

## *Blueface Configuration Guide: SIP Speaker*

Document Part # 931912B

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## **Blueface Configuration Guide: SIP Speaker Document #931912B**

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## Revision Information

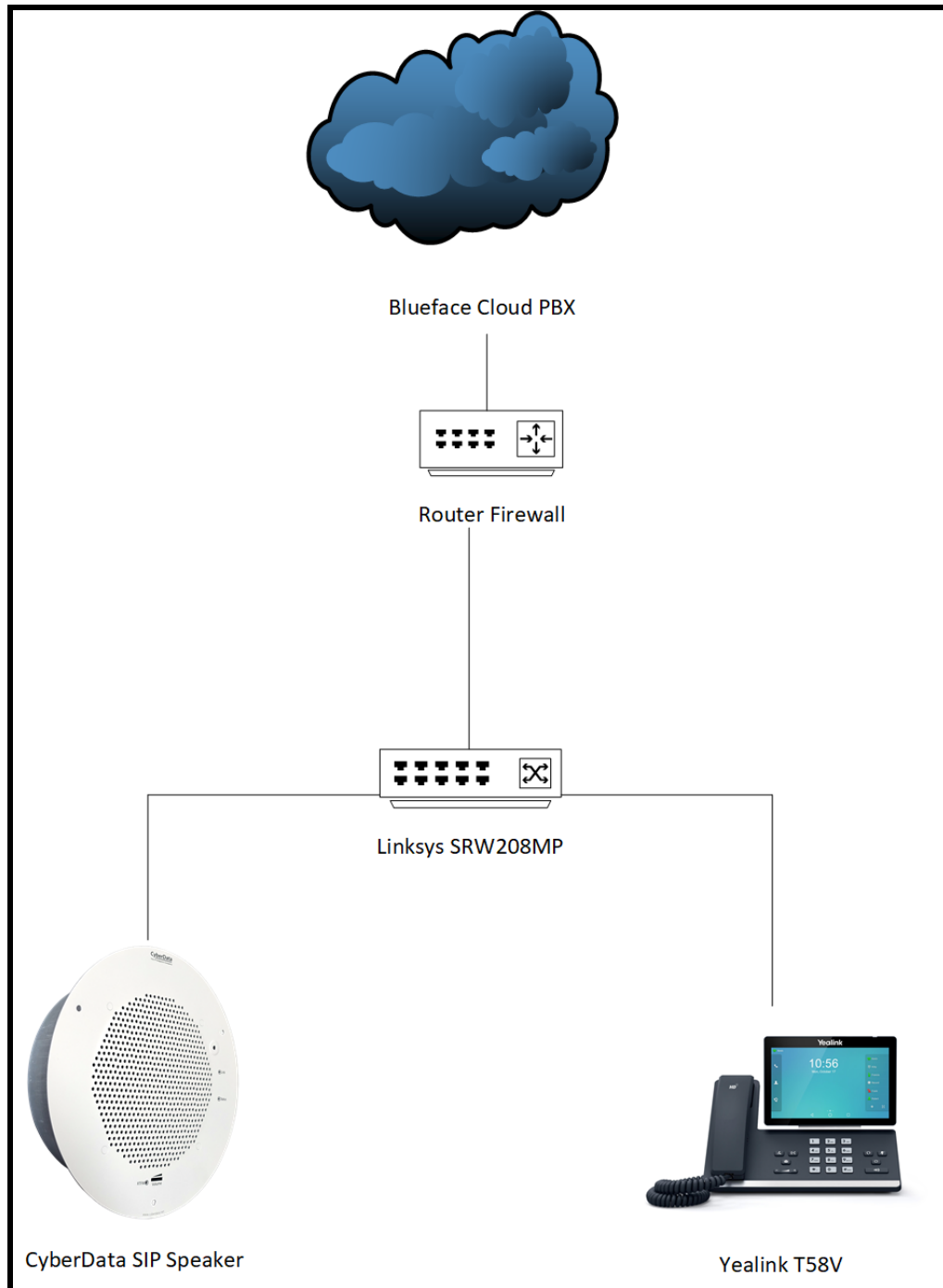
- 1/3/2022 – Initial Release
- 1/5/2022 – Name Update

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## 1.0 Setup Diagram

**Figure 1-1:** Interoperability Test Infrastructure



## 2.0 Test Setup Equipment

This section describes the products used for interoperability testing with Blueface.

**Table 2-1: Setup Equipment**

| EQUIPMENT                         | MODEL or PART NUMBER | FIRMWARE VERSION |
|-----------------------------------|----------------------|------------------|
| CYBERDATA SIP SPEAKER             | 011394               | 12.1.1           |
| CYBERDATA SIP TALKBACK<br>SPEAKER | 011398               | 12.1.1           |

## 3.0 Before You Start

This configuration guide documents the integration process of the CyberData SIP Speaker.

### Network Advisories

Blueface uses a Fully Qualified Domain Name (FQDN) for the SIP server address. The CyberData SIP Speaker's need to perform a DNS A query to resolve the IP address of Blueface's SIP Server FQDN. It is necessary to ensure the configured DNS server(s) have an A record for the SIP Server address.

In addition, be sure to verify the following ports are available for the speaker to use:

- UDP 5062 (SIP)
- UDP 10500 (RTP)

The speaker will need to traverse the public internet in order to operate with Blueface in the cloud.

The speaker's paging extension uses SIP port 5060 to receive SIP messages. The device will send SIP messages to port 5062, the port used by Blueface's SIP Server.

SIP ports 5060 and RTP port 10500 are the default values on all noted firmware levels.

Alternatively, SIP ports for the device are configurable on the **SIP** page of the web interface.

The CyberData Discovery Utility can be used to locate CyberData devices on your network. You may download it from the following web address:

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

***Note:** DHCP addressing mode is enabled on default on all noted firmware levels.*

## **Product Documentation and Utilities**

Before you start, download the Operation and Quick Start guides from the speaker's product webpage:

SIP Speaker [\(011394\)](#):

[https://files.cyberdata.net/assets/011393,011394/011394\\_931181L\\_SIP\\_Speaker\\_Ops\\_Guide.pdf](https://files.cyberdata.net/assets/011393,011394/011394_931181L_SIP_Speaker_Ops_Guide.pdf)

SIP Talkback Speaker [\(011398\)](#):

[https://files.cyberdata.net/assets/011397,011398/011398\\_931191M\\_SIP\\_Talk-Back\\_Speaker\\_Operations\\_Guide.pdf](https://files.cyberdata.net/assets/011397,011398/011398_931191M_SIP_Talk-Back_Speaker_Operations_Guide.pdf)



