



A MITEL
PRODUCT
GUIDE

OpenScape Solution Set V11

Zoom with Mitel OpenScape Voice and Mitel OpenScape SBC

Phone System Integration (PSI) Solution Guide

11/2025

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1 History of Changes

Issue	Date	Summary
1	02/2025	The first issue of the guide.
2	03/2025	Updated and enhanced the following chapters: <ul style="list-style-type: none"> Configuring RTP settings on page 20. Endpoint Overview on page 26. Configuring OpenScape SBC on page 27. Configuring Media Profiles on page 29. Enabling an OpenScape Voice Subscriber for a Zoom connection on page 50. Creating a new SBC configuration for Zoom users on page 53. Configuring an E911 Solution. Appendix A: Restrictions and Known Issues on page 79.
3	03/2025	Updated the following chapters: <ul style="list-style-type: none"> Configuring Media Profiles on page 29. Appendix A: Restrictions and Known Issues on page 79.
4	03/2025	Updates and enhancements in the entire document.
5	04/2025	Updated the Prerequisites on page 7 section.
6	06/2025	Updated the following chapters: <ul style="list-style-type: none"> Licensing on page 19. Enabling an OpenScape Voice Subscriber for a Zoom connection on page 50.
7	06/2025	Updates and enhancements in the entire document.
8	06/2025	Updated the following chapters: <ul style="list-style-type: none"> Adding Zoom integration to a customer account. Adding the Zoom integration to the Customer CloudLink Account on page 15. Enabling an OpenScape Voice Subscriber for a Zoom connection on page 50.
9	06/2025	Added note in Configuring the OpenScape SBC Endpoint on page 21.
10	09/2025	Updated Appendix A: Restrictions and Known Issues on page 79.

History of Changes

Issue	Date	Summary
11	10/2025	<p>Updated the following chapters:</p> <ul style="list-style-type: none">• Configuring Push Notifications on page 34.• Configuring OpenScape Xpressions Visual Voicemail on page 58. <p>Added new chapter Configuring proxy settings in Xpression monitor on page 59.</p> <p>Applied overall enhancements and updates in the document.</p>
12	11/2025	Applied overall enhancements and updates in the document.

2 Introduction

This document outlines how to connect the OpenScape Voice (OSV) and OpenScape SBC (OSSBC) to Zoom.

Zoom is a cloud-based phone system that provides voice communication features such as call management, call forwarding, voicemail, and integration with Zoom Meetings. The Zoom - Mitel Phone System Integration (PSI) solution offers a hybrid communication model that enables users to maintain the telecom functions with their OpenScape Voice system while extending its functionality with Zoom's cloud-based features.

This integration allows Zoom's Phone tab to become a SIP Softphone that registers to the Mitel Calling Platform, utilizing the OpenScape SBC. The OpenScape SBC acts as a secure connection between your on-premises Mitel PBX and Zoom Workplace clients. This allows various Zoom PSI endpoints - including desktop clients, mobile devices, and desk phones - to connect directly to your Mitel system via SIP registration.

The Zoom-Mitel PSI integration requires all key components to be properly configured within the user environment. This guide provides the essential setup steps to establish secure, reliable communication paths that will enhance your organization's collaboration capabilities.

2.1 Target audience

This document is intended for professionals involved in configuring and managing the Zoom-OSV-SBC integration, specifically those working with OpenScape Voice (OSV) and OpenScape Session Border Controller (OSSBC) components. The target audience includes implementation engineers, field technicians, system administrators, business partners, solution providers, and customers directly involved in the deployment and management of the integration.

2.2 Prerequisites

Supported product versions

Product	SW Version (minimum)
Zoom Workplace Client	6.3.6
OpenScape Voice with active SSP / SWA (Unify Software Support / Mitel Software Assurance)	V10R3.33.3
OpenScape SBC	V11 R2.1.1
OpenScape SBC Management Portal	V10R2.4.0
OpenScape UC ISO files	V10R6 FR7 HF1
OpenScape Common Management Platform	V10R6
OpenScape Xpressions	V7 R1.5.0 Zoom Package HF1

System Requirements and Licenses

- **Zoom**
 - Supported License per user which includes Zoom Workplace Business/ Business+, Zoom Workplace Essential/sEnterprise/Enterprise+/ Enterprise Premier, Legacy Meeting Licenses ENH/EAH.
 - Zoom Phone initial setup is completed. See the *How to Complete Initial Setup* section in [Zoom Phone initial setup](#).
 - Automatic Phone assignment for Zoom Workplace Licences is disabled. For more information, refer to [Configuring automatic Zoom Phone activation](#).
 - The *Zoom Phone External Contacts* setup for any non-Zoom, Mitel users is completed.
 - When logged in to the Zoom web portal as an Admin with the privilege to edit account settings, the **Phone System Integration** tab should appear under **Account Management**. If it does not, please contact Zoom.
 - For any additional requirements, refer to the [Zoom-Mitel Phone System Integration support page](#).
- **OSV resource synchronization**

IMPORTANT: Before proceeding with a planned customer migration to Zoom, ensure that a successful OSV resource synchronization has been completed and verified. This step is essential to identify any potential issues early in the process.

When the administrator selects the OpenScape Voice resource, the system displays the available phone numbers retrieved through OSV resource synchronization. OSV resources are synchronized with OpenScape UC during UC startup and periodically thereafter.

For detailed information on resource synchronization, see section **OpenScape Voice Signaling Protocol Data Synchronization** in the [Mitel OpenScape UC Application V11 Configuration and Administration](#) documentation.

2.2.1 User Roles and Permissions

- **Mitel Administration**

To set up the CloudLink and Zoom integration, you must be a CloudLink account admin. This role is assigned to a user by a Mitel Partner or by an Account Admin.

Account Admins can:

- Add, edit, or delete users (including other Account Admins).
- Enable or disable administrative rights for users.
- Configure the integration and connect the on-premise software to CloudLink.

Regular CloudLink users cannot perform this setup. For more information, refer to the [Mitel Administration User Guide](#).

NOTICE:

A Zoom admin does not need to be a CloudLink admin, unless they also need to manage CloudLink settings.

End users authenticate through Zoom and do not interact with CloudLink directly. They will not see or manage CloudLink settings.

After the initial setup, the integration operates using **service accounts**, ensuring continued functionality without requiring individual admin access.

- **Zoom**

- Zoom Account, Business or Enterprise.
- Account Owner or Admin with a Role for managing Users, Phone System Integration, and Zoom Phone.

To manage users, Phone System integrations and maintain a stable and functional integration, you must create and use a dedicated **Admin user** (service account) with a unique email address on your Zoom site specifically for the integration. This account must remain active and should not be deactivated.

Do not use a regular user account, as deactivating it (e.g., if the user leaves) will cause connection failure.

For more information, refer to [Managing users](#).

2.2.2 Network requirements

Zoom Workplace

The Zoom Workplace app uses the standard Firewall ports and IP ranges. To add the required Firewall rules, refer to the **Zoom firewall rules** and **Firewall rules for Zoom website** sections in the [Zoom network firewall or proxy server settings](#) page.

Zoom PSI

Firewall rules for incoming traffic:

When the Client is on the Internet, the following incoming traffic must be allowed for the Zoom PSI client:

- SIP over TLS (TCP port 5061): The Zoom PSI client, like any other SIP remote client, must be able to connect to the external firewall of the OSSBC, which is located in the DMZ. It should be able to connect to the SIP/TLS port of the access interface of OSSBC via the external firewall. The default port value is 5061.
- SRTP media traffic (UDP): SRTP packets should be forwarded to the access interface of the SBC, based on the configured port range of the OSSBC.

Firewall rules for outgoing traffic:

When the Client is on-premise, the following outgoing traffic must be allowed for the Zoom PSI client:

- DNS (UDP/TCP port 53): PSI client needs to resolve the FQDN of the external firewall of the OSSBC using DNS.
- SIP over TLS (TCP port 5061): PSI client must be connected to the external firewall of the OSSBC using the SIP/TLS. The default value is 5061.
- SRTP media traffic (UDP) : SRTP must be allowed to reach the external firewall of the OSSBC, based on the configured port range of the OSSBC.

3 Overview

This chapter provides an overview of the integration solution, detailing the key components, actions, and configurations necessary for phone system integration. It also outlines how the Zoom, SBC, and OSV configurations interact with each other and other critical components of the solution.

3.1 Key Components

The key components of the Zoom-OSV-OSSBC phone system integration solution are as follows:

- **Zoom Web Portal:** Allows you to customize your profile and configure your Zoom settings. When setting up your Zoom-OSV integration, you are granted admin access to the Zoom web portal.
- **OpenScape Voice (OSV):** OpenScape Voice is a scalable SIP-based IP system that provides enterprise-class communications functionality. The OSV platform controls the users' telephony capabilities and manages SIP signaling, call routing, and PSTN and Zoom Phone integration.
- **OpenScape SBC (OSSBC):** OpenScape SBC is a software-based network border element designed to deliver superior Voice over IP (VoIP) security and cost savings. It serves as the secure gateway between OSV and external networks, managing SIP traffic flow and ensuring protected communications.
- **CloudLink Platform:** Mitel CloudLink Platform enables communication between the on-premise PBX (such as OpenScape Voice) and cloud-based applications. Acting as the intermediary, CloudLink platform bridges the Zoom and OSV systems, ensuring seamless account integration.

To properly associate a gateway with a new customer account on the CloudLink platform, the Mitel Partner or the Account Admin must access the Mitel Administration.

- **CloudLink Daemon:** CloudLink Daemon is a software component designed for integration with multiple unified communication platforms. It complements the CloudLink gateway, which connects premise-based PBXs to the CloudLink platform and CloudLink applications, by enabling additional features.

The CloudLink Daemon is embedded in the OpenScape Voice platform. Its primary function is to facilitate the connection with Mitel CloudLink enabled applications such as Zoom PSI.

- **Mitel Administration:** Mitel Administration is a web-based application that enables Mitel Partners to create and manage customer accounts. It also allows the Account Administrator of a customer account to manage the account and its users.

Once CloudLink integration is enabled, users within a customer account can access various Mitel applications and third-party CloudLink applications.

- **Unify Common Management Platform:** This web-based configuration interface supports the OpenScape communications solution. It provides tools for connecting CloudLink, managing Zoom configurations, and integrating Zoom users with OSV users (subscribers).

Overview

Overview of the Phone System Integration (PSI) solution

For an overview of the documentation set for each key component and additional configuration details, please refer to the [Related Documentation](#) section and the references throughout this guide.

3.2 Overview of the Phone System Integration (PSI) solution

OpenScape Voice (OSV) and OpenScape Session Border Controller (SBC) work together to enable seamless integration between Zoom clients and external networks, including the Public Switched Telephone Network (PSTN).

The OpenScape SBC acts as a secure gateway, managing network boundaries, while OpenScape Voice handles core telephony functions including SIP message manipulation and call routing. This integration establishes reliable communication paths between Zoom accounts and OSV subscribers, ensuring smooth call flow in both directions.

The following chapters provide detailed configuration instructions for integrating Zoom, OSV, and OSSBC components into your system.

The guide begins with the configurations needed for the [Zoom-CloudLink integration](#) configurations, followed by the steps required before setting up and provisioning Zoom-OSV users.

IMPORTANT: Before proceeding with a planned customer migration to Zoom, ensure that a successful OSV resource synchronization has been completed and verified. This step is essential to identify any potential issues early in the process.

For detailed information on resource synchronization, see section [OpenScape Voice Signaling Protocol Data Synchronization](#) in the [Mitel OpenScape UC Application V11 Configuration and Administration](#) documentation.

Setting Up Users for the Zoom-OSV Integration

The Zoom-OSV integration involves several key steps:

- 1) Configuring Zoom and CloudLink integration.
- 2) Configuring OpenScape Voice and OpenScape SBC.
- 3) The CloudLink Daemon must be integrated with OpenScape Voice, OpenScape SBC and Common Management Platform.
- 4) Establishing CloudLink connectivity for managing Zoom users associated with OSV subscribers.
- 5) Adding users (if not already added) and assigning licenses in your Zoom account.
- 6) Configuring user mapping between Zoom and OSV through the Unify Common Management Platform.
- 7) Completing user provisioning to finalize the integration.

Your system is fully integrated upon completing the configuration steps outlined or referenced in this document. The integration benefits users with seamless call routing, enhanced communication security, efficient traffic management between Zoom clients and the OSV platform, and a unified user experience across cloud and on-premises systems.

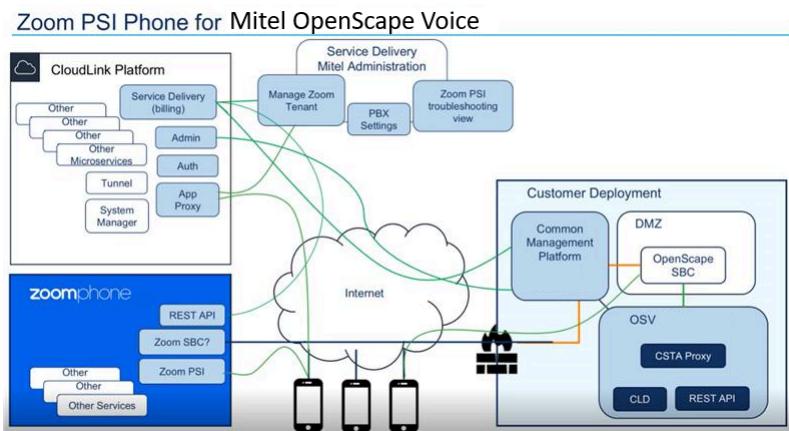


Figure 1: Network Topology Block Diagram

3.3 Related Documentation

Mitel Phone System Integration (PSI) with Zoom Guides

- OpenScape Solution V11, Mitel Phone System Integration (PSI) with Zoom, User Guide
- Mitel Phone System Integration (PSI) with Zoom Troubleshooting Guide, Service Documentation
- OpenScape Solution V11, Zoom with OpenScape Voice and OpenScape SBC BYOC (Bring Your Own Carrier)
- Mitel Phone System Integration (PSI) with Zoom Troubleshooting Guide Service Documentation

OpenScape Voice

- OpenScape Voice V11 Administrator Documentation
- OpenScape Voice V11 Service Manual, Service Documentation
- OpenScape Voice V11 Service Manual: Installation and Upgrades
- OpenScape Voice V11 Interface Manual: Volume 4 CSTA Interface Description
- Mitel OpenScape UC Application V11 Configuration and Administration

OpenScape SBC

- OpenScape SBC V11 Administration Guide
- OpenScape SBC V11 Configuration Guide, Administration Documentation
- OpenScape SBC V11 Installation Guide
- OpenScape SBC V11R2 Security Checklist
- OpenScape SBC Troubleshooting Guide, Service Documentation

CloudLink

- CloudLink Daemon Solution Guide
- CloudLink Gateway User Guide
- Mitel Administration User Guide

OpenScape Common Management Platform

- OpenScape Common Management Platform V10 Administrator Documentation

Zoom

- [Zoom-Mitel Phone System Integration support page](#)

4 Integrating Zoom with CloudLink

This chapter describes the Zoom and CloudLink integration.

Before proceeding with the configurations, ensure you have the required user accounts and permissions. For more information, see [User Roles and Permissions](#) on page 9.

4.1 Adding the Zoom integration to the Customer CloudLink Account

You can configure your Mitel PSI connection via the Zoom web portal.

Prerequisites

You have a Zoom account, Business or Enterprise.

You are an Account Owner or Admin with a role for managing Users, Phone System Integration, and Zoom Phone.

You have obtained the necessary Zoom PSI licensing.

The **Mitel PSI** app is published into the Zoom Marketplace. For more information, refer to [Zoom-Mitel Phone System Integration support page](#). If you cannot see the app, please contact your Zoom administrator or Mitel support representative.

IMPORTANT:

To maintain a stable and functional integration, you must create and use a dedicated admin user (service account) with a unique email address on your Zoom site specifically for the integration. This account must remain active and should not be deactivated. The provisioning of the Mitel PSI app to your CloudLink Platform (CLP) happens automatically once the proper setup is complete.

For more information, see [Prerequisites](#) on page 7.

Step by Step

1) Log in to the [Zoom App Marketplace](#).

2) Search for the **Mitel PSI** app using the search bar or filter options.

3) Click **Add** to install the app to your Zoom account.

You will be redirected to accounts.mitel.io to authorize the connection.

4) Click **Allow** to authorize the Mitel PSI application to access the necessary account information.

Next steps

To complete the integration, you must follow the steps described in [Completing the Mitel PSI App Setup in Mitel Admin](#) on page 16.

