device** page, you may enter values for the parameters indicated in Table 2-6.
- **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.
Volume ?	Volume setting for the line-in (0-127).
	Note: This volume feature is only available on devices with a serial number of 233204XXX or greater.
Clock Settings	
Enable NTP ?	Sync device's local time with the specified NTP Server.
NTP Server ?	Use this field to set the address (in IPv4 dotted decimal notation or as a canonical name) for the NTP Server. This field can accept canonical names of up to 64 characters in length.
Timezone	Enter the tz database string of your timezone.
	Examples:
	America/Los_Angeles
	America/New_York
	Europe/London
	America/Toronto
	See https://en.wikipedia.org/wiki/List of tz database time zones for a full list of valid strings.
Current Time	Displays the current time.
Misc Settings	
Device Name ?	Type the device name. Enter up to 25 characters.
Beep on Init ?	Device will play the user defined "pagetone" audio file when it boots.
Beep on Page <mark>?</mark>	Device will play the user defined "pagetone" audio file before playing a SIP page.
Disable HTTPS (NOT recommended) ?	Disables the encrypted connection to the webpage. We do not recommend disabling HTTPS for security reasons.
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.
DTMF Settings	
DTMF Duration ?	The duration, in milliseconds, of DTMF tones played out of the device's analog audio ports (1-65535).
Bypass DTMF Menus (Go straight to page) ?	When selected, the DTMF menu options are bypassed when a page is sent, and the device begins a live/buffered page no ability to send stored messages).

Table 2-6. Device Configuration Parameters

Web Page Item	Description					
Send pre-configured DTMF for Analog Zone ?	When selected, a pre-configured DTMF sequence is sent to activate a analog zone when Bypass DTMF Menus (Go straight to page) setting i enabled.					
Zone ?	Type the pre-configured DTMF sequence for the analog zone.					
Manual DTMF Entry for Analog Zone ?	When selected, the device will prompt the caller to enter a DTMF sequence to activate an analog zone before prompting the caller to press 1 through 9 to send a stored message or press 0 to page.					
	Note: The user must press the # key after entering the zone.					
Require Security Code ?	When selected, the user will be prompted to enter a Security Code (entered on the Device Page) before being able to execute a page when calling the device.					
Security Code ?	Type the security code in this field.					
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 Relay	Click on the Test Relay button to do a relay test.					
Save	Click the Save button to save your configuration settings.					
Reboot	Click on the Reboot button to reboot the system.					
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.					

Table 2-6. Device Configuration Parameters (continued)

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

2.6.6 Configure the Network Parameters

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

Home	Device Netwo	ork SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
	Cv	berĽ)ata	Pa	ain	g Ac	lapt	ter	
					9	9			
Stored Net	work Setting	<u>ys</u>			VLAN S	ettings			
Addressing Mode:	Static • DHCP				VLAN ID (0-40	95): 0			
	SipDevice04682b				VLAN Priority	and Street and St			
IP Address:	10.10.10.10								
Subnet Mask:	255.0.0.0								
Default Gateway:	10.0.0.1								
DNS Server 1:	10.0.0.1								
DNS Server 2:	10.0.0.1								
IP Address: 1		ıgs			Save	eboot Toggle He	elp		

Figure 2-14. Network Page

On the Network page, enter values for the parameters indicated in Table 2-7.

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 2.5, "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) <mark>?</mark>	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 2-7. Network Configuration Parameters

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

Table 2-7. 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 Adapter's **IP Address**: The SIP Paging Adapter 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.
2.6.7 Configure the SIP Parameters

The SIP parameters enable the SIP Paging Adapter to contact and register with the SIP server. Click on the **SIP** button to open the **SIP** page.

Home Device		re 2-15. SIP Page
Home Device	Network SIP SSL IV	unicast Faunt Audionies Events Autoprov Firmware
С	yberData	Paging Adapter
IP Settings		Nightringer Settings
nable SIP operation:	~	SIP Server:
legister with a SIP Server:	~	SIP Server.
Buffer SIP Calls:		SIP Over ID:
rimary SIP Server:	10.0.0.253	SIP Auth ID: SIP Auth Password:
rimary SIP User ID:	199	
rimary SIP Auth ID:	199	Re-registration Interval (in seconds): 360
rimary SIP Auth Password:		
e-registration Interval (in seco		Call Disconnection
io rogioti attori intervar (in ocoo		
ackup SIP Server 1:		Terminate Call after delay: 0
ackup SIP User ID:		
ackup SIP Auth ID:		Audio Codec Selection
ackup SIP Auth Password:		
Re-registration Interval (in seco	nds): 360	Codec: Auto Select
ackup SIP Server 2:		RTP Settings
ackup SIP User ID:		
ackup SIP Auth ID:		RTP Port (even): 10500
ackup SIP Auth Password:		Asymmetric RTP:
Re-registration Interval (in seco	nds): 360	Jitter Buffer: 50
		RTP Encryption (SRTP): Disabled
temote SIP Port:	5060	
ocal SIP Port:	5060	Save Reboot Toggle Help
IP Transport Protocol:	UDP V	
LS Version:	1.2 only (recommended)	
erify Server Certificate:		
outbound Proxy:		
Outbound Proxy Port:	0	
se Cisco SRST:		
isable rport Discovery:		
nregister on Boot:		
	10000	

On the SIP page, enter values for the parameters indicated in Table 2-8.

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

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 <mark>?</mark>	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 2.6.10, "Configure the Fault Detection Parameters").
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.
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.

