Algo SIP Endpoints and Zoom Phone Interoperability Testing and Configuration Steps

Need Help?

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Introduction

Algo SIP Endpoints can register to Zoom Phone as a third-party SIP Endpoint and provide Paging, Ringing as well as Emergency Alerting capability.

This document provides instructions to add your Algo device to the Zoom web portal. Interoperability testing results are also available.

All testing has been conducted with the Algo 8301 Paging Adapter and Scheduler, 8186 SIP Horn, and 8201 SIP PoE Intercom. These are representative of all Algo SIP speakers, paging adapters and doorphones and similar registration steps would apply. Exceptions are the 8028 SIP Doorphone and 8128 Strobe Light. Please contact Algo Support to inquire.

Please note only one SIP extension may be registered to any given Algo endpoint at a time with Zoom Phone. Multiple Lines feature will be release later in the year. For more information, please contact Zoom support.
Configuration Steps – Zoom Web Portal

To register an Algo SIP Endpoint to Zoom Phone begin by creating a new common area phone in the Zoom web portal. See the Zoom support site for more information.

1. Sign in to the Zoom web portal.
2. In the navigation menu, click Phone System Management then Phones & Devices.
3. Click Add and enter the following information:

<table>
<thead>
<tr>
<th>Field</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>Assigned To</td>
<td>Grant MacLaren</td>
</tr>
<tr>
<td>Display Name</td>
<td>Paging device</td>
</tr>
<tr>
<td>MAC Address</td>
<td>0022ee1e0123</td>
</tr>
<tr>
<td>Device Type</td>
<td>Algo/Cyberdata</td>
</tr>
</tbody>
</table>

- **Assigned To**: Enter the username or email of the phone user to assign the device to. If you have multiple sites, the device will belong to the same site as the phone user.
- **Display Name**: Enter a display name to identify the device.
- **MAC Address**: Enter the 12-digit MAC address of a supported device.
- **Device Type**: Select Algo/Cyberdata and Paging&Intercom.

4. Click Save.
5. Click **Provision** to view the SIP credentials. You will need this information to complete provisioning using the Algo Web Interface.

![Provisioning](image-url)

- **MAC Address**: 00-22-e8-12-00-25
- **Device Type**: Other

You will need to enable TLS1.2 for SIP registration and enable SRTP for secure calling on your IP phone. Please refer to your manufacturer's instructions for these processes.

**You'll need following information for manual provisioning.**

1. **SIP Domain**: 50744587.zoom.us
2. **Outbound Proxy**: us01sipj06.zoom.us:5091
3. **User Name**: [Provide User Name]
4. **Authorization ID**: [Provide Authorization ID]
5. **Password**: [Provide Password]

**Also, download CA certificate and import to trust list on your IP phone.**

**Note:** Please note that Zoom support team will not be able to troubleshoot or configure IP phones that are provisioned in this manner. Some Zoom Phone features may not work on manually provisioned phones. It may vary depending on your desk phone model.
Configuration Steps – Algo Endpoint

To register an Algo SIP Endpoint navigate to the Web Configuration Interface.

1. Open a web browser.
2. Type the IP Address of the endpoint. If you don’t know the address yet, navigate to www.algosolutions.com, find the user guide for your product and go through the Getting Started section.
3. Log in and go to Basic Settings -> SIP tab.
4. Enter the information provided from Zoom as per below:

- SIP Domain (Proxy Server) – Zoom SIP Domain
- Page or Ring Extension – Zoom User Name
- Authentication ID – Zoom Authorization ID
- Authentication Password – Zoom Password
5. Go to Advanced Settings -> Advanced SIP.
6. Set the SIP Transportation protocol to “TLS”.
7. Enter the Outbound Proxy provided by Zoom.

8. Ensure the SIP Registration Status shows “Successful”.

Welcome to the Algo 8301 Paging Adapter & Scheduler Control Panel

Setting up your Paging Adapter & Scheduler:

Step 1: Configure your Paging Adapter & Scheduler
Log in with the default password and use the Basic Settings pages to set up the basic information.