Integrating Zoom with CloudLink

Completing the Mitel PSI App Setup in Mitel Admin

4.2 Completing the Mitel PSI App Setup in Mitel Admin

After adding the Zoom integration to a customer account, you must complete the integration setup.

Prerequisites

You have a Zoom Business or Enterprise account.

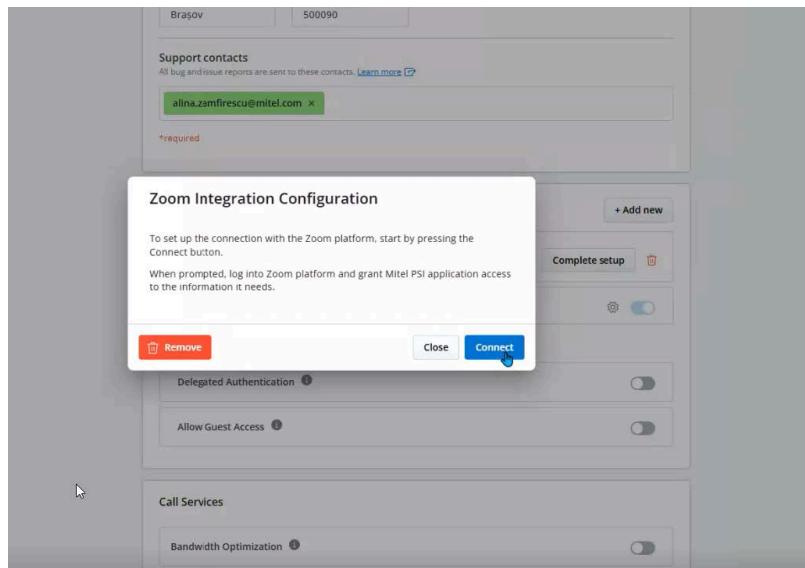
You have obtained the necessary Zoom PSI licensing.

You have created a dedicated service account (admin user) with a unique email address on your Zoom account portal, exclusively for the integration. For more information, see [User Roles and Permissions](#) on page 9.

You have added the Mitel PSI App to your Zoom Account, as described in [Adding the Zoom integration to the Customer CloudLink Account](#) on page 15.

Step by Step

- 1) Log in to [Mitel Administration](#) as an Account Admin.
- 2) Click **Account or Integrations and Applications** from the left main menu. The **Account Information** page of the customer account opens.
- 3) In the **Integrations** section, click **+ Add new**. The **Integrations** pop-up window opens.
- 4) Select the **3rd party** tab. A pop-up screen displays the available third-party integrations.
- 5) Scroll down to locate the **Zoom** integration and click **Add** next to it, then click **Done**.
- 6) The Zoom integration is added to the customer account and it is displayed in the **Integrations** section of the **Account Information** page.
- 7) Go to the newly added Zoom integration and click **Complete Setup**. The **Zoom Integration Configuration** window pops up.

8) Click **Connect**.

A **Zoom login** window will open in a new browser window.

In this window, sign in using the **Zoom Admin account** designated for this integration with CloudLink.

NOTICE: It is strongly recommended that this be a **dedicated service account** in Zoom, not a personal or user-based admin login.

Once signed in, Zoom will redirect the pop-up to the **Mitel PSI authorization page**.

9) Click **Allow** to authorize the Mitel PSI app to access the necessary data for integration.

After authorization is complete, the pop-up will close, and the main CloudLink Administration console will return to the Zoom integration screen.

The integration will now show a **Connected** status.

The **Mitel PSI** application will be automatically added to the Zoom Marketplace account for the customer under “Added Apps.”

10) If you click **Decline**:

- The integration will not proceed.
- You will remain on the **Zoom Authorization** page.

To continue, close the Zoom Authorization window, click **Connect** again on the Zoom Integration Configuration page, and then click **Allow** in the Zoom Authorization window.

NOTICE: Mitel Partner cannot enable integrations in the Partner Account as the integration with other applications is not supported for Partner Accounts. To integrate CloudLink with other applications, a Partner must create a customer account and enable integrations in that account. It is recommended to

disable any existing integrations in the Partner Account to have the full functionality of CloudLink features. For more information about Partner Accounts, see [Log in as a Mitel Partner](#).

Next steps

To remove Zoom integration from a customer account, refer to the [Mitel Administration User Guide](#).

Once the Zoom integration is completed in Mitel Administration, the 'Mitel PSI' application automatically appears in the Zoom Customer's Marketplace under the **Added Apps** section.

5 Licensing

To enable Zoom Workplace integration with OpenScape SBC, the following licenses are required for every user:

- 1) Zoom Hybrid License subscription: Required to support Zoom Phone with OpenScape Voice and OpenScape Session Border Controller.
- 2) OpenScape SBC V11 Zoom BYOC and BYOP Premises License
- 3) OpenScape SBC Base License: Required to support both desktop and mobile Zoom Clients.
- 4) OpenScape SBC Session License
- 5) OpenScape Voice User License (if applicable)
- 6) Voicemail Mailbox License (Optional)

If the above licenses are not readily available, they can be added using the order position specified below:

Table 1:

Description	Order Position	Quantity per User	Notes
Mitel Zoom Hybrid License	See CPQ under Subscription Offers / Mitel Zoom Hybrid License Subscription	1	
OpenScape SBC V11 Zoom BYOC and BYOP Premises License	L30220-D622-B264	1	Required for both new and existing OS SBC deployments. Added at zero cost as part of Zoom Hybrid Subscription.
OS SBC Base License	L30220-D622-B250	1	Required for new OS SBC deployments.
OS SBC Session License	L30220-D622-B251	2	Typically, 2 sessions per Zoom user : one for Zoom to PSTN and one for Zoom to OSV.
OpenScape Voice User License (if applicable)		1	Required per user if OpenScape Voice is used as the call manager.
Voicemail Mailbox License (Optional)		1	Required only if voicemail is needed via OSV.

6 Configuring OpenScape Voice

This chapter describes the Mitel OpenScape Voice configuration for connecting to OpenScape SBC. The purpose of this connectivity is for Mitel OpenScape Voice to provide the necessary SIP message manipulation and call routing facilities to OpenScape SBC so that the latter can interconnect to Zoom.

6.1 Configuring RTP settings

Step by Step

1) Navigate to **Administration > General Settings > RTP**.

2) By default, the **Srx/Main/UseFullIDnForZoom** RTP parameter is set to **RtpTrue**. Optionally, confirm that the default value is correctly set.

NOTICE:

This parameter is required to provide the full DN (E164 format) for calls between two Zoom PSI users.

If **Srx/Main/UseFullDnForZoom** is disabled, the number is displayed in a dialable format, such as an extension or any other format configured through Number Display Modification.

- 3) Click on the **hiQ/CCL/TrackingCategory** RTP parameter to edit its value.
The **Edit RTP parameter** window pops up.
- 4) In the **Value** field, enter **Non-ONS**.
- 5) Set the **Srx/Sip/UpdateMethodSessionTimingEnable** value to **RtpTrue**.

NOTICE:

Zoom does not fully support the *re-Invite* method for Session Refresh. Therefore, to avoid problems with long-duration calls, we recommend enabling the *Update* method option.

However, If your provider does not properly support the *Update* method for Session Refresh, calls longer than 30 minutes may drop. In this case, the administrator should configure the endpoint with the attribute **Do Not Support Update Method for Session Timing**.

6) Click **Save**.

6.2 Configuring Endpoints

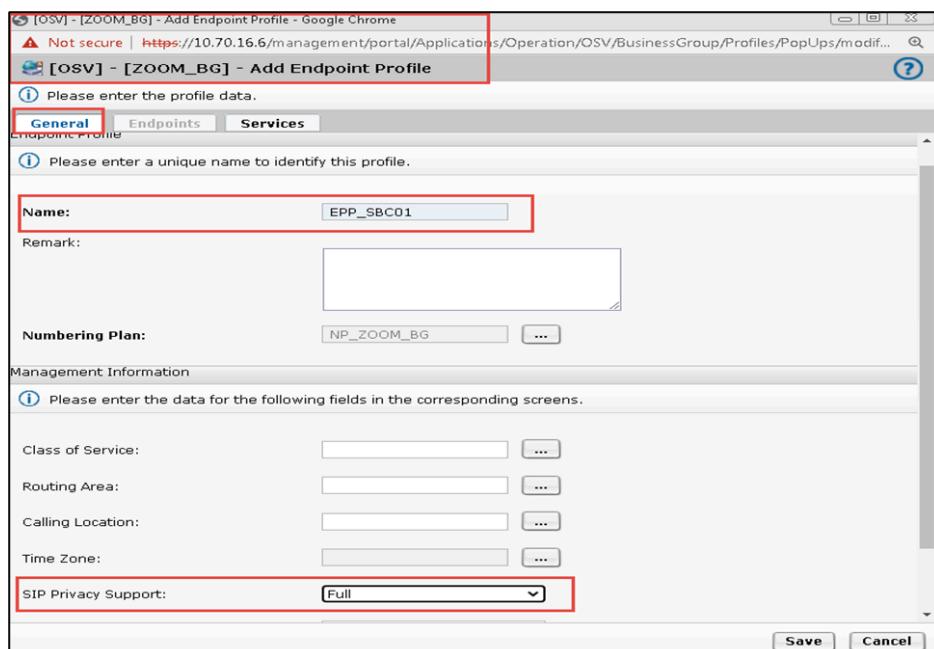
An **Endpoint** is a network component, such as an originating or terminating device and in our case the OpenScape SBC. An endpoint can be a DN (Directory Number) that does not have a number associated with it yet. An **Endpoint Profile** enables the administrator to set parameters for that endpoint.

6.2.1 Configuring the OpenScape SBC Endpoint

To configure the OpenScape SBC Endpoint Profile:

Step by Step

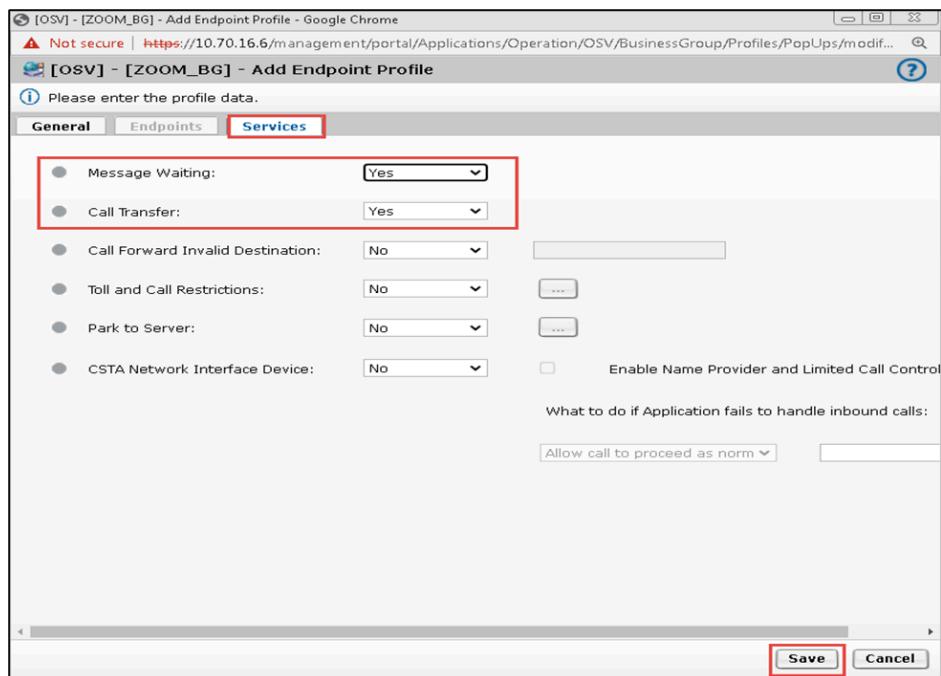
- 1) Navigate to **Unify OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Business Group List**.
- 2) From the **Business Group List** drop-down menu, select your **Business Group**. For example, **Zoom_PSI**.
- 3) In the selected Business Group, navigate to **Profiles > Endpoint** and click **Add**.
- 4) In the **Add Endpoint Profile** window, under the **General** tab, configure the following:
 - a) **Name**: Enter the name of the endpoint profile. For example, **EPP_SBC01**.
 - b) From the **SIP Privacy Support** drop-down menu, select **Full**.



The screenshot shows the 'Add Endpoint Profile' dialog box. The 'General' tab is selected. The 'Name' field contains 'EPP_SBC01'. The 'SIP Privacy Support' dropdown is set to 'Full'. The 'SIP Privacy Support' field is highlighted with a red box. Other fields like 'Numbering Plan', 'Management Information', and 'Class of Service' are also visible but not highlighted.

5) In the **Services** tab, enable the following required services by selecting **Yes** from the following drop-down menus:

- a) **Message Waiting**
- b) **Call Transfer**

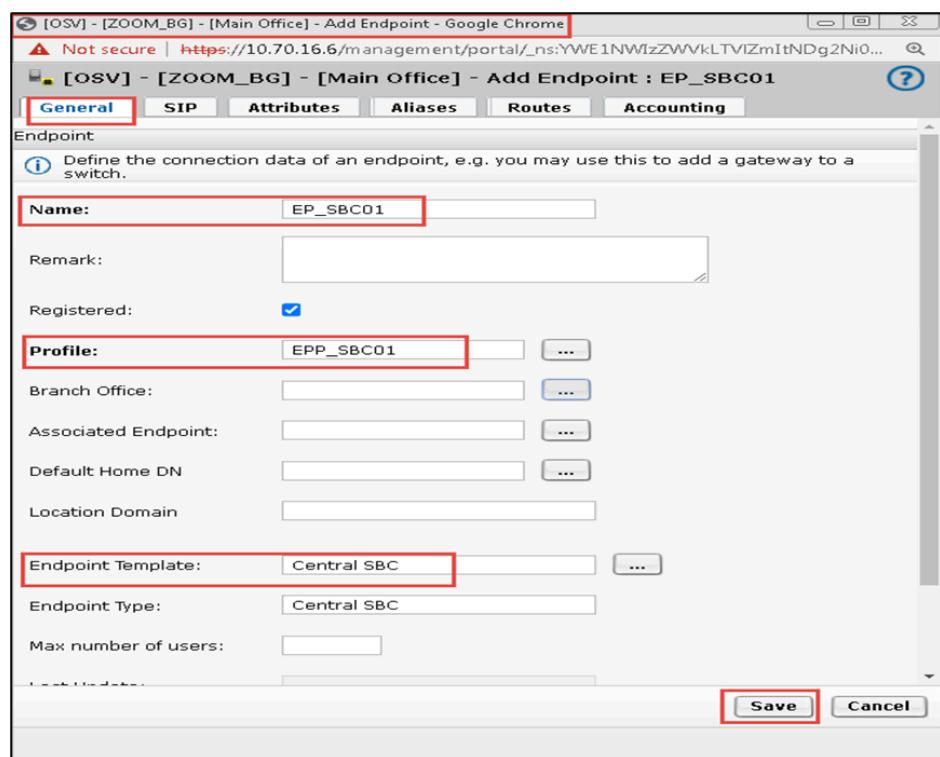


The screenshot shows the 'Add Endpoint Profile' dialog box in the Unify OpenScape Management Platform. The 'Services' tab is selected. The 'Message Waiting' and 'Call Transfer' options are both set to 'Yes' and are highlighted with a red box. Other service options like 'Call Forward Invalid Destination', 'Toll and Call Restrictions', 'Park to Server', and 'CSTA Network Interface Device' are set to 'No'. The 'Save' button at the bottom right is also highlighted with a red box.

- 6) Click **Save**.
- 7) To configure the SBC Endpoint, navigate to **Unify OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Members > Endpoints > .**
- 8) Click **Add**.

