

# *VoIP Flush-Mount Indoor Intercom Operations Guide*

Part #011272

Document Part #930853H  
for Firmware Version 11.4.2

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**VoIP Flush Mount Indoor Intercom Operations Guide 930853H**  
**Part # 011272**

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Technical Support

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

Phone: (831) 373-2601, Ext. 333

Email: [support@cyberdata.net](mailto:support@cyberdata.net)

Fax: (831) 373-4193

Company and product information is at [www.cyberdata.net](http://www.cyberdata.net).

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## Revision Information

Revision 930853H, which corresponds to firmware version 11.4.2, was released on May 15, 2015, and has the following changes:

- Updates [Figure 2-17, "Device Configuration Page"](#)
- Updates [Table 2-7, "Device Configuration Parameters"](#) to add the following settings:
  - Enable Push to Talk
  - Prevent Call Termination
- Updates the DTMF Activation Code setting description in [Table 2-7, "Device Configuration Parameters"](#) and [Table 2-20, "DSR Configuration Parameters"](#) to add the following note:

Note: While Enable DTMF Push to Talk is enabled, the \*, #, and 0 keys cannot be used in the DTMF activation code.
- Updates [Section C.3, "Contact Information"](#) with "http://support.cyberdata.net/"
- Updates [Section C.4, "Warranty and RMA Information"](#) with "http://support.cyberdata.net/"

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## Browsers Supported

The following browsers have been tested against firmware version 11.4.2:

- Internet Explorer (version: 10)
- Firefox (also called Mozilla Firefox) (version: 23.0.1 and 25.0)
- Chrome (version: 29.0.1547.66 m)
- Safari (version: 5.1.7)

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## Pictorial Alert Icons

 <p>GENERAL ALERT</p>	<b>General Alert</b> 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.
	<b>Ground</b> This pictorial alert indicates the Earth grounding connection point.

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## 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.

**14. WARNING: The VoIP Intercom enclosure is not rated for any AC voltages!**

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

<b>Chapter 1 Product Overview</b>	<b>1</b>
1.1 How to Identify This Product .....	1
1.2 Typical System Installation .....	2
1.3 Product Features .....	3
1.4 Supported Protocols .....	4
1.5 Supported SIP Servers .....	4
1.6 Product Specifications .....	5
<b>Chapter 2 Installing the VoIP Flush-Mount Indoor Intercom</b>	<b>6</b>
2.1 Parts List .....	6
2.2 Intercom Components .....	7
2.3 Intercom Setup .....	8
2.3.1 Intercom Connections .....	8
2.3.2 Using the On-Board Relay .....	9
2.3.3 Wiring the Circuit .....	10
2.3.4 Intercom Connectors .....	13
2.3.5 Activity and Link LEDs .....	15
2.3.6 RTFM Button .....	16
2.3.7 Adjusting the Intercom Volume .....	17
2.3.8 Call Button and the Call Button LED .....	18
2.4 Configure the Intercom Parameters .....	19
2.4.1 Factory Default Settings .....	19
2.4.2 Intercom Web Page Navigation .....	20
2.4.3 Using the Toggle Help Button .....	21
2.4.4 Log in to the Configuration Home Page .....	23
2.4.5 Configure the Device .....	27
2.4.6 Configure the Network Parameters .....	35
2.4.7 Configure the SIP Parameters .....	38
2.4.8 Configure the Multicast Parameters .....	45
2.4.9 Configure the Sensor Configuration Parameters .....	47
2.4.10 Configure the Audio Configuration Parameters .....	50
2.4.11 Configure the Events Parameters .....	55
2.4.12 Configure the Door Strike Relay .....	61
2.4.13 Configure the Device (on the DSR page) .....	65
2.4.14 Configure the Autoprovisioning Parameters .....	68
2.5 Upgrade the Firmware and Reboot the Intercom .....	80
2.5.1 Uploading the Firmware .....	80
2.5.2 Reboot the Device .....	82
2.6 Command Interface .....	83
2.6.1 Command Interface Post Commands .....	83
<b>Appendix A Mounting the Intercom</b>	<b>87</b>
A.1 Mounting Components .....	87
A.2 Dimensions .....	88
A.3 Wall Mounting .....	90
A.4 Ground Cable Installation .....	91
<b>Appendix B Setting up a TFTP Server</b>	<b>92</b>
B.1 Set up a TFTP Server .....	92
B.1.1 In a LINUX Environment .....	92
B.1.2 In a Windows Environment .....	92
<b>Appendix C Troubleshooting/Technical Support</b>	<b>93</b>
C.1 Frequently Asked Questions (FAQ) .....	93
C.2 Documentation .....	93
C.3 Contact Information .....	94
C.4 Warranty and RMA Information .....	94



# 1 Product Overview

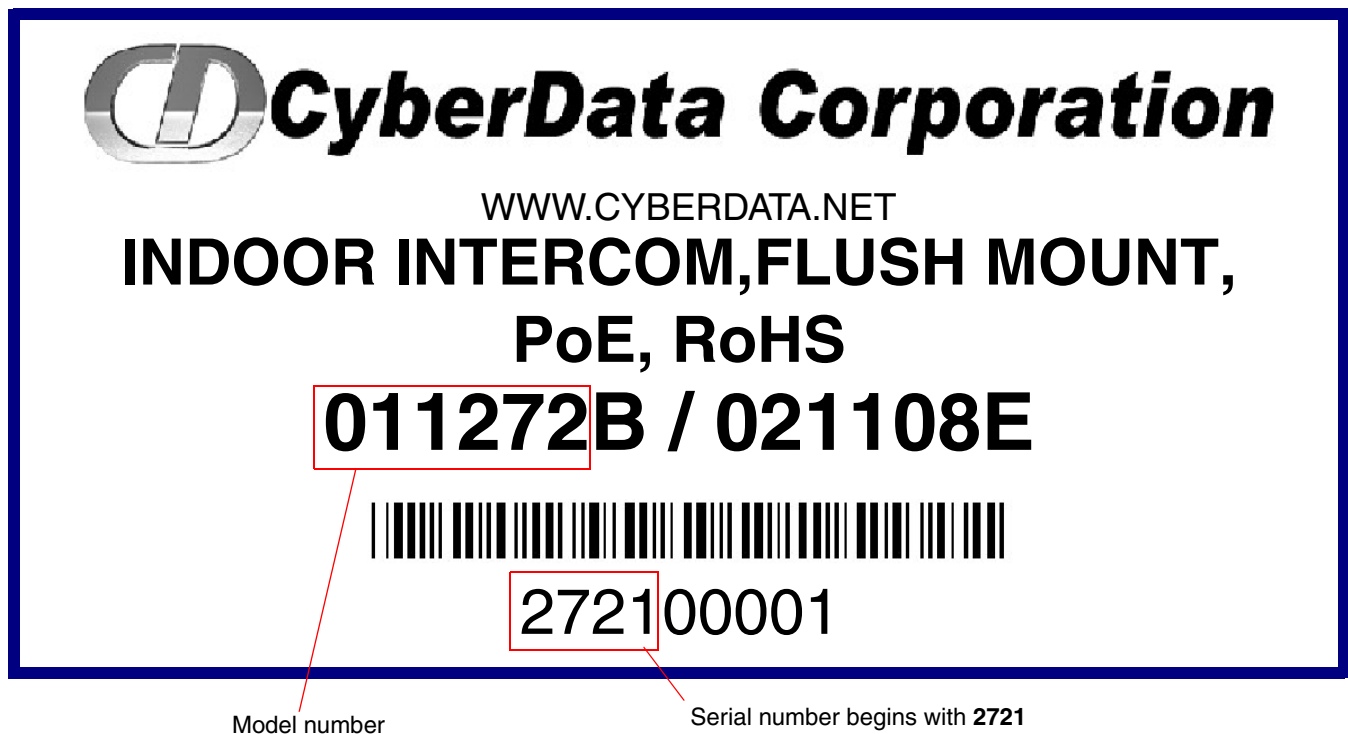
## 1.1 How to Identify This Product

To identify the VoIP Flush-Mount Indoor Intercom, 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 **011272**.
- The serial number on the label should begin with **2721**.

Figure 1-1. Model Number Label



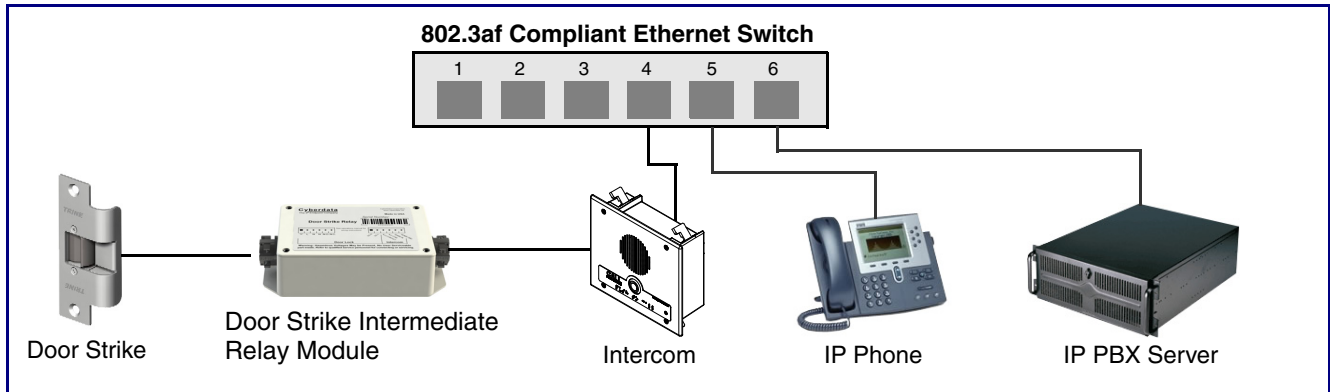


## 1.2 Typical System Installation

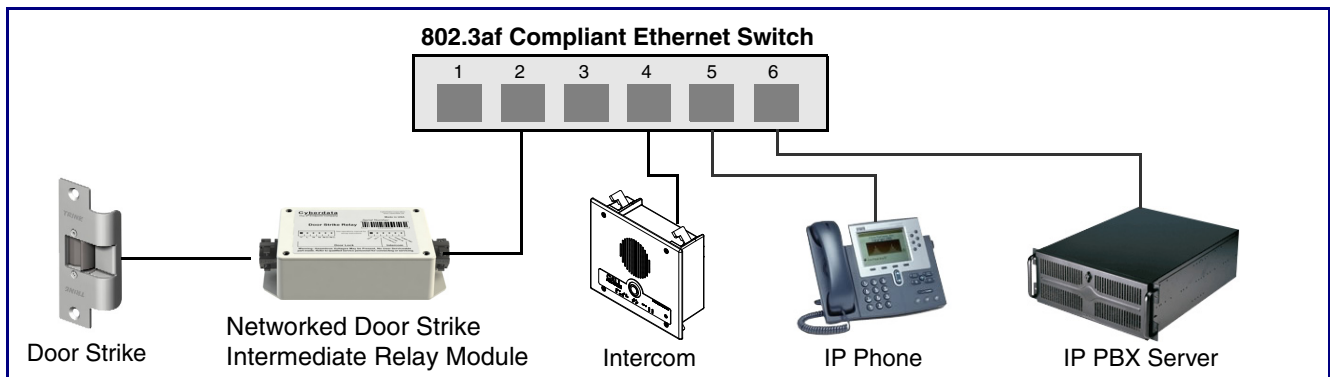
The Voice-over-IP (VoIP) Intercom is a SIP endpoint designed to provide VoIP phone connectivity in a tamper-proof and secure package.

The following figures illustrate how the VoIP Flush-Mount Indoor Intercom can be installed as part of a VoIP phone system.

**Figure 1-2. Typical Installation—Door Strike Intermediate Relay Module**



**Figure 1-3. Typical Installation—Networked Door Strike Intermediate Relay Module**



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## 1.3 Product Features

The VoIP Flush Mount Indoor Intercom has the following features:

- Supports SRST (Survivable Remote Site Telephony) in a Cisco environment. SRST parameters are entered statically into the CyberData product's internal webpage.
- SIP
- Dual speeds of 10 Mbps and 100 Mbps
- 802.3af compliant
- Adaptive full duplex voice operation
- Network/Web management
- Network adjustable speaker volume adjustment
- Network configurable door or intrusion sensor settings
- Network configurable relay activation settings
- Dial Out Extension supports the addition of comma delimited pauses before sending additional DTMF tones
- Network configurable microphone input sensitivity adjustment
- Network downloadable product firmware
- Doubles as a paging speaker
- Call button
- Call activity indicator (light)
- One dry contact relay for auxiliary control
- Autoprovisioning
- Configurable audio files
- Night Ringer
- Peer-to-peer capable
- Door closure and tamper alert signal
- An active call is indicated by the Call Button LED blinking at one second intervals.

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## 1.4 Supported Protocols

The Intercom supports the following protocols:

- SIP
- HTTP Web-based configuration  
Provides an intuitive user interface for easy system configuration and verification of Intercom operations.
- DHCP Client  
Dynamically assigns IP addresses in addition to the option to use static addressing.
- TFTP Client  
Facilitates hosting for the Autoprovisioning configuration file.
- RTP
- RTP/AVP - Audio Video Profile
- Facilitates autoprovisioning configuration values on boot
- Audio Encodings
  - PCMU (G.711 mu-law)
  - PCMA (G.711 A-law)
  - Packet Time 20 ms

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## 1.5 Supported SIP Servers

Go to the following link to find the VoIP Flush-Mount Indoor Intercom product page which will have information on how to configure the device for various supported SIP servers:

<http://www.cyberdata.net/support/server/index.html>

## 1.6 Product Specifications

**Table 1-1. Specifications**

<b>Category</b>	<b>Specification</b>
Speaker Output	1 Watt Peak Power
Ethernet I/F	10/100 Mbps
Protocol	SIP RFC 3261 Compatible
Power Input	PoE 802.3af compliant or 8 to 12VDC at 1000mA
Operating Temperature	-40°C to +55°C (-40°F to 131°F)
Payload Types	G711, A-law and $\mu$ -law
Auxiliary Relay	1A at 30 VDC
Dimensions	5.118" x 5.118" x 2.25" (H x W x D)
Weight	0.71 lbs./shipping weight of 1.1 lbs. (0.32 kg/shipping weight of 0.5 kg)
Part Number	011272

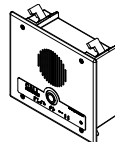


# 2 Installing the VoIP Flush-Mount Indoor Intercom

## 2.1 Parts List

Table 2-1 illustrates the VoIP Flush Mount Indoor Intercom parts.

**Note** See [Appendix A, "Mounting the Intercom"](#) for physical mounting information.

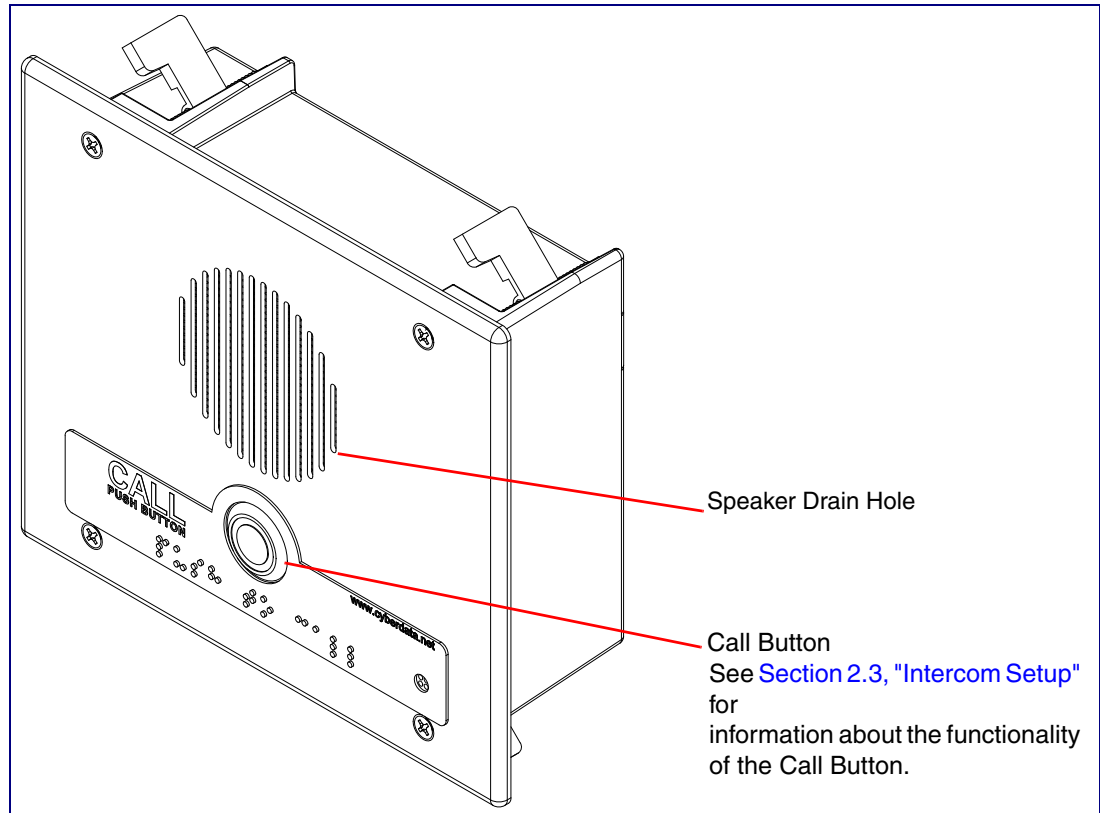
**Table 2-1. Parts List**

Quantity	Part Name	Illustration
1	Intercom Assembly	
1	Installation Quick Reference Guide	
1	Intercom Mounting Accessory Kit	

## 2.2 Intercom Components

Figure 2-1 shows the components of the Intercom.

**Figure 2-1. Intercom Components**



## 2.3 Intercom Setup

### 2.3.1 Intercom Connections

Figure 2-2 shows the pin connections on the J3 (terminal block). This terminal block can accept 16 AWG gauge wire.

**Note** As an alternative to using PoE power, you can supply 8 to 12VDC at 1000mA into the terminal block.


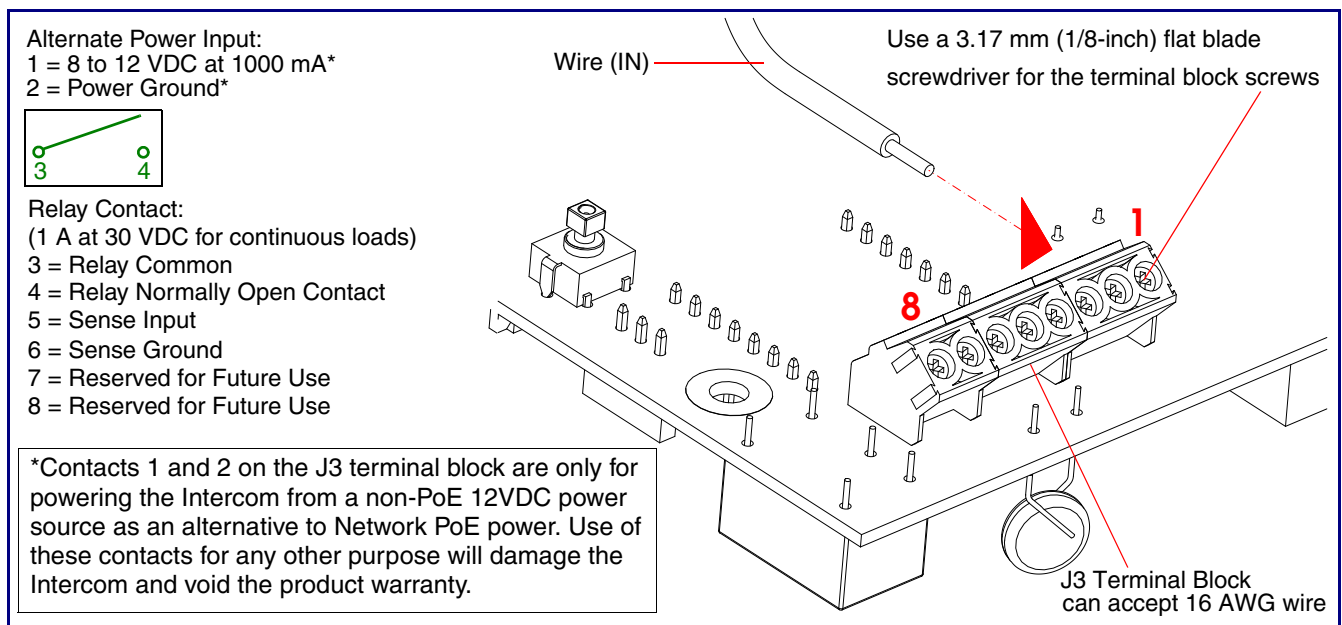



 GENERAL ALERT	<p><b>Caution</b></p> <p><i>Equipment Hazard:</i> Contacts 1 and 2 on the J3 terminal block are only for powering the Intercom from a non-PoE 12 VDC power source as an alternative to Network PoE power. Use of these contacts for any other purpose will damage the Intercom and void the product warranty.</p>
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Figure 2-2. Intercom Connections



## 2.3.2 Using the On-Board Relay

 <p>GENERAL ALERT</p>	<p><b>Warning</b></p> <p><i>Electrical Hazard:</i> This product should be installed by a licensed electrician according to all local electrical and building codes.</p>
 <p>GENERAL ALERT</p>	<p><b>Warning</b></p> <p><i>Electrical Hazard:</i> The relay contacts are dry and provided for a normally open and momentarily closed configuration. Neither the alternate power input nor PoE power can be used to drive a door strike.</p>
 <p>GENERAL ALERT</p>	<p><b>Warning</b></p> <p><i>Electrical Hazard:</i> The relay does not support AC powered door strikes. Any use of this relay beyond its normal operating range can cause damage to the product and is not covered under our warranty policy.</p>

The device has a built-in relay that can be activated by a web configurable DTMF string that can be received from a VoIP phone supporting out of band (RFC2833) DTMF as well as a number of other triggering events. See the [Device Configuration Page](#) on the web interface for relay settings.