Table 2-8. SIP Configuration Parameters

Web Page Item	Description
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.
	Note : Pressing the '#' key while recording a buffered SIP call will end the call and cancel the page before it is sent.
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 2-8. SIP Configuration Parameters (continued)

Web Page Item	Description
Nightringer Settings	
SIP Server ?	Enter the SIP server address as an IPv4 address in dotted decimal notation or a fully qualified domain name. This parameter also becomes the host portion of the SIP-URI for the device's Nightringer extension on the SIP server. This field can accept entries of up to 255 characters in length.
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.
Call Disconnection	
Terminate Call After Delay ?	Automatically terminate an active call after a given delay in seconds. A value of 0 will disable this function. Enter up to 8 digits.
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), G.722, or G.729. Otherwise, the device will perform codec negotiation using the default list of supported codecs.
Codec ?	Select desired codec (only one may be chosen).

Table 2-8. SIP Configuration Parameters (continued)

Web Page Item		Description
RTP Settings		
RTP Port (even) ?		Specify the port number used for the RTP stream after establishing a SIP call. This port number must be an even number and defaults to 10500. The supported range is 0-65536. Enter up to 5 digits.
Asymmetric RTP ?		Specify if the remote endpoint will send and receive RTP packets on different ports. If set to false, the device will track the address/port that is sending RTP packets during a SIP call. If the address/port changes mid-stream, the device will disregard the SDP and send all further RTP packets to this new address.
		If set to true, this device will ignore the sending address/port and send RTP as specified in the SDP. Warning! Enabling asymmetric RTP can cause the RTP stream to be lost.
		Most installations should not enable asymmetric RTP.
Jitter Buffer ?		Specify the size of the jitter buffer (in milliseconds) used for SIP calls. Valid values are 50-1000.
RTP Encryption (SF	RTP) ?	When enabled, a SIP call's audio streams are encrypted using SRTP.
Save		Click the Save button to save your configuration settings.
Reboot		Click on the Reboot button to reboot the system.
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 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. Er	nter the IP address of the SIP Server.
	2. Er	nter the port numbers used for SIP signaling:
	a	. Remote SIP Port
	b	. Local SIP Port
	3. Er	nter the SIP registration parameters:
	a	. SIP User ID
	b	. Authenticate ID
	C	. Authenticate Password
	4. Fc	or SIP Registration, designate whether you want the device to register with your SIP server.
	5. At	Unregister on Reboot:
		 Select Yes to automatically unregister the SIP Paging Adapter when you reboot it. Select No to keep the SIP Paging Adapter registered when you reboot it.

Table 2-8. SIP Configuration Parameters (continued)

- 6. In the **Register Expiration** field, enter the number of seconds the SIP Paging Adapter registration lease remains valid with the SIP Server. The SIP Paging Adapter automatically reregisters with the SIP server before the lease expiration timeout.
- Note You must click on the Save button for the changes to take effect.

2.6.7.1 Point-to-Point Configuration

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

- 1. On the SIP page (Figure 2-16), make sure of the following:
 - The Register with a SIP Server parameter is not selected.
 - The Enable SIP Operation parameter is selected
- 2. Click on the **Save** button to save the changes.
- 3. Click on the **Reboot** button to reboot the device.
- 4. Enter the device's IP address as a "speed dial" (also called "auto-dial") key on the phone(s) from which you want to page.
- **Note** Establishing point-to-point SIP calls may not work with all phones.

Home Device	Network SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
C	yberl	Data	a Pa	ain	αΔn	lant	er	
Ŭ	yberi	Juli		giii	y At	ap		
SIP Settings				Nightrin	ger Setting	IS		
Enable SIP operation:	2			OIT OCTVOL.				
Register with a SIP Server:	.			SIP User ID:				
Buffer SIP Calls:	0			SIP Auth ID:				
Primary SIP Server:	.0.0.253			SIP Auth Pas	sword:			-
Primary SIP User ID:	.99			Re-registratio	n Interval (in seco	1ds): 360		
Primary SIP Auth ID:	199							
Primary SIP Auth Password:								
Re-registration Interval (in secon	ds): 360			Call Dis	connection			
				Terminate Ca	ll after de ay: 0			
Backup SIP Server 1:								
ackup SIP Server 1:								

Figure 2-16. SIP Page

2.6.7.2 Point-to-Point Fault Sense Reporting

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

- 1. On the SIP page (Figure 2-17), make sure of the following:
 - The Register with a SIP Server parameter is not selected.
 - The Enable SIP Operation parameter is selected

Figure 2-17. SIP Pag	е
----------------------	---

Home Device N	etwork SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
C	/berD)ata	a Pa	gin	g Ao	lap	ter	
SIP Settings				Nightrin	ger Setting	gs		
Enable SIP operation: Register with a SIP Server: Buffer SIP Calls:	·			SIP User ID: SIP Auth ID:				
Primary SIP Server: Primary SIP User ID:	.0.0.0.253 .99			SIP Auth Pass		anda): 200		
Primary SIP Auth ID: Primary SIP Auth Password:	.99			Refregistratio	n Interval (in seco	inus). 300		
Primary SIP Auth Password: Re-registration Interval (in seconds				Call Dis	connectior	า		

Register with a SIP Server is not selected

Enable SIP Operation is selected

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

- 4. On the **Fault** page (Figure 2-18) in the **Dial Out Extension** field, enter the IP address of the phone that is to be called when a fault is detected at the Fault Sense Input.
- **Note** Establishing point-to-point SIP calls may not work with all phones.