9) In the **General** tab, configure the following:

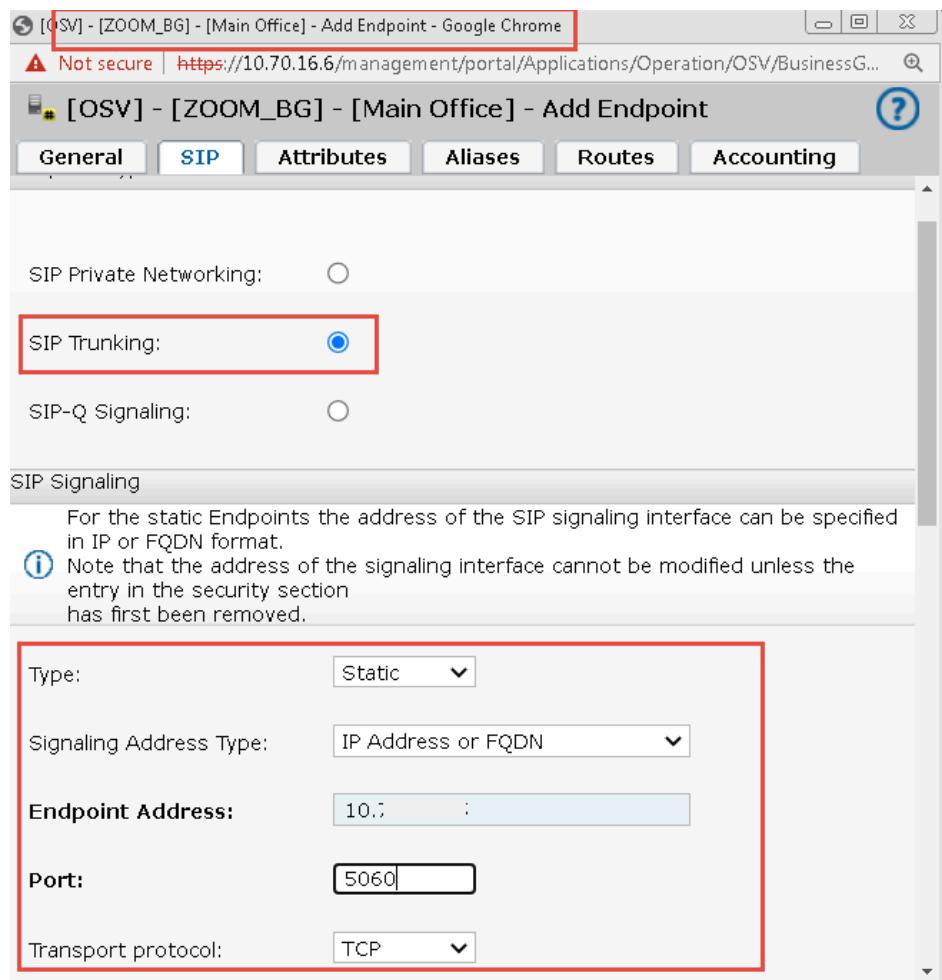
- Name:** Enter the name of the SBC endpoint. For example, **EP_SBC01**
- Profile:** Select the previously created endpoint profile. For example, **EPP_SBC01**.
- Endpoint Template:** Select **Central SBC** (set of pre-configured endpoint attributes).
- Click **Save**.



The screenshot shows the 'Add Endpoint' dialog box. The 'General' tab is active. The 'Name' field contains 'EP_SBC01'. The 'Profile' field contains 'EPP_SBC01'. The 'Endpoint Template' field contains 'Central SBC'. The 'Save' button is highlighted with a red box.

10) Select the **SIP** tab and configure the following:

- Select the **SIP Trunking** option to enable it.
- From the **Type** drop-down menu, select **Static** (it can be enabled only if the **SIP Proxy** attribute is enabled).
- From the **Signaling Address Type** drop-down menu, **select IP Address or FQDN** (route the calls via proxy).
- Endpoint Address:** Enter the SBC address.
- Port:** Enter the port number.
- From the **Transport protocol** drop-down menu, select **TCP**.



11) Locate the **Security** section, click **Edit**, and add the primary SIP port (5060) of the SBC.

The screenshot shows the configuration interface for an SBC endpoint. The 'SIP' tab is active. In the 'Security' section, there is a note: 'Set the Realm, Username and Password for digest authentication or configure the signaling address as a trusted one.' Below this, there is a 'Trusted' section with a 'Ports: N/A' field and an 'Edit...' button. At the bottom of the page are 'Save' and 'Cancel' buttons.

NOTICE:

For security reasons, and to ensure that unauthenticated requests from SBC or OSB do not pass through OSV, the SBC endpoint should **not** be configured as trusted in OSV.

12) Click **Save**.

13) The **Attributes** tab is populated automatically since the "Central SBC" template was selected in the General tab. Ensure that the following are selected:

- SIP Proxy
- Central SBC
- Route via Proxy
- Enable Session Timer

14) Select the **Aliases** tab and click **Add** to enter the SBC LAN interface for incoming SIP traffic.

15) Click **OK** and then click **Save**.

6.2.2 Endpoint Overview

- 1) Navigate to the **Unify OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Members > Endpoints** window.

A list of all the configured endpoints in Unify OpenScape Voice is displayed.

- 2) Select an endpoint and click **Edit**.
- 3) In the **Edit Endpoint** pop-up, under the **General** tab, check the **Registered** checkbox.

The screenshot shows the 'Edit Endpoint' dialog for endpoint EP_SBC01. The 'General' tab is selected. The 'Registered' checkbox is checked. Other tabs include 'SIP', 'Attributes', 'Aliases', 'Routes', and 'Accounting'. The main window shows a list of endpoints under the 'ZOOM_BG' business group, with EP_SBC01 selected.

- 4) Enable the **Registered** option for all the created endpoints.

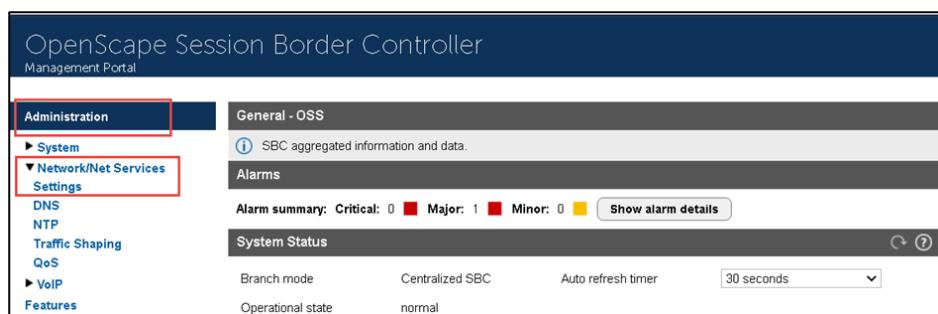
7 Configuring OpenScape SBC

This chapter outlines the configuration of OpenScape SBC for the integration with Zoom, allowing the Zoom PSI to function as a SIP softphone registered to OSV through the SBC over the Internet.

7.1 Configuring Network settings

It is recommended to follow the instructions below to configure the existing core and access interfaces and ensure safety and security.

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to **Administration > Network/Net Services > Settings**.



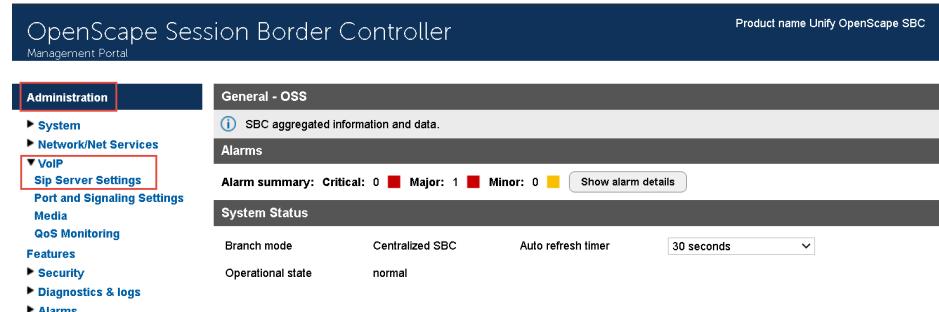
The **Network/Net Services** window opens. By default, the **Settings** tab is displayed.

- 3) Locate the **Access and Admin realm configuration** area and click **Add**.
- 4) Optionally, in the **Network/Net Services** pop-up, edit the following:
 - a) **SIP-UDP**: Set to 0, since it is not recommended to use SIP/UDP for remote subscribers.
 - b) **SIP-TCP**: Set to 0, since it is not recommended to use SIP/TCP for remote subscribers.
 - c) **SIP-TLS**: For security reasons, the default SIP/TLS port (5061) should be changed to a custom port (e.g., 65061).
 - d) Click **OK**.
- 5) Click **Apply Changes**.

7.2 Configuring SIP Server

- 1) Log in to the OpenScape SBC Management Portal.

2) Navigate to **Administration > VoIP > Sip Server Settings**.



The **VOIP** window opens.

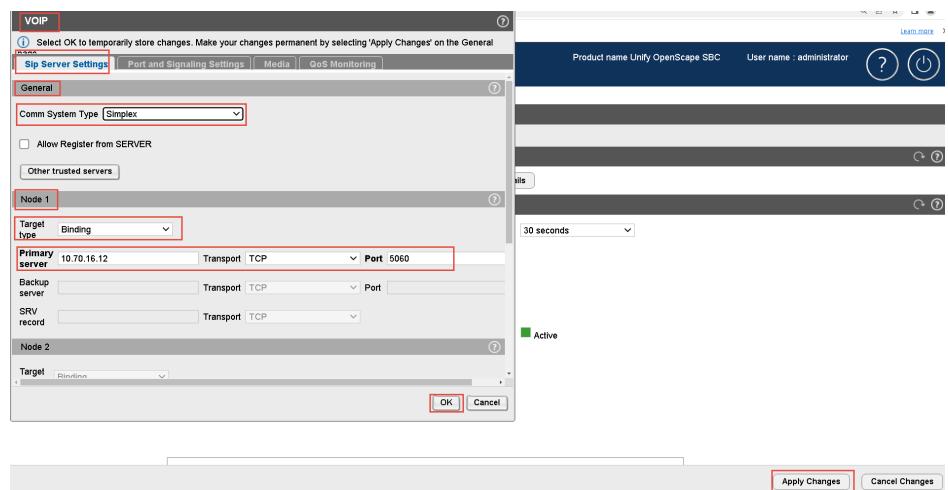
3) In the **Sip Server Settings** tab, enter the following:

a) Under **General**, from the **Comm System Type** drop-down menu, select **Simplex**.

NOTICE: The **Simplex** option depends on your OSV configuration. For example, if you have 2 OSV nodes, you will have to select the **Active-Standby**, **Collocated** or **Clustered** option.

b) Under the **Node 1** section:

- From the **Target type** drop-down menu, select **Binding**.
- Primary server:** Enter the OpenScape Voice IP address.
- From the **Transport** drop-down menu, select **TCP** (for both OS Voice Nodes 1 and 2).
- Port:** Enter **5060** (listening port for both OS Voice Nodes 1 and 2).



4) Click **OK**.

5) Click **Apply Changes**.

NOTICE: The OS Voice SIP Signaling Manager addresses for UDP/TCP/TLS can be found in OS Voice node's **node.cfg** file located in the **/etc/hiq8000** folder (parameters **sipsml_vip** for **OS Voice Node1** and **sipsm2_vip** for **OS Voice**).

Alternatively, the OS Voice SIPSM IP addresses can be found from CMP.

7.3 Configuring Media Profiles

In the **Media Profiles** settings, various SDP messages and audio (RTP) traffic parameters can be configured for the OpenScape SBC SIP endpoints to Zoom PSI client, SSP (PSTN provider), and Unify OpenScape Voice.

7.3.1 Configuring the Zoom PSI Media Profile

As an administrator, you must create a new user-agent header for Zoom PSI users (ZoomPbxPhone) and assign it a media profile, which supports SRTP only with SDES (cipher supported: AES-128). This media profile will be used by Zoom PSI users.

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to the **OpenScape SBC Management Portal > VoIP > Media** window.



Name	Media protocol	SRTP crypto context negotiation	Mark SRTP Call-leg as Secure
default	Best Effort SRTP	mikey + sdes	
webrtc_default	SRTP only	dlls	<input checked="" type="checkbox"/>

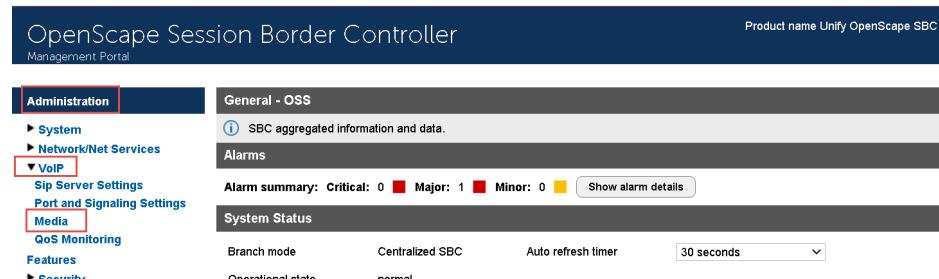
- 3) In the **Media Profiles** area, click **Add** to create the media profile for OpenScape SBC – OpenScape Voice connection.

Name	Media protocol	SRTP crypto context negotiation	Mark SRTP Call-leg as Secure
default	Best Effort SRTP	mikey + sdes	
webrtc_default	SRTP only	dlls	<input checked="" type="checkbox"/>

- 4) In the **Media profile** pop-up, locate the **General** section and configure the following:
 - **Name:** Enter the name of the media profile, such as **SRTP_SDES**.
 - From the **Media protocol** drop-down menu, select **SRTP only**
 - Check the **Keep sendonly attribute on NAT** option to allow the user to see the 'Remote Hold' indication.
- 5) Under the **SRTP configuration** area, select the **SDES** option (cipher supported: AES-128) for the **SRTP crypto context negotiation** configuration.
- 6) Under the **RTCP configuration** area, from the **RTCP Mode** drop-down menu, select **Bypass**.
- 7) In the **RTCP generation timeout** field, enter **4**.
- 8) Under the **Codec configuration** area, check the **Allow unconfigured codecs** checkbox.
- 9) Click **OK** to navigate back to the **VOIP** window.
- 10) Under **User agent**:
 - a) Click **Add**.
 - b) Enter the **ZoomPbxPhone** name.
 - c) Select from the **mediaProfile** drop-down menu the Media profile created for Zoom PSI.
- 11) Click **OK**.
- 12) Click **Apply Changes** on the SBC main page.

7.3.2 Configuring the OpenScape Voice Media Profile

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to the **Unify OpenScape SBC Management Portal > VOIP > Media** window.



- 3) In the **VOIP** pop-up, go to the **Media** tab.

4) Under **Media Handling**, check the following checkboxes:

- **Resend SRTP context upon key change**
- **Use single bridge/port for audio media**

It is recommended to enable the global parameter '**Use single bridge/port for audio media**' to ensure SBC maintains the same audio port with the PSI client during media renegotiation scenarios.



5) Click **OK**.

6) Click **Apply Changes** on the SBC main page.

7.4 Configuring Certificates

For secure communication with Zoom, a Trusted Certificate must be installed in OpenScape SBC.

Zoom Phone System Integration allows only TLS connections for SIP traffic from SBCs with a certificate signed by one of the Zoom-supported Certification Authorities.

The certificate must have the SBC FQDN as the common name (CN) in the subject field. Certificates with a wildcard in the certificate Subject Alternate Name field conforming to RFC2818 are also supported.

For more information about the **Security considerations for the Zoom-Mitel integration** and Zoom certificate management, refer to the [Zoom-Mitel PSI integration](#).

NOTICE: The list of trusted root authorities for Zoom services is maintained by Zoom and may change over time. Including static information from internal documents is not recommended due to potential changes without notice. Always rely on official Zoom documentation or support channels. For the most accurate and up-to-date information, users must contact Zoom Support directly. To contact Zoom Support, visit the [Zoom Support Contact Page](#) or reach out to your Zoom account representative.

For the OpenScape SBC TLS interconnection to Zoom, three files in 'pem' format are required from the Certification Authority:

- A certificate authority or certification authority (CA) certificate (for example, "ssl_root_with_chain.pem").

The CA certificate contains a public key and the owner's identity, ensuring an entity can be trusted.

- Server certificate for OSSBC (for example, "ServerCertificate.pem").
- OSSBC server certificate private key used for the CSR to CA (for example, "PrivateKey.pem").

The files above must be uploaded to OpenScape SBC for the TLS connection with the Zoom interface.

Prerequisites

Adequate administrative permissions.

Adequate knowledge of TLS certificate handling.

At least one OpenScape SBC is configured and in operation.