This relay can be used to trigger low current devices like LED strobes and security camera input signals as long as the load is not an inductive type and the relay is limited to a maximum of 1 Amp @ 30 VDC. Inductive loads can cause excessive “hum” and can interfere with or damage the unit’s electronics.

We highly recommend that inductive load and high current devices use our Door Strike Intermediate Relay product (CD# 011269) (see [Section 2.3.3.2, "Connecting the Door Strike Intermediate Relay Module"](#)).

This relay interface also has a general purpose input port that can be used to monitor an external switch and generate an event.

For more information on the sensor options, see the [Sensor Configuration Page](#) on the web interface.



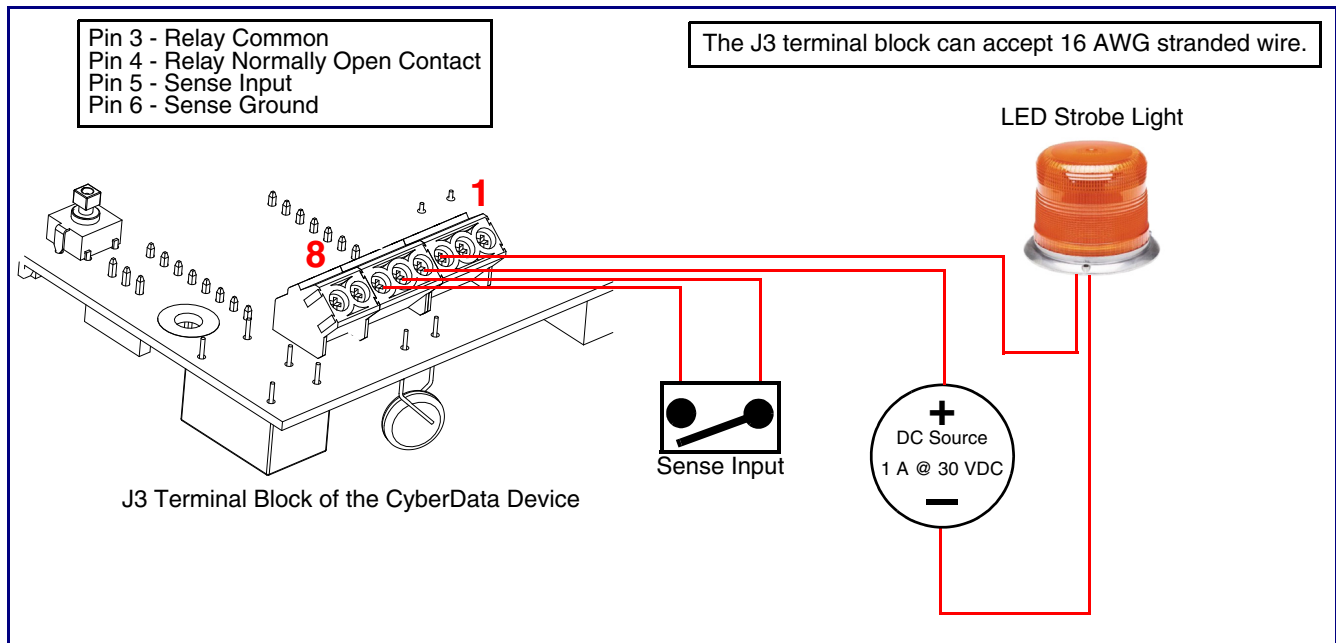
## 2.3.3 Wiring the Circuit

### 2.3.3.1 Devices Less than 1A at 30 VDC

If the power for the device is less than 1A at 30 VDC and is not an inductive load, then see [Figure 2-3](#) for the wiring diagram.

When configuring with an inductive load, please use an intermediary relay with a High PIV Ultrafast Switching Diode. We recommend using the CyberData Door Strike Intermediate Relay Module (CD# 011269) (see [Section 2.3.3.2, "Connecting the Door Strike Intermediate Relay Module"](#)).

**Figure 2-3. Wiring Diagram**

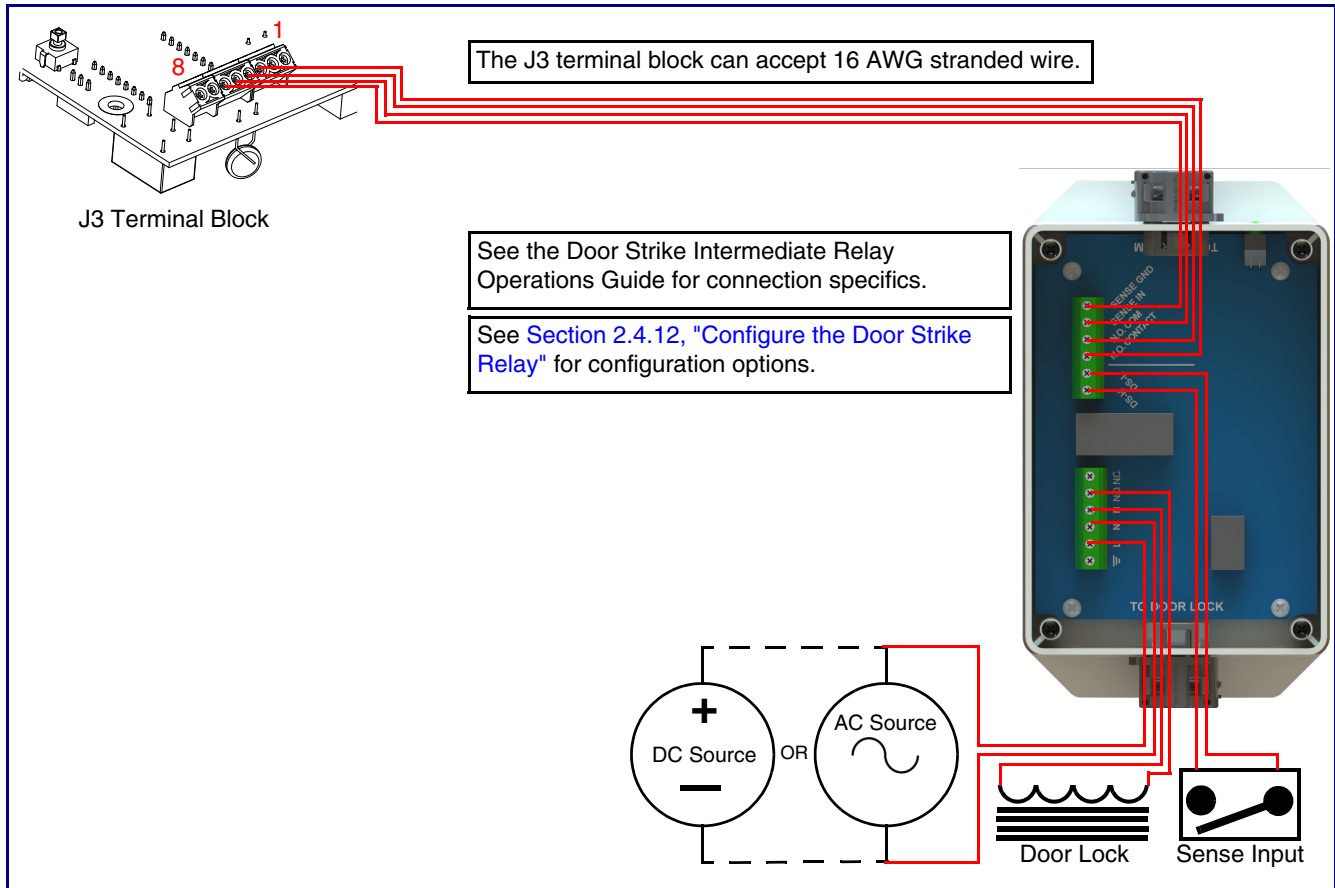


### 2.3.3.2 Connecting the Door Strike Intermediate Relay Module

For wiring an electronic door strike, we recommend the use of our external Door Strike Intermediate Relay (CD# 011269).

This product provides an easier method of connecting standard door strikes as well as AC and higher voltage devices. See [Figure 2-4](#) for the wiring diagram.

**Figure 2-4. Wiring Diagram**



If you have questions about connecting door strikes or setting up the web configurable options, please contact our support department.

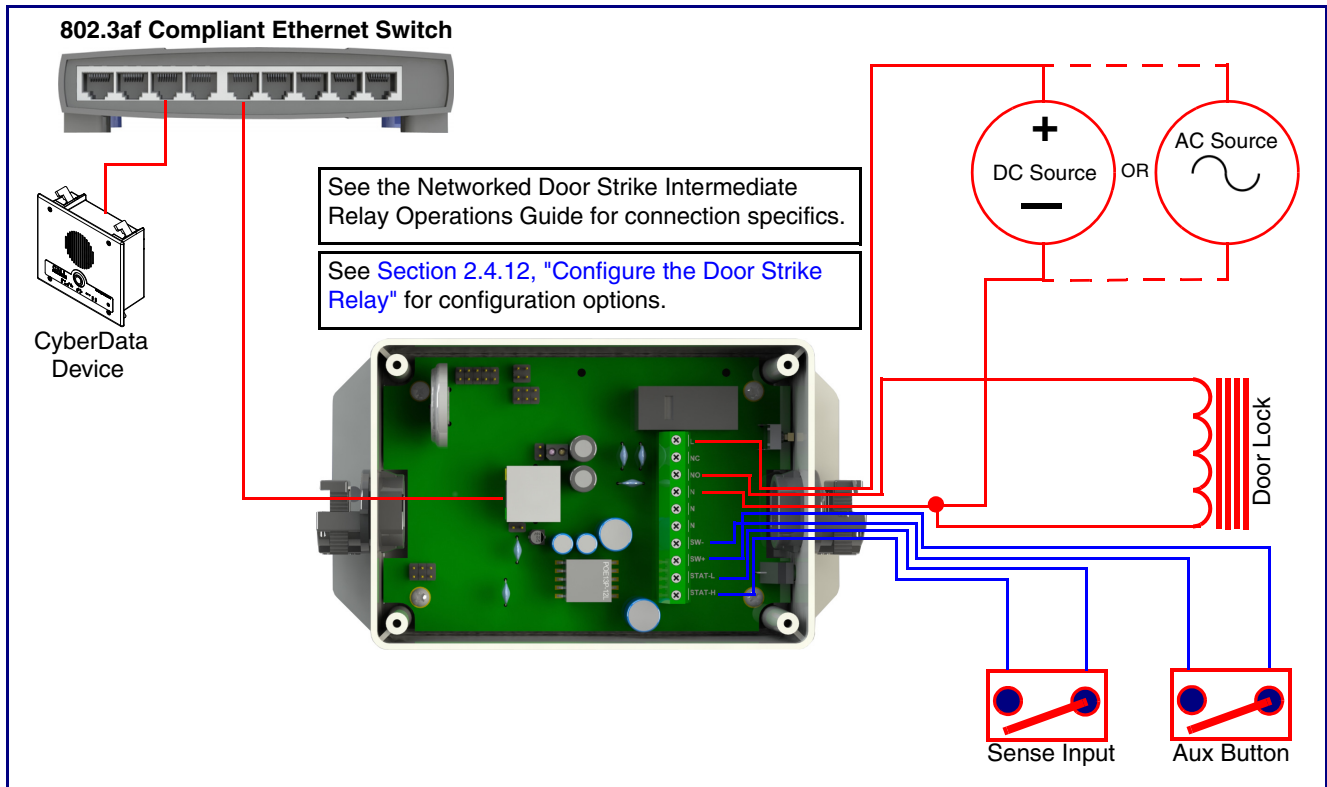
<http://www.cyberdata.net/support/voip/index.html>

### 2.3.3.3 Connecting the Networked Door Strike Intermediate Relay

For wiring an electronic door strike to work over a network, we recommend the use of our external Networked Door Strike Intermediate Relay (CD# 011270).

This product provides an easier method of connecting standard door strikes as well as AC and higher voltage devices. See [Figure 2-5](#) for the wiring diagram.

**Figure 2-5. Wiring Diagram**



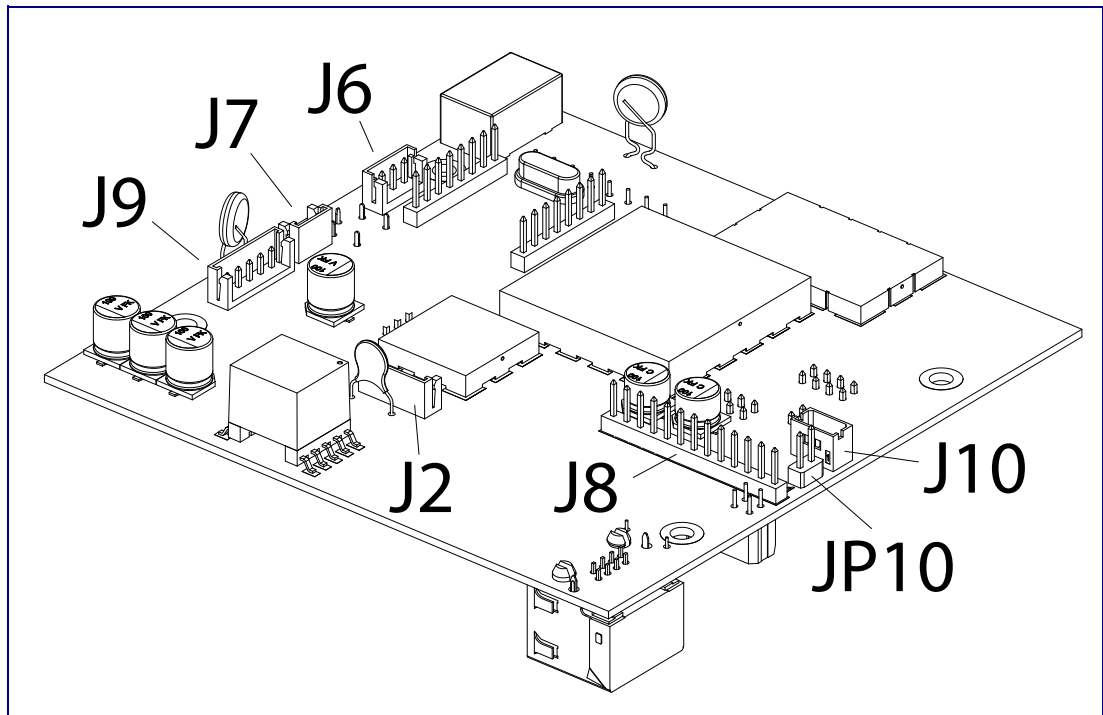
If you have questions about connecting door strikes or setting up the web configurable options, please contact our support department.

<http://www.cyberdata.net/support/voip/index.html>

## 2.3.4 Intercom Connectors

See the following figures and tables to identify the connectors and functions of the Intercom.

**Figure 2-6. Connector Locations**



**Table 2-2. Connector Functions**

Connector	Function
J2	Call Button LED Interface
J6	Microphone Interface
J7	Speaker Interface
J8	Keypad Interface — Not Used
J9	Auxiliary Strobe Connector — Not Used
J10	Proximity Sensor Interface — Not Used
JP10	Disables the intrusion sensor when installed.

Figure 2-7. Connector Locations

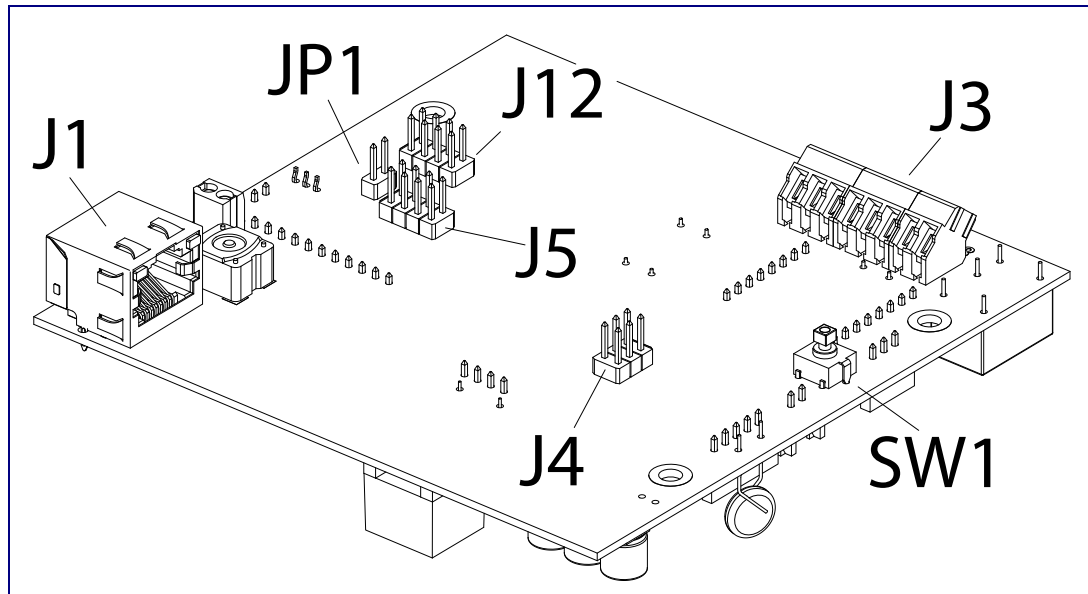


Table 2-3. Connector Functions

Connector	Function
JP1	Reset jumper <sup>a</sup>
J1	PoE Network Connection (RJ-45 ethernet)
J3	Terminal Block (see <a href="#">Figure 2-2</a> )
J4	Console Port (Factory Use Only)
J5	JTAG (Factory Use Only)
J12	Reserved (Factory Use Only)
SW1	See <a href="#">Section 2.3.6, "RTFM Button"</a>

<sup>a</sup>.Do not install a jumper. Momentary short to reset. Permanent installation of a jumper would prevent the board from running all together.

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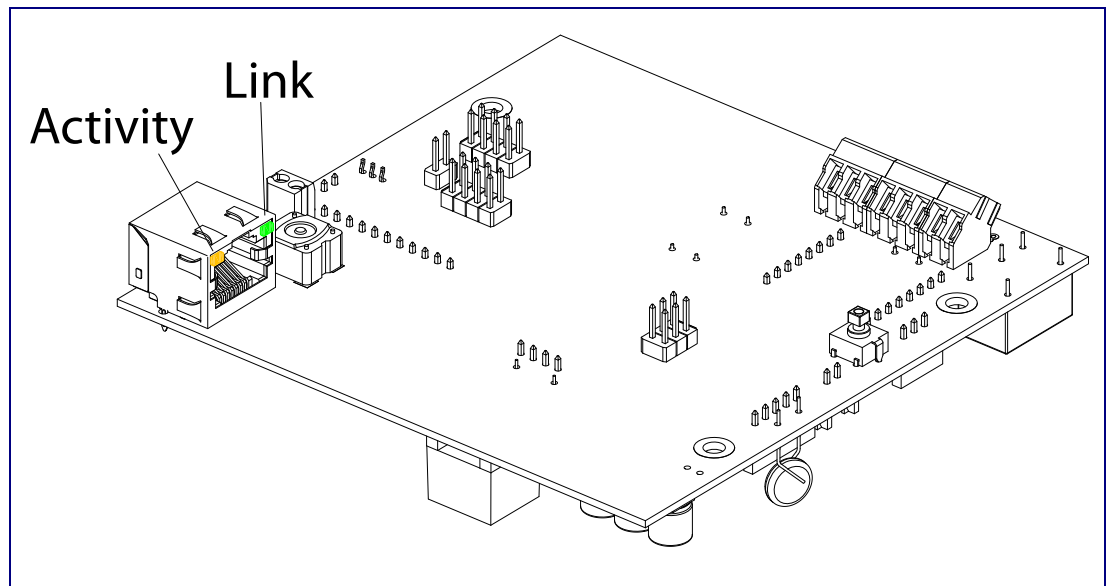
## 2.3.5 Activity and Link LEDs

### 2.3.5.1 Verifying the Network Connectivity and Data Rate

When you plug in the Ethernet cable or power supply to the Intercom, the following occurs:

- The square, **YELLOW Activity** LED blinks when there is network activity (see [Figure 2-8](#)).
- The square, **GREEN Link** LED above the Ethernet port indicates that the network connection has been established (see [Figure 2-8](#)).