Step 2: Check network settings (Optional)
Use the Network page under the Advanced Settings tab to change network settings. The default setting for the device is to obtain its IP address from a DHCP server. Contact your Network System administrator if you plan to assign a static IP address, Mask, and Gateway to the device.

Step 3: Secure your Paging Adapter & Scheduler (Optional)
Use the Admin page under the Advanced Settings tab to change the administrator password.
Changing the password is extremely important if the device is directly connected to a public network.

Step 4: Register your Paging Adapter & Scheduler (Optional)
Please register your product using the link below:
http://www.algosolutions.com/register
Registration ensures your access to the latest upgrades to this product and important service notices.
Note: if registering additional extensions for ringing, paging or emergency alerting, enter the unique credentials for the respective extension in the same way.

Only one SIP extension may be registered to any given Algo endpoint at a time with Zoom Phone. Multiple Lines feature will be release later in the year. For more information, please contact Zoom support.
Interoperability Testing

Register to Zoom Phone

- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn, 8201 SIP PoE Intercom
- Firmware: 1.7.6
- Description: Verify 3rd Party SIP Endpoints are registered successfully.
- Result: **Successful**

Register Multiple SIP Extensions Simultaneously

- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn
- Firmware: 1.7.6
- Description: Verify the server will sustain multiple simultaneous extensions registered to the same endpoint (e.g. page, ring, and emergency alert).
- Result: **Not supported at this time. Please see note below.**

Please note only one SIP extension may be registered to any given Algo endpoint at a time with Zoom Phone. Multiple Lines feature will be release later in the year. For more information, please contact Zoom support.

One-Way Page

- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn
- Firmware: 1.7.6
- Description: Verify one-way page mode functionality, by calling the registered page extension.
- Result: **Successful**

Two-Way Page

- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn, 8201 SIP PoE Intercom
- Firmware: 1.7.6
- Description: Verify two-way page mode functionality, by calling the registered page extension.
- Result: **Successful**
Ringing
- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn
- Firmware: 1.7.6
- Description: Verify ringing mode functionality by calling the registered ring extension.
- Result: **Successful**

Emergency Alerts
- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn
- Firmware: 1.7.6
- Description: Verify emergency alerting functionality by calling the registered extension.
- Result: **Successful**

Outbound Calls
- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn, 8201 SIP PoE Intercom
- Firmware: 1.7.6
- Description: Verify emergency alerting functionality by calling the registered extension.
- Result: **Successful**

TLS for SIP Signaling
- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn, 8201 SIP PoE Intercom
- Firmware: 1.7.6
- Description: Verify TLS for SIP Signaling is supported.
- Result: **Successful**

SDP SRTP Offer
- Endpoints: 8301 Paging Adapter and Scheduler, 8186 SIP Horn, 8201 SIP PoE Intercom
- Firmware: 1.7.6
- Description: Verify support for SRTP calling using Standard (RTP/SAVP) or optional (RTP/AVP) profile.
- Result: **Successful**
Troubleshooting

SIP Registration Status = “Rejected by Server”

**Meaning:** The server received register request from the endpoint and responds with an unauthorized message.

- Ensure the SIP credentials (extension, authentication ID, password) are correct.
- Under Basic Settings > SIP, click on the blue circular arrows to the right of the Password field. If the Password is not what it should be, the web browser is probably auto filling the password field. If so, any change on a page containing a password could be filled in with an undesired string.

SIP Registration Status = “No reply from server”

**Meaning:** the device is not able to communicate across the network to the phone server.

- Double check the "SIP Domain (Proxy Server)", under Basic Settings > SIP tab field is filled out correctly with the address of your server and port number.
- Ensure the firewall (if present) is not blocking the incoming packets from the server.
- Ensure TLS is configured for SIP Transportation Method (Advanced Settings > Advanced SIP).