## 4.0 Configuration Procedure: Callflow Setup

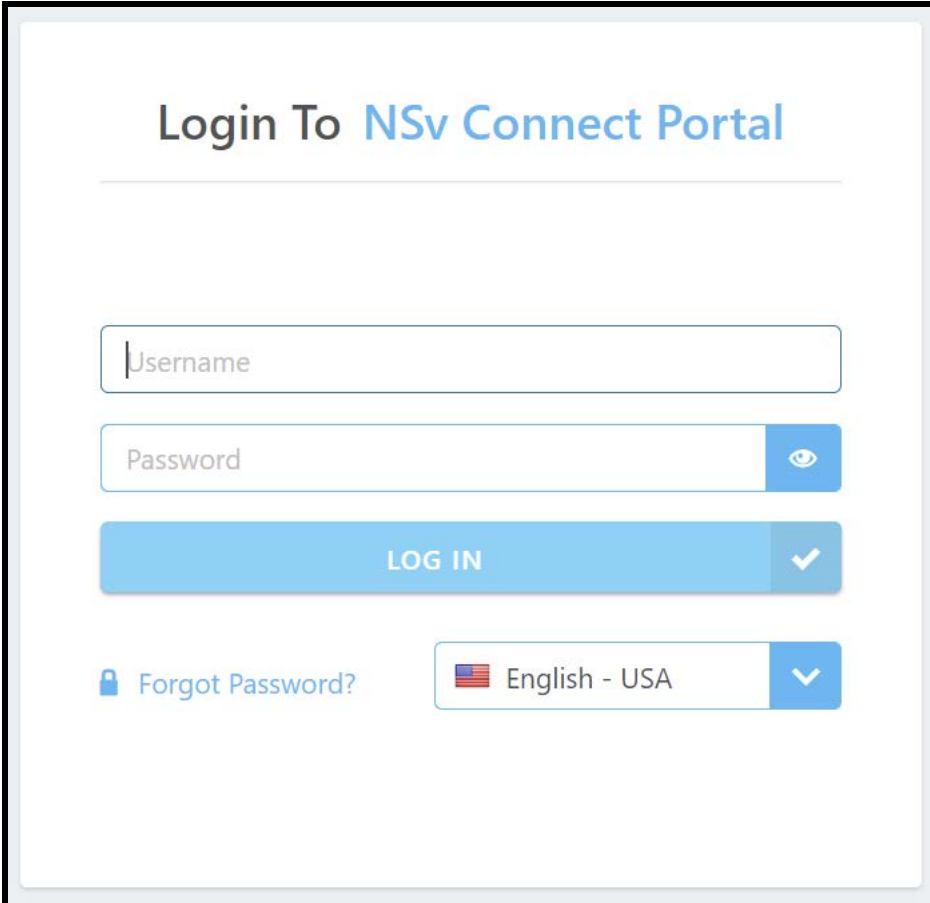
Blueface does not allow users to add their own devices to the platform. The MAC addresses of the devices must be provided to the account manager, who can then add the devices to the platform for you. An email will then be generated and sent to you that will contain the registration information for the CyberData device.

Blueface requires a callflow to be created to call or make a call from any device. This section will outline how to create the dial plan.

1. Log into Blueface.

<https://portal.nsvconnect.com/login>

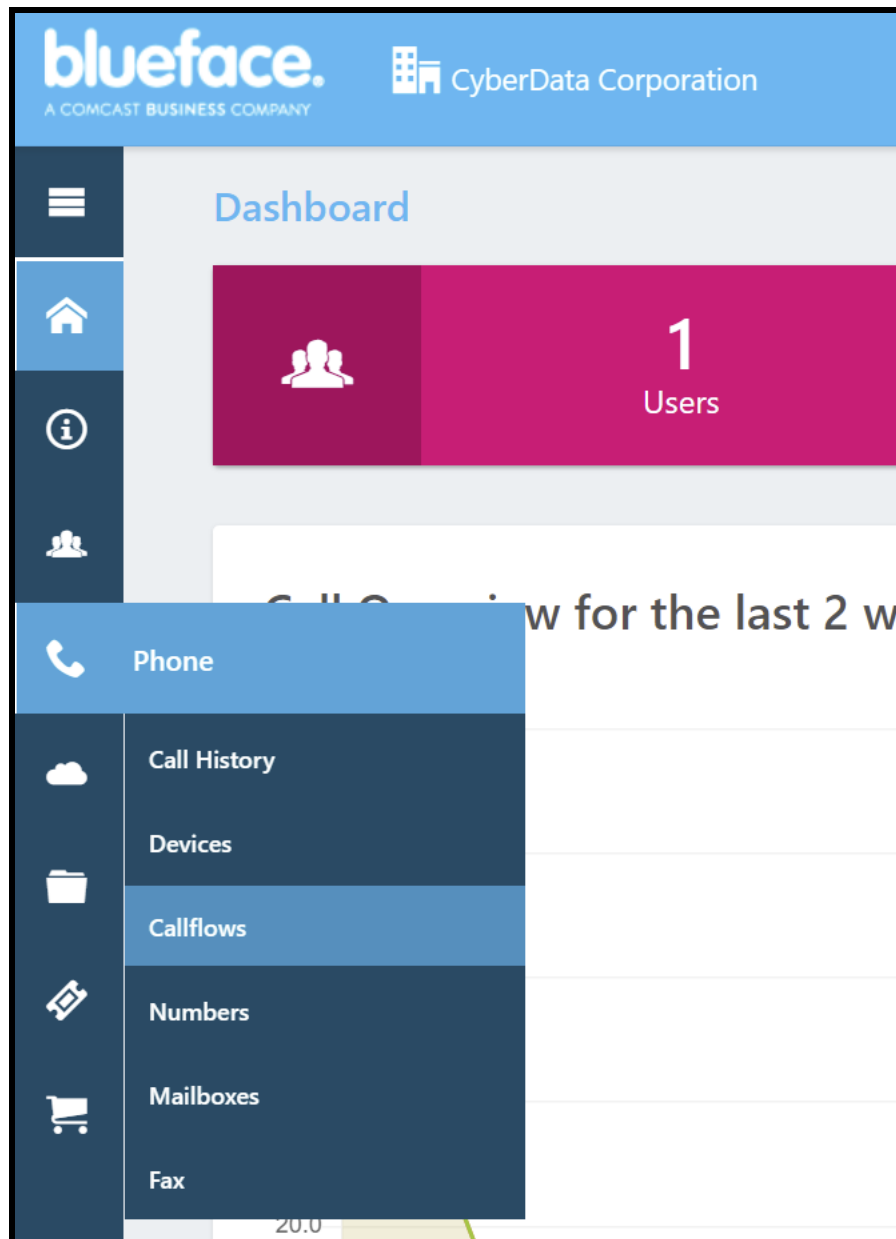
**Figure 4-1:** Login



The screenshot shows the login interface for the NSv Connect Portal. At the top, the text "Login To NSv Connect Portal" is displayed. Below this, there are two input fields: "Username" and "Password". The "Password" field has a blue eye icon on the right side, indicating a toggle for password visibility. Below the input fields is a large blue button labeled "LOG IN" with a white checkmark icon on the right. At the bottom left, there is a link "Forgot Password?" with a lock icon. At the bottom right, there is a language selection dropdown menu showing "English - USA" with a flag icon and a downward arrow.

