



Avaya Solution & Interoperability Test Lab

Application Notes for Aiphone IX Series Audio Door Stations (IX-SS-2G) and Avaya IP Office™ – Issue 1.0

Abstract

These Application Notes describe the procedures for configuring Aiphone IX Series Audio Door Stations (IX-SS-2G) 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-SS-2G) 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 2**, in particular the scope of testing as outlined in **Section 2.1** as well as any observations noted in **Section 2.2**, 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-SS-2G) to interoperate with Avaya IP Office (IP Office). During the compliance testing, Aiphone IX-SS-2G was used.

The Aiphone IX Series Audio Door Stations (IX-SS-2G) are part of Aiphone IX Series 2 Door Stations. The Audio Door Stations act as SIP phones when connected to IP Office. Stations come in both surface mount and flush mount varieties, please see **Appendix A** regarding various versions of IX-SS-2G devices. 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.

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-SS-2G registered as a 3rd 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-SS-2G can register as a SIP endpoint on IP Office, and is able to originate and receive audio calls to and from 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-SS-2G, and exercise basic telephone operations. The main objectives were to verify the following:

- Registration
- Calls to/from Avaya SIP Audio endpoints
- Calls to/from Avaya H.323 Audio endpoints
- Calls to/from Avaya Digital & Analog endpoints
- Calls to/from 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-SS-2G, please contact Aiphone via the following:

Japan

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

USA, Canada

- Web: <https://www.aiphone.com/home>
- Email: tech@aiphone.com
- Phone: 800-692-0200

France

- Web: <https://www.aiphone.fr/>
- Phone: 01 69 11 46 00

Australia, New Zealand

- Web: <https://www.aiphone.com.au/>
- Phone: (02)80364507

Singapore

- Web: <http://www.aiphone.com.sg/>
- Email: admin@aiphone.com.sg
- Phone: 6534-1135

United Kingdom

- Web: <https://www.aiphone.co.uk/>
- Phone: 020-7507-6250

3. Reference Configuration

Figure 1 illustrates a sample configuration consisting of Avaya IP Office components and Aiphone IX-SS-2G.

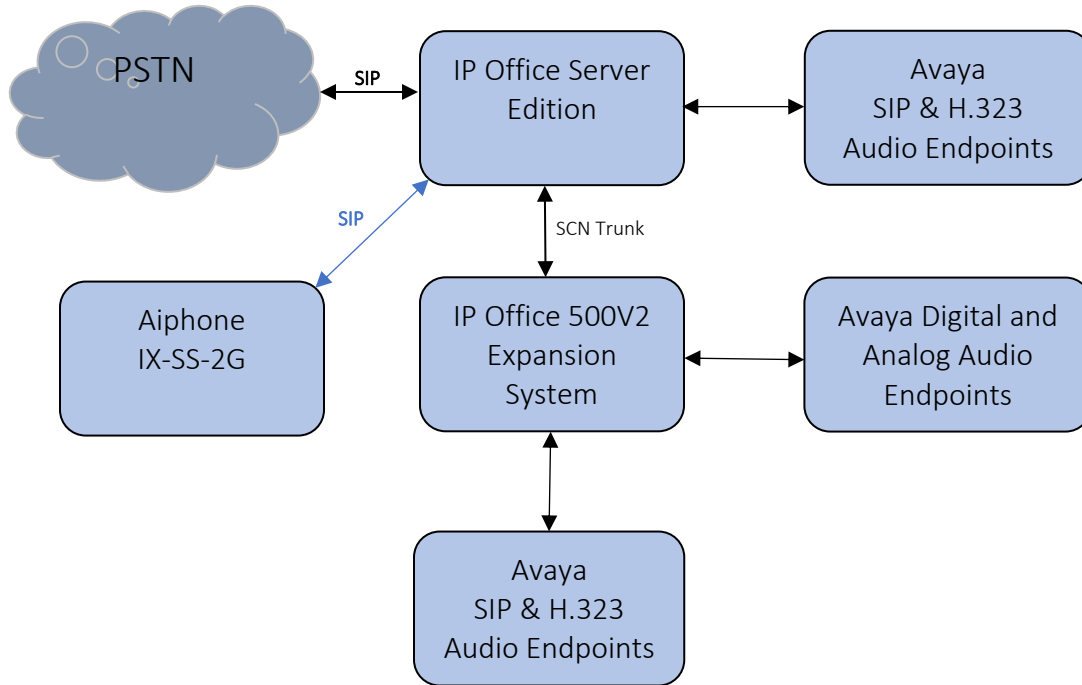


Figure 1: Test Configuration of Aiphone IX-SS-2G with Avaya IP Office

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 Station IX-SS-2G	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:

