



## Avaya Solution & Interoperability Test Lab

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# Application Notes for Aiphone IX Series Audio Door Stations (IX-SPMIC) and Avaya IP Office™ – Issue 1.0

### Abstract

These Application Notes describe the procedures for configuring Aiphone IX Series Audio Door Stations (IX-SPMIC) which were compliance tested with Avaya IP Office™.

The overall objective of the interoperability compliance testing was to verify Aiphone IX Series Audio Door Stations (IX-SPMIC) functionalities in an environment comprised of Avaya IP Office™ and various Avaya endpoints. Aiphone IX Series Audio Door Stations are SIP based door phones.

Readers should pay attention to **Section** Error! Reference source not found., in particular the scope of testing as outlined in **Section** Error! Reference source not found. as well as any observations noted in **Section** Error! Reference source not found., to ensure that their own use cases are adequately covered by this scope and results.

Information in these Application Notes has been obtained through DevConnect compliance testing and additional technical discussions. Testing was conducted via the DevConnect Program at the Avaya Solution and Interoperability Test Lab.

## 1. Introduction

These Application Notes describe the configuration steps required for Aiphone IX Series Audio Door Stations (IX-SPMIC) to interoperate with Avaya IP Office (IP Office). During the compliance testing, Aiphone IX-SPMIC was used.

The Aiphone IX Series Audio Door Stations (IX-SPMIC) are part of Aiphone IX Series 2 Door Stations. The Audio Door Stations, IX-SPMIC, act as SIP phones when connected to IP Office. Stations come in both surface mount and flush mount varieties. All door stations have dry contacts that can be used to release doors when activated by another intercom or phone. The dry contacts can also be used to trigger external signaling devices, such as strobes.

Calls from Aiphone IX-SPMIC are originated and terminated via a URL. Aiphone IX-SPMIC cannot receive calls.

During the compliance test, Avaya IP Office Server Edition was used as a primary system and Avaya IP Office 500V2 as an expansion system. Aiphone IX-SPMIC registered as a 3<sup>rd</sup> party SIP phone using UDP to the Avaya IP Office Server Edition.

## 2. General Test Approach and Test Results

The focus of this interoperability compliance testing was to verify that the Aiphone IX-SPMIC can register as a SIP endpoint on IP Office, and is able to originate audio calls to the IP Office system.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly-defined test plan focuses on exercising APIs and/or standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute full product performance or feature testing performed by DevConnect members, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a DevConnect member's solution.

Avaya recommends our customers implement Avaya solutions using appropriate security and encryption capabilities enabled by our products. The testing referenced in these DevConnect Application Notes included the enablement of supported encryption capabilities in the Avaya products. Readers should consult the appropriate Avaya product documentation for further information regarding security and encryption capabilities supported by those Avaya products.

Support for these security and encryption capabilities in any non-Avaya solution component is the responsibility of each individual vendor. Readers should consult the appropriate vendor-supplied product documentation for more information regarding those products.

For the testing associated with these Application Notes, the interface between Avaya systems and Aiphone did not utilize secure capabilities.

## 2.1. Interoperability Compliance Testing

The general test approach was to place calls to and from, Aiphone IX-SPMIC, and exercise basic telephone operations. The main objectives were to verify the following:

- Registration
- Calls to Avaya SIP Audio endpoints
- Calls to Avaya H.323 Audio endpoints
- Calls to Avaya Digital & Analog endpoints
- Calls to PSTN via SIP Trunks
- Call termination (origination/destination)
- Serviceability

## 2.2. Test Results

The test objectives were verified, and the features tested worked as expected.