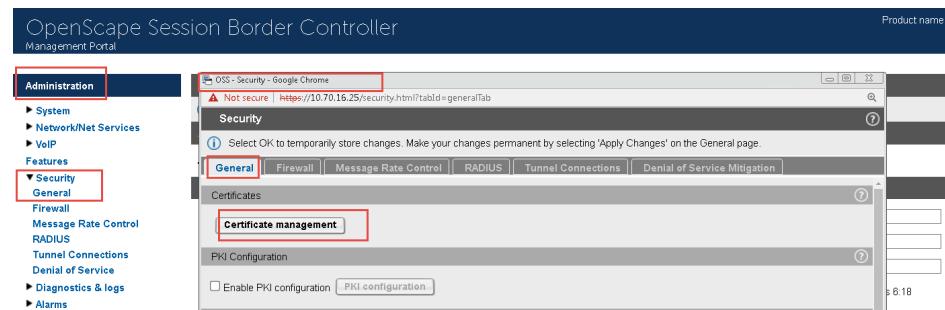
To configure Certificates:

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to **Security > Denial of Service**.
- 3) In the **Security** pop-up, under the **Dynamic Black List** section, check the **Process initial registration** flag to enable it.
- 4) Click **Ok**.
- 5) Navigate to OpenScape SBC Management Portal > **Security > General**.

The **Security** window opens.

- 6) Under the **Certificates** section, click **Certificate Management**.

The **Certificate Management** window appears with the **General Configuration** tab displayed as default.

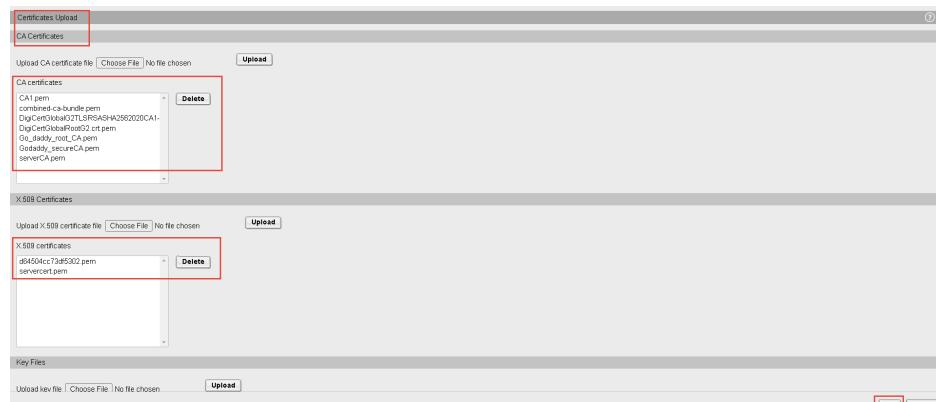


7) Scroll down to locate the **CA Certificate** area and click **Choose File** to browse and select the CA certificates.

Click **Upload**.

Under the **X.509 Certificate** area, click **Choose File** and browse to select the X.509 certificates.

Click **Upload**.



8) Under the **Key Files** section, click **Choose File** and browse to select the OSSBC server certificate private key.

Click **Upload**.

9) Locate the **Certificate Profiles** area and click **Add** to create a new Zoom certificate profile for remote subscribers.

The **Certificate Profile** window opens.

10) Configure the following parameters:

- Enter the **Certificate profile name**.
- From the **Certificate service** drop-down menu, select **SIP-TLS**.
- From the **Local server certificate file** drop-down menu, select the certificate file (for example, "ServerCertificate.pem").
- From the **Local CA file** drop-down menu, select the CA certificate (for example, "ssl_root_with_chain.pem").
- From the **Local key file** drop-down menu, select the private key file (for example, "PrivateKey.pem").

11) Click **OK**.

The **Certificate Profiles** section displays the newly created certificate profile.

12) Click **OK** in the **Certificate Management** window and in the **Security** window.

13) Click **Apply Changes** on the OpenScape SBC main page.

7.5 Configuring Remote Subscribers

Prerequisites

Adequate administrative permissions.

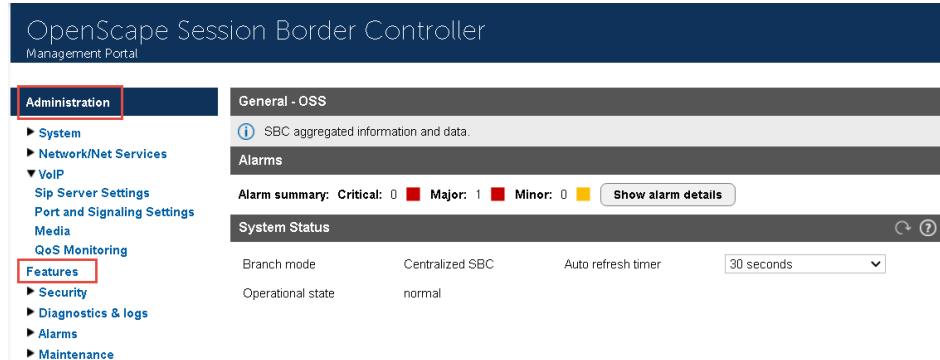
You have configured certificates for secure communication with Zoom.

You have created a Zoom PSI Media Profile.

Configuring OpenScape SBC

Configuring Push Notifications

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to the **Administration > Features**.



- 3) In the **Features** pop-up, check the **Enable Remote Subscribers** checkbox and click **Configure**.
- 4) In the **Remote Subscribers** window, check the **Enable register throttling** checkbox to enable it.
- 5) Check the **Enable register throttling for TLS** checkbox to enable it.

NOTICE:

For Zoom PSI mobile clients (iOS, Android), **Register throttling** must be bypassed because it interferes with the voicemail indicator.

In the **Throttling bypass (UA regex)** field, add the Zoom mobile user agents:

ZoomPbxPhone_Android_Phone | ZoomPbxPhone_IOS_Phone

- 6) Set the **Maximum registration expiry time (sec)** to **2678400** (31 days), to allow the registration from Zoom PSI mobile clients.
- 7) Click **Add** under the **Remote Subscribers configuration** area. The **Remote Subscribers configuration** window opens.
- 8) Configure the following:
 - a) **Subnet IP address**: Enter the SBC Subnet IP address as **0.0.0.0**.
 - b) **Subnet mask**: Enter the subnet mask value as **0.0.0.0**.
 - c) From the **Certificate profile** drop-down menu, select the certificate profile you created for Zoom.
 - d) From the **Media profile** drop-down menu, select the media profile assigned to Zoom (for example, **SRTP_SDES**).
 - e) Click **OK**.
- 9) Click **OK**.
- 10) Click **Apply Changes** on the OpenScape SBC main page.

7.6 Configuring Push Notifications

The mobile push notification service allows users to get notified about an update when they are not actively using their mobile app.

The mobile client will receive a push notification when a call is received.

This feature is supported on Android and iOS Zoom PSI clients.

You can set up mobile push notifications by configuring the required settings in the OpenScape SBC.

7.6.1 Enabling Push Notifications

As an administrator, you can enable the push notifications features required for the Zoom mobile application.

Step by Step

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to **Features**.
The **Features** window pops up.
- 3) Check the **Enable Push Notification Service** checkbox.
- 4) Click **Ok** and **Apply changes**.

8 Configuring CloudLink integrations

The CloudLink Daemon is a software component embedded in the OpenScape Voice platform. Its primary function is to facilitate the connection with Mitel CloudLink enabled applications such as Zoom PSI. This enables the management of Zoom users associated with the OpenScape Voice tenants, ensuring a secure connection through a proxy server. This chapter describes the CloudLink Daemon configuration for Common Management Platform, OpenScape Voice and OpenScape SBC.

IMPORTANT: Do not clone any servers where the CloudLink Daemon is installed (such as CMP, OSV/OS4k, or SBC).

Cloning may cause issues with the interconnection of Cloudlink Daemon with Cloudlink account/tenant and could result in communication or registration failures.

The setup of the CloudLink Daemon is carried out through the following software applications:

- Administration program of the OpenScape SBC: OpenScape SBC Assistant.
- Administration program of the OpenScape Voice: OpenScape Common Management Platform.

For CloudLink troubleshooting information, refer to Chapter 7 of the [OpenScape SBC Troubleshooting Guide, Service Documentation](#).

8.1 Integrating Common Management Platform with CloudLink

As an administrator, you can configure CloudLink to integrate Common Management Platform (CMP) with CloudLink tenants. This integration enables the management of Zoom users associated with those tenants, ensuring a secure connection through a proxy server.

Prerequisites

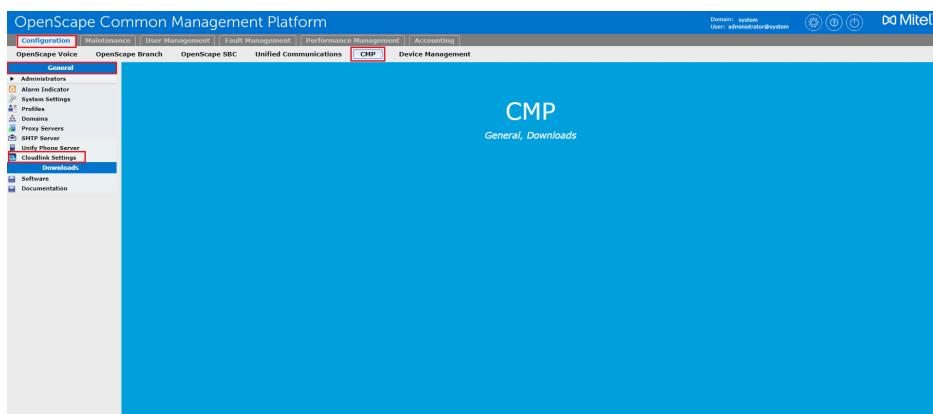
- 1) Adequate administrative permissions.
- 2) Install the `cld-oscmp` CloudLink Daemon RPM package (UC ISOs of V10 R6 FR6). For more information, refer to the "Searching for RPMs" section, included in the **Using osc-setup for handling Repositories** chapter of the [OpenScape UC Application V10, Installation and Upgrade, Installation Guide](#).
- 3) CloudLink requires an internet connection. Set up a proxy server to gain Internet access for CloudLink only if necessary. For instructions on adding a proxy, refer to 'How to create a proxy server' in the [Common Management Platform V10 Administration Guide](#).
- 4) Acquire account for Mitel Administration portal.
- 5) Provision an NTP server for CMP.
- 6) The computer used by the administrator to access the CMP and SBC user interfaces should have internet access.

To set up a connection with CloudLink in the Common Management Platform:

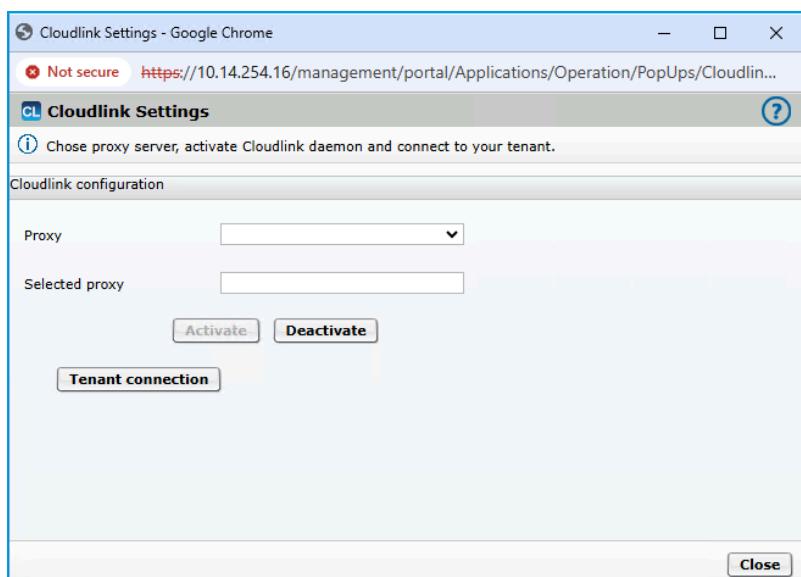
Step by Step

- 1) Log in to the OpenScape Common Management Platform.
- 2) To install the CloudLink Daemon package, execute the following command in cmd: osc-setup in cld-oscmp
- 3) Optional: To check the CloudLink Daemon status, execute the following command in cmd: systemctl status cld
By default, CloudLink Daemon is deactivated.
- 4) To activate CloudLink Daemon, on the **Configuration** navigation tab, click on the **CMP** navigation menu item.
- 5) In the navigation tree, under the **General** area, click **CloudLink Settings**.

NOTICE: The **CloudLink settings** option is visible only if the CloudLink Daemon package installation is successfully completed.



A **CloudLink Settings** window pops up.



6) From the **Proxy** drop-down menu, select a proxy that is configured for CloudLink.

The **Selected Proxy** field (read-only) displays the values that are already stored in CloudLink Daemon.

IMPORTANT: CloudLink requires an internet connection. Set up a proxy for CloudLink only if necessary. For instructions on adding a proxy, refer to 'How to create a proxy server' in the [Common Management Platform V10 Administration Guide](#).

7) Click **Activate** to enable CloudLink Daemon. To deactivate it, click **Deactivate**.

IMPORTANT: If you change the proxy selection after clicking the **Activate** button, you will need to click on it again.

8) Click **Tenant Connection** to set up a connection with a tenant.

NOTICE: The **Tenant Connection** button becomes active when the CloudLink daemon is enabled. Please note that it might take a few seconds for the button to activate after enabling it.

Upon clicking, you will be directed to the CloudLink Daemon window.

NOTICE:

If your CloudLink account is not linked, you will be prompted to log in with your admin credentials to establish the connection.

If the account is already linked, details of the connected account will be displayed.

9) To activate the CMP-CloudLink connection, click **Link to CloudLink**.

10) Enter your credentials in the CloudLink sign-in pop-up window and click **Next**.

The CloudLink Daemon window refreshes to display the CloudLink Tenant Information.

NOTICE: If the process is not complete, then please verify that you have configured a valid NTP (Network Time Protocol) server to CMP server. In case of time offset between CMP and CloudLink the creation of the token will fail.

11) Click **Close**.

8.2 Enabling OpenScape Voice and CloudLink to access the Internet via a proxy

This section describes how to enable OpenScape Voice - CloudLink internet connectivity. System can access the internet via Proxy Server or directly. In case of direct internet connectivity, extra security measures, i.e. DMZ/firewall, are recommended.

IMPORTANT: Before starting this procedure, it is recommended that you back up or save the current `/etc/sysconfig/proxy-osv` to an external server for safekeeping. This ensures the original file can be restored if the OSV behavior following the update necessitates it. Please review this entire procedure before proceeding with the actual updates.

The proxy settings are managed in the `/etc/sysconfig/proxy-osv` file. This file adheres to the same syntax as the standard `/etc/sysconfig/proxy` file, commonly used in Linux environments. The following variables are supported:

HTTP_PROXY: Specifies the proxy server for HTTP connections.

Examples:

- `HTTP_PROXY="http://<proxy-server>:<port>"`
- `HTTP_PROXY="http://123.123.123.123:8080"`

HTTPS_PROXY: Specifies the proxy server for HTTPS connections.

Examples:

- `HTTPS_PROXY="http://<proxy-server>:<port>"`
- `HTTPS_PROXY="http://123.123.123.123:8080"`

NO_PROXY: Defines a comma-separated list of hosts, IPs, or domains to bypass the proxy.

In `NO_PROXY`, the IP of CMP should also be added, using port 8289.

Examples:

- `NO_PROXY="localhost, 127.0.0.1, example.com"`
- `NO_PROXY="localhost, 127.0.0.1"`
- `NO_PROXY="localhost, 127.0.0.1, 11.12.13.14:8289"`

After the modification, duplex OSV nodes must be configured from state 4 to state 3 and then back to state 4 to ensure the caches are refreshed with the new data. This procedure must be performed on both nodes of the duplex OSV.

NOTICE: On OSV upgrades, the `proxy-osv` file will be migrated. Thus, once configured, the proxy configuration will not be affected by any upgrade or reconfiguration actions performed in the system. For detailed information on OSV installation and upgrades, refer to OpenScape Voice Service Manual: Installation and Upgrades.

8.3 Integrating OpenScape Voice with CloudLink

As an administrator, you can configure OSV connection to CloudLink Daemon via the Common Management Platform.

Prerequisites

Adequate administrative permissions.

CMP is connected to CloudLink.

You have enabled [OSV-CloudLink to access the Internet via a proxy](#).

Both CMP and OSV clocks must be synchronized. It is recommended to configure both to use an NTP server for clock synchronization.

To connect OpenScape OSV to CloudLink:

Step by Step

- 1) Log in to the OpenScape Common Management Platform.
- 2) Navigate to the **Configuration > OpenScape Voice** tab and click on the **General** icon.



Name	Switch Type	CHP - OpenScape Voice Compatibility	IP Node1	IP Node2	Version	Connectivity	CloudLink Connectivity
domino4	Simplex	CHP fully compatible with OSV	10.14.254.4	V12 R2.30.0	Reachable	Domained	
94029c	Duplex	CHP fully compatible with OSV	10.9.119.214	V12 R3.33.0	Reachable	Connected & activated	

- 3) Select **Switches**.