**Figure 2-8. Activity and Link LED**

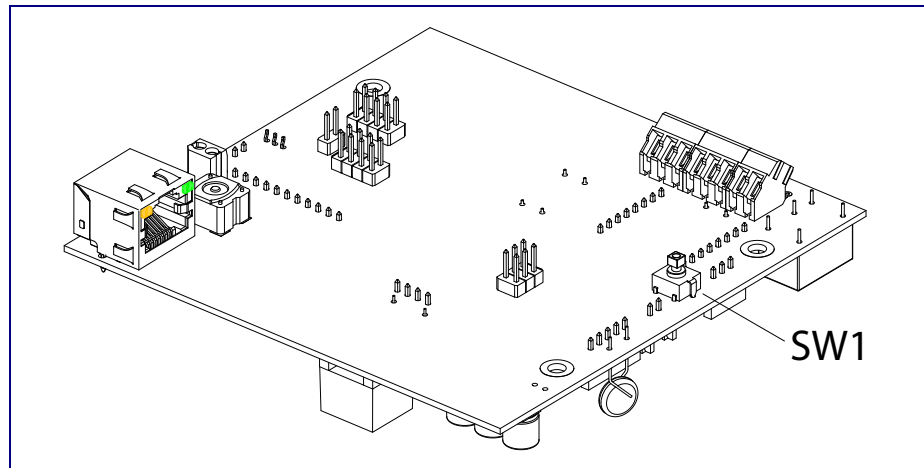


## 2.3.6 RTFM Button

When the Intercom is operational and linked to the network, you can use the Reset Test Function Management (**RTFM**) button (see **SW1** in [Figure 2-9](#)) on the Intercom board to announce and confirm the Intercom's IP Address and test to see if the audio is working.

**Note** You must do these tests prior to final assembly.

**Figure 2-9. RTFM Button (SW1)**



### 2.3.6.1 Announcing the IP Address

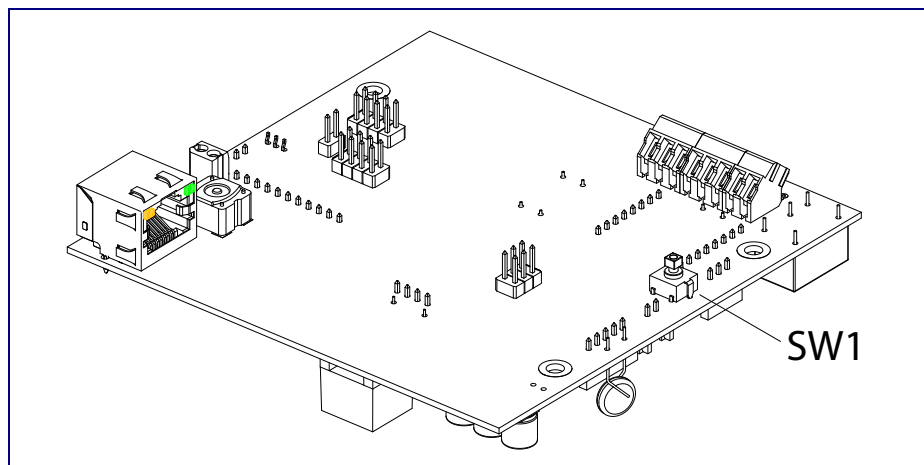
To announce a device's current IP address:

1. Press and release the RTFM button (see **SW1** in [Figure 2-10](#)) within a five second window.

**Note** The device will use DHCP to obtain the new IP address (DHCP-assigned address or default to 10.10.10.10 if a DHCP server is not present).

**Note** Pressing and holding the RTFM button for longer than five seconds will restore the device to the factory default settings.

**Figure 2-10. RTFM Button (SW1)**



### 2.3.6.2 Restoring the Factory Default Settings

When troubleshooting configuration problems, it is sometimes convenient to restore the device to a known state.

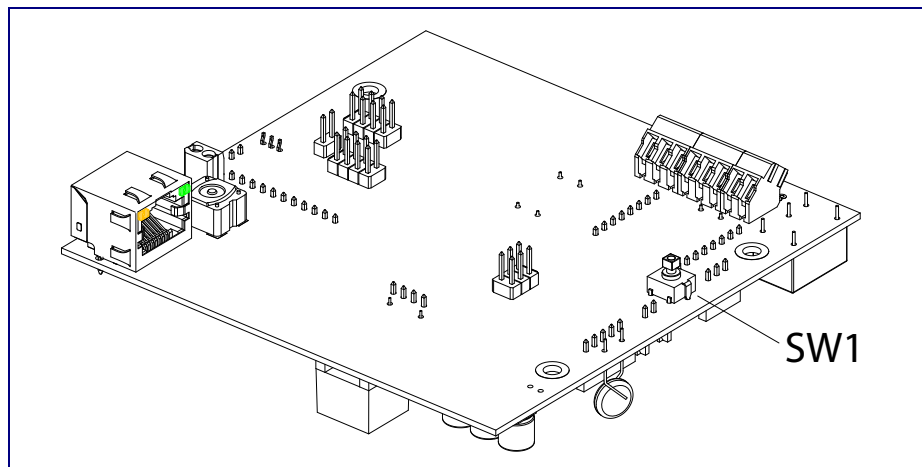
**Note** Each Intercom is delivered with factory set default values.

To restore the factory default settings:

1. Press and hold the **RTFM button** (see **SW1** in [Figure 2-11](#)) for more than five seconds.
2. The device announces that it is restoring the factory default settings.

**Note** The device will use DHCP to obtain the new IP address (DHCP-assigned address or default to 10.10.10.10 if a DHCP server is not present).

**Figure 2-11. RTFM Button**



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### 2.3.7 Adjusting the Intercom Volume

You can adjust the Intercom volume through the [SIP Volume](#), [Multicast Volume](#), [Ring Volume](#), and [Sensor Volume](#) settings on the [Device Configuration Page](#).



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## 2.3.8 Call Button and the Call Button LED

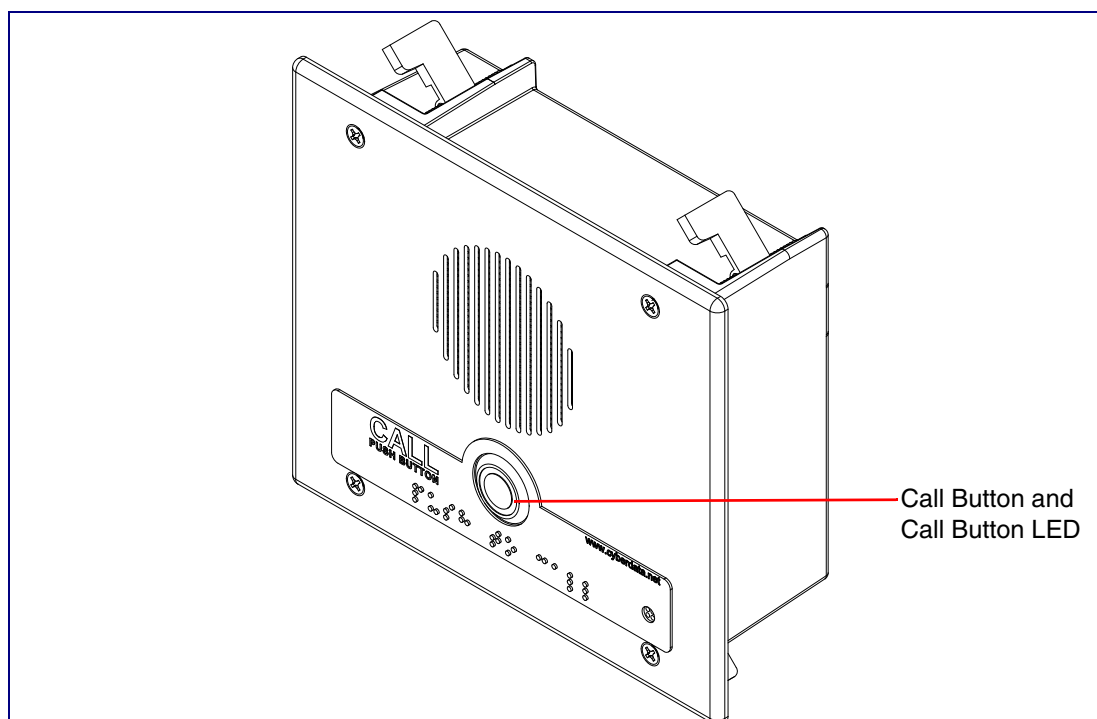
### 2.3.8.1 Calling with the The Call Button

- You may initiate a call by pressing the **Call** button.
- An active call is indicated by the Call Button LED blinking at one second intervals.
- The Intercom can automatically answer an incoming call.
- You can press the Call Button to terminate an active call.

### 2.3.8.2 Call Button LED Function

- Upon initial power or reset, the Call Button LED will illuminate.
- On boot, the Call Button LED will flash ten times a second while setting up the network and downloading autoprovisioning files.
- The device “autoprovisions” by default, and the initial process may take several minutes as the device searches for and downloads updates. The Call Button LED will blink during this process. During the initial provisioning, or after the factory defaults have been reset, the device may download firmware twice. The device will blink, remain solid for 10 to 20 seconds, and then resume blinking. This process will take longer if there are many audio files downloading.
- When the software has finished initialization, the Call Button LED will blink twice.
- When a call is established (not just ringing), the Call Button LED will blink.
- On the [Device Configuration Page](#) (see [Section 2.4.5, "Configure the Device"](#)), there is an option called [Button Lit When Idle](#). This option sets the normal state for the indicator LED. The Call Button LED will still blink during initialization and calls.
- The Call Button LED flashes briefly at the beginning of RTFM mode.

**Figure 2-12. Call Button and Call Button LED**



---

## 2.4 Configure the Intercom Parameters

To configure the Intercom online, use a standard web browser.

Configure each Intercom and verify its operation *before* you mount it. When you are ready to mount an Intercom, refer to [Appendix A, "Mounting the Intercom"](#) for instructions.

---

### 2.4.1 Factory Default Settings

All Intercoms are initially configured with the following default IP settings:

When configuring more than one Intercom, attach the Intercoms to the network and configure one at a time to avoid IP address conflicts.

**Table 2-4. Factory Default Settings**

<b>Parameter</b>	<b>Factory Default Setting</b>
IP Addressing	DHCP
IP Address <sup>a</sup>	10.10.10.10
Web Access Username	admin
Web Access Password	admin
Subnet Mask <sup>a</sup>	255.0.0.0
Default Gateway <sup>a</sup>	10.0.0.1

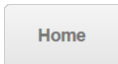

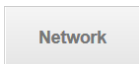


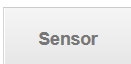
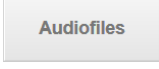
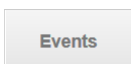


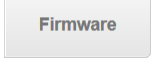
a. Default if there is not a DHCP server present.

---

## 2.4.2 Intercom Web Page Navigation

Table 2-5 shows the navigation buttons that you will see on every Intercom web page.

**Table 2-5. Web Page Navigation**

Web Page Item	Description
	Link to the <b>Home</b> page.
	Link to the <b>Device</b> page.
	Link to the <b>Network</b> page.
	Link to go to the <b>SIP</b> page.
	Link to the <b>Multicast</b> page.
	Link to the <b>Sensor</b> page.
	Link to the <b>Audiofiles</b> page.
	Link to the <b>Events</b> page.
	Link to the <b>Door Strike Relay</b> page.
	Link to the <b>Autoprovisioning</b> page.
	Link to the <b>Firmware</b> page.

## 2.4.3 Using the Toggle Help Button

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

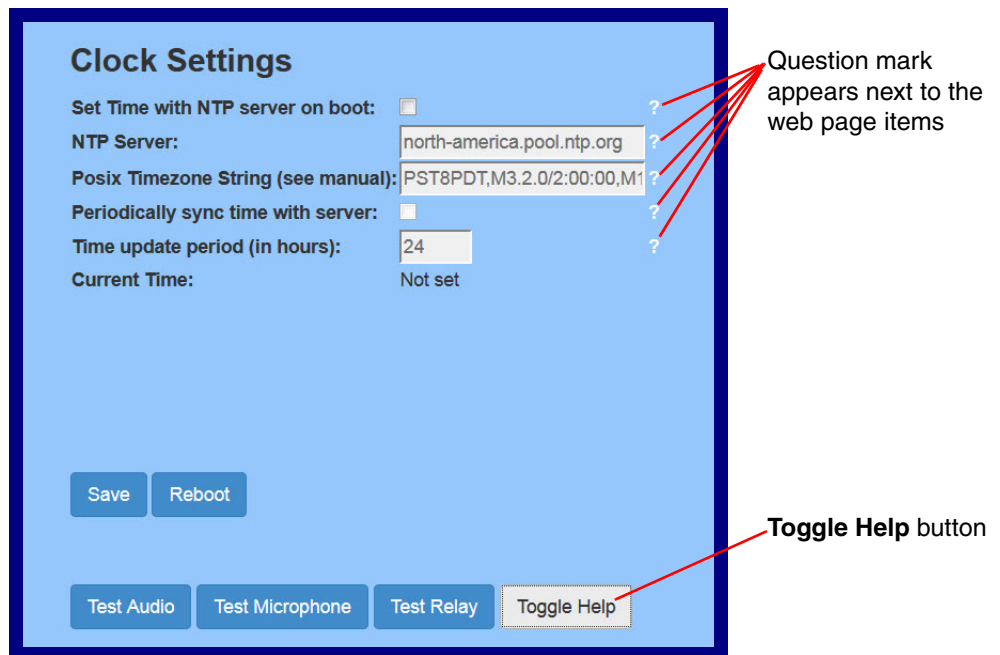
1. Click on the **Toggle Help** button that is on the UI webpage. See [Figure 2-13](#) and [Figure 2-14](#).

**Figure 2-13. Toggle/Help Button**



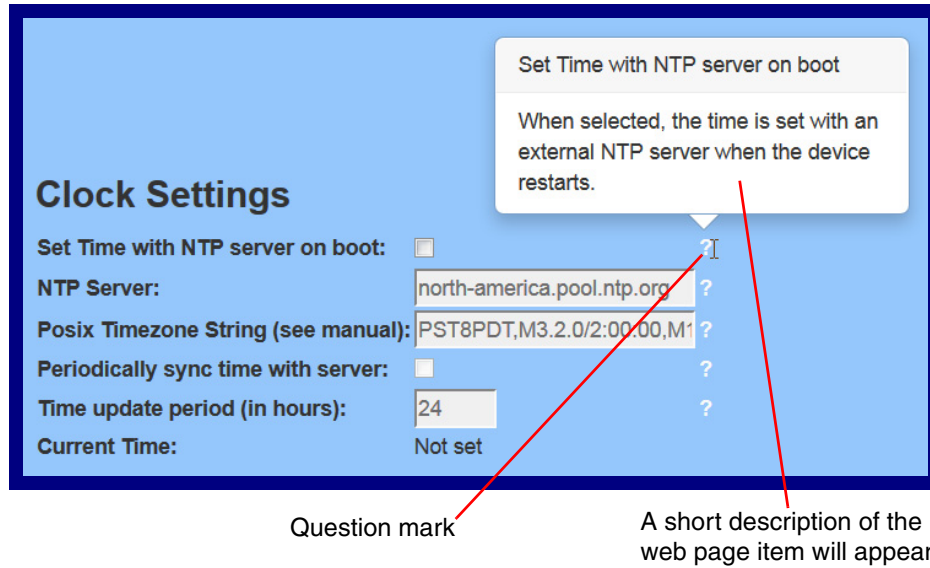
2. You will see a question mark ( ? ) appear next to each web page item that has been provided with a short description by the Help feature. See [Figure 2-14](#).

**Figure 2-14. Toggle Help Button and Question Marks**



3. Move the mouse pointer to hover over the question mark (?), and a short description of the web page item will appear. See [Figure 2-15](#).

**Figure 2-15. Short Description Provided by the Help Feature**



---

## 2.4.4 Log in to the Configuration Home Page

1. Open your browser to the Intercom IP address.

**Note** If the network does not have access to a DHCP server, the device will default to an IP address of 10.10.10.10.

**Note** Make sure that the PC is on the same IP network as the Intercom.

**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:

<http://www.cyberdata.net/support/voip/discovery.html>

**Note** The Intercom 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.

2. When prompted, use the following default **Web Access Username** and **Web Access Password** to access the **Home Page** (Figure 2-16):

Web Access Username: **admin**

Web Access Password: **admin**

Figure 2-16. Home Page

Home Device Network SIP Multicast Sensor Audiofiles Events DSR Autoprov Firmware

# CyberData Indoor Intercom

### Current Status

Serial Number: 272101456  
Mac Address: 00:20:f7:02:4e:bd  
Firmware Version: v11.4.2

IP Addressing: DHCP  
IP Address: 10.10.1.183  
Subnet Mask: 255.0.0.0  
Default Gateway: 10.0.0.1  
DNS Server 1: 10.0.0.252  
DNS Server 2:

SIP Volume: 4  
Multicast Volume: 4  
Ring Volume: 4  
Sensor Volume: 4  
Volume Boost: Off  
Microphone Gain: 4

SIP Mode: Enabled  
Multicast Mode: Disabled  
Event Reporting: Disabled  
Nightringer: Disabled

Primary SIP Server: **Not registered**  
Backup Server 1: Not registered  
Backup Server 2: Not registered  
Nightringer Server: Not registered

### Admin Settings

Username: admin  
Password:  
Confirm Password:

Save Reboot Toggle Help

### Import Settings

Browse... No file selected.

Import Config

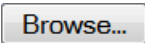

### Export Settings

Export Config

3. On the **Home** page, review the setup details and navigation buttons described 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.

**Table 2-6. Home Page Overview**

Web Page Item	Description
<b>Admin Settings</b>	
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.
<b>Current Status</b>	
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 ( <b>DHCP</b> or <b>static</b> ).
IP Address	Shows the current IP address.
Subnet Mask	Shows the current subnet mask address.
Default Gateway	Shows the current default gateway address.
DNS Server 1	Shows the current DNS Server 1 address.
DNS Server 2	Shows the current DNS Server 2 address.
SIP Volume	Shows the current SIP volume level.
Multicast Volume	Shows the current Multicast volume level.
Ring Volume	Shows the current Ring volume level.
Sensor Volume	Shows the current Sensor volume level.
Volume Boost	Shows the current Volume Boost level.
Microphone Gain	Shows the current microphone gain level.
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.
<b>Import Settings</b>	
	Use this button to select a configuration file to import.
	After selecting a configuration file, click Import to import the configuration from the selected file. Then, click Save and Reboot to store changes.



**Table 2-6. Home Page Overview (continued)**

Web Page Item	Description
<b>Export Settings</b>	
	Click Export to export the current configuration to a file.
	Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.
	Click on the <b>Reboot</b> button to reboot the system.
	Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.

## 2.4.5 Configure the Device

1. Click the **Device** menu button to open the **Device** page. See [Figure 2-17](#).

**Figure 2-17. Device Configuration Page**

The screenshot shows the 'Device' configuration page for a CyberData Indoor Intercom. The page is organized into several sections:

- Volume Settings (0-9):** Includes sliders for SIP Volume, Multicast Volume, Ring Volume, Sensor Volume, and Microphone Gain, all set to 4. A 'No Volume Boost' dropdown is present, with a note: 'Boost operation recommended with volumes set to level 9'.
- Relay Settings:** Includes checkboxes for 'Activate Relay with DTMF code' (checked), 'Play tone during DTMF Activation', 'Activate Relay During Ring', 'Activate Relay During Night Ring', and 'Activate Relay While Call Active'. It also has input fields for 'DTMF Activation Code' (321), 'DTMF Activation Duration (in seconds)' (2), and 'Relay On Button Press Duration' (3).
- Clock Settings:** Includes a checkbox for 'Set Time with NTP server on boot', an 'NTP Server' field (north-america.pool.ntp.org), a 'Posix Timezone String' field (PST8PDT,M3.2.0/2:00:00,M11.1.), a checkbox for 'Periodically sync time with server', a 'Time update period (in hours)' field (24), and a 'Current Time' field (Not set).
- Misc Settings:** Includes a 'Device Name' field (Outdoor Intercom), checkboxes for 'Auto-Answer Incoming Calls', 'Button Lit when Idle', 'Enable Push to Talk', 'Enable DTMF Push to Talk', and 'Prevent Call Termination'. It also has a 'Button Brightness (0-255)' field (255) and a 'Disable HTTPS (NOT recommended)' checkbox.

At the bottom of the page, there are buttons for 'Save', 'Reboot', 'Test Audio', 'Test Microphone', 'Test Relay', and 'Toggle Help'.