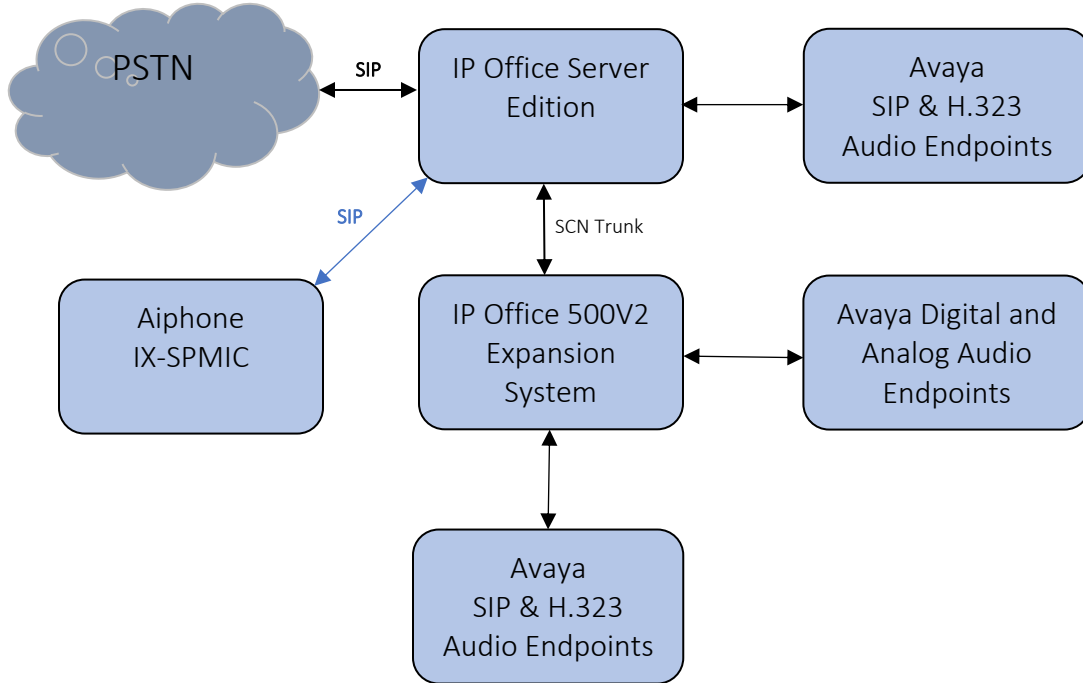
## 2.3. Support

For technical support on Aiphone IX-SPMIC, please contact Aiphone via the following:

- Web: <https://www.aiphone.co.jp/>
- Phone: 052-228-9961

### 3. Reference Configuration

**Figure 1** illustrates a sample configuration consisting of Avaya IP Office components and Aiphone IX-SPMIC.



**Figure 1: Test Configuration of Aiphone IX-SPMIC with Avaya IP Office**

## 4. Equipment and Software Validated

The following equipment and software were used for the test configuration.

Equipment	Software/Firmware
Avaya IP Office Server Edition	11.0.4.0.0 build 74
Avaya IP Office 500V2	11.0.4.0.0 build 74
Avaya IP Office Manager	11.0.4.0.0 build 74
Avaya 9600 Series H.323 IP Deskphones	6.8002
Avaya J129 SIP Phone	4.0.0.0.21
Avaya IX Workspace	3.7.0.102.3
Avaya H175 Collaboration Station	1.0.2.3
Avaya Vantage K175 Phone	3.5.0
Avaya 9504 Digital Phone	0.55
Avaya 6210 Analogue Telephone	-
Aiphone IX Series Audio Door Station IX-SPMIC	3.00.

**Note:** Compliance Testing is applicable when the tested solution is deployed with a standalone IP Office 500 V2 and also when deployed with IP Office Server Edition in all configurations.

## 5. Configure Avaya IP Office™

This section provides the procedures for configuring IP Office. The procedures include the following areas:

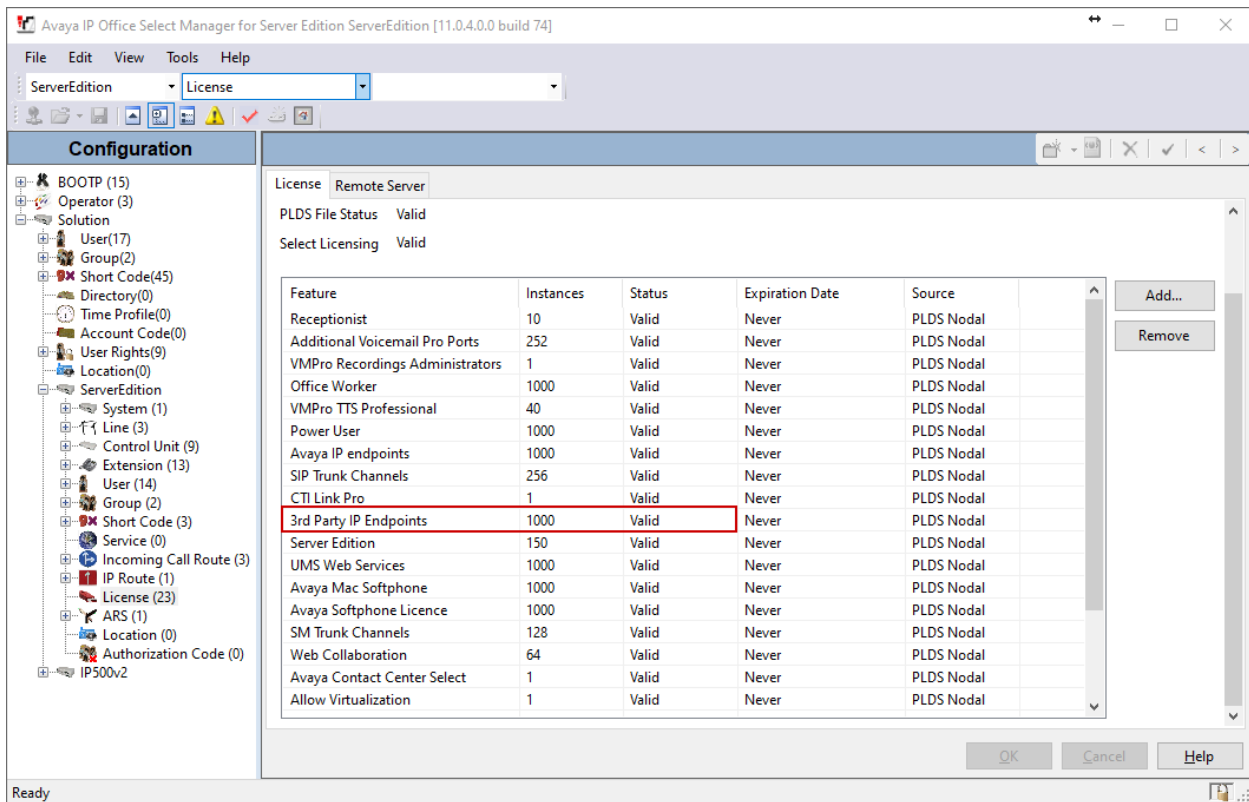
- Verify IP Office license
- Obtain LAN IP address
- Administer SIP registrar
- Administer SIP extensions
- Administer SIP users

These steps are performed from the Avaya IP Office Manager.