The **Switches** dialog opens listing the switches accessible on the system.

- 4) Optional: Select a switch from the list of available switches. If no switch is selected, all listed switches will be included in the OSV-to-CloudLink connection.

NOTICE: The process may take several minutes if one of the switches does not have the Cloudlink Daemon installed.

- 5) Click **Connect to OSV/Cloudlink**.

CMP connects to Cloudlink Daemon on OSV nodes.

The **CloudLink connectivity** column displays the status of each system.

For more information, please refer to the **Status of the OSV system** table below.

- 6) To enable CloudLink features for Zoom on a switch, select the connected switch of your choice and click **Edit**.

The **Edit switch** window pops up.

7) Locate the **CloudLink Configuration** area and click **Enable container**.

NOTICE: The **Enable container** button is active only when the OSV status is **Connected** or **Connected and deactivated**, but NOT **Connected and activated**.

Please refer to the **Status of the OSV system** table below.

8) Click **Save** to save the settings.

The switch status is **Connected and activated**.

Table 2: Status of the OSV system

Status	Description
Disconnected	The OSV system is not connected to CloudLink.
Connected and deactivated	The CloudLink is successfully connected. CloudLink features for Zoom are <u>not</u> enabled.
Connected and activated	<p>The CloudLink is successfully connected and CloudLink features for Zoom are enabled.</p> <hr/> <p>NOTICE: To enable CloudLink features for Zoom, see step 7.</p> <hr/> <p>.</p>
Connected and activated 	<p>One of the OSV nodes is not connected to CloudLink, or there has been an OSV failover.</p> <hr/> <p>NOTICE: To fix this manually, click the Enable container button.</p> <hr/> <p>.</p>

Upon a successful connection, you can open the CloudLink Daemon window and configure the allocated tunnels.

8.4 Integrating OpenScape SBC with CloudLink

OpenScape SBC hosts its own instance of CloudLink Daemon which needs to be enabled and connected to CloudLink. For OpenScape SBC, CloudLink also serves as the platform for transmitting mobile push notification requests to the Zoom Service.

Prerequisites

Adequate administrative permissions.

SBC has internet connectivity.

Acquire credentials for a valid Cloudlink tenant.

To connect OpenScape SBC to CloudLink:

Step by Step

- 1) Log in to the OpenScape SBC Management Portal.
- 2) Navigate to the **System > Settings** tab.

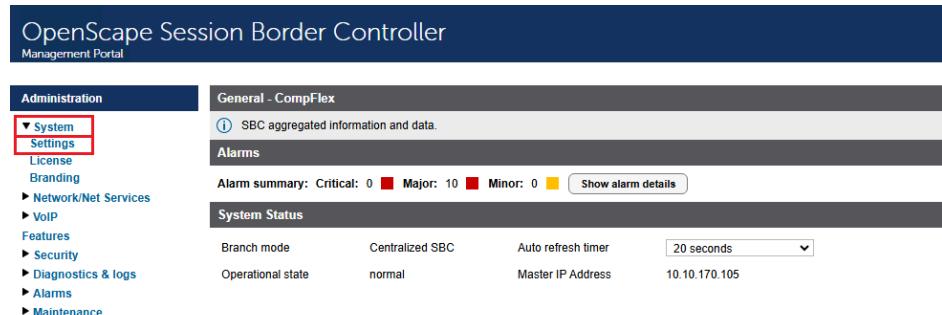
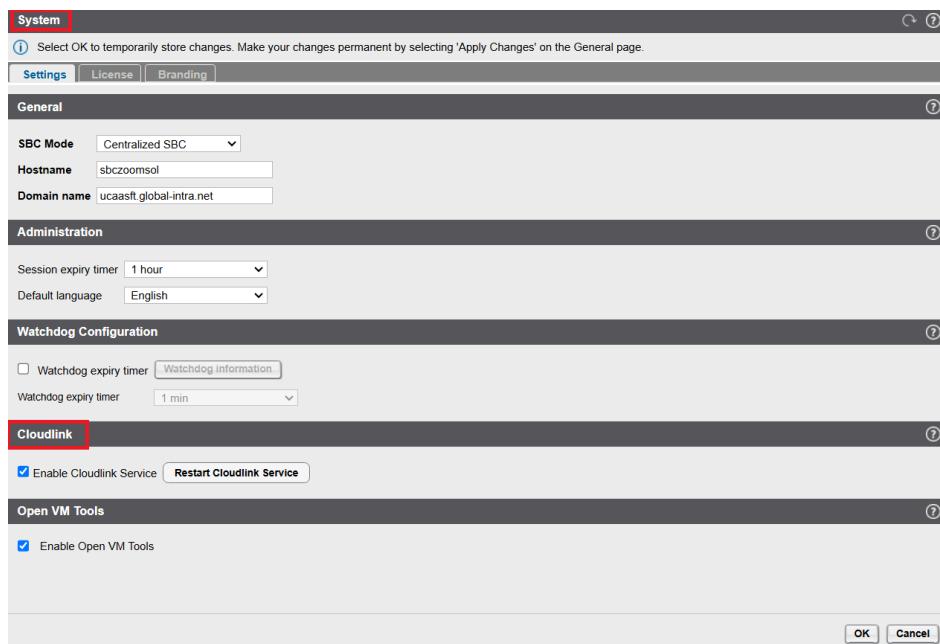


Figure 2:

The **System** window pops up.

- 3) Locate the **CloudLink** section.
- 4) To enable the SBC-CloudLink connection, check the **Enable CloudLink Service** checkbox.

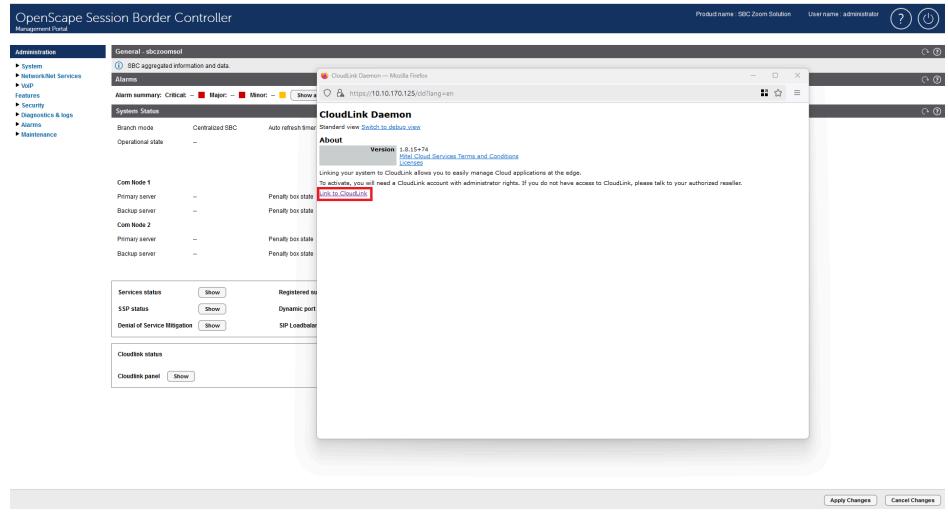


- 5) Click **OK**.

You are redirected to the main page.

- 6) To display the SBC-CloudLink connection information, click **Show** next to the **CloudLink** panel item.

The CloudLink Daemon window pops up. By default, CloudLink is disconnected.

7) Click **Link to CloudLink** to link CloudLink Daemon to CloudLink.8) Enter your admin credentials in the CloudLink sign-in pop-up window and click **Next**.

After the CloudLink Daemon is enabled and linked to the CloudLink platform, the CloudLink status on the SBC Dashboard updates, and a green indicator shows that CloudLink Daemon is running and connected.

The CloudLink Daemon dashboard displays the following details:

- About
- CloudLink Registration
- Inventory Report Submission
- CloudLink Daemon Update
- Tunnels

NOTICE:

Upon a successful connection, an inventory report is generated and refreshed approximately every 30 minutes.

For the details on each of these functionalities, see the [CloudLink Daemon Solution Guide](#).

9) Optionally, to connect the CloudLink Daemon for each component or to enable the tunnel:

- a) In the standard view of the CloudLink Daemon dashboard, under **Tunnels**, select a component, e.g. SBC, to connect or link to, and then click **Start**.
- b) Click **Yes** to confirm.
- c) Log in to CloudLink as an administrator.

Configuring CloudLink integrations

Configuring the PBX system settings in Mitel administration

- 10) Optionally, you can view system information and launch a remote SBC configuration:

- a) Log in to Mitel Administration and navigate to **System Inventory > Platforms**.

The list of Platforms and Applications that are connected from the Tunnel in Server Manager are now populated under the Mitel Administration System Inventory page.

- b) To remotely access the CloudLink Daemon for SBC, click **Launch**.

NOTICE: The **Launch** button becomes active only after the tunnel has been successfully established.

If you want to disconnect from CloudLink, click **Disconnect from CloudLink** in the main CloudLink window.

8.5 Configuring the PBX system settings in Mitel administration

After completing PBX integration with Cloudlink, optionally you can configure the system setting of the PBX.

The PBX system settings configured in the CloudLink Account are not directly synced to the PBX. The settings have to be entered manually in the PBX.

NOTICE: The steps below are mandatory only if you are configuring Emergency calls.

Step by Step

- 1) Log in to Mitel Administration as an Account Admin.
- 2) On the left navigation menu click on the name of the PBX e.g. **OpenScape Voice** and select **System Settings**.
- 3) In the **Voicemail** area add a **Pilot Number** to dial for accessing the voicemail messages.

NOTICE: If using local voicemail, ensure you enter the pilot number of the Expressions voicemail system in the system settings.

4) In the **Feature Codes** area you can configure feature codes for the users. To add a feature code:

- Select a **Feature** from the drop-down menu. The following features are available:
 - Call Forward
 - Disable Call Forward
 - Do not Disturb
 - Disable Do not Disturb
 - Enable Call Forward
 - Enable Do not Disturb

NOTICE: The same access codes can be configured in the PBX.

- Enter the **Dialing access code**.
- To edit a feature code, select the feature code, click on the **Dialing access code** field and edit it as needed.
- To delete a feature code, select the feature code and click  next to the feature code.

5) In the **Emergency Numbers** area you can configure the fallback emergency numbers. To add an emergency number:

- Click **Dialables**.
- Start typing a number for emergency calls, e.g.: 911, 112.
- Press enter, space or add a , (comma) to add the number.
- To edit an emergency number, double click on the number and edit it.
- To delete an emergency number, click X next to the number.

You can add multiple emergency numbers.

6) In the **Dynamic Location provider** area, configure a dynamic location provider for the emergency calls.

Enter the following information:

- Name**
- Type**
- Primary Server**
- Secondary Server**
- Customer ID:** the unique HELD identifier assigned to your organization by the service provider.
- Secret:** the private key or token issued by the service provider to secure communication between Zoom client and the service. This acts as a password and should be treated with high confidentiality.
- Extra Headers:** additional HTTP headers required by the service provider for platform communication. These headers might include custom authentication schemes, API version, or specific configuration options required by the provider. Input must be added in JSON format.

NOTICE:

In some cases, you can retrieve emergency configuration information from your account with your emergency provider.

However, it is highly recommended to always verify the settings with your emergency provider.

7) Click **Save.**

All mandatory fields must be completed before clicking save.

If the Zoom client has an emergency configuration enabled, the user will receive the message **Emergency location detected** upon login.

You can check and troubleshoot the settings in the [Event History](#) page.

9 Provisioning Users

Integration between the CloudLink account and the Zoom tenant is established through a process that links the two accounts. On one side, there is the CloudLink account, and on the other, a corresponding Zoom tenant. These are interconnected via a configuration process performed through CloudLink. This chapter outlines the necessary steps for preparing and setting up OpenScape Voice subscribers, as well as provisioning users, to ensure seamless integration with Zoom.

NOTICE: Ensure OSV resource synchronization has been successfully completed.

For detailed information on resource synchronization, see section **OpenScape Voice Signaling Protocol Data Synchronization** in the [Mitel OpenScape UC Application V11 Configuration and Administration](#) documentation.

9.1 User provisioning in the Zoom tenant

This chapter describes how to add a new Zoom user, set up a new Zoom account, and configure the Zoom-Mitel Phone System Integration.

For more detailed information on managing Zoom users, including deactivating, unlinking, or deleting users from your account, as well as performing actions such as batch importing and user auto-activation, refer to the links below.

- [Zoom-Mitel Phone System Integration support page](#)
- [Managing users](#)
- [Deactivating, unlinking, or deleting users from your account](#)
- [Batch importing, exporting, or updating users on your Zoom account](#)
- [Auto activating added users](#)
- [User Management API's](#)

Zoom single sign-on configuration allows your Zoom users to log in to Zoom using their company credentials.

To configure Zoom single sign-on (SSO), refer to the links below:

- [Quick start guide for single sign-on \(SSO\)](#)
- [SSO with Active Directory](#)
- [Settings and Configuration for SSO](#)

9.1.1 Adding a new Zoom user

An account owner or admin can add users to their account in several ways. This section describes how to add a single new Zoom User, or multiple users by entering their email addresses.

Prerequisites

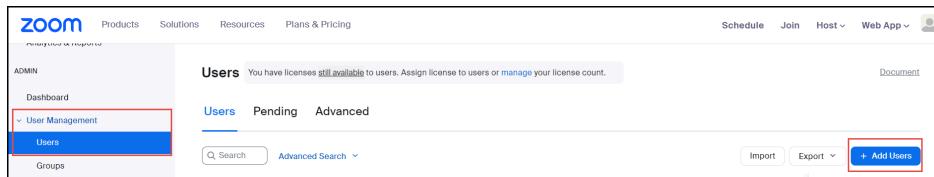
You have a Zoom account, Business or Enterprise.

You are an Account Owner or Admin with a Role for managing Users, Phone System Integration, and Zoom Phone.

You have completed Zoom-CloudLink integration, as described in [Configuring CloudLink integrations](#) on page 36.

Step by Step

- 1) Log in to the Zoom web portal.
- 2) Navigate to **User Management > Users > Add Users**.



- 3) In the **Add Users** pop-up window, enter the user's email address.

To add multiple users with the same settings, enter multiple email addresses separated by commas: , .

NOTICE:

Email address is the unique cross-platform identifier for provisioning Zoom and CloudLink users.

- 4) From the **Zoom Workplace** drop-down menu, select the available Zoom Workplace licenses to assign, such as **Zoom Meetings**.
- 5) Click **Add**.

The new user(s) will appear on the **Pending** tab of the **User Management** section.

New Zoom users will receive an activation email.

If a user already exists in Zoom, you will be prompted to accept the transfer of their account and be assigned to the new Zoom account owner.

Next steps

Activate the user(s) account.

Assign licenses to users. Before assigning a license to a phone user, ensure that automatic phone assignment for Zoom One licenses is disabled for your account. For more information, refer to the [Assigning Zoom licenses](#) page.

9.1.2 Setting up the Zoom account from invitation

Prerequisites

You have received an email invitation from no-reply@zoom.us to set up your Zoom account.

NOTICE: Remember to check your junk or spam folder if you can't find the invitation email in your inbox.

Step by Step

- 1) Open the email and click **Activate your Zoom Account**.
- 2) On the **Activate Your Account** screen, enter the following details:
 - a) First Name
 - b) Last Name
 - c) Password
- 3) Click **Continue**.

The Zoom user account is activated. In the Zoom Web Portal, the new user(s) will now appear under the **Users** tab of the **User Management** section.

To recover a disabled, inactive or locked account, refer to the official [Zoom support](#) page.

9.1.3 Configuring Zoom Phone System Integration

9.1.3.1 Configuring Phone System Integration settings

As an administrator, you can set up users for the Zoom-OSV integration.

Prerequisites

You have a Zoom account, Business or Enterprise.

You are an Account Owner or Admin with a Role for managing Users, Phone System Integration, and Zoom Phone.

You have added Zoom users and assigned licenses to them.

Step by Step

- 1) Log in to the Zoom Admin Portal.
- 2) Navigate to **Account Management > Phone System Integration**.
- 3) Go to the **Settings** tab.
- 4) By default, in the **Integrated calling on Zoom mobile** area, the **Allow use the integrated phone system to phone call on Zoom mobile client** toggle is enabled. If it is not enabled, click to turn it on.

NOTICE:

Ensure that this setting is always enabled. For more information, refer to [Configuring the Zoom-Mitel PSI integration](#).