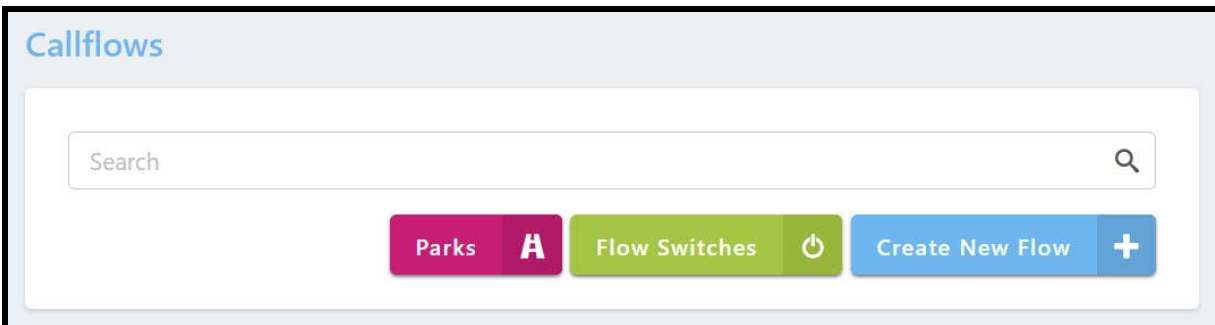
2. From the landing page **Phone** and then **callflows**.

**Figure 4-2:** Dashboard



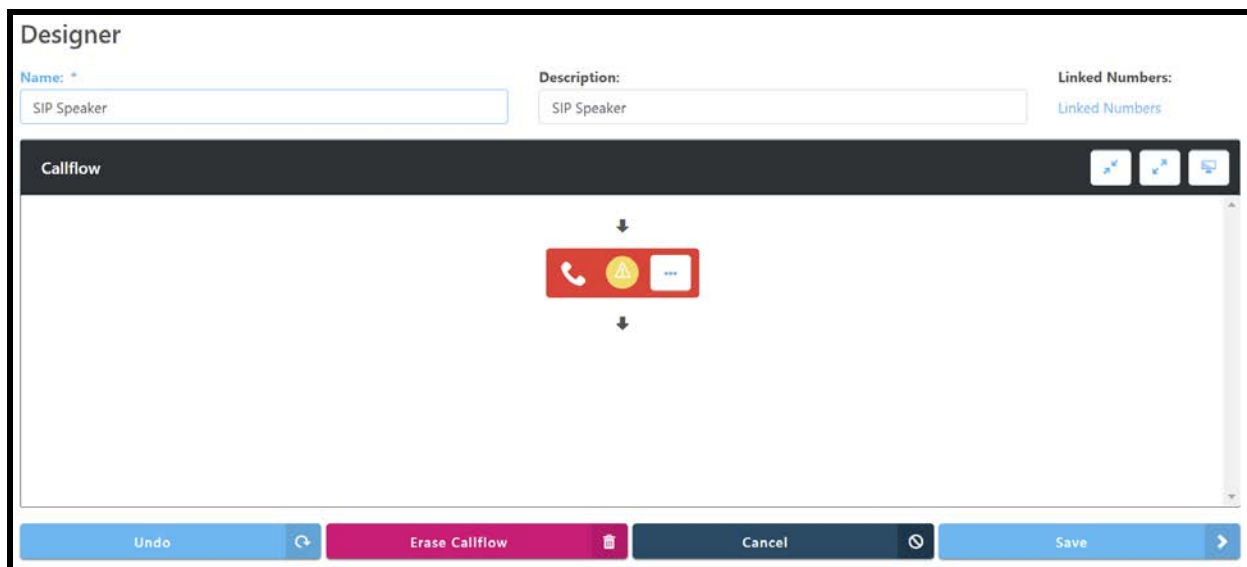
3. On the **Callflows** page press **Create New Flow**.

**Figure 4-3:** Callflow Page



4. Name the new callflow and set a description.

**Figure 4-4:** Callflow designer



5. From Elements drag **Ring** into the Callflow.

6. Click the yellow exclamation point to open the **Ring Element Settings** popup.
7. Select the Phone tab in the popup.
8. Select the Device that will be used in the group.

**Figure 4-5: Ring Element Settings**

Ring Element Settings

Timeout: \*  
20

This configuration rings the destination for 20 seconds

Search

**Devices**

- SIP Paging Adapter  
Generic SIP Device
- SIP Paging Amplifier  
Generic SIP Device
- SIP Paging Server  
Generic SIP Device
- SIP Strobe  
Generic SIP Device
- Video Intercom with Keypad  
Generic SIP Device

< 1 2 3 >

**Selected destinations**

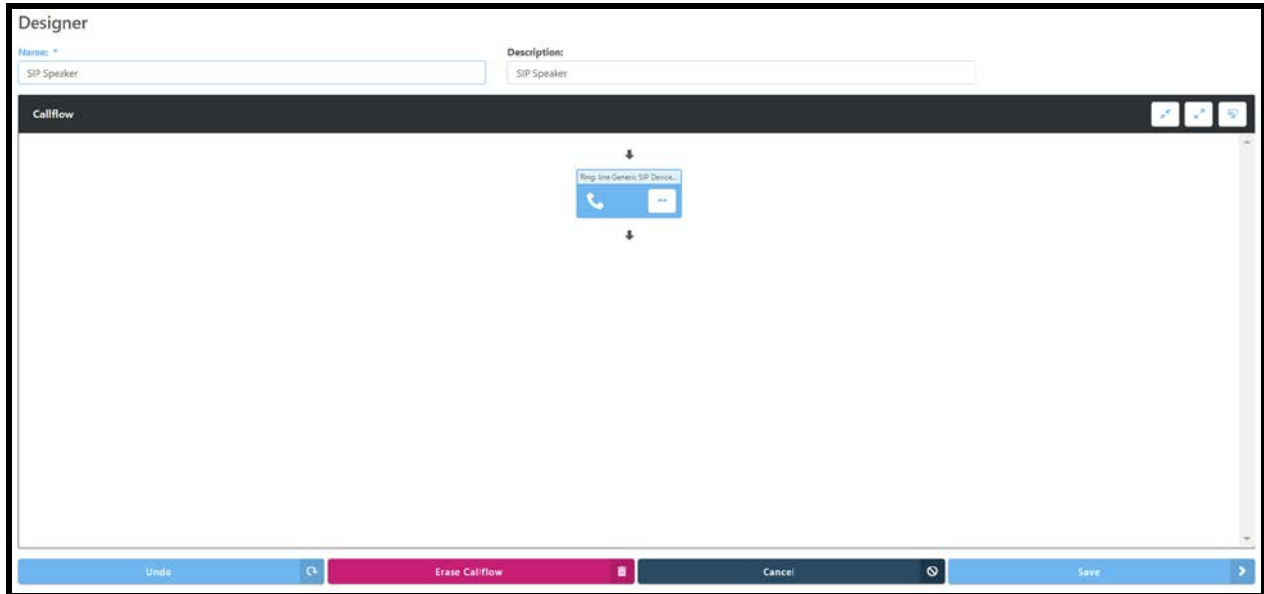
- Generic SIP Device - SIP Speaker - Line 1