_							1				
Hon	ne	Device	Network	SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
										1	
		C	Sybe	erL	Jata	a Pa	gin	g Ac	ap	ier	
Faul	t Dete	ction S	ettings								
Play Au	dio Locall	y: 🗆									
Make Ca	all to Exter	nsion: 🗹									
Dial Out	Extensio	n: 10.10	0.0.93	_							
Dial Out	t ID:	Deteo	cted Fault								
Play Me	ssage:	2									
Save	Reboo	t Toggle	Liele								
Save	Rebuu	loggie	Help								
		_									
Test F	ault Senso	r									

Figure 2-18. Fault Page

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

2.6.8 Configure the SSL Parameters

1. Click SSL menu button to open the SSL page (Figure 2-19 and Figure 2-20).

Figure 2-19. SSL Configuration Page

Home Device Network	SIP SSL Multicast	Fault Audiofiles	Events	Autoprov	Firmware
Cybe Neb Server Certificate	erData Paq		dapt Autoprovision		tificate
<pre>subject= countryName = US stateOrProvinceName = Califor localityName = Montere organizationName = OyDerda commonName = 002007C notBefore=Mar 2 19:55:25 2021 GMT notAfter=Feb 28 19:55:25 2031 GMT</pre>	ey localityName ata organizationName	= US = California = Monterey = Cyberdata = 002017049500 2021 GMT	subject= countryNam stateOrPro localityNa organizati commonName notBefore=Mar	e vinceName me onName	= US = California = Monterey = Cyberdata = 0020f7049506 1 GMT
Browse No file chosen Import Web Certificate Restore Web Certificate	Browse No file chosen Import SIP Certificate Restore SIP Certificate Optional Password:		Import Autoprovis	file chosen sioning Certificate risioning Certificate d:	
Cyberdata CA Save Reboot Toggle H		Connection Test Autoprov	Connection		
	List of Trust	ed CAs			
1 CyberData_CA.pem	No file chosen Import CA C	Certificate Remove All	Restore Defaults	Rem	ove
2 DST_Root_CA_X3.crt			Info	Rem	ove
3 DigiCert_Assured_ID_Root_CA.crt			Info	Rem	ove
4 DigiCert_Assured_ID_Root_G2.crt			Info	Rem	ove

6	DigiCert_Global_Root_CA.crt	Info	Remove
7	DigiCert_Global_Root_G2.crt	Info	Remove
В	DigiCert_Global_Root_G3.crt	Info	Remove
9	DigiCert_High_Assurance_EV_Root_CA.crt	Info	Remove
10	DigiCert_Trusted_Root_G4.crt	Info	Remove
11	GeoTrust_Global_CA.crt	Info	Remove
12	GeoTrust_Primary_Certification_Authority.crt	Info	Remove
13	GeoTrust_Primary_Certification_AuthorityG2.crt	Info	Remove
14	GeoTrust_Primary_Certification_AuthorityG3.crt	Info	Remove
15	GeoTrust_Universal_CA.crt	Info	Remove
16	GeoTrust_Universal_CA_2.crt	Info	Remove
17	Go_Daddy_Class_2_CA.pem	Info	Remove
18	Go_Daddy_Root_Certificate_AuthorityG2.pem	Info	Remove
19	VeriSign_Class_3_Public_Primary_Certification_AuthorityG4.crt	Info	Remove
20	VeriSign_Class_3_Public_Primary_Certification_AuthorityG5.crt	Info	Remove
21	VeriSign_Universal_Root_Certification_Authority.crt	Info	Remove
22	Verisign_Class_1_Public_Primary_Certification_Authority.crt	Info	Remove
23	Verisign_Class_1_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
24	Verisign_Class_2_Public_Primary_Certification_AuthorityG2.crt	Info	Remove
25	Verisign_Class_2_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
26	Verisign_Class_3_Public_Primary_Certification_Authority.crt	Info	Remove
27	Verisign_Class_3_Public_Primary_Certification_AuthorityG3.crt	Info	Remove
28	thawte_Primary_Root_CA.crt	Info	Remove
29	thawte_Primary_Root_CAG2.crt	Info	Remove
30	thawte_Primary_Root_CAG3.crt	Info	Remove

Figure 2-20. SSL Configuration Page

Figure 2-21.

- 2. On the SSL page, enter values for the parameters indicated in Table 2-9.
- **Note** The question mark icon (?) in the following table shows which web page items will be defined after the **Toggle Help** button is pressed.

Tab	le 2-9. SSL Configuration Parameters
Web Page Item	Description
Web Server Certificate	Certificate used by the web server.
Browse	Click Browse to select a certificate to import.
Import Web Certificate	After selecting a certificate, click Import Web Certificate to import it as the certificate used by this device's web server.
Restore Web Certificate	Restore the device's default web server certificate. This will remove the user- uploaded Web Server Certificate.(Server CAs and Trusted CAs are unaffected).
SIP Client Certificate	When doing mutual authentication this device will present a client certificate with these parameters.
Browse	Click Browse to select a certificate to import.
Import SIP Certificate	After selecting a certificate, click Import SIP Certificate to import it as the certificate used by the device during SIP transactions.
Restore SIP Certificate	Restore the device's default sip client certificate. This will remove any user- uploaded sip client certificates (Server CAs and Trusted CAs are unaffected).
Optional Password	Enter the optional password for the SIP certificate's private key.
	Note : When using a password, it must be entered and saved before importing the certificate.
Autoprovisioning Client Certificate	When doing mutual authentication this device will present a client certificate with these parameters.
Browse	Click Browse to select a certificate to import.
Import Autoprovisioning Certificate	After selecting a certificate, click Import Autoprovisioning Certificate to import it as this device's certificate. This certificate will be used when requesting files during autoprovisioning.
Restore Autoprovisioning Certificate	Restore the device's default autoprovisioning certificate. This will remove any user-uploaded autoprovisioning certificates. (Server CAs and Trusted CAs are unaffected).
Optional Password ?	Enter the optional password for the Autoprovisioning certificate's private key.
	Note : When using a password, it must be entered and saved before importing the certificate.
Cyberdata CA 🛜	Right click and Save Link As to get the Cyberdata CA used to sign this client certificate.
Save	Click the Save button to save your configuration settings.