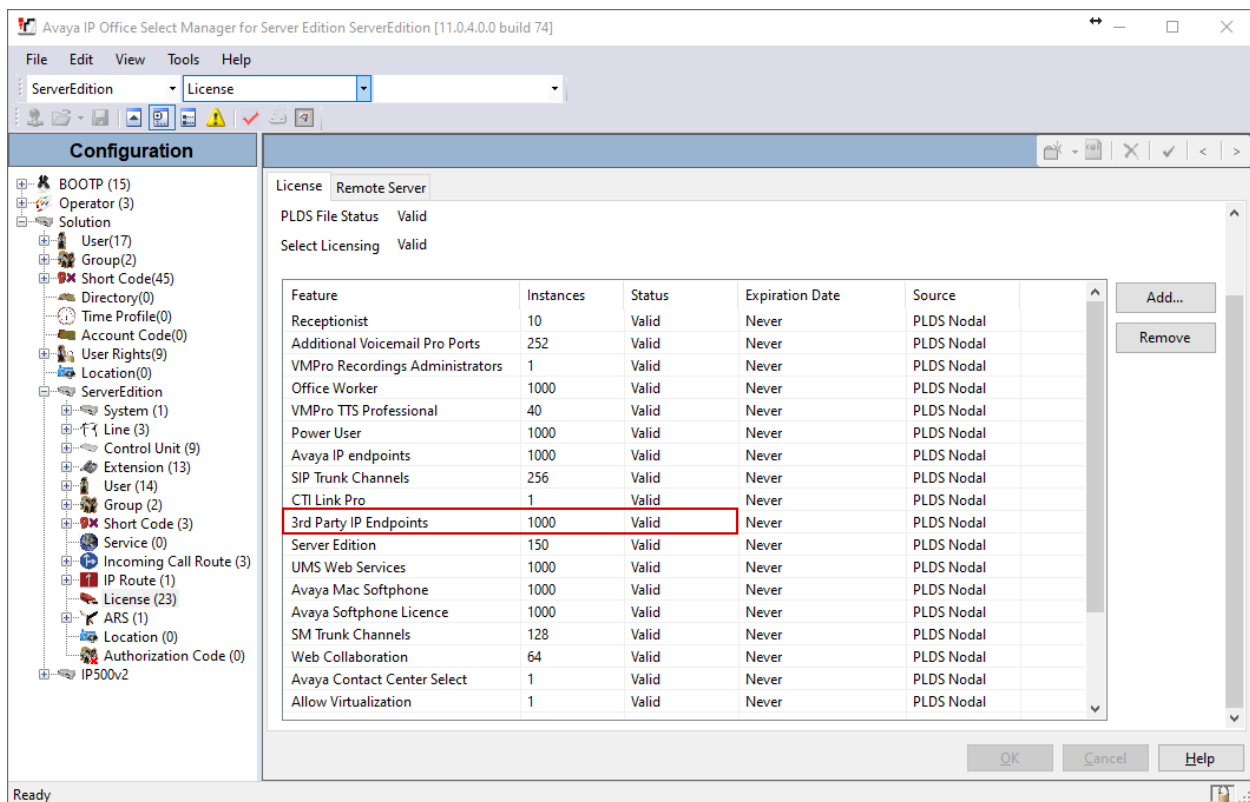
- Verify IP Office license
- Obtain LAN IP address
- Administer SIP registrar
- Administer SIP extensions
- Administer SIP users

These steps are performed from the Avaya IP Office Manager.