Extra options

Cancel Ok

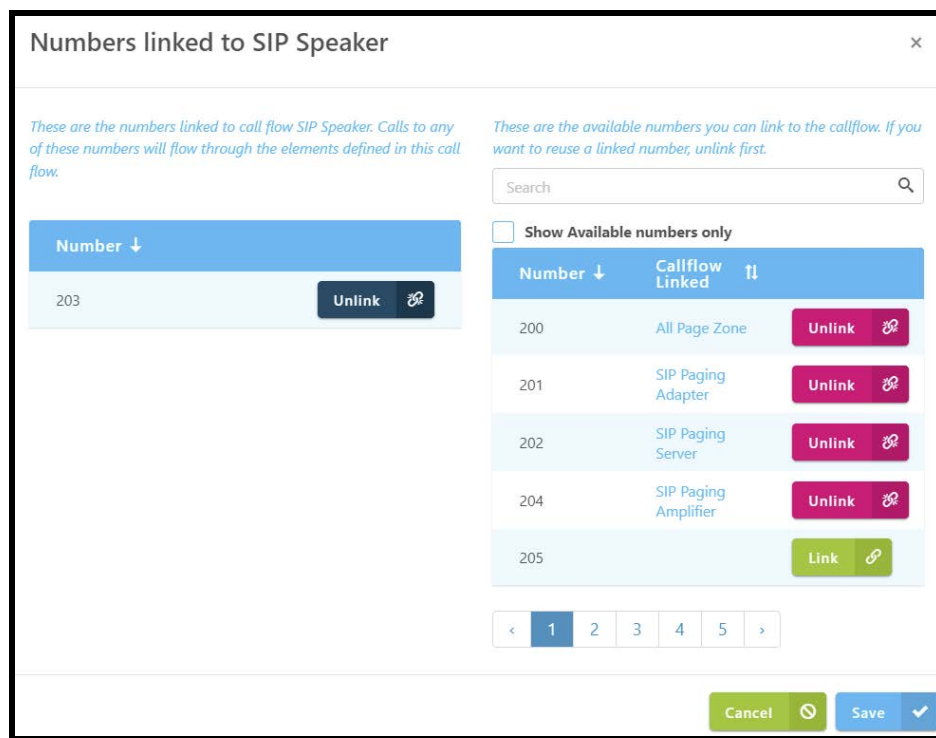
9. Press Ok to save the device to the callflow.
10. Press Save to save the callflow.

**Figure 4-6: Callflow Designer**



11. Click the **Save** button to create the Phone.
12. Next link a number to the new callflow.
13. Save the number to the callflow.

**Figure 4-7: Link a Number**



## 5.0 Configuration Procedure: Setting up the Paging Extension

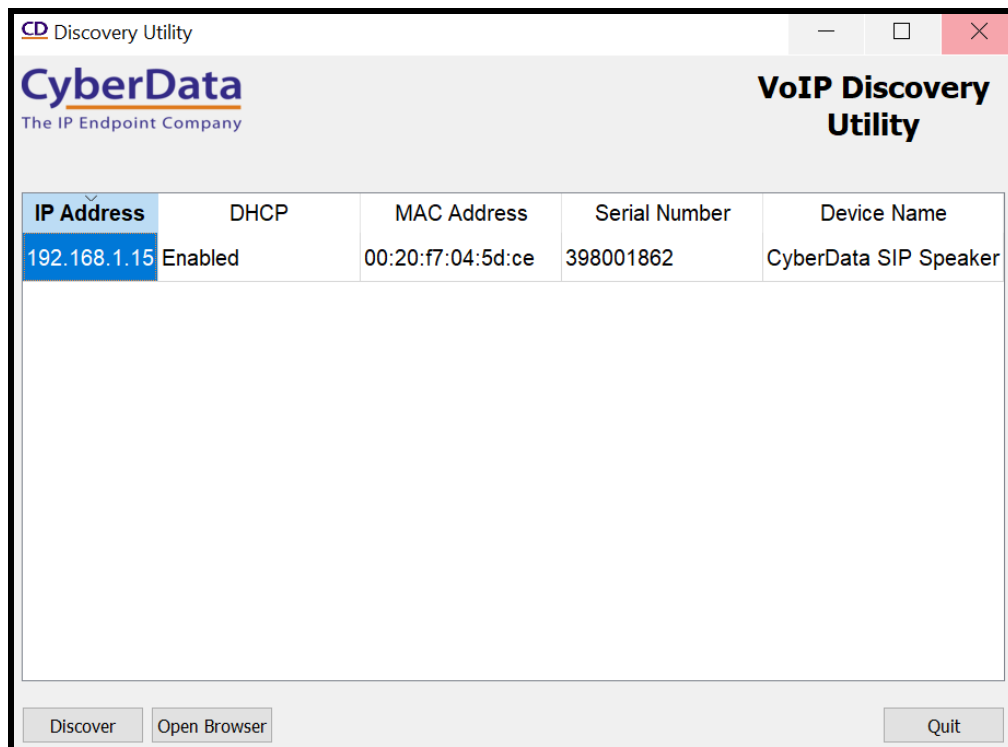
For configuring through the web interface, use the following steps to login to the web interface of your CyberData device.

**Table 5-1: Setting Name correlation**

| CyberData Setting         | Blueface Email    |
|---------------------------|-------------------|
| Primary SIP Server        | SIP Server        |
| Primary SIP User ID       | Username          |
| Primary SIP Auth ID       | Authentication ID |
| Primary SIP Auth Password | Password          |

1. Click **Launch Browser** from the CyberData Discovery Utility or point your browser to the CyberData device's IP address to access the Home Page of the web interface.

**Figure 5-1: CyberData Discovery Utility**



2. Enter the default credentials when prompted and click the **Log In** button.

Username: admin

Password: admin

**Figure 5-2: Home Tab**

The screenshot shows the 'Home' tab of the CyberData SIP Speaker configuration interface. The top navigation bar includes tabs for Home, Device, Audio, Network, SIP, Multicast, SSL, Sensor, Audiofiles, Events, Autopro, and Firmware. The main content area has a light blue background with the title 'CyberData SIP Speaker' in large black font. Below the title, there are three main sections: 'Current Status', 'Admin Settings', and 'Import Settings'. The 'Current Status' section displays various system information including Serial Number, Mac Address, Firmware Version, IP Addressing, IP Address, Subnet Mask, Default Gateway, DNS Server 1, DNS Server 2, SIP Mode, Multicast Mode, Event Reporting, Nightringer, and a list of SIP Servers (Primary, Backup, Nightringer, Monitor) with their registration status. The 'Admin Settings' section includes fields for Username, Password, and Confirm Password, along with 'Save', 'Reboot', and 'Toggle Help' buttons. The 'Import Settings' section has a 'Choose File' button, a 'No file chosen' message, and an 'Import Config' button. Below these is an 'Export Settings' section with an 'Export Config' button.