2. On the **Device** page, you may 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.


**Table 2-7. Device Configuration Parameters**

Web Page Item	Description
<b>Volume Settings (0-9)</b>	
SIP Volume ?	Set the speaker volume for a SIP call. A value of 0 will mute the speaker during SIP calls.
Multicast Volume ?	Set the speaker volume for multicast audio streams. A value of 0 will mute the speaker during multicasts.
Ring Volume ?	Set the ring volume for incoming calls. A value of 0 will mute the speaker instead of playing the ring tone when Auto-Answer Incoming Calls is disabled.
Sensor Volume ?	Set the speaker volume for playing sensor activated audio. A value of 0 will mute the speaker during sensor activated audio.
Microphone Gain ?	Set the microphone gain level.
Volume Boost: ? No Volume Boost Volume Boost 1 Volume Boost 2 Volume Boost 3	<p>Set the Boost level to increase the volume output of the speaker. Using Volume Boost may introduce audio clips or cause the device to drop from full duplex to half duplex operation.</p> <p>Normal operation of the product can be met with volume levels <b>0</b> through <b>9</b>. <b>0</b> being mute and <b>9</b> being the loudest volume that in a normal arm's length and average background noise, will enable full duplex operation and give the best quality of sound output.</p> <p>The volume boost options increase the output of the speaker by:</p> <p>3db for Boost level 1 6db for Boost level 2 9db for Boost level 3</p> <p>If the user would like a higher output from the speaker, the Boost settings are available. However, operation in <b>Boost Mode</b> may overdrive or clip the audio if, for example, the phone that is connected has a high microphone gain or if the person has a loud voice talking too close to the microphone.</p> <p>The acoustic echo canceller also has a harder time maintaining full duplex operation when in the <b>Boost Mode</b>. The product may drop from full duplex operation into half/duplex mode while in <b>Boost Mode</b>.</p> <p>Contact CyberData support for additional information if needed.</p>





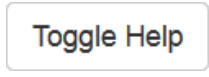
Table 2-7. Device Configuration Parameters (continued)

Web Page Item	Description
<b>Relay Settings</b>	
Activate Relay with DTMF Code ?	Activates the relay when the DTMF Activation Code is entered on the phone during a SIP call with the device. RFC2833 DTMF payload types are supported.
DTMF Activation Code ?	Activation code used to activate the relay when entered on a phone during a SIP call with the device. Activate Relay with DTMF Code must be enabled. Enter up to 25 digits  <b>Note:</b> While <a href="#">Enable DTMF Push to Talk</a> is enabled, the *, #, and 0 keys cannot be used in the DTMF activation code.
DTMF Activation Duration (in seconds) ?	The length of time (in seconds) during which the relay will be activated when the DTMF Activation Code is detected. Enter up to 5 digits.  <b>NOTE:</b> A DTMF activation duration of 0 will toggle the relay indefinitely or until the activation code is sent again
Play tone during DTMF Activation ?	When selected, the device will play a tone out of the speaker upon DTMF relay activation. The tone plays for the DTMF Activation Duration (in seconds).
Activate Relay During Ring ?	When selected, the relay will be activated for as long as the device is ringing. When Auto-Answer Incoming Calls is enabled, the device will not ring and this option does nothing.
Activate Relay During Night Ring ?	When selected, the relay will be activated as long as the Nightringer extension is ringing.
Activate Relay While Call Active ?	When selected, the relay will be activated as long as the SIP call is active.
Activate Relay on Button Press ?	When selected, the relay will be activated when the Call button is pressed.
Relay on Button Press Duration ?	The length of time (in seconds) during which the relay will be activated when the Call button is pressed. Enter up to 5 digits. A Relay on Button Press Duration value of 0 will pulse the relay once when the Call button is pressed.
<b>Clock Settings</b>	
Set Time with NTP Server on boot ?	When selected, the time is set with an external NTP server when the device restarts.
NTP Server ?	Use this field to set the address (in IPv4 dotted decimal notation or as a canonical name) for the NTP Server. This field can accept canonical names of up to 64 characters in length.
Posix Timezone String ?	See <a href="#">Section 2.4.5.1, "Time Zone Strings"</a> for information about how to use the Posix Timezone String to specify time zone and daylight savings time where applicable. Enter up to 63 characters.
Periodically sync time with server ?	When selected, the time is periodically updated with the NTP server at the configured interval below.

**Table 2-7. Device Configuration Parameters (continued)**

Web Page Item	Description
Time update period (in hours) ?	The time interval after which the device will contact the NTP server to update the time. Enter up to 4 digits.
Current Time	Allows you to input the current time. (6 character limit)
<b>Misc Settings</b>	
Device Name ?	Type the device name. Enter up to 25 characters.
Auto-Answer Incoming Calls ?	When selected, the device will automatically answer incoming calls. When Auto-Answer Incoming Calls is disabled, the device will play a ring tone (corresponds to Ring Tone on the Audiofiles page) out of the speaker until someone presses the Call button to answer the call or the caller disconnects before the call can be answered.
Button Lit When Idle ?	When selected, the Call button LED is illuminated while the device is idle (a call is not in progress).
Button Brightness (0-255) ?	The desired Call button LED brightness level. Acceptable values are 0-255, where 0 is the dimmest and 255 is the brightest. Enter up to three digits.
Play Ringback Tone ?	When selected, the device will play a ringback tone (corresponds to Ringback Tone on the Audiofiles page) out of the speaker while placing an outbound call. The Ringback Tone will play until the call is answered.
Enable Push to Talk ?	This option is for noisy environments. When enabled, the microphone will be muted normally. When the Call button is pressed and held, it will unmute the microphone and allow the operator to send audio back. Using Push to Talk prevents the operator from terminating a call by pressing the Call button. The call must be terminated by the phone user.
Enable DTMF Push to Talk ?	This option is for noisy environments. When enabled, in an active call, the remote phone can force receive only audio (setting the mic gain to max and muting the speaker) by pressing the * key.  Pressing the # key will force send only audio (setting the max speaker volume and muting the mic). Pressing the 0 key will restore full duplex operation with the normal microphone and speaker volume.
Prevent Call Termination ?	When this option is enabled, a call cannot be terminated using the call button.
Disable HTTPS (NOT recommended) ?	Disables the encrypted connection to the webpage. We do not recommend disabling HTTPS for security reasons.
	Click on the <b>Test Audio</b> button to do an audio test. When the <b>Test Audio</b> button is pressed, you will hear a voice message for testing the device audio quality and volume.

**Table 2-7. Device Configuration Parameters (continued)**

Web Page Item	Description
	<p>Click on the <b>Test Microphone</b> button to do a microphone test. When the <b>Test Microphone</b> button is pressed, the following occurs:</p> <ol style="list-style-type: none"> <li>1. The device will immediately start recording 3 seconds of audio.</li> <li>2. The device will beep (indicating the end of recording).</li> <li>3. The device will play back the recorded audio.</li> </ol>
	<p>Click on the <b>Test Relay</b> button to do a relay test.</p>
	<p>Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.</p>
	<p>Click on the <b>Reboot</b> button to reboot the system.</p>
	<p>Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> 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.</p>

**Note** You can change the **SIP Volume**, **Multicast Volume**, **Ring Volume**, **Sensor Volume**, and **Microphone Gain** without rebooting the device. You must save and reboot the device for other changes to take effect.

## 2.4.5.1 Time Zone Strings

The posix time zone string tells the internal date and time utilities how to handle daylight savings time for different time zones. [Table 2-8](#) shows some common strings.

**Table 2-8. Common Time Zone Strings**

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

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

[Table 2-9](#) shows a breakdown of the parts that constitute the following time zone string:

- ***CST6DST,M3.2.0/2:00:00,M11.1.0/2:00:00***

**Table 2-9. Time Zone String Parts**

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

Time Zone String Examples [Table 2-10](#) has some more examples of time zone strings.

**Table 2-10. Time Zone String Examples**

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

a. Tokyo does not use daylight savings time.

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

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

**Figure 2-18. Three or Four Character Time Zone Identifier**

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

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

World GMT Table [Table 2-11](#) has information about the GMT time in various time zones.

**Table 2-11. World GMT Table**

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



**Table 2-11. World GMT Table (continued)**

<b>Time Zone</b>	<b>City or Area Zone Crosses</b>
GMT+5	Islamabad, Karachi
GMT+6	Almaty, Dhaka
GMT+7	Bangkok, Jakarta
GMT+8	Hong Kong, Beijing
GMT+9	Tokyo, Osaka
GMT+10	Sydney, Melbourne, Guam
GMT+11	Magadan, Soloman Is.
GMT+12	Fiji, Wellington, Auckland

## 2.4.6 Configure the Network Parameters

1. Click the **Network** menu button to open the **Network** page (Figure 2-19).

Figure 2-19. Network Configuration Page

Home Device **Network** SIP Multicast Sensor Audiofiles Events DSR Autoprov Firmware

# CyberData Indoor Intercom

### Stored Network Settings

Addressing Mode:  Static  DHCP

Hostname:

IP Address:

Subnet Mask:

Default Gateway:

DNS Server 1:

DNS Server 2:

DHCP Timeout in seconds\*:

\* A value of -1 will retry forever

### VLAN Settings

VLAN ID (0-4095):

VLAN Priority (0-7):

Save Reboot Toggle Help

### Current Network Settings

IP Address: 192.168.70.38

Subnet Mask: 255.255.240.0

Default Gateway: 192.168.64.1

DNS Server 1: 192.168.65.20

DNS Server 2: 192.168.65.10



- On the **Network** page, enter values for the parameters indicated in [Table 2-12](#).

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

**Table 2-12. Network Configuration Parameters**

Web Page Item	Description
<b>Stored Network Settings</b>	
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 <a href="#">Section 2.4.1, "Factory Default Settings"</a> for factory default settings. Be sure to click <b>Save</b> and <b>Reboot</b> 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.
<b>VLAN Settings</b>	
VLAN ID (0-4095) ?	Specify the IEEE 802.1Q VLAN ID number. Enter up to 4 digits. <b>Note:</b> The device supports 802.1Q VLAN tagging support. The switch port connected to the device will need to be in "trunking mode" for the VLAN tags to propagate.
VLAN Priority (0-7) ?	Specify the IEEE 802.1p VLAN priority level. Enter 1 digit. A value of 0 may cause the VLAN ID tag to be ignored.
<b>Current Network Settings</b>	
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 2-12. Network Configuration Parameters (continued)**

Web Page Item	Description
	<p>Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.</p>
	<p>Click on the <b>Reboot</b> button to reboot the system.</p>
	<p>Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> 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.</p>

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

## 2.4.7 Configure the SIP Parameters

1. Click on the **SIP** menu button to open the **SIP** page (Figure 2-20).

**Figure 2-20. SIP Configuration Page**

The screenshot shows the SIP Configuration page for a CyberData Indoor Intercom. The page is titled "CyberData Indoor Intercom" and has a navigation bar at the top with tabs for Home, Device, Network, SIP, Multicast, Sensor, Audiofiles, Events, DSR, Autopro, and Firmware. The SIP tab is selected.

The page is divided into several sections:

- SIP Settings:**
  - Enable SIP operation:
  - Register with a SIP Server:
  - Use Cisco SRST:
  - Primary SIP Server: 10.0.0.253
  - Primary SIP User ID: 199
  - Primary SIP Auth ID: 199
  - Primary SIP Auth Password: [masked]
  - Backup SIP Server 1: [empty]
  - Backup SIP User ID 1: [empty]
  - Backup SIP Auth ID 1: [empty]
  - Backup SIP Auth Password 1: [masked]
  - Backup SIP Server 2: [empty]
  - Backup SIP User ID 2: [empty]
  - Backup SIP Auth ID 2: [empty]
  - Backup SIP Auth Password 2: [masked]
  - Remote SIP Port: 5060
  - Local SIP Port: 5060
  - Outbound Proxy: [empty]
  - Outbound Proxy Port: 0
  - Disable rport Discovery:
  - Re-registration Interval (in seconds): 360
  - Unregister on Boot:
  - Keep Alive Period: 0
- Nightringer Settings:**
  - Enable Nightringer:
  - SIP Server: 10.0.0.253
  - Remote SIP Port: 5060
  - Local SIP Port: 5061
  - Outbound Proxy: [empty]
  - Outbound Proxy Port: 0
  - User ID: 241
  - Authenticate ID: 241
  - Authenticate Password: [masked]
  - Re-registration Interval (in seconds): 360
- Dial Out Settings:**
  - Dial out Extension: 204
  - Extension ID: id204
- Call Disconnection:**
  - Terminate Call after delay: 0
- RTP Settings:**
  - RTP Port (even): 10500

At the bottom right of the page, there are three buttons: Save, Reboot, and Toggle Help.

2. On the **SIP** page, enter values for the parameters indicated in [Table 2-13](#).

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













**Table 2-13. SIP Configuration Parameters**

Web Page Item	Description
<b>SIP Settings</b>	
Enable SIP Operation ?	When enabled, the device will transmit, receive, and process SIP messages according to the configured SIP settings below.
Register with a SIP Server ?	When enabled, the device will attempt to register to the configured SIP Server(s) on this page. To configure the device to send and receive point-to-point SIP calls, enable <b>SIP Operation</b> and disable <b>Register with a SIP Server</b> (see <a href="#">Section 2.4.7.2, "Point-to-Point Configuration"</a> ).
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 ?	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.
Backup SIP Server 2 ?	Enter a second backup SIP server address as an IPv4 address in dotted decimal notation or a fully qualified domain name. This parameter also becomes the host portion of the SIP-URI for the device's extension on the second backup SIP server. This field can accept entries of up to 255 characters in length.

**Table 2-13. SIP Configuration Parameters (continued)**




Web Page Item	Description
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.
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.
<b>RTP Settings</b>	
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.
<b>Nightringer Settings</b>	
Enable Nightringer ?	When Nightringer is enabled, the device will attempt to register a second extension with the SIP server. Any calls made to this extension will play a ringtone (corresponds to <b>Night Ring</b> on the <b>Audiofiles</b> page). By design, it is not possible to answer a call to the Nightringer extension.

**Table 2-13. SIP Configuration Parameters (continued)**

Web Page Item	Description
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 <b>Local SIP Port</b> 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.
<b>Dial Out Settings</b>	
Dial Out Extension 	Specify the extension the device will call when someone presses the Call button. Enter up to 64 alphanumeric characters.  <b>Note:</b> For information about dial-out extension strings and DTMF tones, see <a href="#">Section 2.4.7.1, "Dial Out Extension Strings and DTMF Tones (using rfc2833)"</a> .
Extension ID 	A Caller identification string added to outbound calls. Enter up to 64 alphanumeric characters.
<b>Call Disconnection</b>	
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.



**Table 2-13. SIP Configuration Parameters (continued)**

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

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

**Note** For specific server configurations, go to the following website address:  
<http://www.cyberdata.net/support/server/index.html>

#### 2.4.7.1 Dial Out Extension Strings and DTMF Tones (using rfc2833)

On the **SIP Configuration Page**, dial out extensions support the addition of comma delimited pauses and sending additional DTMF tones (using rfc2833). The first comma will pause three seconds after a call is first established with a remote device. Subsequent commas will pause for 2 seconds. A pause of one second will be sent after each numerical digit.

**Table 2-14. Examples of Dial-Out Extension Strings**

Extension String	Resulting Action
302	Dial out extension 302 and establish a call
302,2	Dial out extension 302 and establish a call, wait 3 seconds then send the DTMF tone '2'
302,25,,,4,,1	Dial out extension 302 and establish a call, wait 3 seconds then send the DTMF tone '2', send out DTMF tone 5, wait 6 seconds, send out DTMF tone 4, wait 4 seconds, send out DTMF tone 1

**Note** The maximum number of total characters in the dial-out field is 64.

## 2.4.7.2 Point-to-Point Configuration

When the device is set to not register with a SIP server (see [Figure 2-21](#)), it is possible to set the device to dial out to a single endpoint.

In this case, the dial-out extension should be the IP address of the remote device. The device can also receive Point-to-Point calls. The delayed DTMF functionality is available in the Point-to-Point Mode.

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

**Figure 2-21. SIP Page Set to Point-to-Point Mode**

The screenshot shows the 'SIP' configuration page for a CyberData Indoor Intercom. The page has a navigation bar at the top with tabs for Home, Device, Network, SIP (selected), Multicast, Sensor, Audiofiles, Events, DSR, Autopro, and Firmware. The main title is 'CyberData Indoor Intercom'. The page is divided into several sections:

- SIP Settings:**
  - Enable SIP operation:
  - Register with a SIP Server:  (indicated by a red arrow from the caption below)
  - Use Cisco SRST:
  - Primary SIP Server: 10.0.0.253
  - Primary SIP User ID: 199
  - Primary SIP Auth ID: 199
  - Primary SIP Auth Password: [masked]
  - Backup SIP Server 1: [empty]
  - Backup SIP User ID 1: [empty]
  - Backup SIP Auth ID 1: [empty]
  - Backup SIP Auth Password 1: [masked]
  - Backup SIP Server 2: [empty]
  - Backup SIP User ID 2: [empty]
  - Backup SIP Auth ID 2: [empty]
  - Backup SIP Auth Password 2: [masked]
  - Remote SIP Port: 5060
  - Local SIP Port: 5060
  - Outbound Proxy: [empty]
  - Outbound Proxy Port: 0
  - Disable rport Discovery:
  - Re-registration Interval (in seconds): 360
  - Unregister on Boot:
  - Keep Alive Period: 0
- Nightringer Settings:**
  - Enable Nightringer:
  - SIP Server: 10.0.0.253
  - Remote SIP Port: 5060
  - Local SIP Port: 5061
  - Outbound Proxy: [empty]
  - Outbound Proxy Port: 0
  - User ID: 241
  - Authenticate ID: 241
  - Authenticate Password: [masked]
  - Re-registration Interval (in seconds): 360
- Dial Out Settings:**
  - Dial out Extension: 204
  - Extension ID: id204
- Call Disconnection:**
  - Terminate Call after delay: 0
- RTP Settings:**
  - RTP Port (even): 10500

At the bottom right, there are buttons for 'Save', 'Reboot', and 'Toggle Help'.

Device is set to NOT register with a SIP server

### 2.4.7.3 Delayed DTMF

On the **SIP Configuration** page the dial out extension supports the addition of comma delimited pauses and sending additional DTMF tones (using rfc2833). The first comma will pause three seconds after a call is first established with a remote device. Subsequent commas will pause for 2 seconds. A pause of one second will be sent after each numerical digit.

**Table 2-15. Examples of Dial-Out Extension Strings**

<b>Extension String</b>	<b>Resulting Action</b>
302	Dial out extension 302 and establish a call
302,2	Dial out extension 302 and establish a call, wait 3 seconds then send the DTMF tone '2'
302,25,,,4,,1	Dial out extension 302 and establish a call, wait 3 seconds then send the DTMF tone '2', send out DTMF tone 5, wait 6 seconds, send out DTMF tone 4, wait 4 seconds, send out DTMF tone 1

**Note** The maximum number of total characters in the dial-out field is 25.

## 2.4.8 Configure the Multicast Parameters

The Multicast Configuration page allows the device to join up to ten paging zones for receiving ulaw/alaw encoded RTP audio streams.

A paging zone can consist of one or many CyberData multicast group-enabled products. There is no limit to how many speakers 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 3. The device supports simultaneous SIP and Multicast.

1. Click on the **Multicast** menu button to open the **Multicast** page. See [Figure 2-22](#).

**Figure 2-22. Multicast Configuration Page**

Home Device Network SIP **Multicast** Sensor Audiofiles Events DSR Autoprov Firmware

# CyberData Indoor Intercom

## Multicast Settings

Enable Multicast Operation:

Priority	Address	Port	Name	Beep
9	239.168.3.10	11000	Emergency	<input type="checkbox"/>
8	239.168.3.9	10000	MG8	<input type="checkbox"/>
7	239.168.3.8	9000	MG7	<input type="checkbox"/>
6	239.168.3.7	8000	MG6	<input type="checkbox"/>
5	239.168.3.6	7000	MG5	<input type="checkbox"/>
4	239.168.3.5	6000	MG4	<input type="checkbox"/>
3	239.168.3.4	5000	MG3	<input type="checkbox"/>
2	239.168.3.3	4000	MG2	<input type="checkbox"/>
1	239.168.3.2	3000	MG1	<input type="checkbox"/>
0	239.168.3.1	2000	Background Music	<input type="checkbox"/>



SIP calls are considered priority 4.5  
 Port range can be from 2000-65535  
 Ports must be even numbers  
 Priority 9 is the highest and 0 is the lowest  
 A higher priority audio stream will always supercede a lower one  
 Priority 9 streams will play at maximum volume  
 \* You need to reboot for changes to take effect

Save Reboot

2. On the **Multicast** page, enter values for the parameters indicated in [Table 2-16](#).

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

**Table 2-16. Multicast Configuration Parameters**

Web Page Item	Description
Enable Multicast Operation	Enables or disables multicast operation.
Priority	Indicates the priority for the multicast group. Priority <b>9</b> is the highest (emergency streams). <b>0</b> is the lowest (background music). SIP calls are considered priority <b>4.5</b> . See <a href="#">Section 2.4.8.1, "Assigning Priority"</a> 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]). <b>Note:</b> 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).
Beep	When selected, the device will play a beep before multicast audio is sent.
	Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.
	Click on the <b>Reboot</b> button to reboot the system.