### 5.1. Verify IP Office License

From a PC running the Avaya IP Office Manager application, select **Start → All Programs → IP Office → Manager** to launch the Manager application. Select the proper IP Office system if there are more than one IP Office system, and log in with the appropriate credentials.

The Avaya IP Office Manager screen is displayed. From the configuration tree in the left pane, select **License → 3<sup>rd</sup> Party IP Endpoints** to display available licenses in the right pane. Verify that the License Status field is set to **Valid** for **3<sup>rd</sup> Party IP Endpoints** feature.

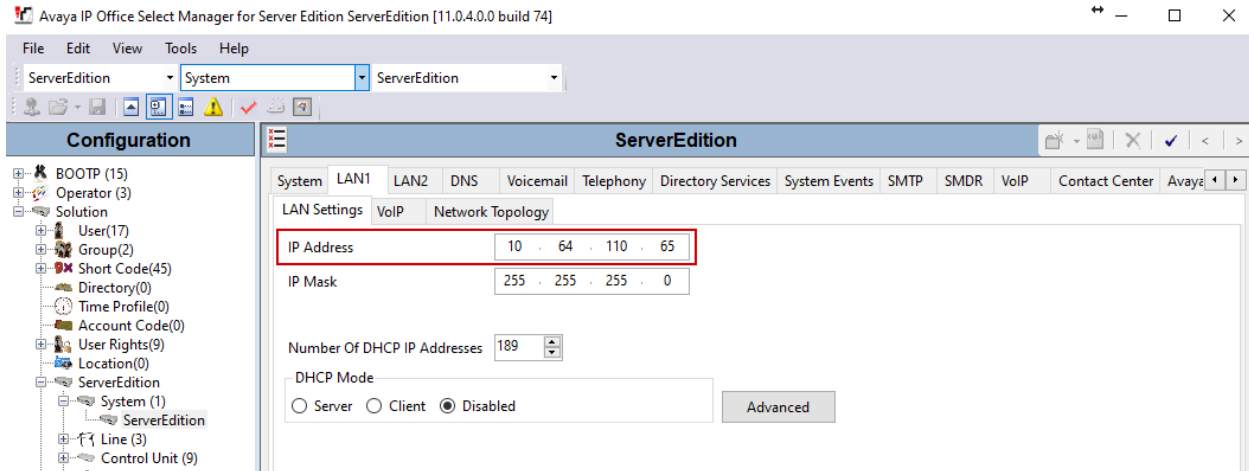


The screenshot shows the Avaya IP Office Select Manager for Server Edition application. The 'License' tab is selected, and the 'Remote Server' section is active. The 'PLDS File Status' is 'Valid' and 'Select Licensing' is 'Valid'. A table displays the following features and their status:

Feature	Instances	Status	Expiration Date	Source
Receptionist	10	Valid	Never	PLDS Nodal
Additional Voicemail Pro Ports	252	Valid	Never	PLDS Nodal
VMPro Recordings Administrators	1	Valid	Never	PLDS Nodal
Office Worker	1000	Valid	Never	PLDS Nodal
VMPro TTS Professional	40	Valid	Never	PLDS Nodal
Power User	1000	Valid	Never	PLDS Nodal
Avaya IP endpoints	1000	Valid	Never	PLDS Nodal
SIP Trunk Channels	256	Valid	Never	PLDS Nodal
CTI Link Pro	1	Valid	Never	PLDS Nodal
<b>3rd Party IP Endpoints</b>	<b>1000</b>	<b>Valid</b>	Never	PLDS Nodal
Server Edition	150	Valid	Never	PLDS Nodal
UMS Web Services	1000	Valid	Never	PLDS Nodal
Avaya Mac Softphone	1000	Valid	Never	PLDS Nodal
Avaya Softphone Licence	1000	Valid	Never	PLDS Nodal
SM Trunk Channels	128	Valid	Never	PLDS Nodal
Web Collaboration	64	Valid	Never	PLDS Nodal
Avaya Contact Center Select	1	Valid	Never	PLDS Nodal
Allow Virtualization	1	Valid	Never	PLDS Nodal