3. Navigate to the SIP tab.

***Note:** All SIP credentials are listed in an email sent by Blueface after the device was added to the platform.*

4. Set the **Primary SIP Server** to the value listed for SIP Server.
5. Set the **Primary SIP User ID** to the value listed for the Username.
6. Set the **Primary SIP Auth ID** to the value listed for the Authentication ID.
7. Set the **Primary SIP Auth Password** to the value listed for the Password.
8. Set the **Remote SIP Port** to 5062.

**Figure 5-2: SIP Tab**

| SIP Settings   |                                     | Nightringer Settings                   |                          |
|--|-------------------------------------|--|--------------------------|
| Enable SIP operation:  | <input checked="" type="checkbox"/> | Enable Nightringer:                    | <input type="checkbox"/> |
| SIP Transport Protocol:  | UDP ▾                               | SIP Server:                            | 10.0.0.253               |
| TLS Version:   | 1.2 only (recommended) ▾            | Remote SIP Port:                       | 5060                     |
| Verify Server Certificate:   | <input type="checkbox"/>            | Local SIP Port:                        | 5061                     |
| Register with a SIP Server:  | <input checked="" type="checkbox"/> | Outbound Proxy:                        |                          |
| Use Cisco SRST:  | <input type="checkbox"/>            | Outbound Proxy Port:                   | 0                        |
| Primary SIP Server:  | cust-uc-us.nsvconnect.com           | User ID:                               | 241                      |
| Primary SIP User ID:   | HwD8a5ZHpfAGKDcGEUyB                | Authenticate ID:                       | 241                      |
| Primary SIP Auth ID:   | HwD8a5ZHpfAGKDcGEUyB                | Authenticate Password:                 | *****                    |
| Primary SIP Auth Password:   | *****                               | Re-registration Interval (in seconds): | 360                      |
| Backup SIP Server 1:   |                                     |  |                          |
| Backup SIP User ID 1:  |                                     |  |                          |
| Backup SIP Auth ID 1:  |                                     |  |                          |
| Backup SIP Auth Password 1:  |                                     |  |                          |
| Backup SIP Server 2:   |                                     |  |                          |
| Backup SIP User ID 2:  |                                     |  |                          |
| Backup SIP Auth ID 2:  |                                     |  |                          |
| Backup SIP Auth Password 2:  |                                     |  |                          |
| Remote SIP Port:   | 5062                                |  |                          |
| Local SIP Port:  | 5060                                |  |                          |
| Outbound Proxy:  |                                     |  |                          |
| Outbound Proxy Port:   | 0                                   |  |                          |
| Monitor User ID:   |                                     |  |                          |
| Monitor Authenticate ID:   |                                     |  |                          |
| Monitor Authenticate Password:   |                                     |  |                          |
| Disable rport Discovery:   | <input type="checkbox"/>            |  |                          |
| Buffer SIP Calls:  | <input type="checkbox"/>            |  |                          |
| Re-registration Interval (in seconds):   | 360                                 |  |                          |
| Unregister on Boot:  | <input type="checkbox"/>            |  |                          |
| Keep Alive Period:   | 10000                               |  |                          |
| <input type="button" value="Save"/> <input type="button" value="Reboot"/> <input type="button" value="Toggle Help"/> |                                     |  |                          |

| RTP Settings     |            |
|------------------|------------|
| RTP Port (even): | 10500      |
| Jitter Buffer:   | 50         |
| SRTP:            | Disabled ▾ |

| Call Disconnection          |   |
|-----------------------------|---|
| Terminate Call after delay: | 0 |

| Codec Selection       |                          |
|-----------------------|--------------------------|
| Force Selected Codec: | <input type="checkbox"/> |
| Codec:                | PCMU (G.711, u-law) ▾    |

| Button Settings     |               |
|---------------------|---------------|
| Dial Out Extension: | 204           |
| Extension ID:       | Classroom 102 |

## 9. Save and Reboot.

Once the speaker finishes rebooting the unit should show Registered on the home tab.



**Figure 5-3: Home Tab – Registered**

The screenshot displays the CyberData SIP Speaker configuration web interface. At the top, there is a navigation bar with tabs: Home, Device, Audio, Network, SIP, Multicast, SSL, Sensor, Audiofiles, Events, Autoprov, and Firmware. The 'Home' tab is selected. Below the navigation bar, the main heading 'CyberData SIP Speaker' is centered. The interface is divided into four main sections: Current Status, Admin Settings, Import Settings, and Export Settings.

**Current Status**

Serial Number: 398001862  
Mac Address: 00:20:f7:04:5d:ce  
Firmware Version: v12.1.1

IP Addressing: DHCP  
IP Address: 192.168.1.15  
Subnet Mask: 255.255.255.0  
Default Gateway: 192.168.1.1  
DNS Server 1: 192.168.1.1  
DNS Server 2:

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

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

**Admin Settings**

Username: admin  
Password:  
Confirm Password:

Buttons: Save, Reboot, Toggle Help

**Import Settings**

Choose File No file chosen  
Import Config

**Export Settings**

Export Config

## 5.1 Configuration Procedure: Setting up the Nightringer Extension

The Nightringer Extension is a secondary extension that will ring when called. This makes the Nightringer extension ideal for use in ring groups.

1. Navigate to the web interface of the speaker.

**Figure 5-4: Home Tab**

The screenshot shows the 'Home' tab of the CyberData SIP Speaker web interface. The top navigation bar includes tabs for Home, Device, Audio, Network, SIP, Multicast, SSL, Sensor, Audiofiles, Events, Autopro, and Firmware. The main content area has a light blue background with the title 'CyberData SIP Speaker' in large black font. Below the title, there are three main sections: 'Current Status', 'Admin Settings', and 'Import Settings'. The 'Current Status' section displays various system information including Serial Number, Mac Address, Firmware Version, IP Addressing, IP Address, Subnet Mask, Default Gateway, DNS Server 1, DNS Server 2, SIP Mode, Multicast Mode, Event Reporting, and Nightringer status. The 'Admin Settings' section includes fields for Username, Password, and Confirm Password, along with 'Save', 'Reboot', and 'Toggle Help' buttons. The 'Import Settings' section has a 'Choose File' button and an 'Import Config' button. Below these sections is an 'Export Settings' section with an 'Export Config' button. At the bottom, there is a list of SIP servers with their registration status.