Provisioning Users

Enabling an OpenScape Voice Subscriber for a Zoom connection

9.1.3.2 Adding Zoom users to the Mitel integration

As an administrator, you can set up users for the Zoom-OSV integration.

Prerequisites

You have a Zoom account, Business or Enterprise.

You are an Account Owner or Admin with a Role for managing Users, Phone System Integration, and Zoom Phone.

You have added Zoom users and assigned licenses to them. Zoom user accounts are activated.

Step by Step

1) Log in to the Zoom Admin Portal.

2) Navigate to **Account Management > Phone System Integration**.

The **Integrated users** tab is displayed.

3) Click **Add users**.

The **Add users** window pops up.

4) Select the user(s) you want to activate.

NOTICE:

You can add a maximum of 50 users at a time.

Ensure that the email address of the user(s) you add matches the email address that was used while creating the Zoom user and the assigned license.

5) Click **Add**.

The new user(s) will be added under the **Integrated Users** tab with the status **Pending SIP credential**.

This status will be updated once the OSV subscriber-Zoom user integration is completed.

To add non-Zoom users to Zoom directory, refer to the [Creating a shared directory of external contacts](#) page and [OpenScape Solution V11, Zoom with OpenScape Voice and OpenScape SBC BYOC \(Bring Your Own Carrier\)](#).

To import users with a CSV file, refer to [Zoom-Mitel Phone System Integration support page](#).

9.2 Enabling an OpenScape Voice Subscriber for a Zoom connection

As a CMP administrator, you must manually configure the OSV subscriber(s) to enable their Zoom connection.

Prerequisites

Adequate administrative permissions.

CMP is connected to CloudLink Daemon, as described in [Integrating Common Management Platform with CloudLink](#) on page 36.

OSV is connected to CloudLink, as described in [Connecting OSV to CloudLink Daemon](#).

You have provisioned users in the Zoom tenant.

To enable an OSV subscriber:

Step by Step

- 1) Log in to the OpenScape Common Management Platform.
- 2) Navigate to **Configuration > OpenScape Voice > Business Group > Members > Subscribers**.

Display Name	Feature Profile	Keyset	Numbering Plan
+302101849102	None	NP_mainAppset05	
+302101849103	None	NP_mainAppset06	
+302101849107	Primary Line	NP_mainAppset07	
+302101849110	None	NP_mainAppset08	
+302101849111	Primary Line	NP_mainAppset09	
+302101849112	Primary Line	NP_mainAppset09	
+302101849114	Primary Line	NP_mainAppset09	
+302101849115	None	NP_mainAppset09	
+302101849116	None	NP_mainAppset09	
+302101849117	None	NP_mainAppset09	

- 3) Click on the OSV subscriber you want to enable. The **Edit subscriber** window pops up.
- 4) Go to the **Routing** tab.
- 5) Scroll down to locate the **Zoom user** checkbox and click on it to enable it.

NOTICE: This attribute controls whether the specific user can accept a CallLogSnapshot. If enabled, OSV will respond with a CallLogEvent. Otherwise, it will return an InternalServiceErrorCode.

If both parties in a call have this attribute enabled, the full number (E164 format) is used instead of the dialable format, such as an extension.

- 6) Go to the **Connection** tab and check the **Registration via Central SBC Allowed** checkbox.
- 7) Go to the **Security** tab and enter the **SIP Authentication** credentials:
 - a) Realm
 - b) User Name
 - c) Password
 - d) Confirm Password
- 8) Go to **Features > Subscriber Features** and click the **Click to select Features** drop-down menu to select and add features to an OpenScape Voice Subscriber.
- 9) Scroll down to locate the **General Features** area and select the **Do Not Disturb** feature.

10) Locate the **Call Forwarding** area and select the following features:

- **Call Forwarding No Reply**
- **Call Forwarding on Busy**
- **Call Forwarding Unconditional**
- **Call Forwarding to Voicemail**

NOTICE: To enable the user to receive traditional Message Waiting Indication (MWI) and/or Visual Voicemail (VVM), click on the **Call forwarding to voicemail** feature. In the window **Call Forwarding to Voicemail** that opens, activate the feature and click **OK** to confirm.

NOTICE:

The Call Forwarding interface in Zoom PSI adapts dynamically to show only the forwarding types assigned to each subscriber.

For example, if an OSV subscriber does not have the Call Forwarding Internal/External feature assigned, Zoom PSI will display only the standard call forwarding options.

The interface can also be managed by CSTA applications through CloudLink gateway.

11) Click **Add**.

The features are automatically added under the **Subscriber Features** list with the status **Assigned**.

NOTICE: Features can also be added from the **Configuration > OpenScape Voice > Business Group > Profiles > Feature** area, by selecting the desired profile to which you want to assign features, as described in *How to Modify a Feature Profile* in the [OpenScape Voice V10 Administrator Documentation](#).

12) In the **Click to select Features** drop-down, go to the **CSTA features** area.

13) Select the **CSTA Access** feature and click **Add**.

14) Click on the name of the **CSTA Access** feature.

The **CSTA Access** window pops up.

15) From the **CSTA Type** drop-down menu, select **CSTA Over SIP** and click **OK**.

16) In the **Click to select Features** drop-down, go to the **Call Forwarding features** area.

17) Select the **Call Forwarding Internal/External** feature and click **Add**.

18) Click on the name of the **Call Forwarding Internal/External** feature.

The **Call Forwarding Internal/External** window pops up.

By default, the **General** tab is displayed.

a) Optionally, enable any of the following flags depending on customer requirements:

- Activate
- Allow Endpoint Control Internal
- Allow Endpoint Control External

The UI in Zoom PSI displays only the

b) Go to the **Advanced > Forward on Busy** tab and enable the following flags:

- Allow Endpoint Control Internal
- Allow Endpoint Control External

c) Go to the **Forward on No Answer** tab and enable the following flags:

- Allow Endpoint Control Internal
- Allow Endpoint Control External

NOTICE:

The Call Forwarding interface in Zoom PSI adapts dynamically to display only the call forwarding types of Call Forwarding Internal/External that are configured with the "Allow Endpoint Control – Internal/External" option enabled.

To hide specific call forwarding types based on customer requirements, administrators can disable the "Allow Endpoint Control – Internal/External" option for those particular forwarding types.

d) Click **OK**.

19) Navigate to **Administration > General Settings > RTP**.

20) Ensure that the value of the `Srx/Main/UseFullDnForZoom` RTP parameter, is set to `RtpTrue`.

21) Set the value of `hiQ/CCL/TrackingCategory` RTP parameter to `Non-ONS`.

22) Optionally, to allow OSV to play its announcement to external callers when the called Zoom user has **Do Not Disturb (DND)** enabled, set the `Srx/Main/DNDAnnAlways` parameter to `RtpTrue`.

23) Click **Save** to save the settings.

You are redirected back to the main **OpenScape Voice** page.

24) Repeat the steps for all subscribers that will be integrated with Zoom users.

9.3 Creating a new SBC configuration for Zoom users

You can create a new Zoom Configuration in OpenScape Common Management Platform.

Prerequisites

Adequate administrative permissions.

CloudLink Daemon configurations are completed.

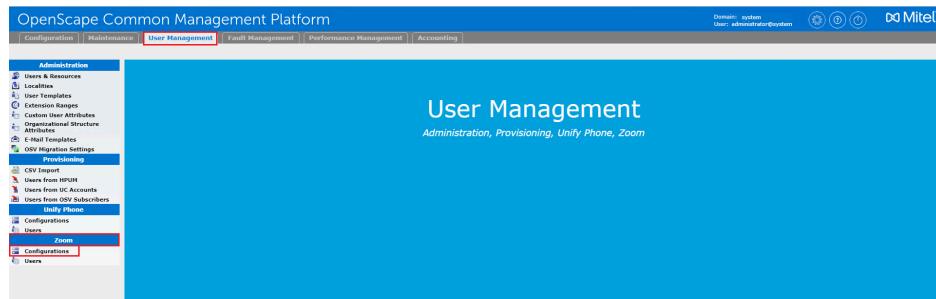
OSV subscribers are enabled for Zoom connection. For more information, refer to [Enabling an OpenScape Voice Subscriber for a Zoom connection](#) on page 50.

Zoom users have been configured in the Zoom tenant, as described in [User provisioning in the Zoom tenant](#) on page 47.

To create a new Zoom configuration:

Step by Step

- 1) Log in to the OpenScape Common Management Platform.
- 2) Click on the **User Management** navigation tab.
- 3) In the navigation tree, click on **Zoom > Configurations**.



A list of the existing SBC configurations is displayed.

- 4) Click **Add** to create a new Zoom configuration.

A new **Add configuration** window appears containing the following fields:

- **Name**
Name of the SBC configuration.
- **IP/FQDN address**
IP or FQDN of the SBC access interface that has a public address, or the external firewall of the SBC if it is located in the DMZ.
- **Port**
Specify the SIP over TLS port, which is used to the SBC access interface, which has internet access.
- **Protocol**
Specify the used protocol (**TLS**, **TCP** or **UDP**).

NOTICE: The PSI client is recommended to be configured to use SIP/TLS. UDP and TCP are not recommended for remote subscribers.

- 5) Click **Save** to save the settings.

The newly created SBC configuration is displayed in the list.

9.4 Integrating OSV subscribers with Zoom users

After creating the Zoom configuration, you can configure mass user provisioning by creating a User Template with Zoom and a Zoom Resource. For details, refer to [OpenScape Common Management Platform V10 Administrator Documentation](#).

Alternatively, you can provision Zoom users individually in OpenScape Voice through the OpenScape Common Management Platform by following the instructions below.

Prerequisites

Adequate administrative permissions.

CloudLink configurations are completed.

Zoom users have been configured in the Zoom tenant, as described in [User provisioning in the Zoom tenant](#) on page 47.

Zoom configurations are completed, as described in [Creating a new SBC configuration for Zoom users](#) on page 53.

General prerequisites regarding subscriber/system configuration from previous chapters are met.

To integrate OSV subscribers with Zoom users:

Step by Step

- 1) Log in to the OpenScape Common Management Platform.
- 2) Click on the **User Management** navigation tab.
- 3) In the navigation tree, click on **Zoom > Users**.

E-mail	First Name	Last Name	Phone Number
admin-user-571052824@mitel-test.com	adminUser		
clsdsupport-rn@mitel-test.com	demoFirst	demoLast	302100801015
cmp-demo1@mitel-test.com	demoFirst2	demoLast2	302100801016
cmp-demo2@mitel-test.com	cloudlinkUserFirst	cloudlinkUserLast	302100801001
cmp-test1@mitel-test.com	test1_2		12345678902
cmp-test2@mitel-test.com	cmp4_First	cmp4_Last	302100801003
cmp-test3@mitel-test.com	cmp4_test5	mitel	302100801004

- 4) Click **Add...** to create a new Zoom User.

A new **Add Zoom User** window pops up containing the following fields:

- **Email:** Enter a valid email address.

IMPORTANT:

The email address is the unique identifier that associates a user with a single Zoom tenant and must be unique.

Ensure that the email address matches the email address that was used while creating the Zoom user and the assigned license.

Once configured, you cannot edit the **Email** of a user.

- **First Name:** Enter the first name of the user.
- **Last Name:** Enter the last name of the user.
- **Configuration:** Select the SBC configuration you want from the dropdown list, as created in [Creating a new SBC configuration for Zoom users](#) on page 53.
- **Phone number:** Click the three-dot icon to add a pre-configured phone number.

The **Select OpenScape Voice resource** window pops up, containing a list of the available phone numbers.

Check the box in front of the phone number you want to select and click **OK**.

IMPORTANT:

When selecting a phone number, all OSV subscribers are visible, not just those with the Zoom user flag enabled.

However, when searching, if the Zoom user flag is not enabled, the subscriber will not be shown.

The administrator must be aware of which OSV subscribers are configured for the Zoom connection. For easier lookup, it is recommended to use search operators such as `*your_subscriber*` to locate your subscriber.

- **VoiceMail:** Callback number of XPR Voicemail server.

5) Click **Save** to save the settings.

NOTICE: If the selected OSV subscriber is not a Zoom User, an error message, The action has failed, is displayed, indicating that the connection has failed.

The OSV subscriber-Zoom user integration is completed:

- In CMP, the new user is added in the **Zoom Users** list.
- In the Mitel Admin **User Management** menu area, the new users is added under the **Users** list.

NOTICE:

The administration of users should not be performed from Mitel Admin. The provisioning of users should be performed from the PBX's management platform.

- In the Zoom **System Integration** menu area, under the **Integrated Users** list, the status of the newly created Zoom user is modified from **Pending SIP credentials to Idle**.

Each user has two instances in total: one for the desktop application and one for the mobile application.

To edit or delete a user, refer to [OpenScape Common Management Platform V10 Administrator Documentation](#).

IMPORTANT:

Editing the first name and/or last name of the user and deleting a Zoom PSI user is not reflected in the Zoom tenant.

Configuring OpenScape Xpressions Visual Voicemail

Setting up Xpressions to access the Internet via a proxy (Optional)

10 Configuring OpenScape Xpressions Visual Voicemail

You can set up the OpenScape Xpressions Visual Voicemail (VVM) system and connect it with CloudLink.

The integration with CloudLink provides a more seamless experience by allowing users to view, manage, and listen to their Visual Voicemail messages in an inbox-like list through the Zoom app interface (desktop or mobile), instead of calling into a voicemail system.

Key features include:

- List view of messages with caller information, timestamps, and message length.
- Playback controls: play, pause, delete, and download messages.
- Callback functionality directly from the message.
- Transcription of voice messages into text.

Mobile clients can also receive push notifications related to Visual Voicemail, ensuring they are promptly alerted when new messages arrive.

Visual Voicemail requires an active Internet connection. You can provide Internet access to the Xpressions server in one of the following ways:

- **Direct access:** Use a dedicated network card to connect the Xpressions server to the Internet.
- **Proxy access:** Configure Xpressions to connect to the Internet through a proxy server (see [Setting up Xpressions to access the Internet via a proxy \(Optional\)](#) on page 58).

10.1 Setting up Xpressions to access the Internet via a proxy (Optional)

Internet access is mandatory for Visual Voicemail to function properly. Optionally, you can set up Xpressions to access the Internet via a proxy.

You can enable proxy server settings on an Xpressions server in one of the following ways:

- Set global variables for the proxy settings to provide internet access to the entire server.
- Provide Internet access only for the Xpression product, not for the entire server.

Both options require administrator privileges on the Xpressions server.

10.1.1 Configuring a global variable for Mitel Cloudlink Daemon

If required, configure a global environment variable that forces the Mitel CloudLink Daemon to use a proxy server.

Prerequisites

- 1) You have administrator privileges in Xpression server.
- 2) The date and time in Xpression server are correct.
- 3) The operating system is **OS Windows Microsoft server 2019 Standard or 2022 Standard**.

- 4) The hotfixes from the latest Zoom Package are installed (CIAPL, CITranscriptAPL and MTA).
- 5) You have CloudLink tenant credentials.

Step by Step

- 1) Search for **Edit the system environment variables** in the Windows Start menu and open it.
- 2) In the **System Properties** window, click **Advanced system settings**.
- 3) In the **Advanced** tab, click **Environment Variables**.

The **Environment Variables** window opens.

- 4) Locate the **System variables** area and click **New** to create the **HTTP_PROXY** variable:

- **Variable name:** **HTTP_PROXY**
- **Variable value:** **192.168.1.1:8080**

NOTICE: Use your actual proxy server address.

- 5) Click **OK** to save the variable.
- 6) Click **New** again to create the **HTTPS_PROXY** variable:

- **Variable name:** **HTTPS_PROXY**
- **Variable value:** **192.168.1.1:8080**

NOTICE: Use your actual proxy server address.

- 7) Click **OK** to save the variable.
- 8) Ensure both variables appear in the **System variables** list.
- 9) Restart the Mitel CloudLink Daemon Service for the changes to take effect.

10.1.2 Configuring proxy settings in Xpression monitor

If required, configure proxy settings in Xpressions monitor.

Prerequisites

- 1) You have administrator privileges in Xpression server.
- 2) The date and time in Xpression server are correct.
- 3) The operating system is **OS Windows Microsoft server 2019 Standard or 2022 Standard**.
- 4) The hotfixes from the latest Zoom Package are installed (CIAPL, CITranscriptAPL and MTA).
- 5) You have CloudLink tenant credentials.

Step by Step

- 1) Open the **Xpression Monitor - System Logging** application (XPR Monitor).
- 2) Log in using administrator credentials.
- 3) In the **Modules of XPR** screen, expand the **Preferences** tree.
- 4) Click **CIAPL** to expand it.