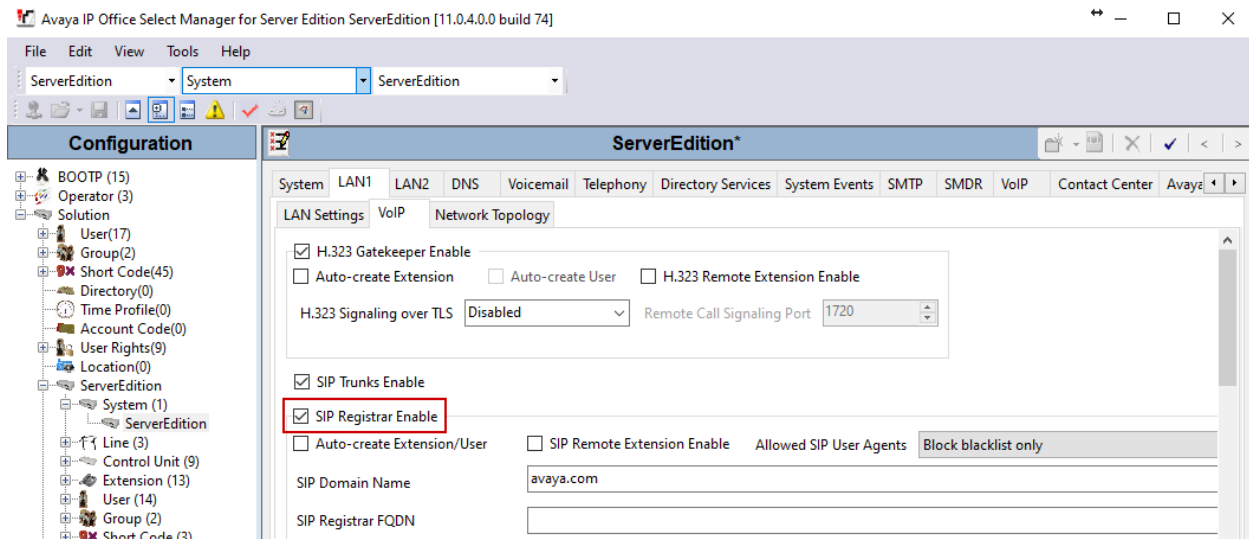
## 5.2. Obtain LAN IP Address

From the configuration tree in the left pane, select **System** to display the System screen in the right pane. Select the **LAN1** tab, followed by the **LAN Settings** sub-tab in the right pane. Make a note of the **IP Address**, which will be used later to configure Aiphone IX-SPMIC.



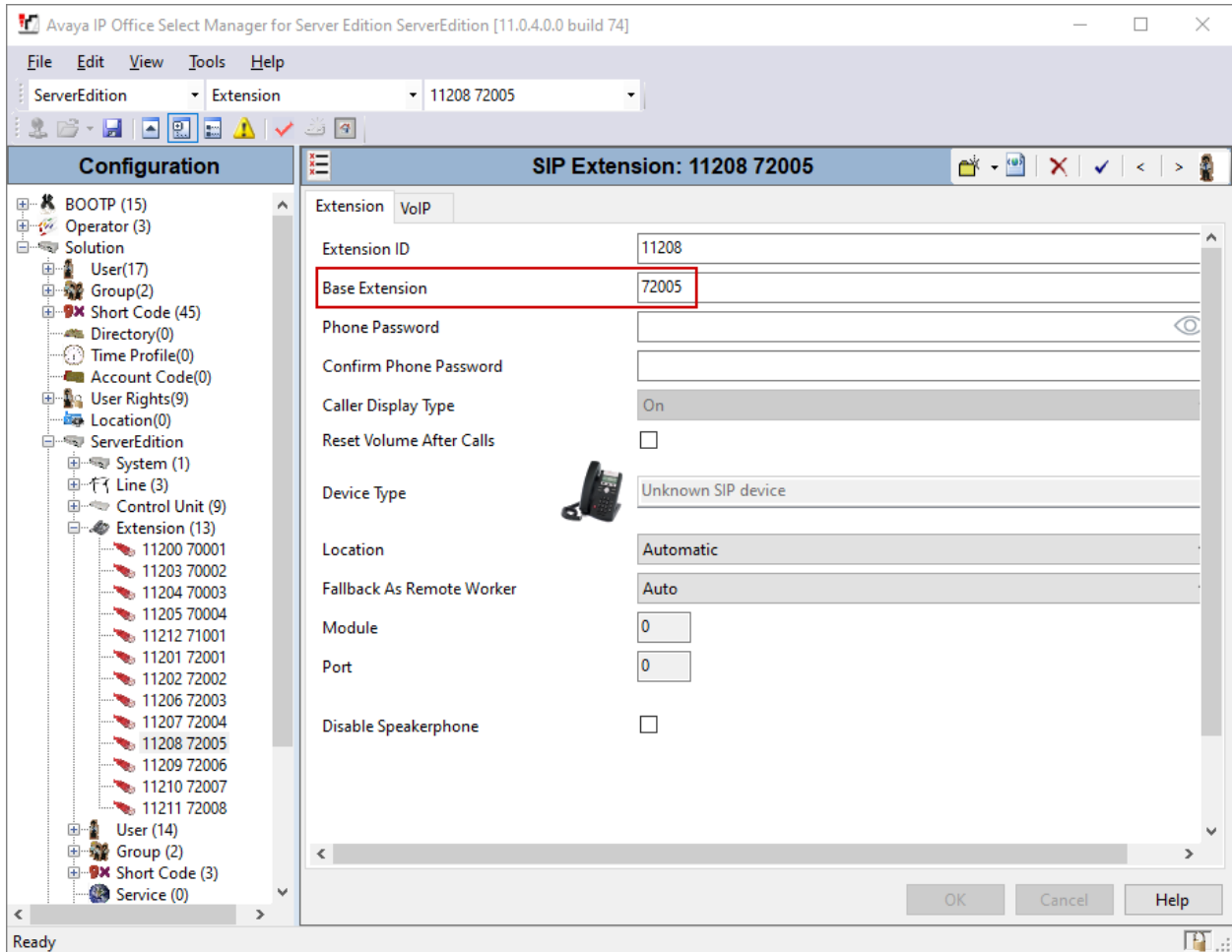
## 5.3. Administer SIP Registrar

Continuing from above, select the **VoIP** sub-tab. Ensure that **SIP Registrar Enable** is checked, as shown below.