5.1. Verify IP Office License

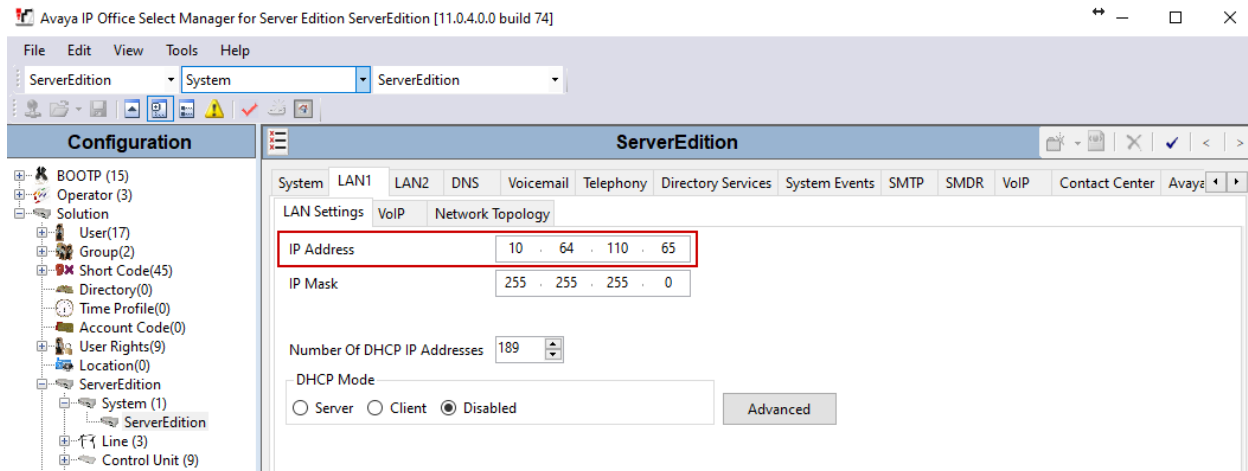
From a PC running the Avaya IP Office Manager application, select **Start → All Programs → IP Office → Manager** to launch the Manager application. Select the proper IP Office system if there are more than one IP Office system, and log in with the appropriate credentials.

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



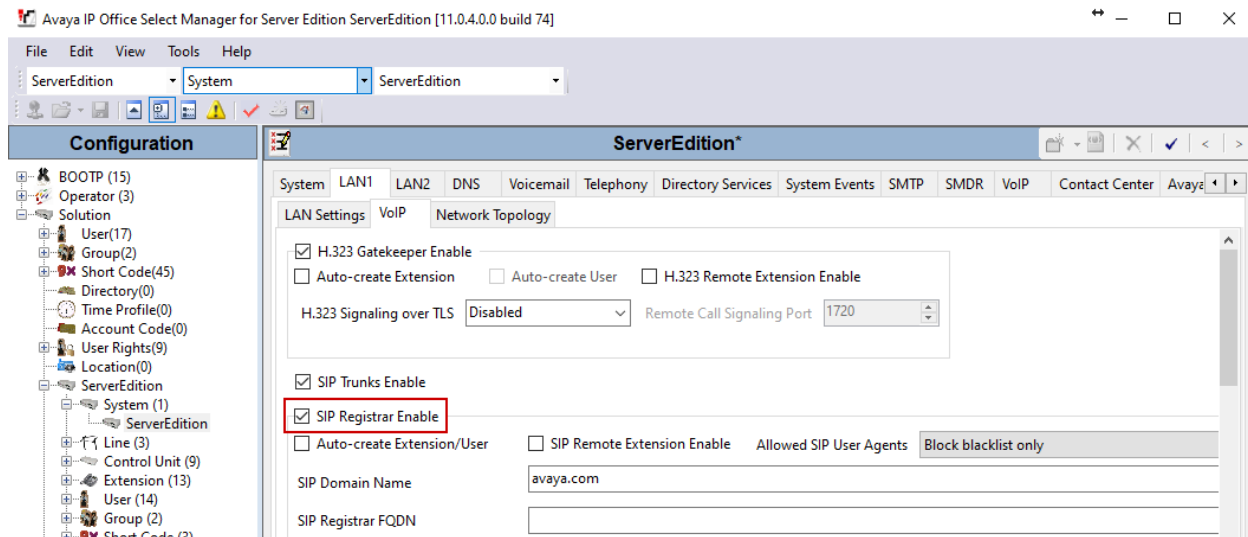
5.2. Obtain LAN IP Address

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-SS-2G.