Configuring OpenScape Xpressions Visual Voicemail

Configuring Xpressions Registry for OSV

- 5) Double-click **Edit Settings**.

The **CloudLink APL configuration** window opens.

- 6) In the **Proxy Server** field, enter the IP address and port of the proxy server in the following format:

xxx.xxx.xxx.xxx:port

For example, 10.10.10.10:5050.

- 7) Click **OK**.

- 8) Save the changes.

- 9) Restart the **XPR CloudLink APL (CIApI)** service.

- 10) Restart the **Mitel Cloudlink Daemon** service.

10.2 Configuring Xpressions Registry for OSV

OSV requires an E.164 number without a leading plus sign (+). To ensure compatibility, a registry setting must be configured in Xpressions.

Prerequisites

- 1) You have administrator privileges in Xpression server.
- 2) The date and time in Xpression server are correct.
- 3) The Xpression server has internet access.
- 4) The operating system is **OS Windows Microsoft server 2019 Standard or 2022 Standard**.
- 5) The hotfixes from the latest Zoom Package are installed (CIApI, CITranscriptApI and MTA).
- 6) You have CloudLink tenant credentials.

Procedure

- Open the Windows Registry Editor.
- Navigate to the following key:
HKLM\Software\Wow6432Node\PP-COM\Mrs\CIApI
- Create a new **REG_DWORD**:
 - Name: **UsingOSV**
 - Value: **1**
- Close the Registry Editor.
- Restart **CIApI** for the change to take effect.

This setting ensures that phone numbers used by OSV are formatted correctly for Xpressions, avoiding issues with the leading plus sign in E.164 numbers.

10.3 Connecting OpenScape Xpressions with CloudLink

Prerequisites

- 1) You have administrator privileges in Xpression server.
- 2) The date and time in Xpression server are correct.
- 3) The Xpression server has internet access.

- 4) The operating system is **OS Windows Microsoft server 2019 Standard or 2022 Standard**.
- 5) The hotfixes from the latest Zoom Package are installed (CIApI, CITranscriptApI and MTA).
- 6) You have CloudLink tenant credentials.

Step by Step

- 1) Open a browser and navigate to: <http://localhost:8288/cld>.
- 2) Click **Connect to CloudLink**.
- 3) Log in using your CloudLink tenant credentials.
After successful login, you will be redirected to the **Mitel Administration** portal.
- 4) To manage the Xpression server via the CloudLink portal, start the tunnels.
- 5) Go to **Mitel Administration > System Inventory > Platforms** to verify that the Xpression server is listed in the CloudLink portal.

10.4 Activating an Xpressions user

You can activate an Xpressions user. When activated, they can receive voicemails, listen to them, and get alerts.

Prerequisites

- 1) You have administrator privileges in Xpression server.
- 2) The date and time in Xpression server are correct.
- 3) The Xpression server has internet access.
- 4) The operating system is **OS Windows Microsoft server 2019 Standard or 2022 Standard**.
- 5) The hotfixes from the latest Zoom Package are installed (CIApI, CITranscriptApI and MTA).
- 6) You have CloudLink tenant credentials.

Procedure

- For each user whose primary SMTP address matches, if the CLOUDLINK property is set to X, the user will have the new features enabled.

Next steps

Once the CLOUDLINK property is enabled, the user will have access to the new features.

10.5 Configuring Zoom integration and transcription

After a user is activated, the following steps are required to allow voicemail access and transcription in Zoom.

Prerequisites

The user must be activated in Xpressions (see [Activating an Xpressions user](#) on page 61) and have transcription privileges.

The user's primary email in Xpressions must match their Zoom account email.

Procedure

- The **CLOUDLINK** property must remain enabled in **Webassistant > User administration**.
- **Voicemail transcript** privilege must be permitted in **Webassistant > User administration**.

NOTICE: You can permit the **Voicemail transcript** privilege only for an individual user. If it is permitted for the group, it will affect all users that belongs to that group.

Transcription occurs only when a new message is sent; existing messages cannot be transcribed.

The transcription language is based on the recipient's language configured in Xpressions. The supported languages are:

- en-US
- de-DE
- en-GB
- fr-FR
- it-IT
- nl-NL
- es-ES
- pt-PT

For more information, see also [OpenScape Solution V11, Mitel Phone System Integration \(PSI\) with Zoom, User Guide](#).

11 Monitoring and Troubleshooting Zoom Integration in Mitel CloudLink Admin Portal

11.1 Viewing the Zoom integration status

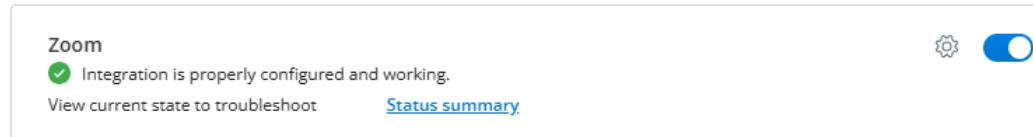
Once the Zoom integration is added to a customer account, you can check its status to ensure it is set up properly. The Zoom integration can have one of the following statuses:

- Connected
- Error
- Pending

Viewing a summary of the Zoom integration status

To view a summary of the Zoom integration status, follow the steps below:

- 1) Log in to Mitel Administration as an Account Admin.
- 2) Access the **Integrations** panel from the **Accounts Information** page or from the **Integrations & Apps** option.
- 3) In the **Integrations** panel, locate the **Zoom** integration. Check the status icon and message next to it.



The icon indicates the current status of the integration, while the status message provides additional information about the overall status.

Viewing detailed information about the Zoom integration status

For a more in-depth view of the Zoom integration status, especially for troubleshooting, you can one of the following:

- Click **Status summary** next to the **Zoom** integration in the **Integrations** panel.
- Navigate to **Support > Zoom**.

You can then view detailed information about the Zoom integration status, including the following:

- **OAuth status:** Displays the OAuth authorization status (Authorized, Failed), indicating whether the Zoom OAuth token is valid, expired, or needs re-authorization. If the OAuth status is Failed, error messages associated with the most recent OAuth failure will also be displayed below the status.
- **Integration status:** Indicates the current status of the Zoom integration (Connected, Error, or Pending).
- **Sync status:** Indicates the synchronization status between CloudLink and Zoom. If the last sync was unsuccessful, error messages associated with the most recent failed sync attempt will also be displayed below the status.
- **Last successful sync:** Date and time of the last successful synchronization between CloudLink and Zoom.

The following image shows an example of detailed information about the Zoom integration status when the integration is set up properly.

Zoom Integration Status

The current state of the integration with the Zoom platform.

OAuth status:	✓ Authorized
Integration status:	✓ Connected
Sync status:	✓ Successful
Last successful sync:	1/31/2025, 6:02:55 PM

Done

The following image shows an example of detailed information about the Zoom integration status when the integration is not set up properly.

Zoom Integration Status

The current state of the integration with the Zoom platform.

OAuth status:	⚠ Failed
Details:	Zoom(GET /v2/users): ZoomAuthService.refreshAuthCredentials => Error refreshing Zoom PSI user: failed to post auth resource: InvalidRequest
Integration status:	✓ Connected
Sync status:	⚠ Failed
Details:	Zoom(GET /v2/users): ZoomAuthService.refreshAuthCredentials => Error refreshing Zoom PSI user: failed to post auth resource: InvalidRequest
Last successful sync:	2025-01-27, 1:46:31 p.m.

Done

In the second example, as shown in the details section below the failed **OAuth status** and **Sync status**, an error occurred while attempting to obtain a new refresh token from Zoom.

Refreshing the Zoom integration status

To refresh the Zoom integration status, follow the steps below:

- 1) Navigate to **Support > Zoom**.
- 2) In the **Status** tab, click **Refresh**.

11.2 Generating a User Comparison Report

The User Comparison Report analyzes user data across multiple systems to identify inconsistencies. It consolidates user information from four sources, using the email address as the unique identifier:

- CloudLink User Database (CL User DB)
- Service Delivery License Database
- Zoom User List
- Zoom Phone List

The User Comparison Report helps identify mismatches and missing data that may impact the proper provisioning of services.

You can generate and download a report comparing users' information between Zoom and CloudLink.

Step by Step

- 1) Log in to Mitel Administration as an Account Admin.
- 2) Click **Support > Zoom** from the left main menu.
The **Zoom Sync & Provisioning Errors** page of the customer account opens.
- 3) Select the **User Comparison Report** tab.
- 4) Click **Generate** to compare users' information between Zoom and CloudLink.
The system initiates an asynchronous request for generating the report.
A report is generated in a csv format.
- 5) Click **Download** next to the csv file.

The User Comparison Report contains the following information:

Field	Description
email	The primary identifier.
name	User's display name.
clUserId	The user's ID in CloudLink (if found).
licenses	Assigned licenses (e.g., ["ZoomPSI"]).
zmUserId	The user's ID in Zoom (if found).
zmUserStatus	The current status of the user in Zoom (active, inactive, pending).
zmSipPhoneId	The ID of the user's assigned Zoom desktop client SIP phone (if found).
zmSipPhoneNumber	The assigned Zoom desktop client SIP phone number.
zmSipPhoneMobileId	The ID of the user's assigned Zoom mobile SIP phone (if found).

Monitoring and Troubleshooting Zoom Integration in Mitel CloudLink Admin Portal

Troubleshooting common issues identified in the User Comparison Report

Field	Description
zmSipPhoneMobileNumber	The assigned Zoom mobile phone number.
issues	A list of identified inconsistencies.

11.3 Troubleshooting common issues identified in the User Comparison Report

If any issue is identified in the User Comparison Report, it is recorded in the issue column of the User Comparison Report.

Below are the potential issues and the recommended resolution:

Issue	Cause	Resolution
CloudLinkUserNotFound	The user is not found in the CloudLink User Database.	Ensure the user is provisioned in CloudLink. Verify that their email address is correct.
ZoomUserNotFound	The user does not exist in Zoom.	Confirm that the user has been added to the Zoom tenant. Verify the email address that is used.
ZoomSipPhoneNotFound	The user does not have a Zoom SIP phone assigned.	Assign a SIP phone to the user in the Zoom Admin Portal.
ZoomUserStatusInactive	The user's Zoom status is inactive.	Reactivate the user in the Zoom Admin Portal.
ZoomUserStatusPending	The user's Zoom status is pending activation.	Ensure the user completes the activation process by following the Zoom invite email.
NoCIZoomPsiLicense	The user does not have the required "ZoomPSI" license in CloudLink.	Assign the "ZoomPSI" license to the user in the management Portal. If this issue is detected, no further checks are performed.

Steps to Validate and Fix Issues

- 1) Open the User Comparison Report.
- 2) Locate users with issues in the issues column.
- 3) Identify the corresponding inconsistency from the list above.
- 4) Follow the resolution steps for each detected issue.
- 5) After making corrections, regenerate the report to verify the fixes.

If the issues persist after resolving them, contact the appropriate system administrator for further investigation.

NOTICE: If a user does not have a "ZoomPSI" license, no further checks are performed.

NOTICE: Email addresses must match exactly across all sources for proper data joining.

11.4 Viewing the Event History table (Zoom integration)

The Event History provides insight to Mitel Partners and Account Admins regarding events that occurred within an account with Zoom integration.

Step by Step

- 1) Log in to Mitel Administration as an Account Admin.
- 2) Click **Support > Zoom** from the left main menu.
The **Zoom Sync & Provisioning Errors** page of the customer account opens.
- 3) Select the **Event History** tab.
- 4) Click on an event in the Event History table to view the event details.
The **Event Details** popup window is displayed.
- 5) Click **Copy** to copy the event details of the following tabs:
 - **Core details**
 - **Properties Changed**
 - **Extra Details**
 - **Log tags**
- 6) Click **Export** to export all data in a csv format.

NOTICE:

Actions performed in Mitel Administration (MA) will only appear in the **Event History** after a 24-hour delay. This delay is expected and does not indicate a failure or issue with the action itself.

12 Configuring E911 Calls

This chapter provides information on the necessary configurations to ensure that the E911 solution can successfully determine the physical location of a registered user during an emergency call. Once the exact location is identified, the E911 solution routes the E911 call to the appropriate Public Safety Answering Point (PSAP) and notifies security personnel.

E911 Solutions must comply with E911 legislation. The Federal Communications Commission (FCC) developed [Kari's Law and the RAY BAUM's Act](#), which comprise a set of rules and regulations that specify direct dialing, notification, and dispatchable location minimum requirements for all Multi-line Telephone System (MLTS) platforms. All organizations across the US must comply with both Kari's Law and the RAY BAUM's Act.

OSV, as a Multi-line Telephone System (MLTS), implements Section 506 of RAY BAUM Act and Kari's Law support in conjunction with third-party Next Generation of 911 emergency services providers in the USA.

For OSV, we have the following device categories:

- Fixed MLTS Devices. For example, Analog DevicesTDM devices (Analog Devices, Digital Devices, and Integrated DECT).
- Non-Fixed MLTS devices. For example, IP Devices, SIP Devices, softphones, all teleworkers, and so on.

To fully support the requirements above, OSV is integrated with [Intrado](#) in USA and with [Redsky](#) in USA and Canada. A valid service agreement with either RedSky or Intrado is necessary for the E911 Solution.

NOTICE: Mitel does not provide this service agreement directly. To support local notifications compliant with Kari's law compliant, the solution will use the E911 Provider's notification application.

RedSky and Intrado use SIP trunks to route E911 calls to the appropriate Public Safety Answering Points (PSAPs) based on the civic address. Both providers pass callback information from the call-server to enable the PSTN to route the call back from the PSAP to the specified callback number.

NOTICE: Intrado also offers a function called Extension bind for non-DID numbers. This function, when enabled, assigns a temporary valid Direct Inward Dialing (DID) callback number for the extension number (non 10-digits number) that made the 911 call. In this case, if the call gets disconnected the Emergency Response Team can call back the person that called the Emergency Service.

Emergency Call Flow

Emergency calls are **only supported** from the **Zoom desktop client**. If you attempt to place an emergency call from the **Zoom mobile client**, the call will automatically be redirected to the mobile cellular network.

Additionally, the emergency location is provided by RedSky. The process for retrieving the emergency location is as follows:

- When a user logs into the Zoom desktop client, Zoom sends a request to CloudLink.
- CloudLink, using the Emergency Provider tenant information, forwards the request to Emergency Provider to retrieve the user's emergency location.

For OpenScape Voice Emergency Calling information, refer to chapter 4.9 *Emergency Calling* of the [OpenScape Voice V10 Administrator Documentation](#).

For CloudLink Emergency configurations, refer to [Configuring the PBX system settings in Mitel administration](#) on page 44 (steps 5-6).

To complete the Mitel OpenScape Voice and Mitel OpenScape SBC configurations required for an Emergency Solution, follow the instructions provided in the following chapters.

12.1 OpenScape SBC E911 Configuration

This section describes how to configure Mitel OpenScape SBC for emergency calls used by Zoom users.

Prerequisites

- 1) You must request an account from your Emergency provider (Redsky or Intrado).
- 2) You must have a **CloudLink account** with **CloudLink account admin** privileges.
- 3) The IP address from which the SBC will send traffic must be added to the provider's whitelist. Please contact your emergency provider.
- 4) Proper firewall rules must be created to allow traffic for signaling and RTP ports configured for emergency calls.
- 5) You must enable the NG911 service in Mitel OpenScape Voice. To do this, refer to chapters 4.9.5 *How to Route Emergency Calls to NG911 Service Providers* and 4.9 *Emergency Calling* in the [OpenScape Voice V10 Administrator Documentation](#).

12.1.1 Configuring an E911 Media Profile

Follow the steps below to create a new media profile for your Emergency Provider (Intrado or Redsky).

Step by Step

- 1) Log in to the SBC management portal.
- 2) Navigate to **VoIP > Media** in the navigation tree under Administration.
- 3) Under **Media Profiles**, click **Add**.

The **Media Profiles** window pops up.

- 4) Under the **General** section:
 - a) Enter an E911 Media Profile name. For example, RedSky.
 - b) From the **Media protocol** drop-down menu, select **RTP only**.

NOTICE:

The **Media Protocol** is specified by your Emergency Provider. To ensure compliance with their requirements, please contact your Emergency Provider's support.

- 5) If codec configuration is required by your Emergency Provider, do the following:

NOTICE: In some cases, codec configuration from an Emergency Provider (such as Redsky) is necessary to align technical specifications and ensure that emergency calls can be handled efficiently within the organization's communication infrastructure.