## 5.4. Administer SIP Extensions

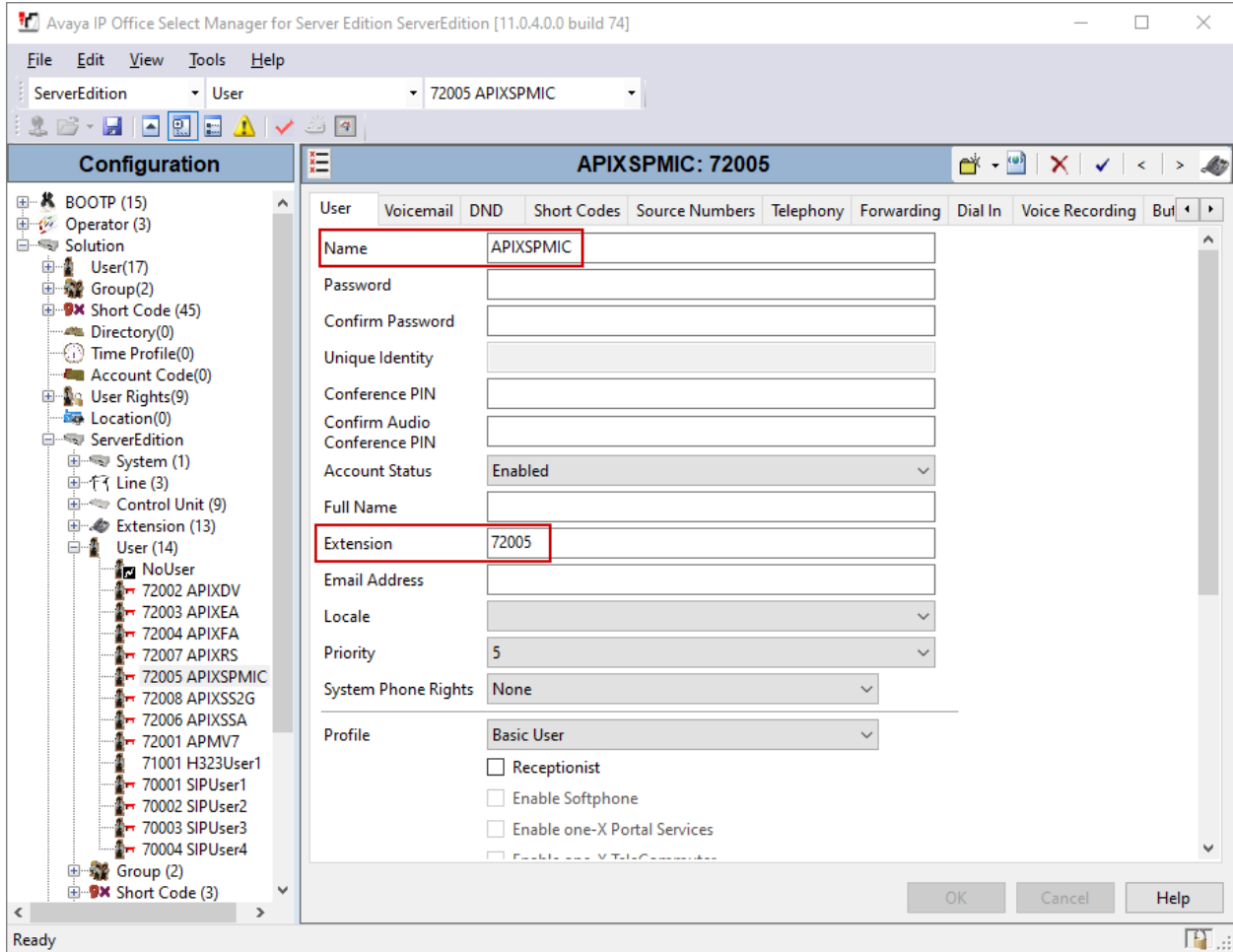
To create a new SIP Extension, from the configuration tree in the left pane, right-click on **Extension**, and select **New** → **SIP Extension** from the pop-up list (not shown). Enter desired digits for the **Base Extension** field. This is the Extension that will be used for Aiphone IX-SPMIC to log in.