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

### 2.4.8.1 Assigning Priority

The device will prioritize simultaneous audio streams according to their priority in the list.

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**.

During priority **9** multicast streams, the volume is set to maximum.

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

Ringtones and Nightringtones

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

## 2.4.9 Configure the Sensor Configuration Parameters

The door sensor (pins 5 and 6) on the header can be used to monitor a door's open or closed state. There is an option on the **Sensor** page to trigger on an open or short condition on these pins. The door sensor alarm will be activated when the **Door Open Timeout** parameter has been met.

The intrusion sensor is an optical sensor installed on the Intercom board and will be activated when the Intercom is removed from the case.

Each sensor can trigger up to five different actions:

- Flash the LED until the sensor is deactivated (roughly 10 times/second)
- Activate the relay until the sensor is deactivated
- Loop an audio file out of the Intercom speaker until the sensor is deactivated
- Call an extension and establish two way audio
- Call an extension and play a pre-recorded audio file

**Note** Calling a preset extension can be set up as a point-to-point call, but currently can't send delayed DTMF tones.

1. Click **Sensor** menu button to open the **Sensor** page (Figure 2-23).

Figure 2-23. Sensor Configuration Page

The screenshot displays the 'Sensor' configuration page for the CyberData Indoor Intercom. The page has a blue header with a navigation menu containing: Home, Device, Network, SIP, Multicast, Sensor (selected), Audiofiles, Events, DSR, Autopro, and Firmware. The main title is 'CyberData Indoor Intercom'. Below the title, there are two columns of settings:

- Door Sensor Settings:**
  - Door Sensor Normally Closed:  Yes  No
  - Door Open Timeout (in seconds):
  - Flash Button LED:
  - Activate Relay:
  - Play Audio Locally:
  - Make call to extension:
  - Dial Out Extension:
  - Dial Out ID:
  - Play recorded audio:
- Intrusion Sensor Settings:**
  - Flash Button LED:
  - Activate Relay:
  - Play Audio Locally:
  - Make call to extension:
  - Dial Out Extension:
  - Dial Out ID:
  - Play recorded audio:

At the bottom of the page, there are three buttons: 'Save', 'Reboot', and 'Toggle Help'. Below these, there are two buttons: 'Test Door Sensor' and 'Test Intrusion Sensor'.





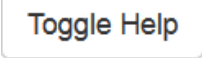
2. On the **Sensor** page, enter values for the parameters indicated in [Table 2-17](#).

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

**Table 2-17. Sensor Configuration Parameters**

Web Page Item	Description
<b>Door Sensor Settings</b>	
Door Sensor Normally Closed ?	Select the inactive state of the door sensor. The door sensor is also known as the Sense Input on the device's terminal block.
Door Open Timeout (in seconds) ?	The time (in seconds) the device will wait before it performs an action when the on-board door sensor is activated. The action(s) performed are based on the configured Door Sensor Settings below. Enter up to 5 digits.
Flash Button LED ?	When selected, the Call button LED will flash until the on-board door sensor is deactivated (roughly 10 times/second).
Activate Relay ?	When selected, the device's on-board relay will be activated until the on-board door sensor is deactivated.
Play Audio Locally ?	When selected, the device will loop an audio file out of the speaker until the door sensor is deactivated.
Make call to extension ?	When selected, the device will call an extension when the on-board door sensor is activated. Use the <b>Dial Out Extension</b> field below to specify the extension the device will call.
Dial Out Extension ?	Specify the extension the device will call when the on-board door sensor is activated. Enter up to 64 alphanumeric characters.
Dial Out ID ?	An additional Caller identification string added to outbound calls. Enter up to 64 alphanumeric characters.
Play recorded audio ?	When selected, the device will call the <b>Dial Out Extension</b> and play an audio file to the phone answering the SIP call (corresponds to <b>Door Ajar</b> on the <b>Audiofiles</b> page).
<b>Intrusion Sensor Settings</b>	
Flash Button LED ?	When selected, the Call button LED will flash until the intrusion sensor is deactivated (roughly 10 times/second).
Activate Relay ?	When selected, the device's on-board relay will be activated until the intrusion sensor is deactivated.
Play Audio Locally ?	When selected, the device will loop an audio file out of the speaker until the intrusion sensor is deactivated.
Make call to extension ?	When selected, the device will call an extension when the intrusion sensor is activated. Use the <b>Dial Out Extension</b> field below to specify the extension the device will call.
Dial Out Extension ?	Specify the extension the device will call when the intrusion sensor is activated. Enter up to 64 alphanumeric characters.
Dial Out ID ?	An additional Caller identification string added to outbound calls. Enter up to 64 alphanumeric characters.
Play recorded audio ?	When selected, the device will call the <b>Dial Out Extension</b> and play an audio file (corresponds to <b>Intrusion Sensor Triggered</b> on the <b>Audiofiles</b> page) to the phone answering the SIP call when the intrusion sensor is activated.

**Table 2-17. Sensor Configuration Parameters (continued)**

Web Page Item	Description
	Click the <b>Test Door Sensor</b> button to test the door sensor.
	Click the <b>Test Intrusion Sensor</b> button to test the Intrusion sensor.
	Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.
	Click on the <b>Reboot</b> button to reboot the system.
	Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> button, and you will see a question mark (?) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.

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



## 2.4.10 Configure the Audio Configuration Parameters

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 Intercom.

1. Click on the **Audiofiles** menu button to open the **Audiofiles** page (Figure 2-24).

**Figure 2-24. Audiofiles Configuration Page**



**Figure 2-25. Audiofiles Page**



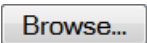



2. On the **Audiofiles** page, enter values for the parameters indicated in [Table 2-18](#).

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

**Table 2-18. Audiofiles Configuration Parameters**

Web Page Item	Description
Available Space	Shows the space available for the user to save custom audio files if they want to change the message when the door or sensor is triggered.
0-9	<p>The name of the audio configuration option is the same as the spoken audio that plays on the board (24 character limit).</p> <p>'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."</p>

**Table 2-18. Audiofiles Configuration Parameters (continued)**

Web Page Item	Description
0-9	'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)
Audiotest	Corresponds to the message " <b><i>This is the CyberData IP speaker test message...</i></b> " (24 character limit)
Page tone	Corresponds to a simple tone used for beep on initialization and beep on page (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).
Ringback tone	This is the ringback tone that plays when calling a remote extension (24 character limit).
Ring tone	This is the tone that plays when set to ring when receiving a call (24 character limit).
Intrusion Sensor Triggered	Corresponds to the message "Intrusion Sensor Triggered" (24 character limit).
Door Ajar	Corresponds to the message "Door Ajar" (24 character limit).
Night Ring	Specifies the ringtone for nightring. By default this parameter uses the same audio file that is selected for the <b>Ring Tone</b> parameter.
	Click on the <b>Browse</b> button to navigate to and select an audio file.
	The <b>Play</b> button will play that audio file.
	The <b>Delete</b> button will delete any user uploaded audio and restore the stock audio file.
	The <b>Save</b> button will download a new user audio file to the board once you've selected the file by using the <b>Browse</b> button. The <b>Save</b> button will delete any pre-existing user-uploaded audio files.

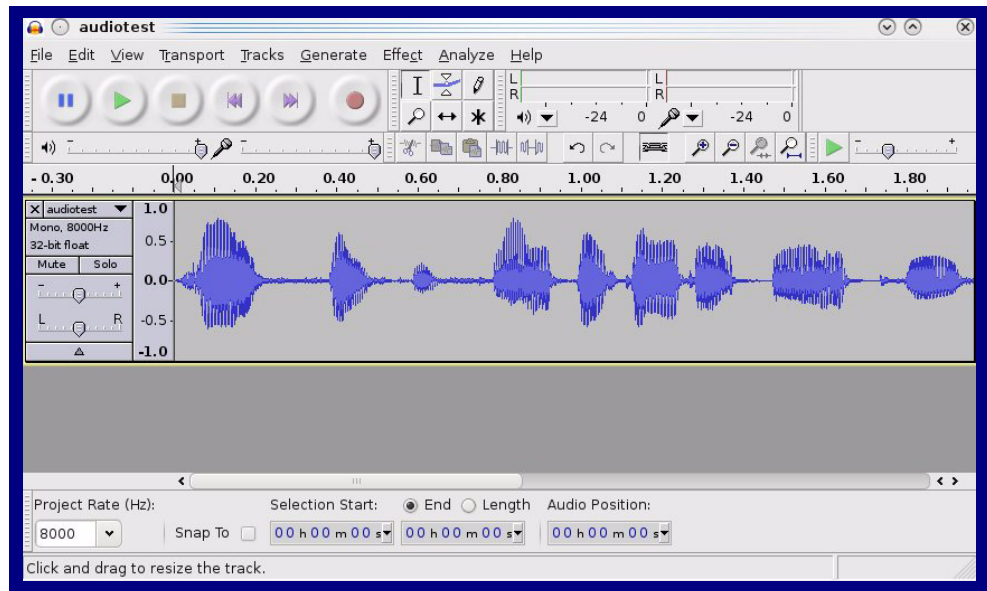
### 2.4.10.1 User-created Audio Files

User created audio files should be saved in the following format:

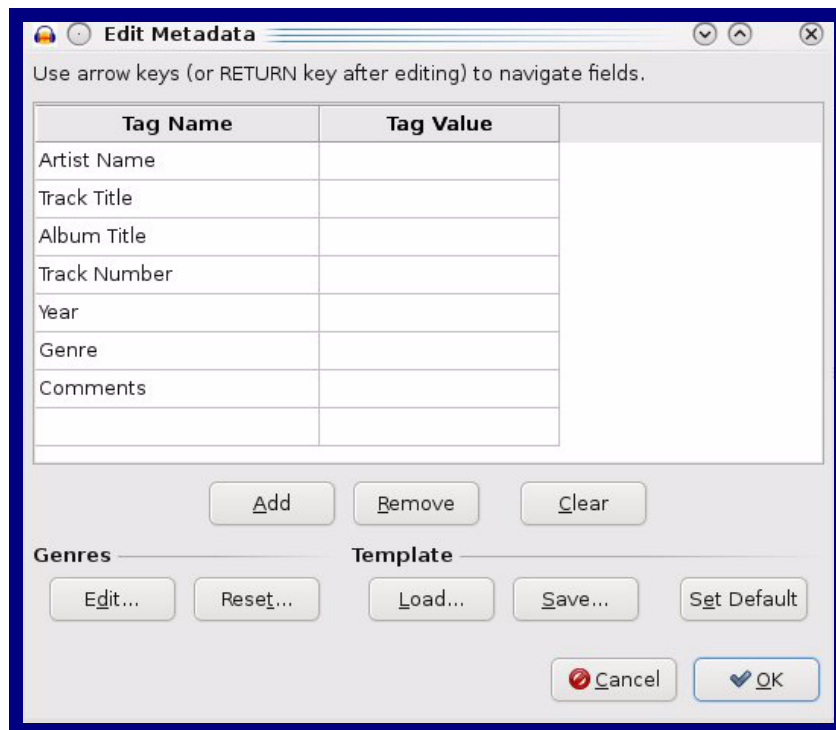
RIFF (little-endian) data, WAVE audio, Microsoft PCM, 16 bit, mono 8000 Hz

You can use the free utility *Audacity* to convert audio files into this format. See [Figure 2-26](#) through [Figure 2-28](#).

**Figure 2-26. Audacity 1**



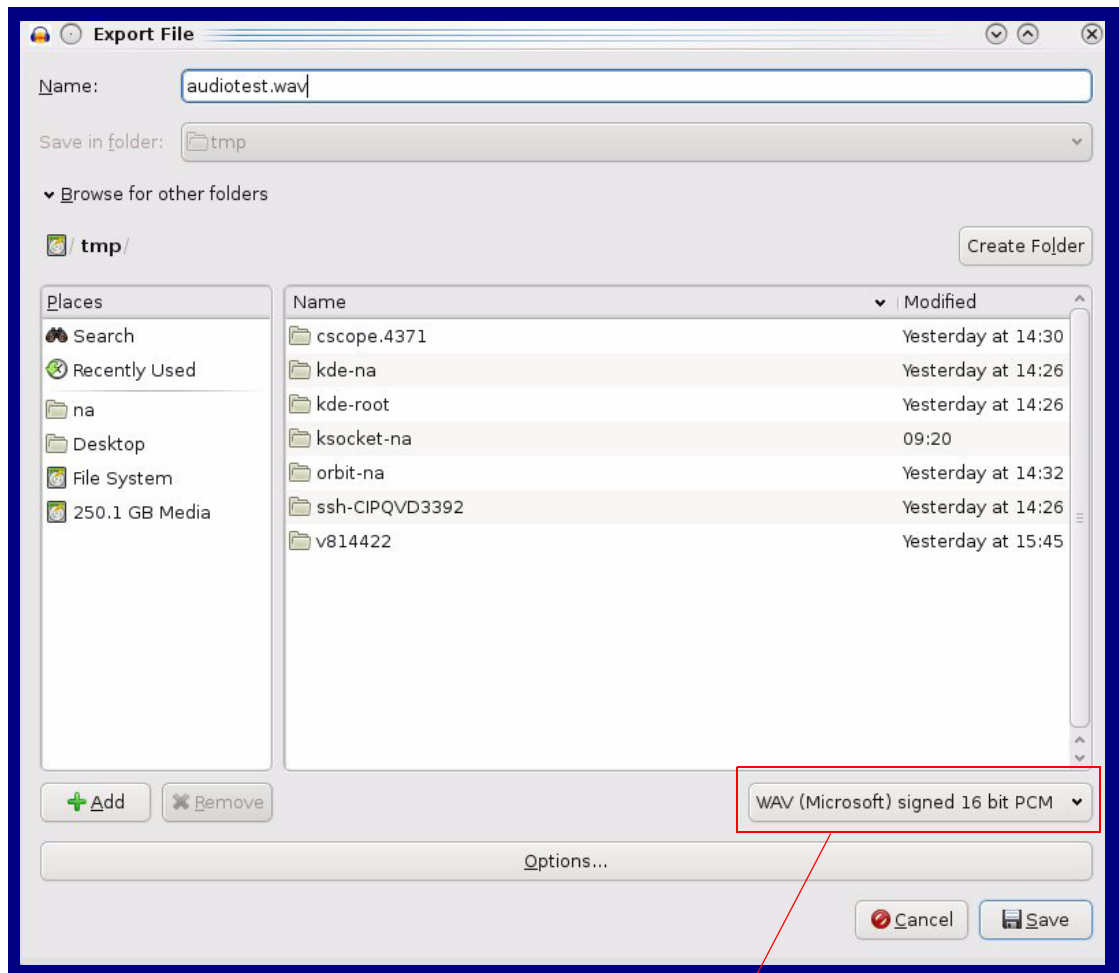
**Figure 2-27. Audacity 2**



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

- **WAV (Microsoft) signed 16 bit PCM.**

**Figure 2-28. WAV (Microsoft) signed 16 bit PCM**



WAV (Microsoft) signed 16 bit PCM

## 2.4.11 Configure the Events Parameters

The **Events** page specifies a remote server that can be used to receive HTTP POST events when actions take place on the board.

1. Click on the **Events** menu button to open the **Events** page (Figure 2-29).

**Figure 2-29. Event Configuration Page**

Home Device Network SIP Multicast Sensor Audiofiles **Events** DSR Autopro Firmware

# CyberData Indoor Intercom

Enable Event Generation:

### Events

- Enable Button Events:
- Enable Call Start Events:
- Enable Call Terminated Events:
- Enable Relay Activated Events:
- Enable Relay Deactivated Events:
- Enable Ring Events:
- Enable Night Ring Events:
- Enable Multicast Start Events:
- Enable Multicast Stop Events:
- Enable Power On Events:
- Enable Sensor Events:
- Enable Remote Relay Events:
- Enable Security Events:
- Enable 60 Second Heartbeat:

[Check All](#) [Uncheck All](#)

### Event Server

Server IP Address:

Server Port:

Server URL:






2. On the **Events** page, enter values for the parameters indicated in [Table 2-19](#).

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

**Table 2-19. Events Configuration Parameters**

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.
<b>Events</b>	
Enable Button Events ?	When selected, the device will report Call button presses.
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 Ring Events ?	When selected, the device will report when it starts ringing upon an incoming SIP call. A Ring Event will not be generated when <b>Auto-Answer Incoming Calls</b> is enabled on the <b>Device</b> page.
Enable Night Ring Events ?	When selected, the device will report when it starts ringing upon an incoming SIP call to the Nightringer extension. As a reminder, the Nightringer extension always rings upon an incoming SIP call and it is not possible to alter this behavior.
Enable Multicast Start Events ?	When selected, the device will report when the device starts playing a multicast audio stream.
Enable Multicast Stop Events ?	When selected, the device will report when the device stops playing a multicast audio stream.
Enable Power On Events ?	When selected, the device will report when it boots.
Enable Sensor Events ?	When selected, the device will report when the on-board sensor is activated.
Enable Remote Relay Events ?	When selected, the device will report when the remote relay (DSR) is activated.
Enable Security Events ?	When enabled, the device will report when the intrusion sensor is activated.
Enable 60 Second Heartbeat Events ?	When enabled, the device will report a Heartbeat event every 60 seconds. SIP registration is not required to generate Heartbeat events.
Check All	Click on <b>Check All</b> to select all of the events on the page.
Uncheck All	Click on <b>Uncheck All</b> to de-select all of the events on the page.
<b>Event Server</b>	
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.

**Table 2-19. Events Configuration Parameters(continued)**

Web Page Item	Description
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.
	Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.
	Click on the <b>Reboot</b> button to reboot the system.
	Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> button, and you will see a question mark (  ) appear next to some of the web page items. Move the mouse pointer to hover over a question mark to see a short description of a specific web page item.