- a) Locate the **Codec Configuration** area.
- b) Check the **Allow unconfigured codecs** checkbox.
- c) From the **Codec** drop-down menu, select the codec as specified by your Emergency Provider, according to the region where they are located. For example, select G711U 8kHz - 64 kbps (for US-NA) or G711A 8kHz - 64 kbps (for Europe).

NOTICE: Redsky accepts only US & Canada locations. In addition, it prefers G711U codec.

- 6) Click **OK** to save the configuration.
- 7) Click **Apply changes**.

12.1.2 Configuring Remote Endpoints

To configure Emergency remote endpoints, you must first create SIP Service Provider Profiles (SSPs) and then proceed with configuring the remote endpoints.

12.1.2.1 E911 SIP Service Provider Profile Configuration

The following configuration must be applied to the E911 Remote Endpoint Profile to handle Zoom > E911 calls.

Step by Step

- 1) Log in to the SBC management portal.
- 2) Navigate to **Features** in the navigation tree under Administration.

The **Features** window pops up.

- 3) Check the **Enable Remote Endpoints** checkbox and click **Configure**.

The **Remote endpoints** window pops up.

- 4) Under the **SIP Service Provider Profile** area, click **Add**

NOTICE: The **SIP Service Provider Profiles** window pops up.

- 5) In the **Name** field, enter the name of your E911 Provider. For example, RedSky.
- 6) Check the **Enable SSP Privacy and Complementary Flags** checkbox.
- 7) Click **OK** to save the configuration.
- 8) Click **Apply Changes** in the main window to confirm the changes to the OpenScape SBC appliance.

12.1.2.2 E911 Remote Endpoint Configuration

Follow the steps below to configure an E911 Provider remote endpoint.

Prerequisites

You have created a E911 SIP Service Provider Profile.

Step by Step

- 1) Log in to the SBC management portal.
- 2) Navigate to **Features** in the navigation tree under Administration.

The **Features** window pops up.

- 3) Check the **Enable Remote Endpoints** checkbox and click **Configure** next to it.

The **Remote endpoints** window pops up.

- 4) Scroll down to locate the **Remote endpoint configuration** area and click **Add**.

The **Remote Endpoint configuration** window pops up.

- 5) Under the **Remote Endpoint Settings** area:
 - a) In the **Name** field, enter a unique name for the E911 Provider remote endpoint. For example, RedSky.
 - b) From the **Type** drop-down menu, select **SSP**.
 - c) From the **Profile** drop-down menu, select the E911 SIP Service Provider Profile you created in E911 SIP Service Provider Profile configuration.
 - d) From the **Access realm profile** drop-down menu, select the network ID that has access to Internet. For example, **Main-access-Realm-ipv**.

IMPORTANT:

For security purposes, IP whitelisting is used by E911 Providers to block network access to all IPs except those in the whitelist. To ensure the public Firewall IP

Configuring E911 Calls

OpenScape Voice E911 Configuration

you are using will be whitelisted, share it with your E911 Provider.

- e) From the **Core realm profile** drop-down menu, select the core realm profile. For example, **Main-core-realm-ipv4**.
- 6) Under the **Remote Location domain list** area, click **Add**

The **Remote Location Domain** window pops up.

NOTICE: The settings presented below are provided by your E911 Provider.

- a) In the **Remote URL** field, enter the URL of the remote endpoint for E911.
- b) In the **Remote port** field, enter the remote port for communication between E911 and OSSBC.
- c) From the **Remote transport** drop-down menu, select the remote transport protocol provided by your E911 Provider (TCP, UDP, or TLS).
- 7) Under the **Media Configuration** area, from the **Media Profile** drop-down menu, select the Media profile of your E911 Provider created in *Configuring an E911 Media Profile*.
- 8) Click **OK**.

You are directed back to the **Remote Endpoint Configuration** window.

- 9) Locate the **Remote Location Identification Routing** area.
- 10) In the **Core realm port**, enter a unique value.
- 11) Click **OK**.
- 12) Click **Apply changes**.

You are directed back to the **Remote Endpoints** window. The E911 Provider Remote endpoint is shown under the **Remote endpoint configuration** table.

12.2 OpenScape Voice E911 Configuration

This section describes how to configure Mitel OpenScape Voice for emergency calls used by Zoom users.

Prerequisites

- 1) You must request an account from your Emergency provider (Redsky or Intrado).
- 2) You must have a **CloudLink account** with **CloudLink account admin** privileges.
- 3) The IP address from which the SBC will send traffic must be added to the provider's whitelist. Please contact your emergency provider.
- 4) Proper firewall rules must be created to allow traffic for signaling and RTP ports configured for emergency calls.
- 5) You must enable the NG911 service in Mitel OpenScape Voice. To do this, refer to chapters *4.9.5 How to Route Emergency Calls to NG911 Service Providers* and *4.9 Emergency Calling* in the [OpenScape Voice V10 Administrator Documentation](#).

12.2.1 Configuring an E911 Provider Remote Endpoint

To create a new Endpoint Profile for your Emergency Provider:

Step by Step

- 1) Navigate to **Unify OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Business Group List**.
- 2) From the **Business Group List** drop-down menu, select your **Business Group**. For example, Zoom_BG.
- 3) In the selected Business Group, navigate to **Profiles > Endpoint** and click **Add**.
- 4) In the **Add Endpoint Profile** window, under the **General** tab, configure the following:
 - a) **Name**: Enter the name of the Endpoint Profile. For example, Redsky.
 - b) Select the **Profile** for Zoom PSI.
- 5) Click **Save**.
- 6) Select the **SIP** tab and configure the following:
 - a) Select the **SIP Trunking** option to enable it.
 - b) From the **Type** drop-down menu, select **Static** (it can be enabled only if the **SIP Proxy** attribute is enabled).
 - c) From the **Signaling Address Type** drop-down menu, **select IP Address or FQDN** (route the calls via proxy).
 - d) **Endpoint Address**: Enter the SBC address.
 - e) **Port**: Enter the port number, as configured in OpenScape SBC [E911 Remote Endpoint Configuration](#) on page 71 (step 10).
 - f) From the **Transport protocol** drop-down menu, select **TCP**.
- 7) Click **Save**.
- 8) Go to the **Attributes** tab and select the following attributes to enable them:
 - a) Public/Offnet Traffic
 - b) Enable Session Timer
- 9) Go to the **Aliases** tab and click **Add** to add an Alias with the format **SBC IP : port**.
- 10) Click **OK** and then click **Save**.

12.2.2 Translation Configuration

With **Translation**, the administrator configures the routing of outgoing calls based on the dialed digits from OS Voice subscribers. A call can only be routed if the dialed digits match a PAC (Prefix Access Code).

The **Destination Code** feature provides destination codes for basic telephone service. A destination code will be applied to a call if the dialed or modified (via PAC) digits and the nature of the address match.

12.2.2.1 Configuring the E911 Call Routing

This section describes how to create a prefix to route the call to Emergency service.

- 1) Navigate to **OpenScape Common Management Platform > Configuration > OpenScape Voice > Business Group > Translation > Prefix Access Codes**.
- 2) Click **Add**.
- 3) In the **Add Prefix Access Code** pop-up, configure the following parameters:
 - a) **Prefix Access code**: Enter the prefix. For example, 933.
 - b) **Minimum length**: Enter the minimum expected length of the Emergency number.
 - c) **Maximum length**: Enter the maximum expected length of the Emergency number.
 - d) **Digit Position**: Configure as 0, which implies not removing any digits from the dialed number before sending it to the destination.
 - e) From the **Prefix Type** drop-down menu, select **Off-net Access** to permit access to remote destinations.
 - f) From the **Nature of Address** drop-down menu, select **Unknown**.
 - g) From the **Destination Type** drop-down menu, select **None**. The resulting digits will be processed in the user's numbering plans destination codes table.
 - h) Click **Save**.
- 4) To create a Destination code for 911:
Navigate to **OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Translation > Destination Codes**.
- 5) Click **Add**.
- 6) In the **Add Destination Code** pop-up, configure the following:
 - a) **Destination Code**: Select the previously created Prefix Access Code (PAC).
 - b) Click the **Select Traffic Type** button and then click the three-dot icon (only enabled if the option is selected) to open the traffic type selection window.
Select the **Emergency** traffic type.
Click **OK**.
 - c) From the **Destination Type** drop-down menu, select **Service**.
 - d) Click the three-dot icon next to **Service** and select **Emergency** from the Service List pop-up.
Click **OK**.
 - e) Click **Save**.

12.2.3 Configuring an Access code

As an administrator, you can create an Access code to route the emergency call to a Destination.

Step by Step

- 1) Navigate to **Unify OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Business Group List**.
- 2) From the **Business Group List** drop-down menu, select your **Business Group** to edit it. For example, **Zoom_PSI**.
The **Edit Business Group** window pops up. The General tab is displayed by default.
- 3) Go to the **Services** tab.
- 4) Under the **Next Generation 911** area, check the **Activate** option.
- 5) Enter an access code in the **Access Code** text field. For example, 333.
- 6) Click **Save**.

12.2.4 Destinations and Routes Configuration

Destinations are logical targets for off-net or on-net routing. When a destination is created, its name is bound to the numbering plan where it is made.

Destinations are used to route a call to an endpoint representing a gateway.

Each **Route** is a collection of groups or addresses providing a destination path.

12.2.4.1 Configuring the Emergency Destination

To add a Destination that will be used to route the emergency calls to SBC:

- 1) Navigate to **Unify OpenScape Common Management Platform > Configuration > Unify OpenScape Voice > Business Group > Destinations and Routes > Destinations**.
- 2) Click **Add**.
- 3) In the **Add Destination** pop-up, under the **General** tab, enter the name of the Zoom destination. For example, **DEST_NG911**.
- 4) Click **Save**.
- 5) Select the destination you created in the previous step and click **Edit**.
- 6) In the **Edit Destination** pop-up, select the **Routes** tab and click **Add**.
- 7) In the **Add Route** pop-up, configure the following:
 - a) **ID**: Enter the priority level of this route ID as 1 (if there are multiple routes to a destination and route prioritization is selected, the route with the lowest numbered route ID has the highest priority and will be selected first).
 - b) From the **Type** drop-down menu, select **SIP Endpoint**.

- c) Click the three-dot icon at the right side of the **SIP Endpoint** and select the Emergency Endpoint (for example, Redsky) created in [Configuring an E911 Provider Remote Endpoint](#) on page 73.

NOTICE: The emergency provider requires three digits, thus the route should remove all digits and send a three-digit number (e.g., 933).

- 8) Configure the Destination Directory Number settings as below:
 - a) From the **Modification Type** drop-down menu, select **Number Manipulation**.
 - b) In the **Number of digits** field, enter the number of digits to cut off from the directory number. For the Redsky configuration, enter **6**.
 - c) In the **Digits to insert** field, enter the digit string which gets added to the beginning of the remaining digits. For the Redsky configuration, enter, for example, **933**.
 - d) From the **Nature of Address** drop-down menu, select **Unknown**.
 - e) Click **Save**.

12.2.5 Configuring a new Destination Code

As an administrator, you can create a new Destination Code, based on the configurations described in [Configuring the Emergency Destination](#) on page 75.

Example

In case you have created a prefix for emergency calls (933) and an access code (333) in [Configuring the Emergency Destination](#) on page 75, you will need to create a new destination code: 933333.

The first 3 digits (933) will route the call to Emergency Service of **OSV**.

The next three digits (333) will route the call to the **SBC**.

Finally, you can manipulate the number (displayed when selecting to edit the pre-configured emergency destination under the **Routes** tab) so that only **933** is sent to the SBC, ensuring the correct number is sent to the SIP Endpoint created for the Emergency Endpoint, which is actually the SBC, since the emergency call will be routed to Emergency provider via SBC.

- 1) To create a new Destination code:

 Navigate to **OpenScape Common Management Platform** > **Configuration** > **Unify OpenScape Voice** > **Business Group** > **Translation** > **Destination Codes**.

- 2) Click **Add**.

- 3) In the **Add Destination Code** pop-up, configure the following:
 - a) **Destination Code**: Enter the destination code. For example, 933333.
 - b) Click the **Select Traffic Type** button and then click the three-dot icon (only enabled if the option is selected) to open the traffic type selection window.
Select the **Emergency** traffic type.
 - c) Click **OK**.
 - d) From the **Destination Type** drop-down menu, select **Destination**.
 - e) Click the three-dot icon next to **Destination** and select the destination configured in [Configuring the Emergency Destination](#) on page 75. For example, **DEST_NG911**.
 - f) Click **OK**.
 - g) Click **Save**.

12.3 CloudLink E911 Configuration

For CloudLink Emergency configurations, refer to [Configuring the PBX system settings in Mitel administration](#) on page 44 (steps 5-6).

12.4 Adding IP Range Mapping (Redsky)

Prerequisites

You have an administrator account from your Emergency provider (Redsky).

Step by Step

- 1) Log in to your Redsky account.
- 2) Select **Network Discovery** from the left-side Configuration menu.
The **Network Discovery** page is displayed.
- 3) Go to the **IP Ranges** tab.
- 4) Click **Add IP Range Mapping** at the top right of the **Network Discovery** page to add the IP Range mapping configurations.
- 5) Configure the following fields as required:
 - Range Start
 - Range End
 - Building
 - Location
 - Description

NOTICE:

If the Building or Location you want to select does not appear in the dropdown, you must add it as a new Building or Location entry.

6) Click **Save.**

For more information, refer to the [Redsky Online Help](#) page.

NOTICE:

The IP range created should be mapped with an E911 location created in the Redsky portal.

13 Appendix

13.1 Appendix A: Restrictions and Known Issues

The following table lists the tested features when Zoom is integrated with OpenScape Voice and OpenScape SBC.

Feature	Description
Do Not Disturb (DND)	<p>When the user logs out, DND is automatically deactivated from client side.</p> <p>Upon login, the Zoom PSI Client re-enables Do Not Disturb (DND) if it was active before logout.</p>
Local Conference	A local conference is limited to 3 members (including the initiator).
Local tones on Zoom PSI client	<p>The Zoom PSI client does not play a busy tone when the called party is busy. Instead, it only displays the message: <i>"The number you are calling is currently busy, please try again later."</i></p> <p>Limitation: This occurs when the PSI client receives a 486 Busy Here SIP message from OSV.</p>
Managing Zoom PSI Users via CMP	<p>Zoom PSI user deletion or name modification via CMP is not synchronized with the Zoom tenant.</p> <p>Limitation: When a Zoom PSI user is deleted from CMP, the user remains in the Zoom tenant and continues to consume a PSI license. Similarly, if an administrator modifies a user's first or last name in CMP, the changes are not propagated to the Zoom tenant.</p>
Session timers	<p>Zoom PSI client does not fully support SIP session refresh using the INVITE method (RFC 4028), which may lead to unexpected SDP renegotiation and audio issues. To address this limitation, OSV is configured to use the UPDATE method for session refresh by changing <code>Srx/Sip/UpdateMethodSessionTimingEnable</code> to true.</p> <p>Administrators should verify whether the SIP service provider or any other third-party endpoint supports the UPDATE method for session refresh.</p> <p>If the UPDATE method is not supported but is included in the Allow header, the endpoint should be configured with the attribute Do Not Support Update Method for Session Timing.</p>

Appendix

Appendix B: Default User Name and Password

Feature	Description
SIP error message (487)	The SIP error message (487) appears in the Zoom client when the user dials a service code to activate or deactivate a service. Despite the error code, the user's action is successfully applied.
Support for Storing Message Waiting Indicator (MWI) Message Counters from Xpressions Voicemail Server	OSV does not support storing the Message Waiting Indicator (MWI) message counters sent by the Xpressions voicemail server. Zoom PSI users cannot see the number of waiting voicemail messages upon login or registration. MWI notifications do not reflect the actual number of unread voicemail messages.
User status not shown as Busy/DND on Desk Phone calls after 24h inactivity on Zoom Mobile	Media presence and DND status are not updated if the user uses only the Zoom mobile client (iOS or Android) and the last mobile login was more than 24 hours ago. This means that if the user is busy with a call on Deskphone or DND is enabled, other users will still see this user as available on mobile.
Xpressions Voicemail	If the sender is not configured with an Xpressions account, and there is no corresponding Zoom user or Zoom contact for the sender, the receiver of the Visual Voicemail (VVM) will only see the sender's number instead of their name. This happens because Xpressions Voicemail is not currently able to store the name provided in the PAI header.

13.2 Appendix B: Default User Name and Password

The following table lists the default user name and password for the OpenScape SBC Management Portal.

User Name	Password
administrator	Asd123!

For information on OpenScape SBC Security Checklist, refer to [OpenScape SBC V11R2 Security Checklist](#).