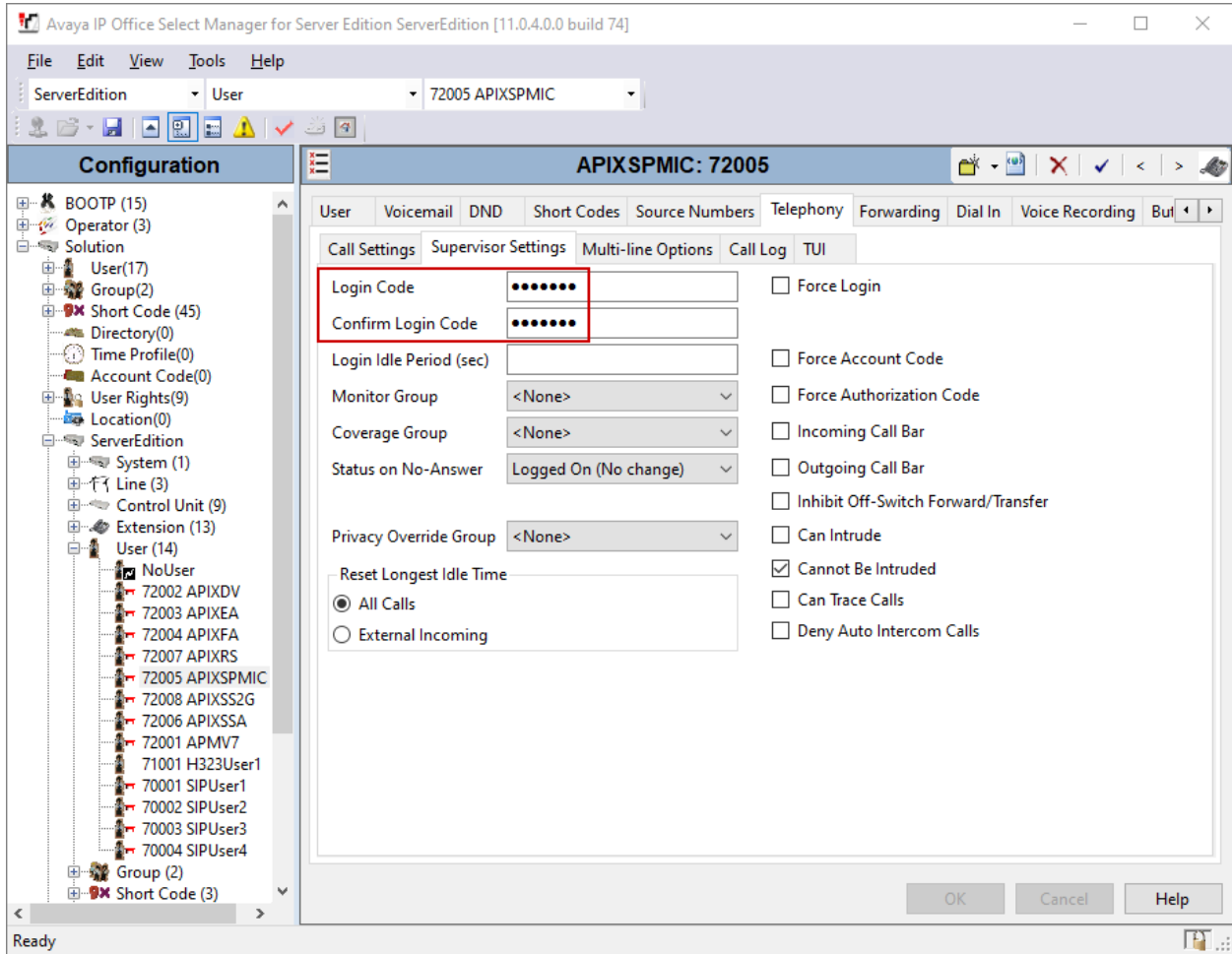


## 5.5. Administer SIP Users

To create a new SIP User, from the configuration tree in left pane, right-click on **User**, and select **New** from the pop-up list (not shown). Enter desired values for the **Name** field. For the **Extension** field, enter the SIP extension created in **Section 5.4**.



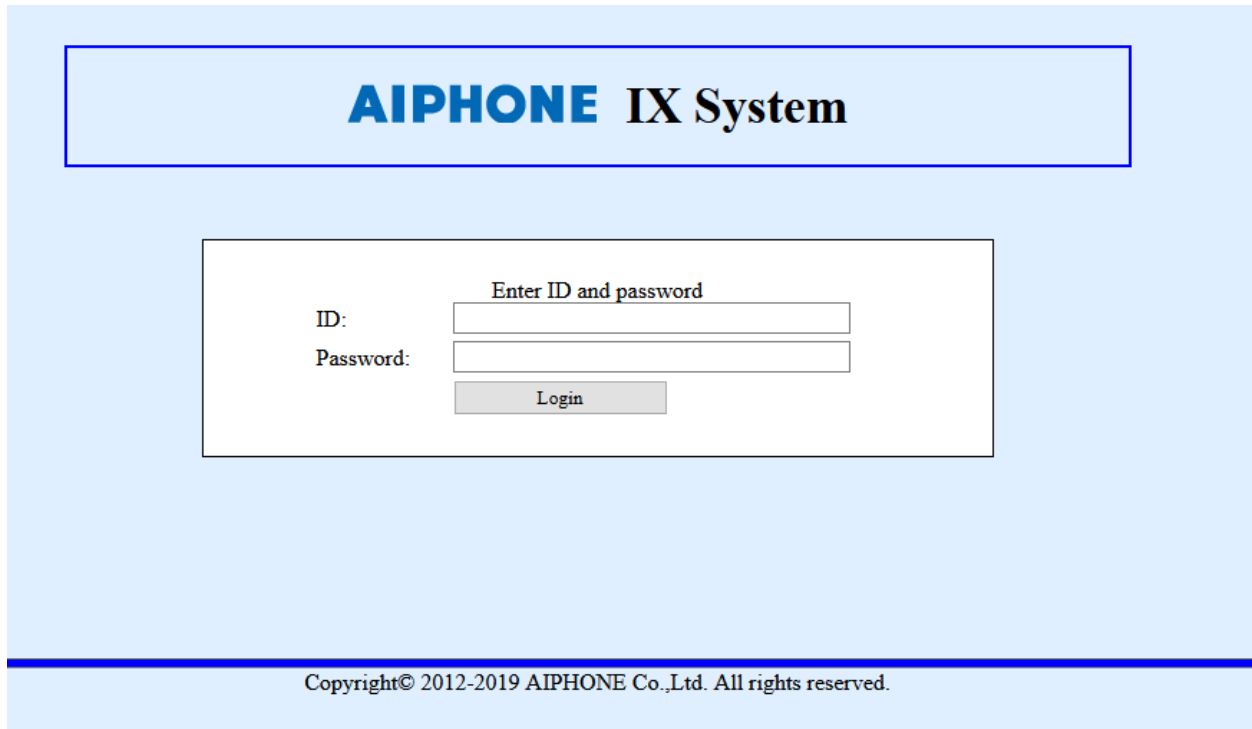
Select the **Supervisor Settings** tab, and enter a desired **Login Code** and **Confirm Login Code**. This code will be used as a password for Aiphone IX-SPMIC.