Table 2-9. SSL 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.
Test TLS Connection	
Server	The ssl test server address as a fully qualified domain name or in IPv4 dotted decimal notation.
Port	The supported range is 0-65536. SIP connections over TLS to port 5060 are modified to connect to port 5061. This test button will do the same.
Test SIP Connection	Use this button to test a TLS connection to a remote server using the sip client key and password. This will attempt to make a socket connection to the configured test server and port and report the success or failure. This can be used to debug TLS connection issues separate from SIP registration issues.
Test Autoprov Connection	Use this button to test a TLS connection to a remote server using the autoprovisioning client key and password. This will attempt to make a socket connection to the configured test server and port and report the success or failure. This can be used to debug TLS connection issues with secure autoprovisioning.
List of Trusted CAs	
Browse	Use this button to select a configuration file to import.
Import CA Certificate	Click Browse to select a CA certificate to import. After selecting a server certificate authority (CA), click Import CA Certificate to import it to the list of trusted CAs. CAs are used to validate the certificate presented by the server when establishing a TLS connection.
Restore Defaults	Restore Defaults will restore the default list of registered CAs and Remove All will remove all registered CAs.
Remove All	Restore Defaults will restore the default list of registered CAs and Remove All will remove all registered CAs.
Info	Provides details of the certificate. After clicking on this button, the Certificate Info Window appears. See Section 2.6.8.1, "Certificate Info Window".
Remove	Removes this certificate from the list of trusted certificates. After clicking on this button, the Remove Server Certificate Window appears. See Section 2.6.8.2, "Remove Server Certificate Window".

Table 2-9. SSL Configuration Parameters (continued)

2.6.8.1 Certificate Info Window

The **Certificate Info Window** provides details of the certificate. This window appears after clicking on the **Info** button. See Figure 2-22.

Figure 2-22. Certificate Info Window

Cer	rtificate Info	
organizationalUnitName	= ACCV = ES 11 GMT	
		ОК

2.6.8.2 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 2-23.



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

2.6.9 Configure the Multicast Parameters

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

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

1. Click on the **Multicast** button to open the **Multicast** page. See Figure 2-24.

Figure 2-24. Multicast Page

Home	Device	Network	SIP	SSL	Multicast	Fault	Audiofile	es	Events	Autoprov	Firmware	1
	(Cybe	er[Data	a Pa	adin	a A	\d	apt	ter		
		- J				.9	5.					
					A	Cattinger						
						t Settings						
					Enable Multica	ast Operation: 🗆						
			Priority A	ddress	Port	Name	В	ufferBe	ер			
			3 3 3 3 3	39.168.3.1	2000	Background Mus	ic					
				39.168.3.2	3000	MG1						
				39.168.3.3 39.168.3.4	4000 5000	MG2 MG3						
				39.168.3.5	6000	MG4						
				39.168.3.6	7000	MG5						
			6 2	39.168.3.7	8000	MG6)			
			7 2	39.168.3.8	9000	MG7)			
			8 2	39.168.3.9	10000	MG8			1			
			9 2	39.168.3.10	11000	Emergency		L				
Polycom Default Channel 1 Polycom Priority Channel 24 Polycom Emergency Channel 25												
				SI	P calls are con	sidered priority 4.	5					
				Po	ort range can b	e from 2000-6553	5					
				Priori	ty 9 is the high	est and 0 is the lo	west					
				A higher priority	audio stream w	vill always superse	de a lower on	е				
					Save	Reboot						

- 2. On the Multicast 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
Enable Multicast Operation	Enables or disables multicast operation.
Priority	Indicates the priority for the multicast group. Priority 9 is the highest (emergency streams). 0 is the lowest (background music). SIP calls are considered priority 4.5 See Section 2.6.9.1, "Assigning Priority" for more details.
Address	Enter the multicast IP Address for this multicast group (15 character limit).
Port	Enter the port number for this multicast group (5 character limit [range can be from 2000 to 65535]).
	Note : The multicast ports have to be even values. The webpage will enforce this restriction.
Name	Assign a descriptive name for this multicast group (25 character limit).
Buffer	Device will buffer up to four minutes of audio and then play back the recording after the multicast stream finishes or after the buffer is full.
Веер	When selected, the device will play a beep before multicast audio is sent.
Polycom Default Channel	When a default Polycom channel/group number is selected, the SIP Paging Adapter will subscribe to the default channel for one-way group pages. Group Numbers 1-25 are supported. Or, select Disabled to disable this channel.
Polycom Priority Channel	When a priority Polycom channel/group number is selected, the SIP Paging Adapter will subscribe to the priority channel for one-way group pages. Group Numbers 1-25 are supported. Or, select Disabled to disable this channel.
Polycom Emergency Channel	When an emergency Polycom channel/group number is selected, the SIP Paging Adapter will subscribe to the default channel for one-way group pages. Group Numbers 1-25 are supported. Or, select Disabled to disable this channel.
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.