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



### 2.4.11.1 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</event>
<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</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>RELAY_DEACTIVATED</event>
</cyberdata>
```

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

## 2.4.12 Configure the Door Strike Relay

The Door Strike Relay (DSR) is a network device designed to control an electronic door strike. The DSR is meant to be used as a replacement for (or an addition to) the on-board relay. In addition to being a drop-in 12 Amp relay, the DSR can monitor and record when the door is open or closed.

The DSR can be configured to trigger in the following ways: on the entry of a DTMF code, during different call events, manually through the web interface, or by using a Windows application.

1. Click on the **DSR** menu button to open the **DSR** page (Figure 2-30).

Figure 2-30. DSR Page

Home Device Network SIP Multicast Sensor Audiofiles Events **DSR** Autopro Firmware

# CyberData Indoor Intercom

### Remote Relay Settings

**Not associated with any DSRs**

Activate Remote Relay with DTMF code:

DTMF Activation Code:

DTMF Activation Duration (in seconds):

Activate Remote Relay During Ring:

Activate Remote Relay During Night Ring:

Activate Remote Relay While Call Active:

Activate Remote Relay On Button Press:

Remote Relay On Button Press Duration (in seconds):

Listen Port for Remote Relay Status:

### Remote Door Sensor Settings

Door Open Timeout (in seconds):

Flash Button LED:

Activate Local Relay:

Play Audio Locally:

Make call to extension:

Play recorded audio:

Dial Out Extension:

Dial Out ID:

### Discovered Remote Relays

Product Type	IP Address	MAC Address	Serial Number	Name	Version		
DoorLock	10.10.1.147	00:20:F7:02:6C:F7	270000001	LOCK270000001	V1.6A	<input type="button" value="View"/>	<input type="button" value="Associate"/>
DoorLock	10.10.1.66	00:20:F7:02:A7:A2	270000021	LOCK270000021	V1.9A	<input type="button" value="View"/>	<input type="button" value="Associate"/>
DoorLock	10.10.0.242	00:20:F7:02:A7:A3	270000022	LOCK270000022	V1.9A	<input type="button" value="View"/>	<input type="button" value="Associate"/>
DoorLock	10.10.1.102	00:20:F7:02:A7:E2	270000078	LOCK270000078	V1.9A	<input type="button" value="View"/>	<input type="button" value="Associate"/>

Cache age: 01:08







2. On the **DSR** page, enter values for the parameters indicated in [Table 2-20](#).

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





**Table 2-20. DSR Configuration Parameters**

Web Page Item	Description
<b>Remote Relay Settings</b>	The settings in this section will activate an associated door strike relay.
Activate Relay with DTMF Code ?	Activates the remote relay (DSR) when the DTMF Activation Code is entered on the phone during a SIP call with the device. RFC2833 DTMF payload types are supported.
DTMF Activation Code ?	Activation code used to activate the remote relay (DSR) when entered on a phone during a SIP call with the device. Activate Remote Relay with DTMF Code must be enabled. Enter up to 25 digits.  <b>Note:</b> While <a href="#">Enable DTMF Push to Talk</a> is enabled, the *, #, and 0 keys cannot be used in the DTMF activation code.
DTMF Activation Duration (in seconds) ?	The length of time (in seconds) during which the remote relay (DSR) will be activated when the DTMF Activation Code is detected. Enter up to 5 digits.
Activate Remote Relay During Ring ?	When selected, the remote relay (DSR) will be activated for as long as the device is ringing. When Auto-Answer Incoming Calls is enabled, the device will not ring and this option does nothing. When selected, the network relay will be activated for as long as the call is active.
Activate Remote Relay During Night Ring ?	When selected, the remote relay (DSR) will be activated as long as the Nightringer extension is ringing.
Activate Remote Relay While Call Active ?	When selected, the remote relay (DSR) will be activated as long as the call is active.
Activate Remote Relay on Button Press ?	When selected, the remote relay (DSR) will be activated when the Call Button is pressed.
Remote Relay on Button Press Duration (in seconds) ?	The length of time (in seconds) during which the remote relay (DSR) will be activated when the Call button is pressed. Enter up to 5 digits. A Remote Relay on Button Press Duration value of 0 will pulse the remote relay (DSR) once when the Call button is pressed.
Listen Port for Remote Relay Status ?	Specify the port to listen for remote relay (DSR) status packets.
<b>Remote Door Sensor Settings</b>	
Door Open Timeout (in seconds) ?	The time (in seconds) the device will wait before it performs an action when the remote (DSR) door sensor is activated. The action(s) performed are based on the configured Remote Door Sensor Settings below.
Flash Button LED ?	When selected, the Call button LED will flash until the remote (DSR) door sensor is deactivated (roughly 10 times/second).
Activate Local Relay ?	When selected, the device's on-board relay will be activated until the remote (DSR) door sensor is deactivated.
Play Audio Locally ?	When selected, the device will loop an audio file out of the speaker until the remote (DSR) door sensor is deactivated.

**Table 2-20. DSR Configuration Parameters (continued)**

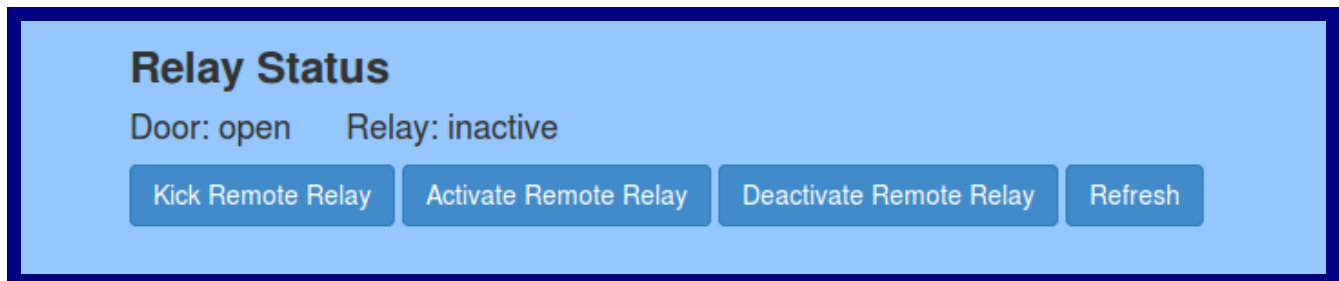
Web Page Item	Description
Make call to extension ?	When selected, the device will call an extension when the remote (DSR) door sensor is activated. Use the 'Dial Out Extension' field below to specify the extension the device will call.
Play recorded audio ?	When selected, the device will call the Dial Out Extension and play an audio file to the phone answering the SIP call (corresponds to Door Ajar on the Audiofiles page) when the remote (DSR) door sensor is activated.
Dial Out Extension ?	Specify the extension the device will call when the remote (DSR) door sensor is activated. Enter up to 64 alphanumeric characters.
Dial Out ID ?	An additional Caller identification string added to outbound calls. Enter up to 64 alphanumeric characters.
	Click the <b>Save</b> button to save your configuration settings. <b>Note:</b> You need to reboot for changes to take effect.
	Click on the <b>Reboot</b> button to reboot the system.
	Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> 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.
<b>Discovered Remote Relays</b>	The <b>Discovered Remote Relays</b> section lists all of the networked door strike relays on the network. To associate your device with a door strike relay, click on the <b>Associate</b> button. This action allows the user to configure the door strike relay. Keep in mind that a device may only be associated with one door strike relay.
Product Type	Displays the product type of the remote relay.
IP Address	Displays the IP address of the remote relay.
MAC Address	Displays the MAC address of the remote relay.
Serial Number	Displays the serial number of the remote relay.
Name	Displays the name of the remote relay.
Version	Displays the version of the remote relay.
	Use this button to search for and find any remote relays that are available on the network.
	Use this button to view the settings of a remote relay that has been "discovered" after pressing the <b>Discover</b> button.
	Use this button to associate the remote relay with the device. Only one relay may be associated with a device.

**Table 2-20. DSR Configuration Parameters (continued)**

Web Page Item	Description
<b>Relay Status</b>	<b>Note:</b> The <b>Relay Status</b> section and settings (Figure 2-31) only appear on the webpage when there is an associated door strike relay.
Door	Shows the status of the door.
Relay	Shows the status of the remote relay.
	Click on the <b>Kick Remote Relay</b> button to activate the remote relay for a specified time. The time is equal to the DTMF timeout.
	Click on the <b>Activate Remote Relay</b> button to activate the remote relay until the <b>Deactivate Remote Relay</b> button is pressed.
	Click on the <b>Deactivate Remote Relay</b> button to deactivate the remote relay.
	Click on the <b>Refresh</b> button to refresh the web page and accurately display the status of the remote relay (active/inactive) and door (open/closed).

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

**Figure 2-31. Relay Status Section**



## 2.4.13 Configure the Device (on the DSR page)

1. Click the **View** button on the **DSR** page to open the **Configure Device** page (Figure 2-32).

**Figure 2-32. DSR Page Configure Device Page**

### Configure Device

<b>Serial Number</b>	<input type="text" value="270000002"/>	<input type="button" value="Refresh"/>
<b>MAC Address</b>	<input type="text" value="00:20:F7:02:6C:F8"/>	<input type="button" value="Get Log"/>
<b>Version</b>	<input type="text" value="V1.2A"/>	<input type="button" value="Clear Log"/>
<b>Device Name</b>	<input type="text" value="LOCK270000003"/>	<input type="button" value="Reboot"/>
<b>Addressing Mode</b>	<input type="radio"/> Static <input checked="" type="radio"/> DHCP	<input type="button" value="Set Time"/>
<b>IP Address:</b>	<input type="text" value="192.168.70.74"/>	<input type="button" value="Save Changes"/>
<b>Subnet Mask:</b>	<input type="text" value="255.255.240.0"/>	<input type="button" value="Cancel"/>
<b>Default Gateway:</b>	<input type="text" value="192.168.64.1"/>	
<b>Command Port:</b>	<input type="text" value="59999"/>	
<b>Send Events</b>	<input checked="" type="radio"/> Off <input type="radio"/> On	
<b>Event IP Address:</b>	<input type="text" value="192.168.79.255"/>	
<b>Event Port:</b>	<input type="text" value="49999"/>	
<b>Energize Time:</b>	<input type="text"/>	
<b>DST</b>	<input checked="" type="radio"/> Off <input type="radio"/> On	
<b>DST Start:</b>	<input type="text" value="M3.2.0/02.00.00"/>	
<b>DST End:</b>	<input type="text" value="M11.1.0/02.00.00"/>	
<b>Current Time:</b>	<input type="text" value="17:45:26 08182014"/>	
<b>Encryption:</b>	<input checked="" type="radio"/> None <input type="radio"/> AES-256	
<b>Encryption Key:</b>	<input type="text"/>	
<b>Door State</b>	<input type="text" value="open"/>	
<b>Relay State</b>	<input type="text" value="inactive"/>	
<b>Button State</b>	<input type="text" value="inactive"/>	
<b>LED</b>	<input type="text" value="red"/>	
<b>Alarm State</b>	<input type="text" value="alarm"/>	
<b>JP4, 6, 9, 10</b>	<input type="text" value="0000"/>	
<input type="button" value="Browse..."/>	No file selected.	<input type="button" value="Upgrade"/>












2. On the **Configure Device** page, enter values for the parameters indicated in [Table 2-21](#).

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

**Table 2-21. DSR Page Configure Device Parameters**

Web Page Item	Description
Serial Number	Displays the serial number of the door strike relay.
MAC Address	Displays the mac address of the door strike relay.
Version	Displays the firmware version of the door strike relay.
Device Name	Displays the name of the door strike relay. The default name is "LOCK," followed by the 9 digit ASCII serial number. The maximum name length is 13 characters. The unit will always respond to its default name.
Addressing Mode	Determines whether an IP address will be manually assigned through Static mode or dynamically assigned through a DHCP server.
IP Address	Displays the IP address of the door strike relay.
Subnet Mask	Displays the subnet mask of the door strike relay.
Default Gateway	Displays the default gateway of the door strike relay.
Command Port	This shows the port on which the door strike relay sends status packets to the device (defaults to 49999).
Send Events	When enabled, events can be sent to the associated device.
Event IP Address	The IP address of the associated device.
Event Port	This is the port by which the door strike relay receives commands (defaults to 59999).
Energize Time	This is the number of seconds that the relay will be energized.
DST	Allows you to either enable or disable the Daylight Savings Time feature.
DST Start	Sets the Daylight Savings Time starting time in the following format: <b>M3.2.0/02:00:00</b> <b>M3</b> is the third month (March). <b>.2</b> is the second occurrence of the day in the month. <b>.0</b> is Sunday. <b>/02:00:00</b> is the time. <b>Note:</b> When the occurrence is set to <b>5</b> , the final occurrence of the day in the specified month is used.
DST End	Sets the Daylight Savings Time ending time in the following format: <b>M11.1.0/02:00:00</b> <b>M11</b> is the eleventh month (November). <b>.1</b> is the first occurrence of the day in the month. <b>.0</b> is Sunday. <b>/02:00:00</b> is the time. <b>Note:</b> When the occurrence is set to <b>5</b> , the final occurrence of the day in the specified month is used.

**Table 2-21. DSR Page Configure Device Parameters (continued)**

<b>Web Page Item</b>	<b>Description</b>
Current Time	Sets the current time. <b>Note:</b> Be sure to save the current time by clicking on the <b>Set Time</b> button.
Encryption	Encryption can either be set to <b>None</b> or <b>AES-256</b> .
Encryption Key	Sets the AES encryption key. If encryption is currently enabled, the response to this command will be sent using the “old” key. The new key should be sent as 64 ASCII hexadecimal characters.
Door State	This field displays the current door state and is not configurable.
Relay State	This field displays the current relay state and is not configurable.
Button State	This field displays the current button state and is not configurable.
LED	This field displays the current LED state and is not configurable.
Alarm State	This field displays the current alarm state and is not configurable.
JP4, 6, 9, 10	This shows whether jumpers JP4, JP6, JP9, or JP10 are either enabled or disabled through the four digit sequence ( <b>0000</b> ). The <b>0</b> turns to <b>1</b> for an enabled jumper. For example, <b>0011</b> would mean jumpers JP9 and JP10 are activated, but JP4 and JP9 are not.
	Click on the <b>Refresh</b> button to refresh the <b>Device Configuration</b> page.
	Click on the <b>Get Log</b> button to get a log of the associated door strike relay activity. The door strike relay has 128Kb non-volatile storage for log data, storing an average of 10 days' worth of log data before it is overwritten.
	Click on the <b>Clear Log</b> button to clear the log from the door strike relay
	Click on the <b>Reboot</b> button to reboot any “discovered” remote relays and clear any associated devices.
	Click on the <b>Set Time</b> button to change the time.
	Click on the <b>Save Changes</b> button to save any changes that are made to the Device Configuration page. <b>Note:</b> The time setting must be saved by pressing the <b>Set Time</b> button.
	Click on the <b>Cancel</b> button to cancel any changes that were made to the <b>Configure Device</b> page and return to the <b>DSR</b> page.
	Click on the <b>Browse</b> button to navigate through your computer and find firmware files.
	Click on the <b>Upgrade</b> button to upgrade the firmware of the door strike relay.

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

## 2.4.14 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**

Home Device Network SIP Multicast Sensor Audiofiles Events DSR Autoprov Firmware

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Disable Autoprovisioning:

Autoprovisioning Server:

Autoprovisioning Filename:

Use tftp:

Username:

Password:

Autoprovisioning autoupdate (in minutes):

Autoprovision at time (HHMMSS):

Autoprovision when idle (in minutes > 10):

See the manual to learn how to use autoprovisioning to configure your device.

Autoprovisioning happens on boot.

The device will first look for a configured server address and filename.

If these haven't been configured, it will look for an autoprovisioning server in your list of DHCP options and try to download '0020f7024ebd.xml' and if this fails, '000000cd.xml'.

Save Reboot Toggle Help

Download Template


### Autoprovisioning log

```
00:00 Autoprovisioning Device...
00:00 Autoprov found option 43 in DHCP server="http://chalmers.cyberdata.net"
00:00 Autoprov looking for 0020f7024ebd.xml at http://chalmers.cyberdata.net
00:00 Got autoprov file. Parsing "0020f7024ebd.xml"
00:00 Autoprov found option 72 in DHCP server="10.0.0.252"
00:00 Autoprov looking for 0020f7024ebd.xml at 10.0.0.252
00:00 Got autoprov file. Parsing "0020f7024ebd.xml"
00:00 Autoprov found option 150 in DHCP server="10.0.0.252"
00:00 Autoprov looking for 0020f7024ebd.xml at 10.0.0.252
00:00 Got autoprov file. Parsing "0020f7024ebd.xml"
00:01 Autoprov found option 66 in DHCP server="http://chalmers.cyberdata.net"
```




- On the **Autoprovisioning** page, you may enter values for the parameters indicated in [Table 2-22](#).

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

**Table 2-22. Autoprovisioning Configuration Parameters**

Web Page Item	Description
Disable Autoprovisioning ?	Prevent the device from automatically trying to download a configuration file. See <a href="#">Section 2.4.14.1, "Autoprovisioning"</a> for more information.
Autoprovisioning Server ?	Enter the IPv4 address of the provisioning server in dotted decimal notation.
Autoprovisioning Filename ?	<p>The autoprovisioning filename is the configuration filename. The default autoprovisioning filename is in the format of <b>&lt;mac address&gt;.xml</b>.</p> <p>Supported filename extensions are .txt, and .xml. The current filename is denoted by an asterisk at the bottom of the <a href="#">Autoprovisioning Page</a>. Enter up to 256 characters.</p> <p>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.</p>
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) ?	<p>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.</p> <p><b>Note:</b> To use the auto update options, enable the <a href="#">Set Time with NTP Server on boot</a> setting on the <a href="#">Device Configuration Page</a> page (see <a href="#">Table 2-7</a>).</p>
Autoprovision at time (HHMMSS) ?	<p>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.</p> <p><b>Note:</b> To use the auto update options, enable the <a href="#">Set Time with NTP Server on boot</a> setting on the <a href="#">Device Configuration Page</a> page (see <a href="#">Table 2-7</a>).</p>
Autoprovision when idle (in minutes > 10) ?	<p>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.</p> <p><b>Note:</b> To use the auto update options, enable the <a href="#">Set Time with NTP Server on boot</a> setting on the <a href="#">Device Configuration Page</a> page (see <a href="#">Table 2-7</a>).</p>
	<p>Click the <b>Save</b> button to save your configuration settings.</p> <p><b>Note:</b> You need to reboot for changes to take effect.</p>

**Table 2-22. Autoprovisioning Configuration Parameters (continued)**

Web Page Item	Description
	Click on the <b>Reboot</b> button to reboot the system.
	Click on the <b>Toggle Help</b> button to see a short description of some of the web page items. First click on the <b>Toggle Help</b> 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.
	Press the <b>Download Template</b> button to create an autoprovisioning file for the device. See <a href="#">Section 2.4.14.3, "Download Template Button"</a>
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).

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

### 2.4.14.1 Autoprovisioning

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

1. The file configured on the autoprovisioning page.
2. A file named according to it's mac address (for example: 0020f7350058.xml).
3. The file 000000cd.xml

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

By default, the device will get its autoprovisioning server from the DHCP options. See [Section 2.4.14.2, "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 auto provisioning template file from the [Auto provisioning Page](#) using the **Download Template** button (see [Table 2-22](#)). This file contains every configuration option that can be set on the board.

Auto provisioning files can contain the whole configuration or a subset of this file. The first auto provisioning file can also contain links to other auto provisioning files.

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

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

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

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

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

The file 0020f7123456.xml contains:

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

1. The device will first set its 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](http://example.com/sip_reg0020f7123456.xml).
4. It will try to download <http://example.com/audio0020f7123456>.
5. It will try to download <http://example.com/device.xml>.

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

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

Checking for New  
Autoprovisioning  
Files after Boot

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

The  
Autoprovisioning  
Filename

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

**Table 2-23. Autoprovisioning File Name**

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

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

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

For example:

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

The autoprovisioning filename is set to “cyberdata/”

On boot, the device will try to download:

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

...and if this fails:

https://www.example.com/cyberdata/000000cd.xml

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



```

Autoprovisioning <FirmwareSettings>
Firmware Updates <FirmwareFile>505-uImage-ceiling-speaker</FirmwareFile>
                  <FirmwareServer>10.0.1.3</FirmwareServer>
                  <OutdoorIntercom30>firmware_file_v9.3.0</OutdoorIntercom30>
                  <OutdoorIntercom31>firmware_file_v10.3.0</OutdoorIntercom31>
                  <CallButton31>firmware_file_v10.3.0</CallButton31>
                  </FirmwareSettings>

```

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

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

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

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

The list of valid product strings:

```

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

```

Autoprovisioning  
Example 1

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

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

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

**000000cd.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**.
2. Device1 finds an AutoprovFile element containing the filename **sip\_common.xml**. The device downloads **sip\_common.xml** from "https://autoprovtest.server.net," and imports this configuration, setting the sip server to **10.0.0.253** and the remote port to **5060.3**.
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://autoprovttest.server.net.” Device2 sets the SIP User ID to **500**, the password to **ext500**, and the dialout extension to **555**.

#### Autoprovisioning Example 2

Here is another example of setting up your autoprovisioning files:

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

##### **0020f7020001.xml**

```
<MiscSettings>
<AutoprovFile>common_settings.xml</AutoprovFile>
</MiscSettings>
<SIPSettings>
<SIPUserID>198</SIPUserID>
<SIPAuthPassword>ext198</SIPAuthPassword>
<DialoutExtension0>204</DialoutExtension0>
</SIPSettings>
```

##### **0020f7020002.xml**

```
<MiscSettings>
<AutoprovFile>common_settings.xml</AutoprovFile>
</MiscSettings>
<SIPSettings>
<SIPUserID>500</SIPUserID>
<SIPAuthPassword>ext500</SIPAuthPassword>
<DialoutExtension0>555</DialoutExtension0>
</SIPSettings>
```

##### **common\_settings.xml**

```
<MiscSettings>
<DeviceName>CyberData Autoprovisioned</DeviceName>
</MiscSettings>
<SIPSettings> <SIPServer>10.0.0.253</SIPServer>
<RemoteSIPPort>5060</RemoteSIPPort>
</SIPSettings>
```

1. On boot, Device1 downloads **0020f7020001.xml** from **10.0.1.3** and imports these values. The SIP User ID is **198**, the password is **ext198**, and the dialout extension is **204**.

2. Device1 then gets the filename **common\_settings.xml** from the AutoprovFile element and downloads this file from the TFTP server at **10.0.1.3**. and imports these settings. The device name is set to **CyberData Autoprovisioned**, the SIP server is set to **10.0.0.253**, and the port is set to **5060**.

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

## 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)

## Autoprovisioned Audio Files

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

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

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

## 2.4.14.2 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;

    option domain-name            "voiplab";
    option domain-name-servers    10.0.0.252;

    option time-offset            -8;                # Pacific Standard Time

#   option www-server              99.99.99.99;        # OPTION 72

#   option tftp-server-name        "10.0.1.52";        # OPTION 66
#   option tftp-server-name        "http://test.cyberdata.net"; # OPTION 66

#   option option-150              10.0.0.252;        # OPTION 150

# These two lines are needed for option 43
#   vendor-option-space VendorInfo;                # OPTION 43
#   option VendorInfo.text "http://test.cyberdata.net"; # OPTION 43

    range 10.10.0.1 10.10.2.1; }
}
```

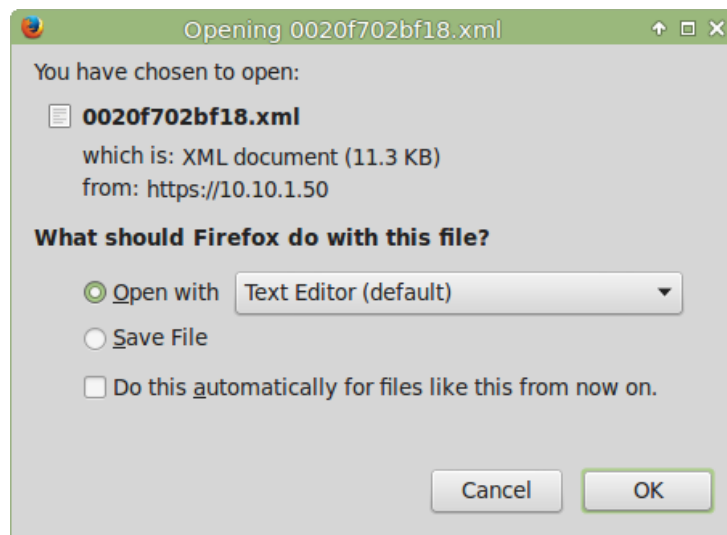
### 2.4.14.3 Download Template Button

The **Download Template** button allows the user to generate, download, edit, and then store an auto provisioning template on the server that serves the auto provisioning files for devices.