## 6. Configure Aiphone IX Series Audio Door Station

This section provides steps to configure Aiphone IX-SPMIC.

To configure Aiphone IX-SPMIC, using a web browser, navigate to <https://<IP Address of IX-SPMIC>/webset.cgi?login> and log in using appropriate credentials.



**AIPHONE IX System**

Enter ID and password

ID:

Password:

Login

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Once logged in, for the **Number** field, type in the SIP extension that is being configured (from **Section 5.4**), and a desired **Name**. Select **Update** to save change.

**AIPHONE IX System Setting** Update

Category: Audio Sub Station Station Type: IX-SPMIC

**Station Information**

Identification  
ID and Password  
Language  
Time  
Expanded System

**Network Settings**

IP Address  
DNS  
SIP  
Audio

**Station Information**

◆ Identification

Number ◆	<input type="text" value="72005"/>	3-5 digits
Name	<input type="text" value="IX-SPMIC"/>	1-24 alphanumeric characters
Location	<input type="text"/>	1-24 alphanumeric characters

◆ Required Setting

From the left, select **Network Settings** → **SIP** and configure as follows:

- **SIP Signaling Port:** Set to **5060**.
- **User Agent:** Type in a desired value.
- **ID:** SIP Extension number from **Section 5.4**.
- **Password:** SIP Extension password from **Section 5.4**.
- **IPv4 Address:** LAN IP Address of IP Office from **Section 5.2**.
- **Port:** Set to **5060**.

Once done, select **Update** to save changes.

**AIPHONE IX System Setting** Update

Category: Audio Sub Station Station Type: IX-SPMIC

**Network Settings**

◆ SIP

**SIP Connections**

SIP Signaling Port ◆	<input type="text" value="5060"/>	1-65535
User Agent	<input type="text" value="IX-SPMIC"/>	1-36 alphanumeric characters

**SIP Server**

Primary Server

ID	<input type="text" value="72005"/>	1-24 alphanumeric characters
Password	<input type="text" value="●●●●●●"/>	1-24 alphanumeric characters
IPv4 Address	<input type="text" value="10.64.110.65"/>	1.0.0.1-223.255.255.254 or hostname(1-64 alphanumeric)
IPv6 Address	<input type="text"/>	::FF:0-FE:FF:FFFF:FFFF:FFFF:FFFF:FFFF:FFFF:FF
Port ◆	<input type="text" value="5060"/>	1-65535

From the left, select **Call Settings** → **Station Settings (for Door)** and configure as follows:  
The numbers configured here will be dialed when the button on the IX-SPMIC is pressed.

- **Station Number:** Type in an extension number on IP Office that will be called for a given line.
- **IPv4:** Type in the LAN IP Address from **Section 5.2**.
- **Station Type:** Set to **VoIP Phone**.

Select **Update** to save changes.

**AIPHONE IX System Setting**  
Category: Audio Sub Station Station Type: IX-SPMIC

**Call Settings**

•Called Stations (for Door)

Option Input #:

Station Number must be 3-5 digits. (3-32 digits for VoIP Phone)  
IPv4 must be 1.0.0.1-223.255.255.254 or hostname(1-64 alphanumeric characters).  
IPv6 must be :FF:0:FEFF:FFFF:FFFF:FFFF:FFFF:FFFF or hostname(1-64 alphanumeric characters).  
Enter SIP Primary Server IP address for VoIP Phone, set only one VoIP Phone per call group.  
Station Type must be "VoIP Phone" when calling via SIP server.  
U = Unicast, M = Multicast

#	Station Number	IPv4 Address	IPv6 Address	Station Type
1	70001	10.64.110.65		VoIP Phone
2				
3				

From the left, select **Function Settings** → **CGI** and select the **Enable** radio button for **CGI Functionality**. This enables Aiphone IX-SPMIC to place calls via a CGI URL.