Table 2-10. Multicast Configuration Parameters

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

2.6.9.1 Assigning Priority

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

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

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

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

2.6.9.2 Polycom Paging

Page your entire paging infrastructure, including legacy analog paging systems, using Polycom IP phones and a CyberData SIP Paging Adapter. Simultaneously paging your IP phones and overhead speakers can be as simple as pressing the Paging soft key on a Polycom IP phone.

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 SIP Paging Adapter.

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 SIP Paging Adapter. 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 SIP Paging Adapter can subscribe to Group Numbers 1 through 25 for Group Paging one-way audio pages. You may configure up to three group numbers or "channels", which are labeled **Polycom Default Channel**, **Polycom Priority Channel**, and **Polycom Emergency Channel** on the **Multicast Page**. Each of the three available channels can be disabled.

It is important to note the SIP Paging Adapter assigns a priority to each multicast group, as referenced in Section 2.6.9.2, "Polycom Paging". Polycom priority assignments by channel are ignored.

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 SIP Paging Adapter supports Polycom Group Paging multicasts that are G.711Mu encoded with a payload size of 20 ms.

Use the following steps to configure Polycom Group Paging on the SIP Paging Adapter:

1. Identify the Polycom multicast IP address and port number used by the Polycom phones.

- 2. Check the box to Enable Multicast Operation on the Multicast Page.
- 3. Choose a priority group and enter the Polycom IP address and port number into the **Priority**, **Address**, and **Port** fields on the **Multicast Page**.
- 4. Select up to three channel/group numbers for Group Paging subscriptions at the bottom of the Multicast Page.
- 5. Save and reboot to store changes.

2.6.10 Configure the Fault Detection Parameters

1. Click on the **Fault** button to open the **Fault** page. See Figure 2-25.

Figure 2-25. Fault Page

Home	Device	Network	SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
	6	2vh	orD	hat		ain		lant	tor	
	, c	~yD	erd	al	a ra	ym	g Ac	Japi	lei	
Fault De	etection S	Settings								
Play Audio Lo										
Make Call to E Dial Out Exter										
Dial Out ID:	id204									
Repeat Messa										
Save Re	boot Toggle	Help								
Test Fault Se	ensor									

- 2. On the **Fault Detection** 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
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.
Test Fault Detection	Click on the Test Fault Detection button to test the fault detection feature.
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) 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 2-11. Fault Detection Configuration Parameters

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

2.6.11 Configure the Audio Parameters

Click on the **Audiofiles** button to open the **Audiofiles** page. See Figure 2-26. 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 SSL N	Aulticast Fault	Audiofiles	Events	Autoprov Firmware
CyberData	Pagin	g Ac	lapte	er
	ored Messages			
Stored Message 1:Currently set to: default	No file chosen	Play Delete	Save Repeat: 0	Infinite:
Stored Message 2:Currently set to: default Browse Stored Message 3:Currently set to: default	No file chosen	Play Delete	Save Repeat: 0	Infinite:
Stored Message 4:Currently set to: default	No file chosen No file chosen	Play Delete Play Delete	Save Repeat: 0	Infinite:
Stored Message 5:Currently set to: default Browse	No file chosen	Play Delete	Save Repeat: 0	Infinite:
Stored Message 6:Currently set to: default Browse Stored Message 7:Currently set to: default	No file chosen	Play Delete	Save Repeat: 0	Infinite: 🗆
Stored Message 8:Currently set to: default	No file chosen No file chosen	Play Delete Play Delete	Save Repeat: 0 Save Repeat: 0	Infinite:
Stored Message 9:Currently set to: default Browse	No file chosen	Play Delete	Save Repeat: 0	Infinite:
	Audio Files			
0: Currently set to: default	Browse No file chos	sen Pl	lay Delete Sa	ve
Currently set to: default Currently set to: default	Browse No file chos	sen Pl	lay Delete Sa	ve
3: Currently set to: default	Browse No file chose Browse No file chose Browse		lay Delete Sa lay Delete Sa	

Figure 2-26. Audiofiles Page

4:	Currently set to: default	Browse No file chosen	Play Delete Save	
5:	Currently set to: default			
6:	Currently set to: default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
7:	Currently set to: default	Browse No file chosen	Play Delete Save	
8:	Currently set to: default			
9:	Currently set to: default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
Dot:	Currently set to: default	Browse No file chosen	Play Delete Save	
Audio Test:	Currently set to: explosion	.wav		
Page Tone:	Currently set to: default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
Your IP Add	ress Is:Currently set to: default	Browse No file chosen	Play Delete Save	
Rebooting:	Currently set to: default			
Restoring D	efault: Currently set to: default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
Ringback To	one: Currently set to: default	Browse No file chosen	Play Delete Save	
Ring Tone:	Currently set to: default			
Night Ring:	Currently set to: default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
Sensor Trig	gered: Currently set to: default	Browse No file chosen	Play Delete Save	
		Menu Audio Files		
Cancel:	Currently set to:defau	Ilt Browse No file chosen	Play Delete Save	
Currently	Playing: Currently set to:defau	ult		
Invalid Er	try: Currently set to:defau	Browse No file chosen	Play Delete Save	
	our			