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

1. On the **Auto provisioning** page, click on the **Download Template** button.
2. 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](#).

**Figure 2-34. Configuration File**



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

---

## 2.5 Upgrade the Firmware and Reboot the Intercom



### Caution

**Equipment Hazard:** Devices with a serial number that begins with 2721xxxxx can only run firmware versions 10.0.0 or later.


---

### 2.5.1 Uploading the Firmware

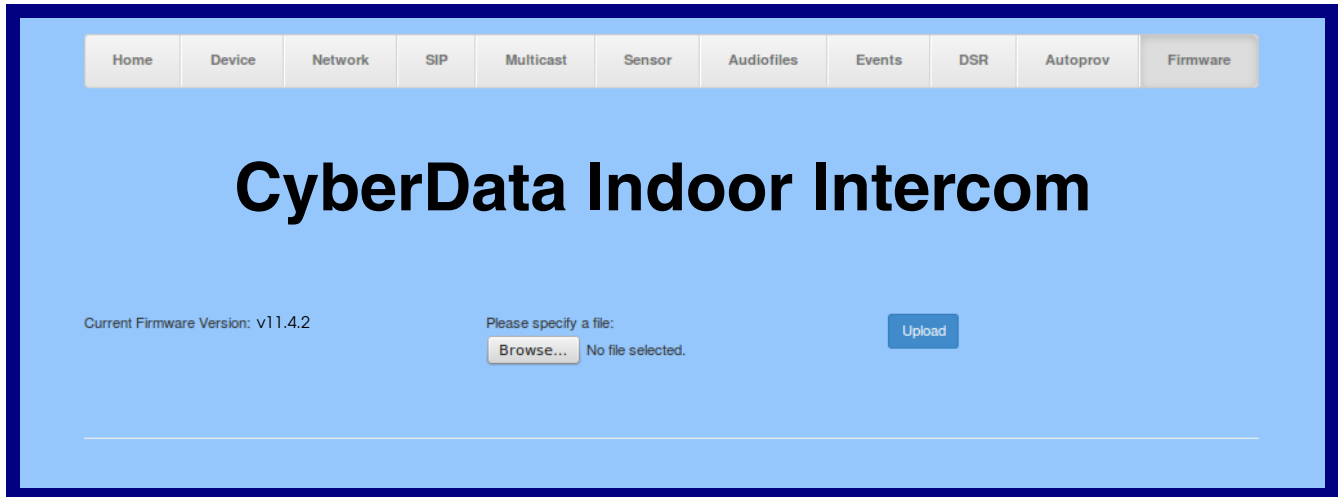
To upload the firmware from your computer:

1. Retrieve the latest Intercom firmware file from the VoIP Flush-Mount Indoor Intercom **Downloads** page at:  
<http://www.cyberdata.net/products/voip/digitalanalog/intercomindoorflush/downloads.html>
2. Unzip the firmware version file. This file may contain the following:
  - Firmware file
  - Release notes
3. Log in to the Intercom home page as instructed in [Section 2.4.4, "Log in to the Configuration Home Page"](#).

- Click the **Firmware** menu button to open the **Firmware** page. See [Figure 2-35](#).

 <small>GENERAL ALERT</small>	<p><b>Caution</b></p> <p><b>Equipment Hazard:</b> CyberData strongly recommends that you first reboot the device before attempting to upgrade the firmware of the device. See <a href="#">Section 2.5.2</a>, "<a href="#">Reboot the Device</a>".</p>
---	---

**Figure 2-35. Firmware Page**



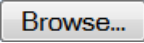

- Click on the **Browse** button, and then navigate to the location of the firmware file.
- Select the firmware file.
- Click on the **Upload** button.

**Note** Do not reboot the device after clicking on the **Upload** button.

**Note** This starts the upgrade process. Once the Intercom has uploaded the file, the **Uploading Firmware** countdown page appears, indicating that the firmware is being written to flash. The Intercom will automatically reboot when the upload is complete. When the countdown finishes, the **Firmware** page will refresh. The uploaded firmware filename should be displayed in the system configuration (indicating successful upload and reboot).

- [Table 2-24](#) shows the web page items on the **Firmware** page.

**Table 2-24. Firmware Parameters**

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

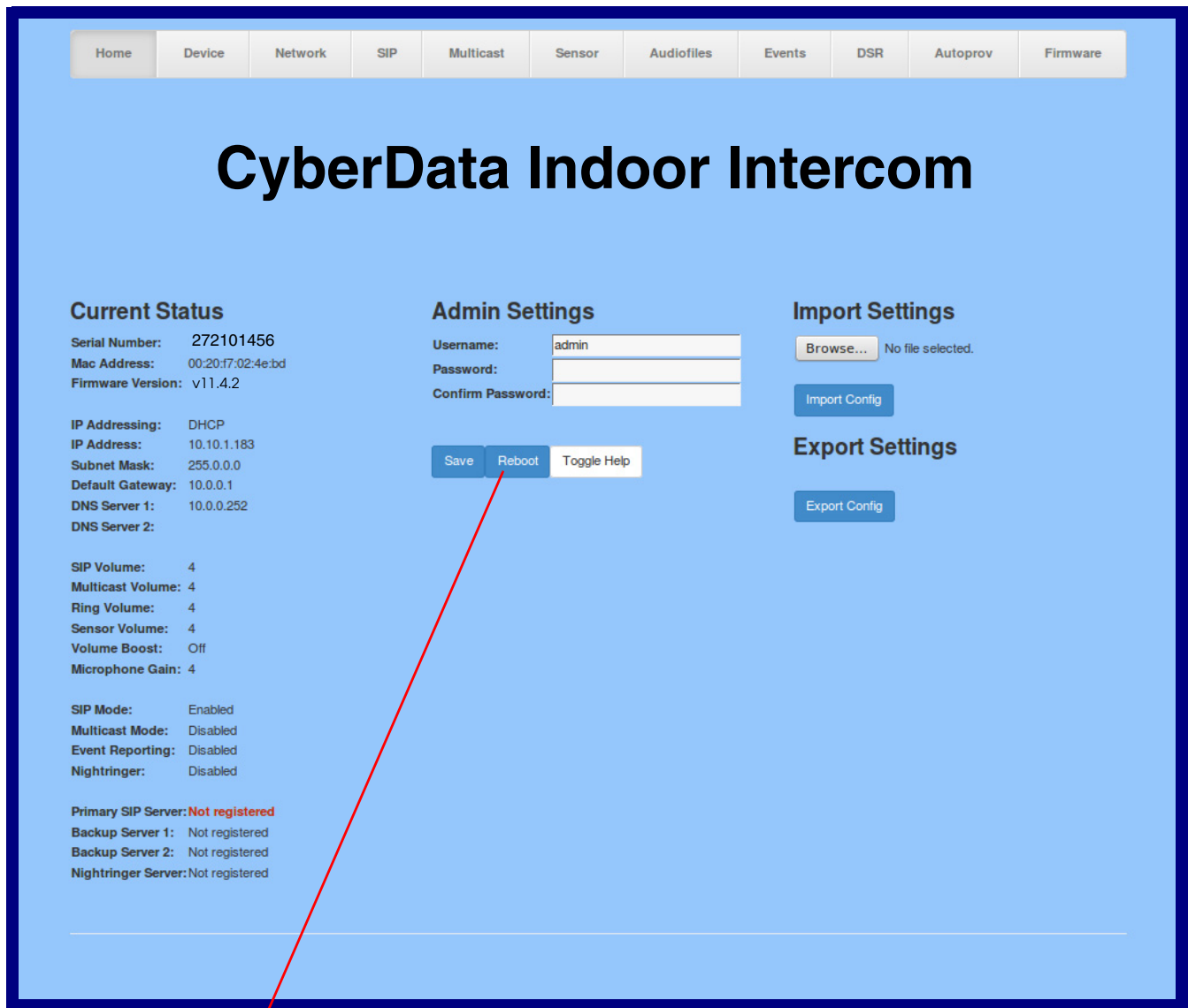


## 2.5.2 Reboot the Device

To reboot a Intercom, log in to the web page as instructed in [Section 2.4.4, "Log in to the Configuration Home Page"](#).

1. Click on the **Reboot** button on the **Home** page ([Figure 2-36](#)). A normal restart will occur.

**Figure 2-36. Reboot System Section**



Reboot

## 2.6 Command Interface

Some functions on the device can be activated using simple POST commands to the web interface. The examples in [Table 2-25](#) use the free unix utility, **wget** commands. However, any program that can send HTTP POST commands to the device should work.

### 2.6.1 Command Interface Post Commands

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

**Table 2-25. Command Interface Post Commands**

Device Action	HTTP Post Command <sup>a</sup>
Trigger relay (for configured delay)	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "test_relay=yes"
Place call to extension (example: extension 130)	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "call=130"
Place point-to-point call <sup>b</sup> (example: IP phone address = 10.0.3.72)	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "call=10.0.3.72"
Terminate active call	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "terminate=yes"
Force reboot	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "reboot=yes"
Test Audio button	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "test_audio=yes"
Announce IP address	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/command.cgi" --post-data "speak_ip_address=yes"
Play the "0" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_0=yes"
Play the "1" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_1=yes"
Play the "2" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_2=yes"
Play the "3" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_3=yes"

**Table 2-25. Command Interface Post Commands (continued)**

<b>Device Action</b>	<b>HTTP Post Command<sup>a</sup></b>
Play the "4" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_4=yes"
Play the "5" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_5=yes"
Play the "6" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_6=yes"
Play the "7" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_7=yes"
Play the "8" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_8=yes"
Play the "9" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_9=yes"
Play the "Dot" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_d=yes"
Play the "Audio Test" audio file (from Audio Config)	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_audiotest=yes"
Play the "Page Tone" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_pagetone=yes"
Play the "Your IP Address Is" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_youripaddressis=yes"
Play the "Rebooting" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_rebooting=yes"
Play the "Restoring Default" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_restoringdefault=yes"
Play the "Ringback tone" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_ringback=yes"
Play the "Ring tone" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_ringtones=yes"
Play the "Intrusion Sensor Triggered" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_intrusionsensortriggered=yes"
Play the "Door Ajar" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_doorajar=yes"

**Table 2-25. Command Interface Post Commands (continued)**

<b>Device Action</b>	<b>HTTP Post Command<sup>a</sup></b>
Play the "Night Ring" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "play_nightring=yes"
Delete the "0" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_0=yes"
Delete the "1" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_1=yes"
Delete the "2" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_2=yes"
Delete the "3" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_3=yes"
Delete the "4" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_4=yes"
Delete the "5" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_5=yes"
Delete the "6" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_6=yes"
Delete the "7" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_7=yes"
Delete the "8" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_8=yes"
Delete the "9" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_9=yes"
Delete the "Audio Test" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_audiotest=yes"
Delete the "Page Tone" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_pagetone=yes"
Delete the "Your IP Address Is" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_youripaddressis=yes"
Delete the "Rebooting" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_rebooting=yes"
Delete the "Restoring Default" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_restoringdefault=yes"

**Table 2-25. Command Interface Post Commands (continued)**

<b>Device Action</b>	<b>HTTP Post Command<sup>a</sup></b>
Delete the "Ringback tone" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_ringback=yes"
Delete the "Ring tone" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_ringtone=yes"
Delete the "Intrusion Sensor Triggered" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_intrusionsensortriggered=yes"
Delete the "Door Ajar" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_doorajar=yes"
Delete the "Night Ring" audio file	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/audioconfig.cgi" --post-data "delete_nightring=yes"
Trigger the Door Sensor Test (Sensor Config page)	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/sensorconfig.cgi" --post-data "doortest=yes"
Trigger the Intrusion Sensor Test (Sensor Config page)	wget --user admin --password admin --auth-no-challenge --no-check-certificate --quiet -O /dev/null "https://10.0.3.71/cgi-bin/sensorconfig.cgi" --post-data "intrusiontest=yes"

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




b. Must be in point-to-point mode see [Section 2.4.7.2, "Point-to-Point Configuration"](#)

# Appendix A: Mounting the Intercom

## A.1 Mounting Components

Before you mount the Intercom, make sure that you have received all the parts for each Intercom. Refer to the following tables.

**Table A-1. Mounting Components (Part of the Accessory Kit)**

Quantity	Part Name	Illustration
1	T15 Security Pin Torx Key	
4	#6 X 3/8-inch, 100 Deg., Flat Head, Self-Tapping Screw	
4	#6 X 3/8-inch, 100 Deg., Flat Head T15 Security Pin Torx Screw	

## A.2 Dimensions

Figure A-1. Unit Dimensions

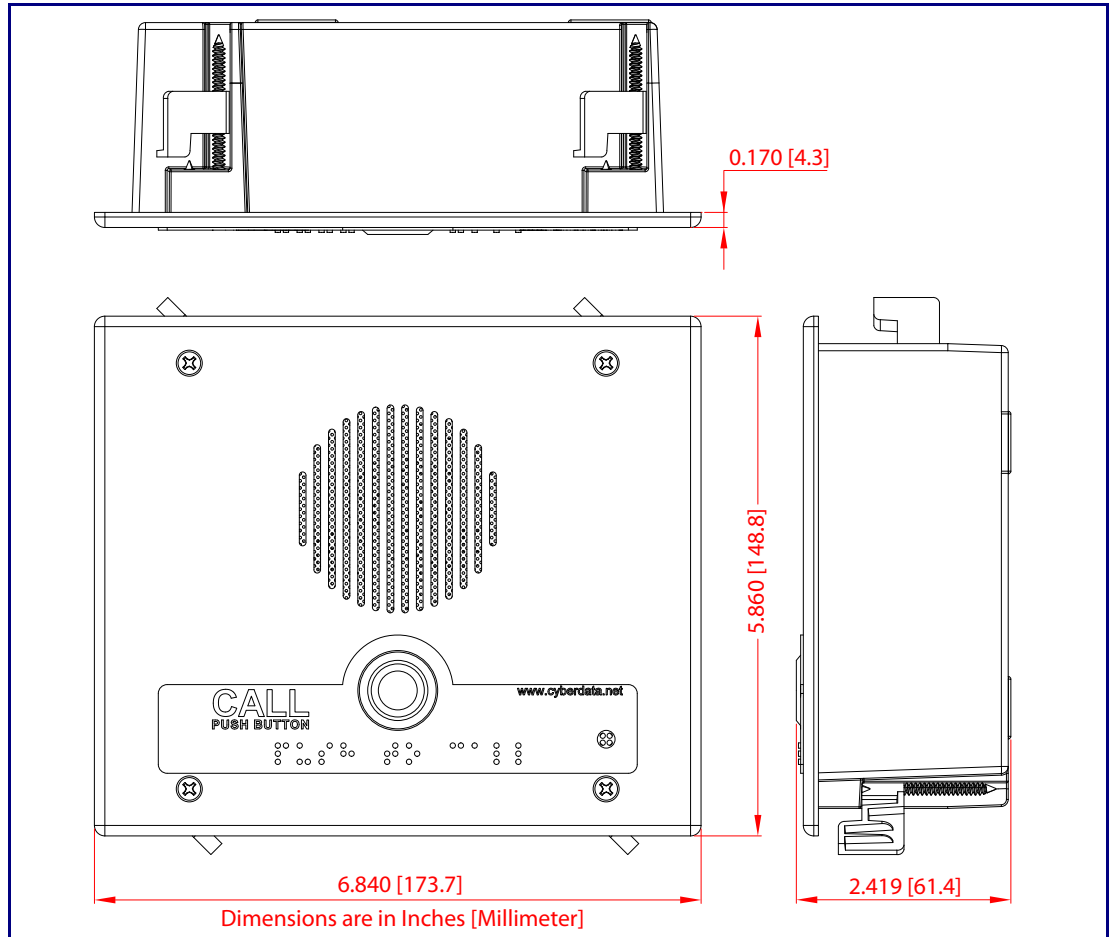


Figure A-2. Faceplate Dimensions and Intrusion Sensor Range

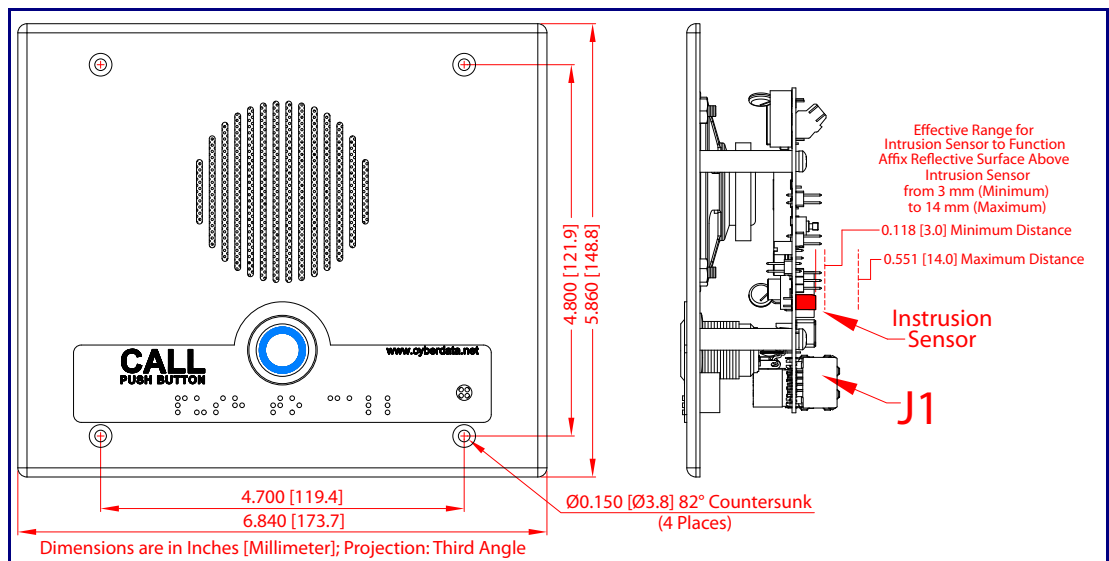
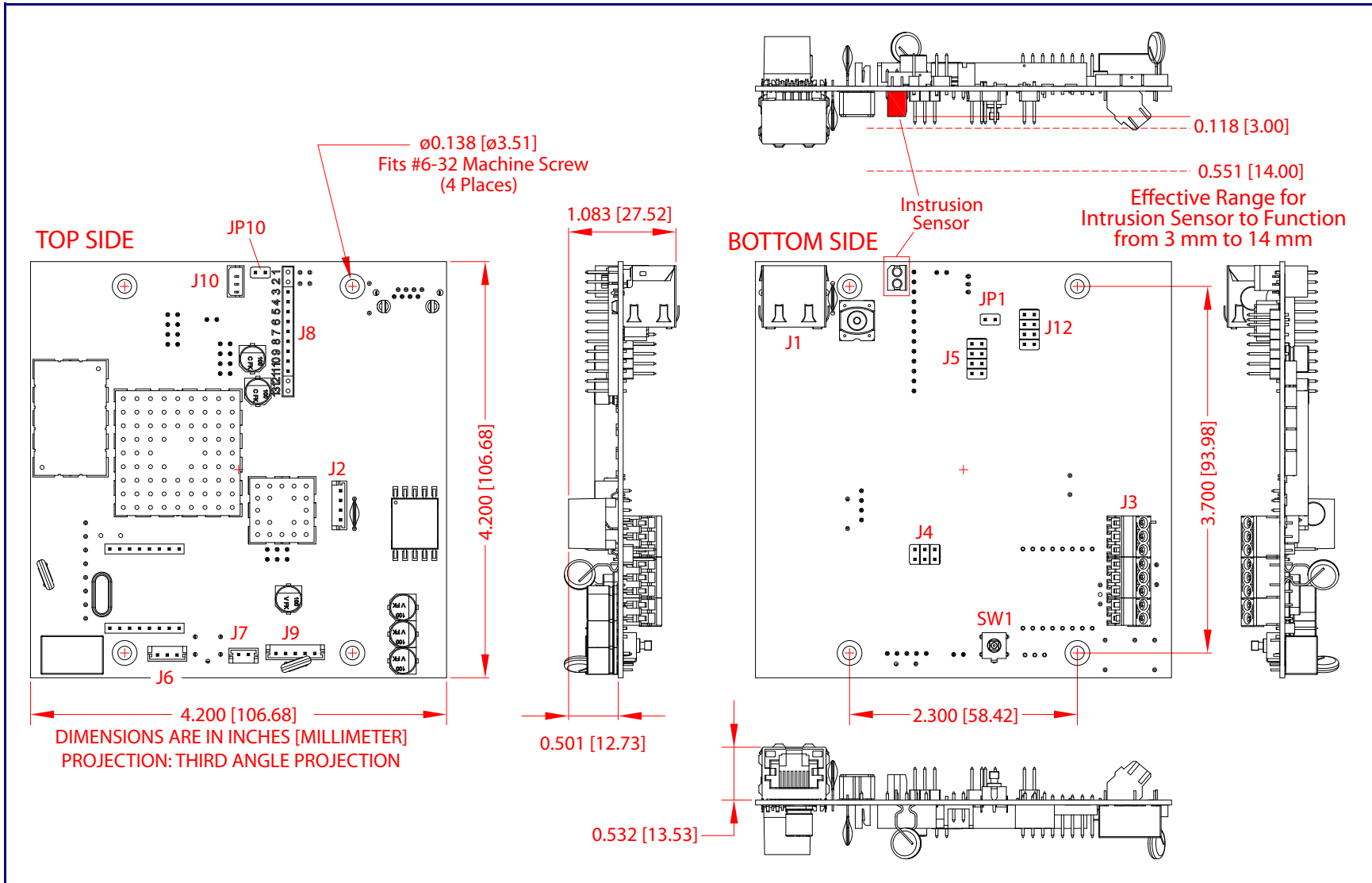