| Current Status      |                   |
|---------------------|-------------------|
| Serial Number:      | 398001862         |
| Mac Address:        | 00:20:f7:04:5d:ce |
| Firmware Version:   | v12.1.1           |
| IP Addressing:      | DHCP              |
| IP Address:         | 192.168.1.15      |
| Subnet Mask:        | 255.255.255.0     |
| Default Gateway:    | 192.168.1.1       |
| DNS Server 1:       | 192.168.1.1       |
| DNS Server 2:       |                   |
| SIP Mode:           | Enabled           |
| Multicast Mode:     | Enabled           |
| Event Reporting:    | Disabled          |
| Nightringer:        | Disabled          |
| Primary SIP Server: | Not registered    |
| Backup Server 1:    | Not registered    |
| Backup Server 2:    | Not registered    |
| Nightringer Server: | Not registered    |
| Monitor SIP Server: | Not registered    |

| Admin Settings   |       |
|--|-------|
| Username:  | admin |
| Password:  |       |
| Confirm Password:  |       |
| <input type="button" value="Save"/> <input type="button" value="Reboot"/> <input type="button" value="Toggle Help"/> |       |

| Import Settings                              |                |
|--|----------------|
| Choose File                                  | No file chosen |
| <input type="button" value="Import Config"/> |                |

| Export Settings                              |  |
|--|--|
| <input type="button" value="Export Config"/> |  |

2. Navigate to the SIP tab.

**Note:** All SIP credentials are listed in an email sent by Blueface after the device was added to the platform.

3. Check the box **Enable Nightringer**.
4. Set the **SIP Server** to the value listed for SIP Server.
5. Set the **User ID** to the value listed for the Username.
6. Set the **Authenticate ID** to the value listed for the Authentication ID.
7. Set the **Authenticate Password** to the value listed for the Password.
8. Set the **Remote SIP Port** to 5062.

**Figure 5-5: SIP Tab - Nightringer**

**SIP Settings**

Enable SIP operation: ☒

SIP Transport Protocol:

TLS Version:

Verify Server Certificate: ☐

Register with a SIP Server: ☒

Use Cisco SRST: ☐

Primary SIP Server:

Primary SIP User ID:

Primary SIP Auth ID:

Primary SIP Auth Password:

Backup SIP Server 1:

Backup SIP User ID 1:

Backup SIP Auth ID 1:

Backup SIP Auth Password 1:

Backup SIP Server 2:

Backup SIP User ID 2:

Backup SIP Auth ID 2:

Backup SIP Auth Password 2:

Remote SIP Port:

Local SIP Port:

Outbound Proxy:

Outbound Proxy Port:

Monitor User ID:

Monitor Authenticate ID:

Monitor Authenticate Password:

Disable rport Discovery: ☐

Buffer SIP Calls: ☐

Re-registration Interval (in seconds):

Unregister on Boot: ☐

Keep Alive Period:

**Nightringer Settings**

Enable Nightringer: ☒

SIP Server:

Remote SIP Port:

Local SIP Port:

Outbound Proxy:

Outbound Proxy Port:

User ID:

Authenticate ID:

Authenticate Password:

Re-registration Interval (in seconds):

**RTP Settings**

RTP Port (even):

Jitter Buffer:

SRTP:

**Call Disconnection**

Terminate Call after delay:

**Codec Selection**

Force Selected Codec: ☐

Codec:

**Button Settings**

Dial Out Extension:

Extension ID:

**9. Save and reboot the speaker.**

If the credentials were added correctly, when the unit finishes rebooting Registered in Green should appear next to Nightringer Status on the Home Tab.

**Figure 5-6: Nightringer Registered**

Home Device Audio Network SIP Multicast SSL Sensor Audiofiles Events Autoprov Firmware

# CyberData SIP Speaker

## Current Status

Serial Number: 398001862  
Mac Address: 00:20:f7:04:5d:ce  
Firmware Version: v12.1.1

IP Addressing: DHCP  
IP Address: 192.168.1.15  
Subnet Mask: 255.255.255.0  
Default Gateway: 192.168.1.1  
DNS Server 1: 192.168.1.1  
DNS Server 2:

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

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

## Admin Settings

Username:   
Password:   
Confirm Password:

## Import Settings

No file chosen

## Export Settings

## 6.0 Using CyberData SIP Speakers.

CyberData Speakers are designed for one-way communication. When a call is made to the device an announcement can be made. The units can be used by directly calling the SIP extension, in a page group, or with multicast. This makes the speaker's extremely versatile paging endpoints

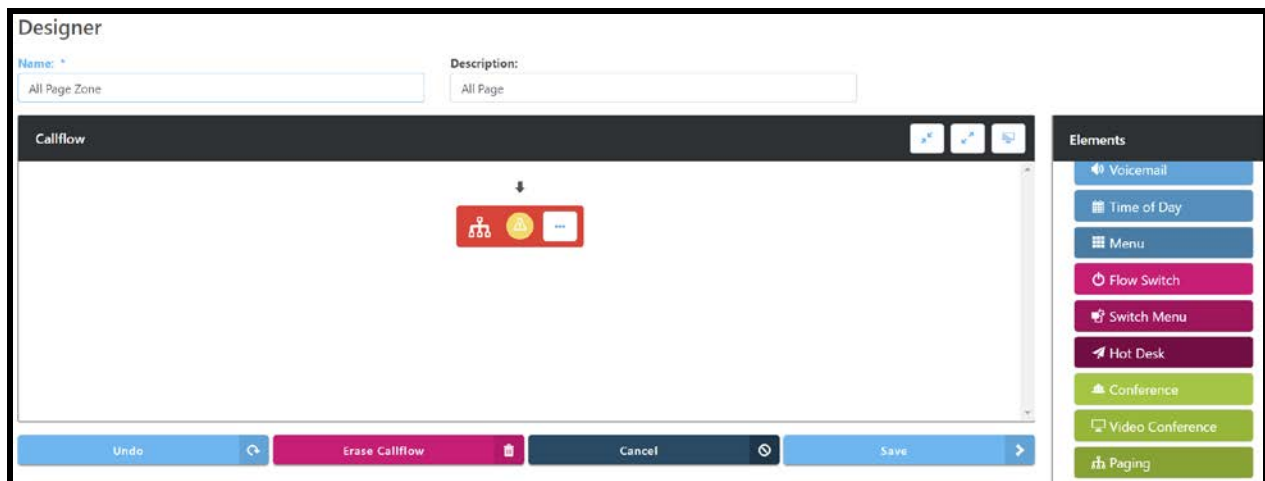
CyberData SIP Speakers also come in a talkback format which allows the speaker to be used for two-way communication. This will typically require the use of the accessory Remote Call Button [\(011508\)](#). The addition of the accessory call button allows a call to be made from the speaker, although the speaker can still be called and used in a two-way conversation without the call button.

### 6.1 Setting up a page group

After registering the device to Blueface, a page group can be created which allows a call to be made which can reach multiple endpoints simultaneously. This allows for zoned paging directly through the service and does not require additional hardware.

1. Select **Callflows** in Phones on Blueface.
2. Name the new callflow and set a description.
3. In the callflow designer select the Paging Element.