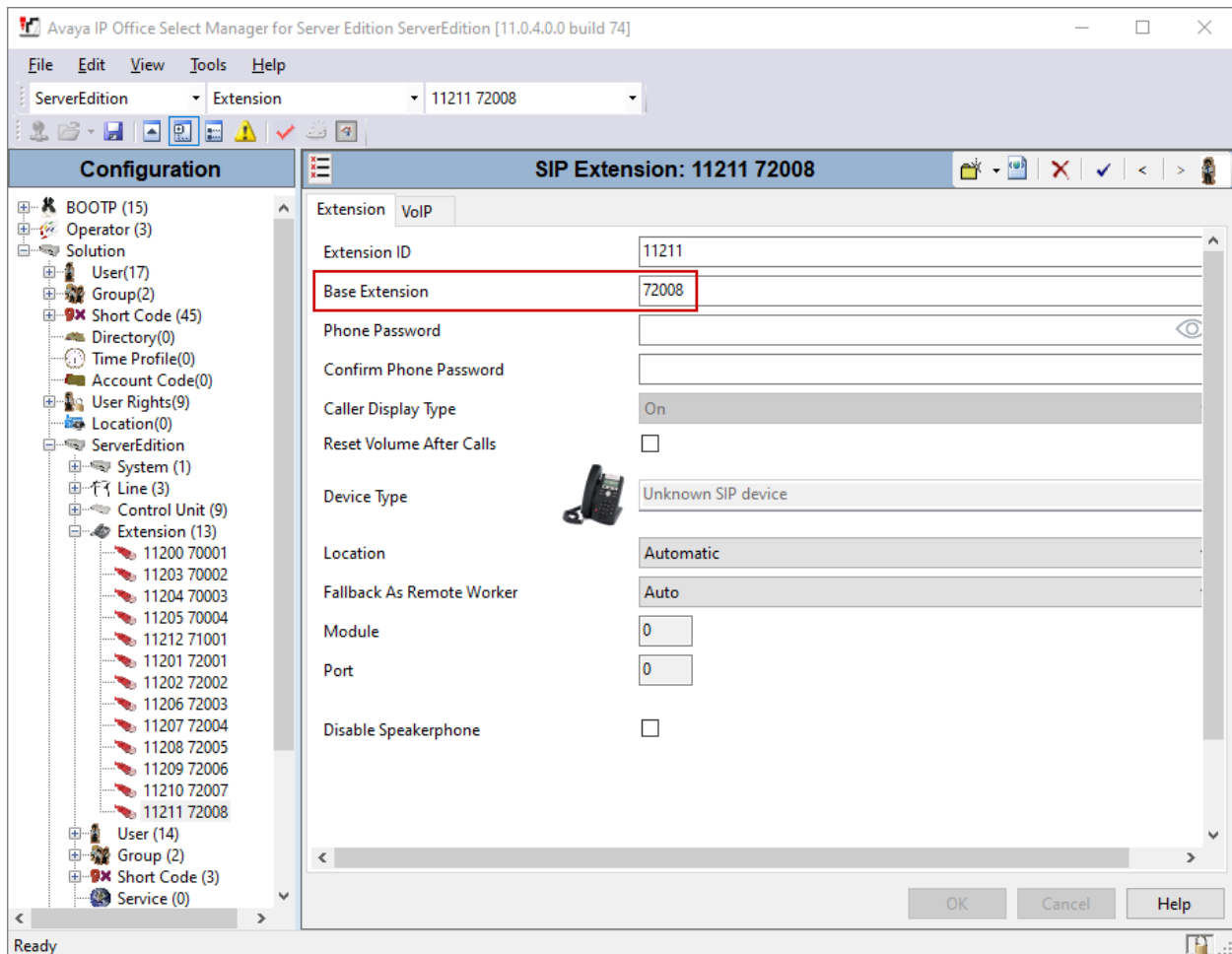
5.3. Administer SIP Registrar

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-SS-2G 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**.