Figure A-3. PCB Dimensions and Intrusion Sensor Range

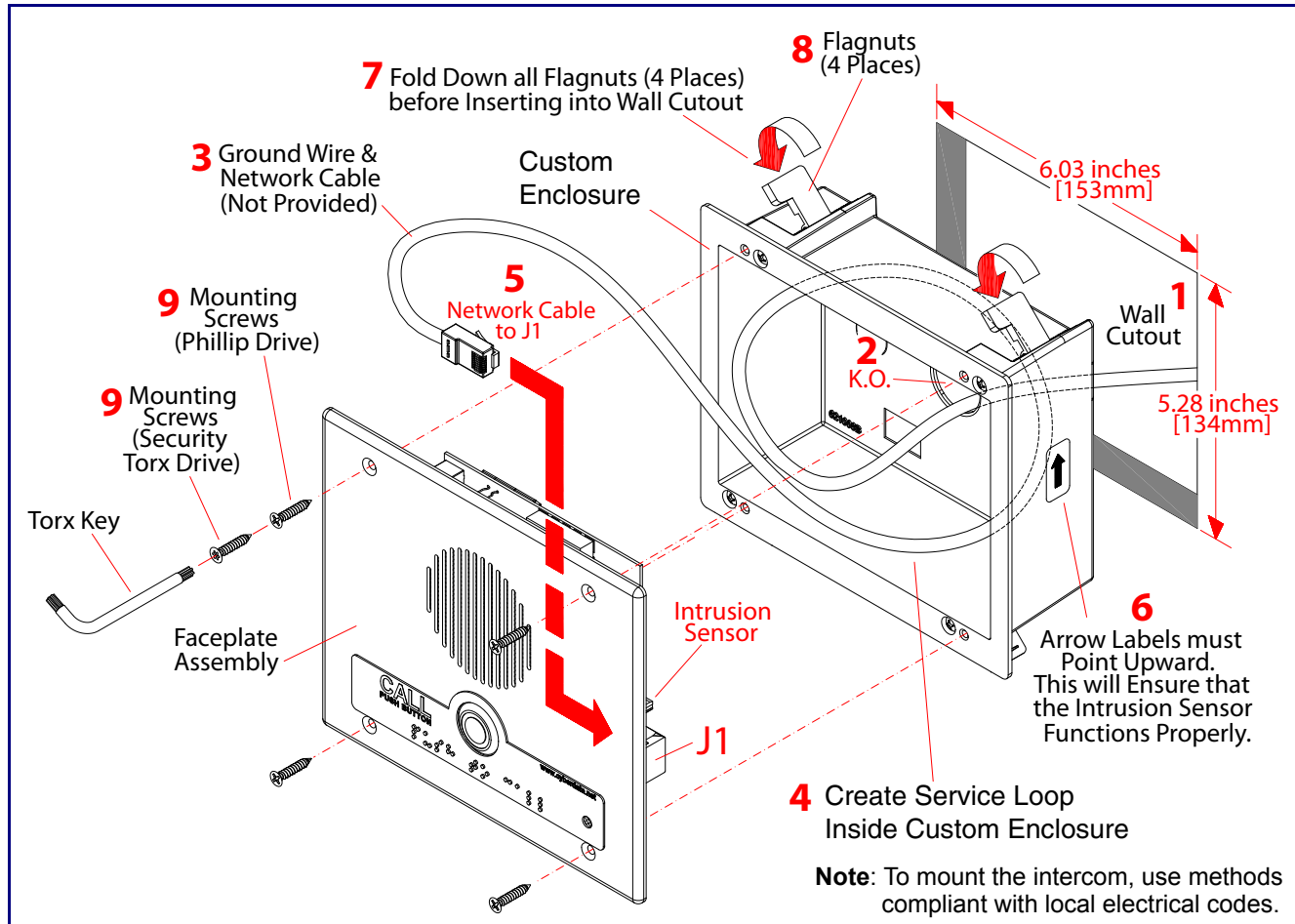




## A.3 Wall Mounting

Figure A-4 illustrates a wall mounting option for the VoIP Flush-Mount Indoor Intercom.

**Figure A-4. Wall Mounting**



**To mount the Intercom:**

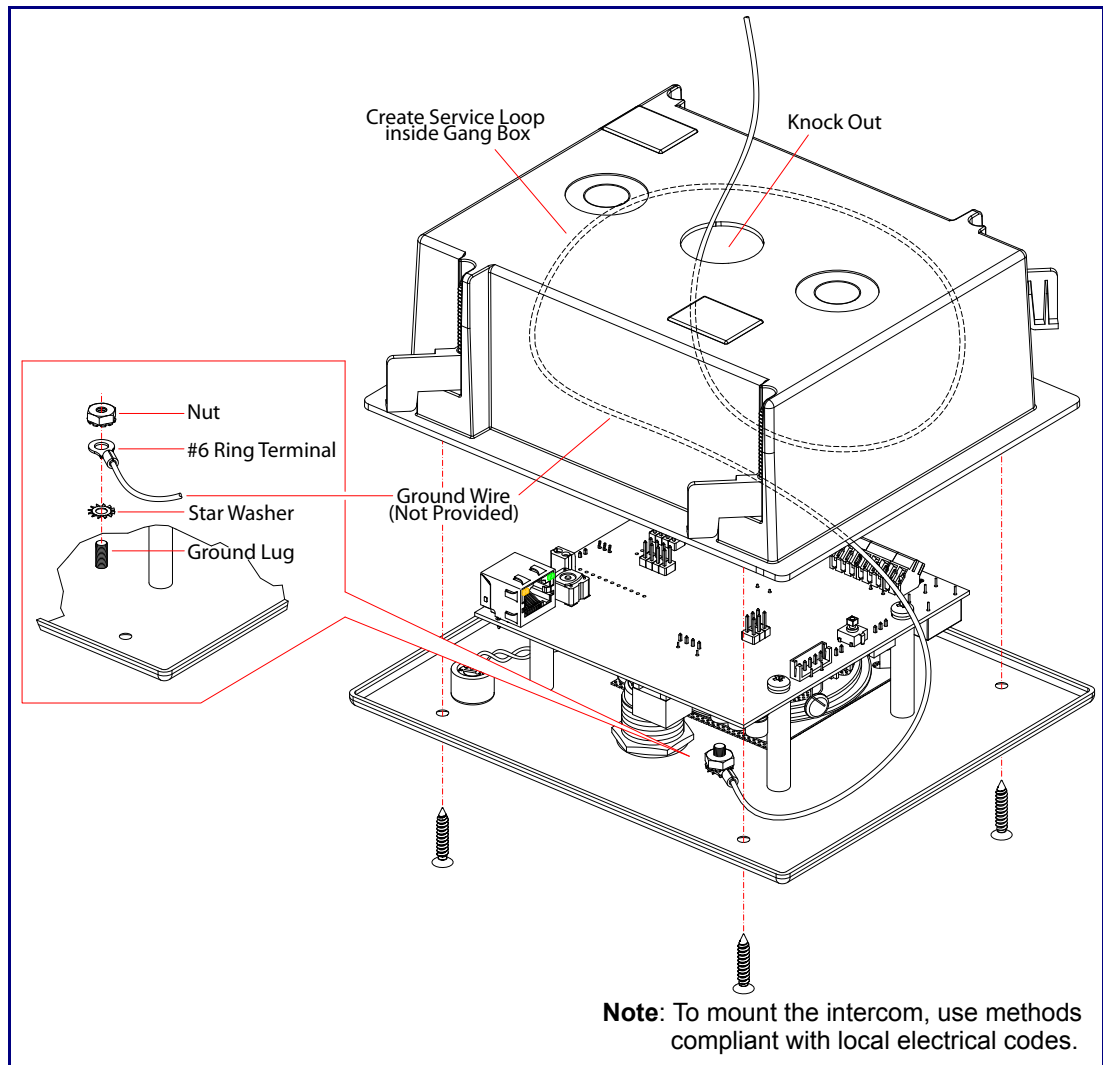
1. Make a wall cutout as shown in the picture.
2. Use a flat blade screwdriver to remove the knockout (KO) of the gang box.
3. Feed the ground wire (shown in [Section A.4, "Ground Cable Installation"](#)) and the network cable from the wall cutout through the knockout hole of the gang box.
4. Create a service loop for both the ground wire and network cable.
5. Plug the network cable into the J1 connector.
6. Make sure that the arrow labels are pointing up. This will ensure that the intrusion sensor functions properly.
7. Fold down all of the flagnuts, and then insert the gang box into the wall cutout.

8. Tighten the flagnuts with a Phillips screwdriver.
9. Secure the Intercom faceplate assembly to the gang box with either Phillips screws or security Torx screws.

## A.4 Ground Cable Installation

Figure A-5 illustrates how to connect a ground cable to the VoIP Flush-Mount Indoor Intercom.

**Figure A-5. Ground Cable Installation**



# Appendix B: Setting up a TFTP Server

---

## B.1 Set up a TFTP Server

Autoprovisioning requires a TFTP server for hosting the configuration file.

---

### B.1.1 In a LINUX Environment

To set up a TFTP server on LINUX:

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

```
in.tftpd -l -s /tftpboot/your_directory_name
```

---

### B.1.2 In a Windows Environment

You can find several options online for setting up a Windows TFTP server. This example explains how to use the Solarwinds freeware TFTP server, which you can download from the following website address:

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

To set up a TFTP server on Windows:

1. Install and start the software.
2. Select **File/Configure/Security** tab/**Transmit Only**.
3. Make a note of the default directory name, and then move the firmware files to be uploaded to that directory.

# Appendix C: Troubleshooting/Technical Support

---

## C.1 Frequently Asked Questions (FAQ)

A list of frequently asked questions (FAQs) are available on the VoIP Flush-Mount Indoor Intercom product page at:

<http://www.cyberdata.net/products/voip/digitalanalog/intercomindoorflush/faqs.html>

Select the support page for your product to see a list of frequently asked questions for the CyberData product:

---

## C.2 Documentation

The documentation for this product is released in an English language version only. You can download PDF copies of CyberData product documentation from the VoIP Flush-Mount Indoor Intercom product page at:

<http://www.cyberdata.net/products/voip/digitalanalog/intercomindoorflush/docs.html>

---

## C.3 Contact Information

Contact	<p>CyberData Corporation 3 Justin Court Monterey, CA 93940 USA <a href="http://www.CyberData.net">www.CyberData.net</a> Phone: 800-CYBERDATA (800-292-3732) Fax: 831-373-4193</p>
Sales	<p>Sales 831-373-2601 Extension 334</p>
Technical Support	<p>The fastest way to get technical support for your VoIP product is to submit a VoIP Technical Support form at the following website:</p> <p><a href="http://support.cyberdata.net/">http://support.cyberdata.net/</a></p> <p>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 <b>Comments</b> section of the Support Form.</p> <p>Phone: (831) 373-2601, Ext. 333 Email: support@cyberdata.net</p>
Returned Materials Authorization	<p>To return the product, contact the Returned Materials Authorization (RMA) department:</p> <p>Phone: 831-373-2601, Extension 136 Email: RMA@CyberData.net</p> <p>When returning a product to CyberData, an approved CyberData RMA number must be printed on the outside of the original shipping package. Also, RMA numbers require an active VoIP Technical Support ticket number. A product will not be accepted for return without an approved RMA number. Send the product, in its original package, to the following address:</p> <p>CyberData Corporation 3 Justin Court Monterey, CA 93940 Attention: RMA "your RMA number"</p>
RMA Status Form	<p>If you need to inquire about the repair status of your product(s), please use the CyberData RMA Status form at the following website address:</p> <p><a href="http://support.cyberdata.net/">http://support.cyberdata.net/</a></p>

---

## C.4 Warranty and RMA Information

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

<http://support.cyberdata.net/>

# Index

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## Numerics

16 AWG gauge wire 8

## A

activate local relay (door sensor) 62  
 activate relay (door sensor) 48  
 activate relay (intrusion sensor) 48  
 activity LED 15  
 address, configuration login 23  
 announcing a device's IP address 16  
 audio configuration 50  
   night ring tone parameter 52  
 audio configuration page 50  
 audio encodings 4  
 audio files, user-created 53  
 autoprovision at time (HHMMSS) 69  
 autoprovision when idle (in minutes > 10) 69  
 autoprovisioning 70  
   download template button 70  
   setting up a TFTP server 92  
 autoprovisioning autoupdate (in minutes) 69  
 autoprovisioning configuration 68, 69  
 autoprovisioning filename 69  
 autoprovisioning server (IP Address) 69  
 auxiliary relay 5

## B

backup SIP server 1 39  
 backup SIP server 2 39  
 backup SIP servers, SIP server  
   backups 39  
 boost (volume) 28

## C

call button 18  
 call button LED 18  
 call termination 30  
 changing  
   the web access password 27  
 Chrome (web browser) 3  
 Cisco SRST 39  
 command interface 83

commands 83  
 configurable parameters 28, 36, 39, 66  
 configuration  
   audio 50  
   default IP settings 19  
   door sensor 47  
   intrusion sensor 47  
   network 35, 65  
   SIP 38  
   using Web interface 19  
 configuration home page 23  
 configuration page  
   configurable parameters 28, 36, 66  
 contact information 94  
 contact information for CyberData 94  
 current network settings 36  
 CyberData contact information 94

## D

default  
   gateway 19  
   intercom settings 95  
   IP address 19  
   subnet mask 19  
   username and password 19  
   web login username and password 23  
 default gateway 19, 36  
 default intercom settings 17  
 default IP settings 19  
 default login address 23  
 device configuration 27  
   device configuration parameters 69  
   the device configuration page 68  
 device configuration page 27  
 device configuration parameters 28  
 device configuration password  
   changing for web configuration access 27  
 DHCP Client 4  
 dial out extension (door sensor) 48, 63  
 dial out extension (intrusion sensor) 48  
 dial out extension strings 42, 44  
 dimensions 5, 88  
   pcb dimensions and intrusion sensor range 89  
   unit dimensions—front and side view 88  
 discovery utility program 23  
 DNS server 36  
 door sensor 47, 48, 62  
   activate local relay 62  
   activate relay 48

- dial out extension 48, 63
- door open timeout 48, 62
- door sensor normally closed 48
- flash button LED 48, 62
- play audio locally 48, 62
- door strike intermediate relay 11, 12
- download autoprovisioning template button 70
- DTFM
  - play tone during DTMF activation 29
- DTMF push to talk 30
- DTMF tones 42, 44
- DTMF tones (using rfc2833) 42

## E

- enable night ring events 56
- event configuration
  - enable night ring events 56
- expiration time for SIP server lease 40, 41
- export settings 25, 26

## F

- factory default settings 17
  - how to set 17
- Firefox (web browser) 3
- firmware
  - where to get the latest firmware 80
- flash button LED (door sensor) 48, 62
- flash button LED (intrusion sensor) 48

## G

- get autoprovisioning template 70
- GMT table 33
- GMT time 33
- ground cable installation 90

## H

- home page 23
- http POST command 83
- http web-based configuration 4

## I

- identifier names (PST, EDT, IST, MUT) 33

- identifying your product 1
- import settings 25, 26
- import/export settings 25, 26
- installation, typical intercom system 2
- intercom configuration
  - default IP settings 19
- intercom configuration page
  - configurable parameters 39
- Internet Explorer (web browser) 3
- intrusion sensor 47, 48
  - activate relay 48
  - dial out extension 48
  - flash button LED 48
  - play audio locally 48
- IP address 19, 36
- IP addressing
  - default
    - IP addressing setting 19

## J

- J3 terminal block, 16 AWG gauge wire 8

## L

- lease, SIP server expiration time 40, 41
- LED
  - green link LED 15
  - yellow activity LED 15
- lengthy pages 46
- link LED 15
- Linux, setting up a TFTP server on 92
- local SIP port 40
- log in address 23

## M

- MGROUP
  - MGROUP Name 46
- mounting 87
  - ground cable installation 90
  - network cable entry restrictions 90
  - overview of installation types 87
  - service loop cable routing 91
- mounting components 87
- Mozilla Firefox (web browser) 3
- multicast configuration 50
- Multicast IP Address 46

## N

- navigation (web page) 20
- navigation table 20
- network cable entry restrictions 90
- network configuration of intercom 35, 65
- network rate 5
- Network Setup 35, 65
- nightring tones 46
- Nightringer 8, 79
- nightringer settings 40
- NTP server 29

## O

- on-board relay 9
- operating temperature 5

## P

- packet time 4
- pages (lengthy) 46
- part number 5
- password
  - for SIP server login 39
  - login 23
  - restoring the default 19
- payload types 5
- pcb dimensions and intrusion sensor range 89
- play audio locally (door sensor) 48, 62
- play audio locally (intrusion sensor) 48
- play tone during DTMF activation 29
- point-to-point configuration 43
- port
  - local SIP 40
  - remote SIP 40
- posix timezone string
  - timezone string 29
- POST command 83
- power requirement 5
- priority
  - assigning 46
- product
  - configuring 19
  - parts list 6
- product features 3
- product overview
  - product features 3
  - product specifications 5
  - supported protocols 4
  - supported SIP servers 4
  - typical system installation 2

- product specifications 5
- protocol 5
- protocols supported 4
- push to talk, DTMF 30

## R

- reboot 81, 82
- regulatory compliance 5
- remote SIP port 40
- reset test function management button 16
- resetting the IP address to the default 87, 93
- restoring factory default settings 17, 95
- restoring the factory default settings 17
- ringtones 46
  - lengthy pages 46
- RJ-45 14
- RMA returned materials authorization 94
- RMA status 94
- rport discovery setting, disabling 40
- RTFM button 16
- RTFM jumper 16, 17
- RTP/AVP 4

## S

- Safari (web browser) 3
- sales 94
- sensor setup page 47, 61
- sensor setup parameters 47
- sensors 48, 62
- server address, SIP 39
- service 94
- service loop cable routing 91
- set time with external NTP server on boot 29
- setting up an intercom 8
- settings, default 17
- SIP
  - enable SIP operation 39
  - local SIP port 40
  - user ID 39
- SIP (session initiation protocol) 4
- SIP configuration 38
  - SIP Server 39
- SIP configuration parameters
  - outbound proxy 40, 41
  - registration and expiration, SIP server lease 40, 41
  - unregister on reboot 40
  - user ID, SIP 39
- SIP registration 39
- SIP remote SIP port 40
- SIP server 39



- password for login 39
- SIP servers supported 4
- unregister from 40
- user ID for login 39
- SIP volume 28
- speaker output 5
- SRST 39
- subnet mask 19, 36
- supported protocols 4

- navigation 20
- web page navigation 20
- web-based intercom configuration 19
- weight 5
- wget, free unix utility 83
- Windows, setting up a TFTP server on 92
- wiring the circuit 10
  - devices less than 1A at 30 VDC 10

## T

- tech support 94
- technical support, contact information 94
- terminal block, 16 AWG gauge wire 8
- TFTP server 4, 92
- time zone string examples 33

## U

- unit dimensions—front and side view 88
- user ID
  - for SIP server login 39
- username
  - changing for web configuration access 27
  - default for web configuration access 23
  - restoring the default 19

## V

- VLAN ID 36
- VLAN Priority 36
- VLAN tagging support 36
- VLAN tags 36
- volume
  - microphone gain 28
  - multicast volume 28
  - ring volume 28
  - sensor volume 28
  - SIP volume 28
- volume boost 28

## W

- warranty policy at CyberData 94
- web access password 19
- web access username 19
- web configuration log in address 23
- web page