**Figure 6-1: Page Group Designer**



4. After adding the paging element, click on it to assign users.
5. Add all necessary users for the paging group.

**Figure 6-2: Paging Element Creation**

**Paging Element Settings**

① PIN:

① Display Name:

**Devices**

Search

Yealink VP-T49G IP Video Phone

Generic SIP Device

< 1 2 >

**Selected destinations**

- Generic SIP Device - SIP Paging Amplifier - Line 1
- Generic SIP Device - SIP Paging Adapter - Line 1
- Generic SIP Device - SIP Speaker - Line 1
- Generic SIP Device - SIP Strobe - Line 1
- Generic SIP Device - Office Ringer - Line 1
- Generic SIP Device - SIP Paging Server - Line 1

**Extra options**

No media available

**Media Uploader**

① Broadcast media:

Drag & Drop Files or Browse

Maximum file size 5mb

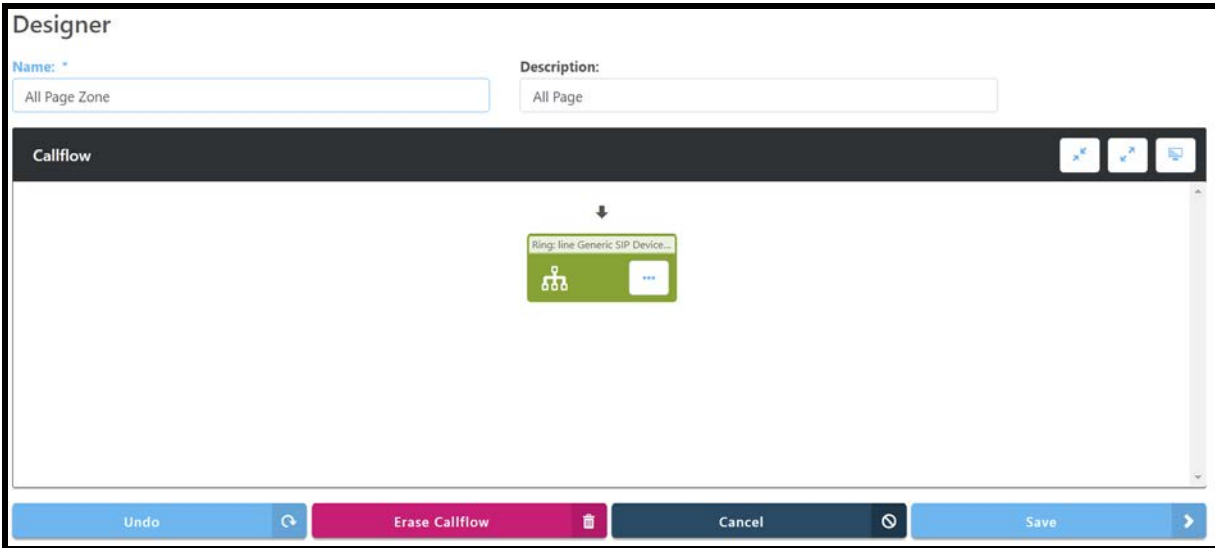
Acceptable file formats: wav, ogg, aac, wma, flac, mp1, mp2, mp3, mp4, opus, ra

① Accessible From Off Net: ☐ OFF

① Bypass on Call Protection: ☐ OFF

6. Once all the desired users are added, press Ok.
7. Next save the new call flow.

**Figure 6-3: Callflow Created**



8. After saving the callflow click **Linked Numbers** to set an extension number for the paging group callflow.

**Figure 6-4: Linked Numbers**



9. Set a number for the paging group.

**Figure 6-5: Linking Number**

### Numbers linked to All Page Zone

These are the numbers linked to call flow All Page Zone. Calls to any of these numbers will flow through the elements defined in this call flow.

| Number ↓ |        |
|----------|--------|
| 200      | Unlink |

These are the available numbers you can link to the callflow. If you want to reuse a linked number, unlink first.

Search

☐ Show Available numbers only

| Number ↓ | Callflow Linked      |        |
|----------|----------------------|--------|
| 201      | SIP Paging Adapter   | Unlink |
| 202      | SIP Paging Server    | Unlink |
| 203      | SIP Speaker          | Unlink |
| 204      | SIP Paging Amplifier | Unlink |
| 205      |                      | Link   |

<

1

2

3

4

5

>

Cancel

Save

**10.** Press Save to save the number to the callflow.

The callflow is now ready to be used. When called it will send a SIP call to all group elements and allow a page to be made.



## 6.2 Multicast Setup

Most CyberData devices support Multicast which is a protocol that allows for easy paging on a local area network (LAN). This section will illustrate how to setup the device to listen for multicast and the different settings that work with multicast.

**Figure 6-6: Multicast Tab**

### Multicast Settings

Enable Multicast Operation: ☒

| Priority | Address      | Port  | Name              | Buffer                   | Beep                                | Relay                    |
|----------|--------------|-------|-------------------|--------------------------|-------------------------------------|--------------------------|
| 9        | 239.168.3.10 | 11000 | Emergency Warning | <input type="checkbox"/> | <input checked="" type="checkbox"/> | <input type="checkbox"/> |
| 8        | 239.168.3.9  | 10000 | All Page          | <input type="checkbox"/> | <input checked="" type="checkbox"/> | <input type="checkbox"/> |
| 7        | 239.168.3.8  | 9000  | Warehouse Only    | <input type="checkbox"/> | <input checked="" type="checkbox"/> | <input type="checkbox"/> |
| 6        | 239.168.3.7  | 8000  | Unused            | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |
| 5        | 239.168.3.6  | 7000  | Unused            | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |
| 4        | 239.168.3.5  | 6000  | Unused            | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |
| 3        | 239.168.3.4  | 5000  | Unused            | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |
| 2        | 239.168.3.3  | 4000  | Unused            | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |
| 1        | 239.168.3.2  | 3000  | Unused            | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |
| 0        | 239.168.3.1  | 2000  | Background Music  | <input type="checkbox"/> | <input type="checkbox"/>            | <input type="checkbox"/> |