The screenshot displays the Avaya IP Office Select Manager for Server Edition [11.0.4.0.0 build 74] window. The left pane shows the Configuration tree with the following structure:

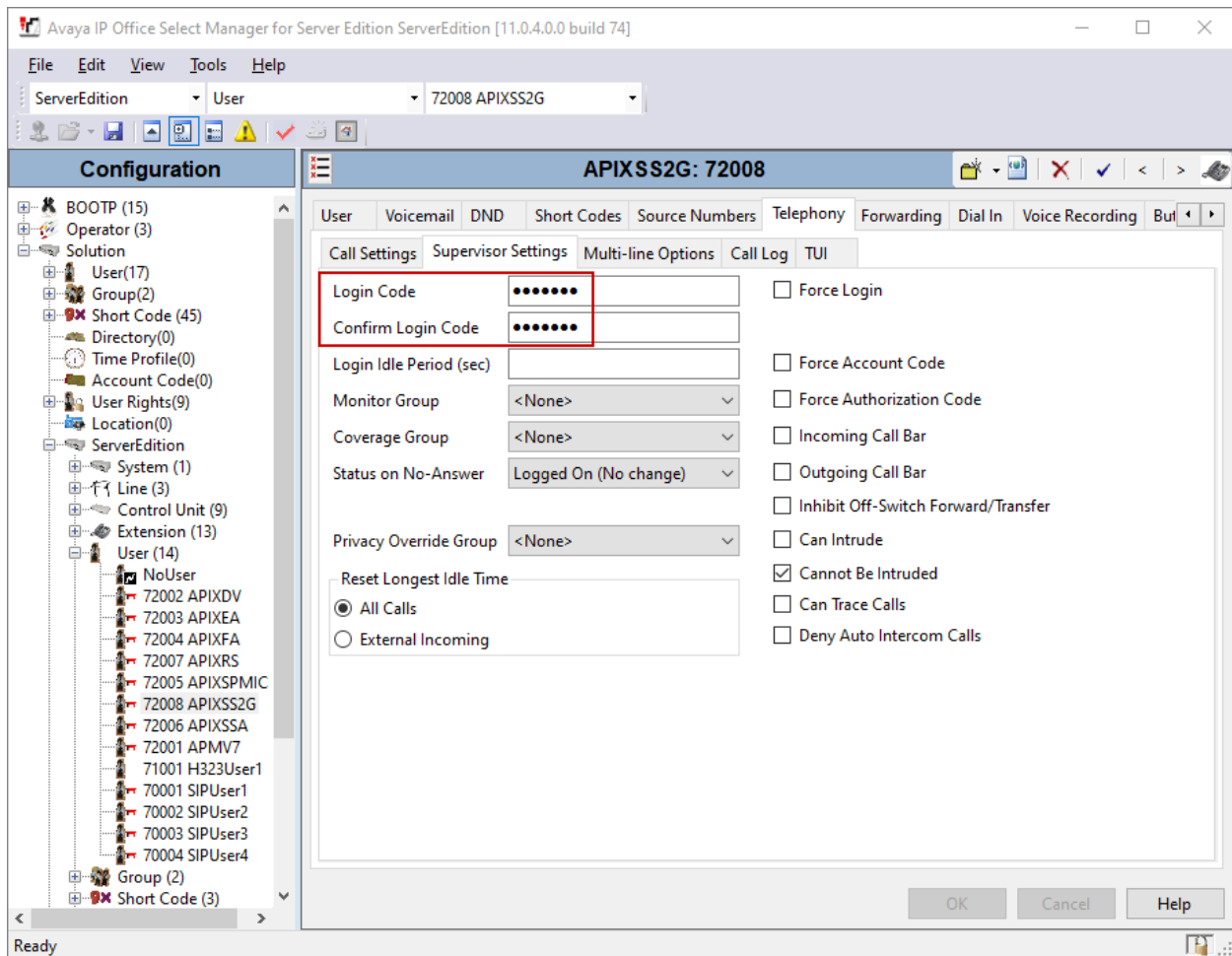
- BOOTP (15)
- Operator (3)
- Solution
 - User(17)
 - Group(2)
 - Short Code (45)
 - Directory(0)
 - Time Profile(0)
 - Account Code(0)
 - User Rights(9)
 - Location(0)
 - ServerEdition
 - System (1)
 - Line (3)
 - Control Unit (9)
 - Extension (13)
 - User (14)
 - NoUser
 - 72002 APIXDV
 - 72003 APIXEA
 - 72004 APIXFA
 - 72007 APIXRS
 - 72005 APIXSPMIC
 - 72008 APIXSS2G
 - 72006 APIXSSA
 - 72001 APMV7
 - 71001 H323User1
 - 70001 SIPUser1
 - 70002 SIPUser2
 - 70003 SIPUser3
 - 70004 SIPUser4
 - Group (2)
 - Short Code (3)