Figure 2-27. Audiofiles Page

-				
Restoring De	efault: Currently set to: default			
Ringback To	ne: Currently set to: default	Browse No file chosen	Play Delete Save	
Ring Tone:	Currently set to: default	Browse No file chosen	Play Delete Save	
Night Ring:	Currently set to: default	Browse No file chosen	Play Delete Save	
Sensor Trigg	ered: Currently set to: default	Browse No file chosen	Play Delete Save	
		Browse No file chosen	Play Delete Save	
	l	Menu Audio Files		
Cancel:	Currently set to:defaul	t		
Currently	Playing: Currently set to:defaul	Browse No file chosen	Play Delete Save	
Invalid En	try: Currently set to:defaul	Browse No file chosen	Play Delete Save	
Page:	Currently set to:defaul	Browse No file chosen	Play Delete Save	
Play Store	d Message:Currently set to:defaul	Browse No file chosen	Play Delete Save	
Pound (#):	Currently set to:defaul	Browse No file chosen t	Play Delete Save	
Press:	Currently set to:defaul	Browse No file chosen	Play Delete Save	
Stored Me	ssage: Currently set to:defaul	Browse No file chosen	Play Delete Save	
Through:	Currently set to:defaul		Play Delete Save	
То:	Currently set to:defaul		Play Delete Save	
Enter Cod	e: Currently set to:defaul		Play Delete Save	
Invalid Co	de: Currently set to:defaul		Play Delete Save	
Enter Zon	e: Currently set to:defaul		Play Delete Save	
		Browse No file chosen	Play Delete Save	

Figure 2-28. Audiofiles Page

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

- **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 Adapter 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
Stored Messages	
Stored Message 1	Stored Message 1 corresponds to the message played after pressing 1 on a phone keypad.
through 9	Stored Message 2 corresponds to the message played after pressing 2 on a phone keypad.
	Stored Message 3 corresponds to the message played after pressing 3 on a phone keypad.
	Stored Message 4 corresponds to the message played after pressing 4 on a phone keypad.
	Stored Message 5 corresponds to the message played after pressing 5 on a phone keypad.
	Stored Message 6 corresponds to the message played after pressing 6 on a phone keypad.
	Stored Message 7 corresponds to the message played after pressing 7 on a phone keypad.
	Stored Message 8 corresponds to the message played after pressing 8 on a phone keypad.
	Stored Message 9 corresponds to the message played after pressing 9 on a phone keypad.
Repeat	Type the number of times that you want the specific Stored Message to repeat. A value of 0 means the message will play once (no repeat). A value of 1 means the message will play twice (one repeat)
Infinite	When selected, the specific Stored Message will repeat indefinitely after pressing the specific number key on a phone keypad.
	Note: The repeatedly playing audio can be canceled by calling, selecting the message, and pressing the # key.
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).
	Corresponde to the measure "This is the CyberDate IP appalent test measure," (24 abarrator limit)
Audio Test	Corresponds to the message "This is the CyberData IP speaker test message" (24 character limit)

Table 2-12. Audiofiles Configuration Parameters

Web Page Item	Description
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.
Menu Audio Files	Menu Audio Files are user-uploadable messages that create the audio menu played to the caller.
Cancel	Corresponds to the word "Cancel" used in the audio menu played to the caller. (24 character limit).
Currently Playing	Corresponds to the words "Currently Playing" used in the audio menu played to the caller. (24 character limit).
Fault Detection Message	Corresponds to the words "Fault Detection Message" used in the audio menu played to the caller. (24 character limit).
Invalid Entry	Corresponds to the words "Invalid Entry" used in the audio menu played to the caller. (24 character limit).
Page	Corresponds to the word "Page" used in the audio menu played to the caller. (24 character limit).
Play Stored Message	Corresponds to the words "Play Stored Message" used in the audio menu played to the caller. (24 character limit).
Pound (#)	Corresponds to whatever word or phrase the user wishes to call the pound key in the audio menu played to the caller (24 character limit).
Press	Corresponds to the word "Press" used in the audio menu played to the caller. (24 character limit).
Stored Message	Corresponds to the words "Stored Message" used in the audio menu played to the caller. (24 character limit).
Through	Corresponds to the word "Through" used in the audio menu played to the caller. (24 character limit)
То	Corresponds to the word "To" used in the audio menu played to the caller. (24 character limit).
Enter Zone	Corresponds to the words "Enter Zone" used in the audio menu played to the caller. (24 character limit).
Browse	The Browse button will allow you to navigate to and select an audio file.
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.
Save	Click the Save button to save your configuration settings.

Table 2-12. Audiofiles Configuration Parameters (continued)

Web Page Item	Description
Reboot	Click on the Reboot button to reboot the system.

Table 2-12. Audiofiles Configuration Parameters (continued)

2.6.11.1 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 2-29 through Figure 2-31.



Figure 2-29. Audacity 1

Tag Name	Tag Value	
Artist Name		
Track Title		
Album Title		
Track Number		
Year		
Genre		
Comments		
Add	Bemove	<u>C</u> lear
Genres	Template	
E <u>d</u> it Rese <u>t</u>	Load	Save S <u>e</u> t Default

Figure 2-30. Audacity 2

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

• WAV (Microsoft) signed 16 bit PCM.