Polycom Default Channel

Polycom Priority Channel

Polycom Emergency Channel

SIP calls are considered priority 4.5

Port range can be from 2000-65535

Priority 9 is the highest and 0 is the lowest

A higher priority audio stream will always supersede a lower one

\* You need to reboot for changes to take effect

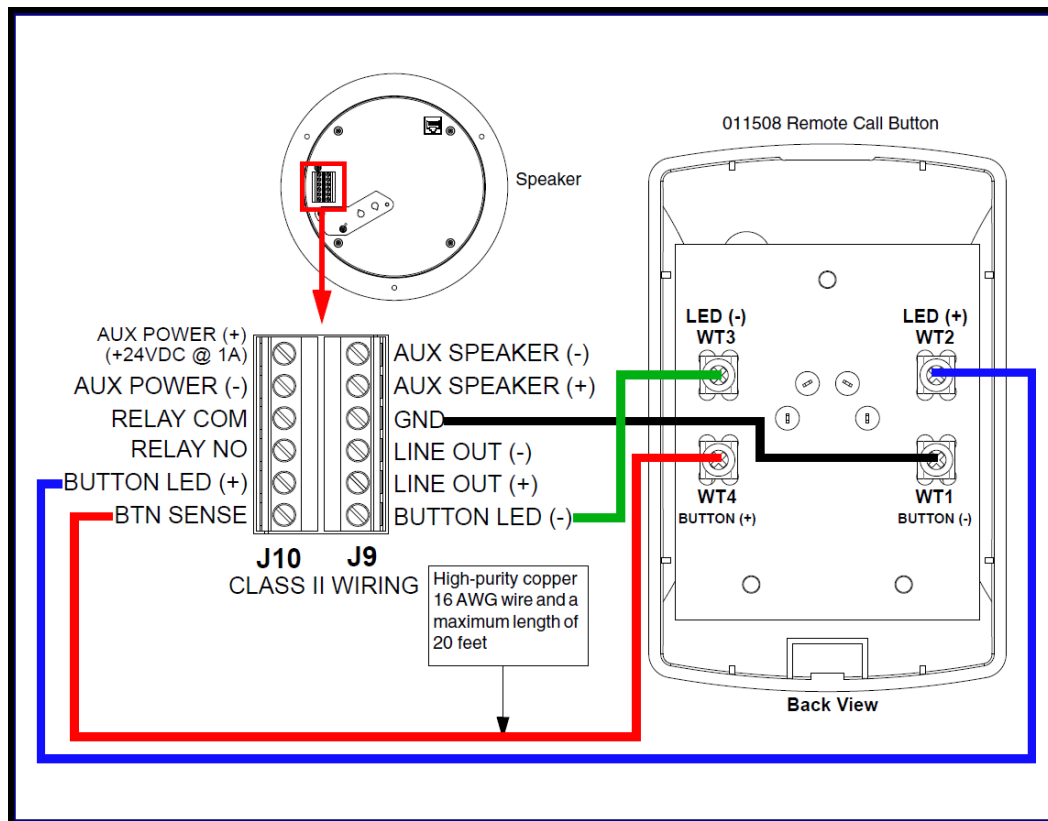
The multicast engine workings on priority, higher priority supersedes a lower priority. CyberData recommends setting all pages or emergency pages to a higher priority, this will prevent a non-emergency message playing over any emergency notifications. There are also options to Buffer the message, play a beep tone before the message or enable the onboard relay for the duration of the message.

## 6.3 Remote Call Button Setup

This section walks through the process of wiring and enabling the Remote Call Button ([011508](#)). While this accessory is not required to use the speaker in a two-way call, the button is typically installed to allow a call to be made from the speaker.

To use the button it must first be wired to the speaker. The button uses a four-wire connection which notifies the speaker of the call button being pressed but also applies power to the button to illuminate the LED.

**Figure 6-7: Button Wiring Diagram**



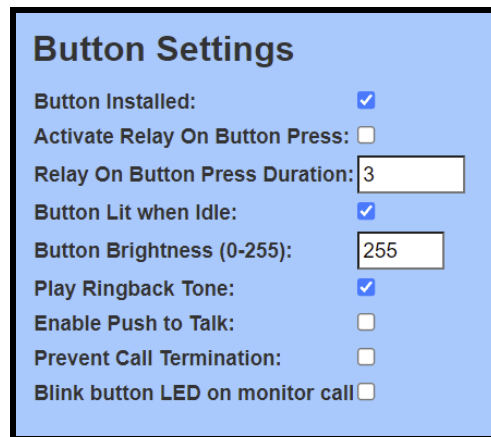
1. With the speaker disconnected from power use a small screwdriver to wire the button to the speaker.

**Note:** CyberData recommends a 16 AWG wire with a maximum length of 20 feet.

2. After wiring the button to the speaker, connect the speaker back to power and mount both the speaker and button in place.
3. As the speaker boots up the button's LED ring should illuminate.

4. Navigate to the SIP Talkback Speaker's web interface.
5. From the Home tab, click on the Device Tab.
6. On the Device Tab, check the box for **Button Installed** located in the Button Settings section.
7. Adjust any settings as necessary, CyberData recommends using the setting **Play Ringback Tone** to alert the user of the speaker that the call is being setup.

**Figure 6-8:** Button Settings

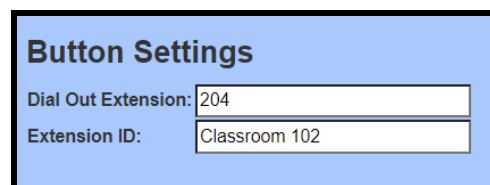


| Button Settings                  |                                     |
|----------------------------------|-------------------------------------|
| Button Installed:                | <input checked="" type="checkbox"/> |
| Activate Relay On Button Press:  | <input type="checkbox"/>            |
| Relay On Button Press Duration:  | <input type="text" value="3"/>      |
| Button Lit when Idle:            | <input checked="" type="checkbox"/> |
| Button Brightness (0-255):       | <input type="text" value="255"/>    |
| Play Ringback Tone:              | <input checked="" type="checkbox"/> |
| Enable Push to Talk:             | <input type="checkbox"/>            |
| Prevent Call Termination:        | <input type="checkbox"/>            |
| Blink button LED on monitor call | <input type="checkbox"/>            |

***Note:** The setting **Prevent Call Termination** is useful in some scenarios where a call should not be ended from the speaker. CyberData finds this setting to be used frequently with the talkback speakers when located in schools. This will prevent a second press of the button on the speaker from ending the call.*

8. Save.
9. Navigate to the SIP Tab.
10. Set the **Dialout Extension** to the number that will be called when the accessory button is pressed.
11. Set the **Extension ID** to the location of the speaker

**Figure 6-9:** Button Settings – Dialout



| Button Settings     |  |
|---------------------|--|
| Dial Out Extension: | <input type="text" value="204"/>           |
| Extension ID:       | <input type="text" value="Classroom 102"/> |

12. Save.
13. Reboot.

Once the speaker has booted back up, pressing the call button accessory will trigger a call from the speaker.

## 7.0 Contact CyberData Corporation

### Sales

For sales-related questions, please visit our [Contact CyberData Sales](#) web page for more information.

### Technical Support

For CyberData Technical Support, please submit a [Contact CyberData VoIP Technical Support](#) form on our website.

The CyberData VoIP Technical Support Contact form initiates a troubleshooting ticket which CyberData uses for quality assurance purposes.

Additionally, the Contact VoIP Tech Support form tells us which phone system you are using, the make and model of the network switch, and other essential troubleshooting information we need to efficiently assist with a resolution. Please also include as much detail as possible in the Describe Problem section of the form. Your installation is extremely important to us.

### Documentation Feedback

We realize changes to the software or hardware of the PBX solution may render this document obsolete. We welcome and encourage documentation feedback to ensure continued applicability.