**AIPHONE IX System Setting**  
Category: Audio Sub Station Station Type: IX-SPMIC

**Function Settings**

•CGI

CGI Functionality  Enable  Disable

## 7. Verification Steps

The following steps may be used to verify the configuration:

- From a PC running the Avaya IP Office Monitor application. The **Avaya IP Office SysMonitor** screen is displayed (not shown). Select **Status** → **SIP Phone Status** from the top menu. Verify the SIP extension added from **Section 5.4** is displayed and the Status is **SIP: Registered**.

SIPPhoneStatus

Total Configured: 27      Waiting 0 secs for update

Total Registered: 12      Registered Status: [Progress Bar]

Extn Num	User Num	Phone Type	Security	IP Address	Transport	User Agent	Licensed	SIP Options	Status
\$1.70001	70001	EQN_X_D...	disable	10.64.10.48	TCP	Avaya Communicator/3.0 (3.6.4.31....	Avaya Softph...	RU	SIP: Registered
70001	70001	J129 SIP	disable	10.64.10.224	TCP	Avaya J129 IP Phone 4.0.0.0.21 (21)	Avaya IP	RU	SIP: Registered
70002	70002	VANTAG...	disable	10.64.10.223	TCP	Avaya Communicator Android/3.5.0...	Avaya IP	RU	SIP: Registered
70003	70003	H175 SIP	disable	10.64.10.222	TCP	Avaya H175 Collaboration Station ...	Avaya IP	RU	SIP: Registered
72001	72001	SIP	disable	10.64.10.40	UDP	IX-MV7	3rd Party IP	RU	SIP: Registered
72002	72002	SIP	disable	10.64.10.41	UDP	IX-DV	3rd Party IP	RU	SIP: Registered
72003	72003	SIP	disable	10.64.10.43	UDP	IX-EA	3rd Party IP	RU	SIP: Registered
72004	72004	SIP	disable	10.64.10.44	UDP	IX-FA	3rd Party IP	RU	SIP: Registered
72005	72005	SIP	disable	10.64.10.45	UDP	IX-SPMIC	3rd Party IP	RU	SIP: Registered
72006	72006	SIP	disable	10.64.10.46	UDP	IX-SSA	3rd Party IP	RU	SIP: Registered
72007	72007	SIP	disable	10.64.10.49	UDP	IX-RS	3rd Party IP	RU	SIP: Registered
72008	72008	SIP	disable	10.64.10.42	UDP	IX-SS-2G	3rd Party IP	RU	SIP: Registered

Display Options:  Show All  Registered  UnRegistered      Page 1      Save Page      Reset Phones      Reregister Phones      Cancel

- Place a call from Aiphone IX-SPMIC to an Avaya endpoint. The state of the call be viewed on a PC running the **Avaya IP Office System Status** application. Select **Extensions** → Aiphone IX-SPMIC extension.

Calls be placed by using the following URL:

<https://<IP Address of IX-SPMIC>/CallUP.cgi?ID=admin&PW=admin>

Once calls are connected, calls can be terminated by using the following URL:

<https://<IP Address of IX-SPMIC>/CallTalkEnd.cgi?ID=admin&PW=admin>

The screenshot shows the Avaya IP Office System Status application interface. The title bar indicates the server edition (10.64.110.65) and the IP Office Linux PC version (11.0.4.0.0 build 74). The main window is titled "IP Office System Status" and features a navigation menu on the left and a central content area.

The navigation menu includes sections for System, Alarms, Extensions, Trunks, Active Calls, Resources, Voicemail, IP Networking, and Locations. The "Extensions (13)" section is expanded, showing a list of extensions from 70001 to 72008, with 72005 selected.

The "Extension Status" section displays the following details for extension 72005:

- Extension Number: 72005
- IP address: 10.64.10.45
- Standard Location: None
- Registrar: Primary
- Telephone Type: Unknown SIP Device
- User-Agent SIP header: IX-SPMIC
- Media Stream: RTP
- Layer 4 Protocol: UDP
- Current User Extension Number: 72005
- Current User Name: APIXSPMIC
- Forwarding: Off
- Twinning: Off
- Do Not Disturb: Off
- Message Waiting: Off
- Phone Manager Type: None
- SIP Device Features: REFER,UPDATE
- License Reserved: No
- Last Date and Time License Allocated: 11/14/2019 2:54:41 AM
- DTMF Required: No
- Packet Loss Fraction: Connection Type: Direct Media
- Jitter: Codec: G711 Mu
- Round Trip Delay: Remote Media Address: 10.64.10.224

Below the extension status, a call log table shows an active call:

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
495	Connected	00:00:16		Outgoing	Extn 70001, SIPUser1

At the bottom of the application, there are buttons for Trace, Trace All, Pause, Ping, Call Details, Print..., and Save As... The status bar at the bottom right shows the time as 11:37:03 PM and the system is Online.

## 8. Conclusion

Aiphone IX-SPMIC was compliance tested with Avaya IP Office. Aiphone IX-SPMIC functioned properly for feature and serviceability.

## 9. Additional References

Avaya IP Office product documentation can be found at: <https://ipofficekb.avaya.com/>

Documentation related to Aiphone IX-SPMIC can be found at:  
<https://www.aiphone.co.jp/products/business/ix/>



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