🔒 💽 Export File		\odot \odot \otimes
Name: audiote	st.wav	
Save in <u>f</u> older:		*)
✓ Browse for other folde	rs	
🔯/ tmp/		Create Fo <u>l</u> der
Places	Name	✓ Modified
🆚 Search	🛅 cscope.4371	Yesterday at 14:30
🛞 Recently Used	🛅 kde-na	Yesterday at 14:26
🛅 na	🛅 kde-root	Yesterday at 14:26
🛅 Desktop	🛅 ksocket-na	09:20
🐻 File System	🛅 orbit-na	Yesterday at 14:32
👩 250.1 GB Media	ssh-CIPQVD3392	Yesterday at 14:26
	▶ v814422	Yesterday at 15:45
		\$
<u> </u>	re	WAV (Microsoft) signed 16 bit PCM 🗸
	Q	otions
		<u>⊘ cancel</u> <u>Save</u>

Figure 2-31. WAV (Microsoft) signed 16 bit PCM

WAV (Microsoft) signed 16 bit PCM

2.6.12 Configure the Event Parameters

Click on the **Events** button to open the **Events** page (Figure 2-32). The **Events** page specifies a remote server that can be used to receive HTTP POST events when actions take place on the board.

С	yb	erD	ata	a Pa	ging	g Ac	lapt	ter	
Enable Event Generation:									
Events					Event Ser	ver			
					Server IP Addres	s: 10.0.0.250			
Enable Call Start Events:					Server Port:	8080			
Enable Call Terminated Events:					Server URL:	xmlparse_eng	ine		
Enable Relay Activated Events:									
Enable Relay Deactivated Event									
Enable Night Ring Events:									
Enable Multicast Start Events:									
Enable Multicast Stop Events: Enable Power On Events:									
Enable Power On Events: Enable Fault Events:									
Enable Fault Events:									
Enable 60 Second Heartbeat:									
inable of Second HeartDeat.									

Figure 2-32. Events Page

Table 2-13 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 2.6.12.1, "Example Packets for Events" for sample packets.					
Events						
Enable Call Start Events 🛜	When selected, the device will report the start of a SIP call.					
Enable Call Terminated Events ?	When selected, the device will report the end of a SIP call.					
Enable Relay Activated Events ?	When selected, the device will report relay activation.					
Enable Relay Deactivated Events ?	When selected, the device will report relay deactivation.					
Enable Night Ring Events 🛜	When selected, the device will report when it starts ringing upon an incoming SIP cal 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 🛜	Specify the event server port number. The supported range is 0-65536. Enter up to 5 digits.					
Server URL 🛜	Generally, the destination URL is the name of the application that receives the events and the string in the HTTP POST command. It can be a script used to parse and process the HTTP POST events. Enter up to 127 characters.					
Check All	Click on Check All to select all of the events on the page.					
Uncheck All	Click on Uncheck All to de-select all of the events on the page.					
Save	Click the Save button to save your configuration settings.					
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 2-13. Events Configuration

button if you are specifying a server.

2.6.12.1 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>
```

2.6.13 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 2-33.

Figure 2-33. Autoprovisioning Page

				αια	ιга	gin	g Ac	lap		
						5	5			
Enable Autop		Z								
Autoprovisio	and the second	_								
	ning Filename:									
Use tftp: Verify Server	Certificate									
Username:	Certificate	i i i i i i i i i i i i i i i i i i i								
Password:		_								
Autoprovisio	ning autoupdate	(in minutes): 0								
	n at time (HHMM									
	n when idle (in m									
See the manu	al to learn how to	use autoprovisior	ning to configu	ire your devid	e.					
Autoprovision	ing happens on bo	oot.								
The device wi	ll first look for a co	nfigured server a	ddress and fil	ename.						
If these haven	't been configured	l, it will look for ar	n autoprovisio	ning server in	your list of DHCP	options and try	y to download '0020i	f704682b.xml' an	d if this fails, '0000	00cd.xml'.
Save R	eboot Toggle	Help								
Save	eboot Toggle	Help								
Save	eboot Toggle	Help								
Save R Download T		Help								
		Help								
	ïemplate	Help								
Download T	ïemplate	Help								
Download T	ïemplate	Help								
Download T	ïemplate	Help								
Download T	ïemplate	Help								
Download T	ïemplate	Help								

- 2. On the **Autoprovisioning** page, you may 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
Disable Autoprovisioning ?	Prevent the device from automatically trying to download a configuration file. See Section 2.6.13.1, "Autoprovisioning" for more information.
Autoprovisioning Server ?	Enter the address of the provisioning server as a fqdn or IPv4 address in dotted decimal notation.
Autoprovisioning Filename ?	The name of the configuration file. 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 Enable NTP setting on the Device Page page (see Table 2-6).
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 Enable NTP setting on the Device Page page (see Table 2-6).
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 Enable NTP setting on the Device Page page (see Table 2-6).
Save	Click the Save button to save your configuration settings.
Reboot	Click on the Reboot button to reboot the system.
Toggle Help	Click on the Toggle Help button to see a short description of some of the web page items. First click on the Toggle Help button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.

Table 2-14. Autoprovisioning Configuration Parameters
Web Page Item	Description					
Download Template	Press the Download Template button to create an autoprovisioning file for the device. See Section 2.6.13.3, "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 2-14. Autoprovisioning Configuration Parameters (continued)

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

2.6.13.1 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 2.6.13.2, "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 2-14). 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:

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 2-15. 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>OutdoorKeypad31</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></productString></productString> Autoprovisioning He Example 1

During 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

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.

2.6.13.2 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; }

2.6.13.3 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.
- You will see a window prompting you to save a configuration file (.xml) to a location on your computer (Figure 2-34). 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 2-34.

😉 Opening 0020f702bf18.xml 🔶 🗖 🗙
You have chosen to open:
0020f702bf18.xml which is: XML document (11.3 KB) from: https://10.10.1.50
What should Firefox do with this file?
O Open with Text Editor (default)
○ <u>S</u> ave File
Do this <u>a</u> utomatically for files like this from now on.
Cancel OK

Figure 2-34. 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.

2.7 Upgrade the Firmware

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

To upgrade the firmware of your device:

- 1. Download the latest firmware file from the **Downloads** tab at the following webpage: <u>https://www.cyberdata.net/products/011233</u>
- 2. Unzip the firmware version file. This file may contain the following:
- Firmware file
- Release notes
- Autoprovisioning template
- 3. Log in to the Home page as instructed in 2.6.4 "Log in to the Configuration GUI".
- 4. Click on the Firmware menu button to open the Firmware page (Figure 2-35).



Caution

Equipment Hazard: CyberData strongly recommends that you first reboot the device before attempting to upgrade the firmware of the device. See Section 2.7, "Upgrade the Firmware".

Figure 2-35. Firmware Page

Home	Device	Network	SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
	C	Cvb	erD	ata	a Pa	ain	g Ac	lapt	ter	
Browse						J	9			
Upload	Progress									
Upload	Post Proc	cessing								
	Messages	6								
Status Socket conn	100	5								

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

6. Select the firmware file. This reveals the **Upload** button (Figure 2-36).

Figure	2-36.	Upload	Button
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Home	Device	Network	SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware	
	(Cyb	erD	ata	. Pa	ain	g Ac	lant	er		
Browse		<i>y</i> , , , , ,		ull		giii	970	ap			
	Progress										
Upload	Post Pro	cessing									
		-									
Status Socket conr	Messages	S									
pload button	Statu	us Messad	185	Uni	oad Post I	Processi	na har	Unioa	d Progress	har	

- 7. Click on the **Upload** button. After selecting the **Upload** button, you will see the progress of the upload in the **Upload Progress** bar.
- 8. When the upload is complete, you will see the words Upload finished under Status Messages.
- 9. At this point, you will see the progress of the upload's post processing in the **Upload Post Processing** bar.
- **Note** Do not reboot the device before the upgrading process is complete.
- 10. When the process is complete, you will see the words **SWUPDATE Successful** under **Status Messages**.
- 11. The device will reboot automatically.
- 12. The **Home** page will display the version number of the firmware and indicate which boot partition is active.

Table 2-16 shows the web page items on the **Firmware** page.

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

Table 2-16. Firmware Page Parameters

2.7.1 Reboot the SIP Paging Adapter

To reboot a SIP Paging Adapter, log in to the web page as instructed in Section 2.6.4, "Log in to the Configuration GUI".

1. Click **Reboot** (Figure 2-37). A normal restart will occur.

Figure 2-37. Home Page

Home	Device	Network	SIP	SSL	Multicast	Fault	Audiofiles	Events	Autoprov	Firmware
	6	Nh	orF)ata	Da	ain	g Ao	lant	or	
		J y D		λαιά	ига	giii	y A	Japi		
Current St	tatus			Admin	Settings		Ii	mport Set	tings	
Serial Number:		233100001		Username:	admin			Browse No f		
Mac Address:	0	0:20:f7:04:68:2b		Password:	•••••					
Firmware Version		v20.2.0		Confirm Pas				Import Config		
Partition 2: Partition 3:		∨20.2.0 ∨20.2.0		Confirm Pas	swora:					
Booting From:		artition 2								
Boot From Othe				Save F	Reboot Toggle	Help	E	xport Set	tings	
								Export Config		
IP Addressing:		HCP								
IP Address: Subnet Mask:		0.10.1.81 55.0.0.0								
Default Gateway:		0.0.0.1								
DNS Server 1:		0.0.1.56								
DNS Server 2:										
SIP Mode:	E	nabled								
Multicast Mode:		isabled								
Event Reporting:		lisabled								
Nightringer:	C	isabled								
Primary SIP Serv	er: N	lot registered								
Backup Server 1:	: N	lot registered								
Backup Server 2: Nightringer Serve		lot registered								
	A NORTH A	lot registered								

Reboot

2.8 Command Interface

Some functions on the device can be activated using simple POST commands to the web interface. The examples in Table 2-17 use the free unix utility, **wget commands**. However, any program that can send HTTP POST commands to the device should work.

2.8.1 Command Interface Post Commands

The commands in Table 2-17 require an authenticated session (a valid username and password to work).

HTTP Post Command ^a
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=reboot"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=call&extension=600"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=terminate"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=test_relay"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=activate_relay"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=deactivate_relay"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=speak_ip_address"
wgetuser adminpassword adminauth-no-challengequiet -O /dev/nullno-check-certificate "https://10.10.1.247/command" post-data "request=test_audio"
wgetuser adminpassword adminauth-no-challengeno- check-certificatequiet -O /dev/null "https://10.10.1.81/command" post-data "request=swap_boot_partition"

Table 2-17. Command Interface Post Commands

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

Appendix A: Troubleshooting/Technical Support

A.1 Frequently Asked Questions (FAQ)

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

https://www.cyberdata.net/products/011233

A.2 Documentation

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

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

https://www.cyberdata.net/products/011233

A.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:

https://support.cyberdata.net/

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

Phone: (831) 373-2601, Extension 333

A.4 Warranty and RMA Information

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

https://support.cyberdata.net/

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