The right pane shows the configuration for the selected user, **APIXSS2G: 72008**. The tabs are: User, Voicemail, DND, Short Codes, Source Numbers, Telephony, Forwarding, Dial In, Voice Recording, and But. The fields are:

- Name: APIXSS2G
- Password:
- Confirm Password:
- Unique Identity:
- Conference PIN:
- Confirm Audio Conference PIN:
- Account Status: Enabled
- Full Name:
- Extension: 72008
- Email Address:
- Locale:
- Priority: 5
- System Phone Rights: None
- Profile: Basic User
 - ☐ Receptionist
 - ☐ Enable Softphone
 - ☐ Enable one-X Portal Services
 - ☐ Enable one-X Telephony

Buttons at the bottom: OK, Cancel, Help.

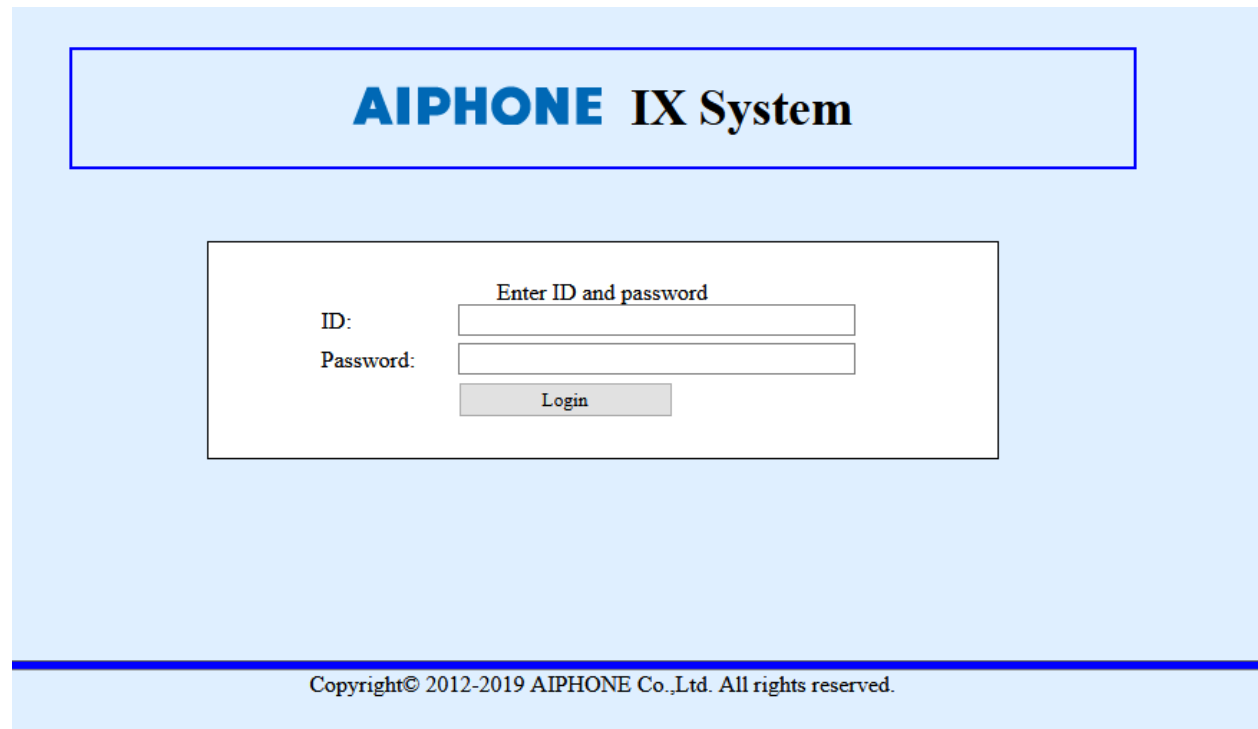
Select the **Supervisor Settings** tab, and enter a desired **Login Code** and **Confirm Login Code**. This code will be used as a password for Aiphone IX-SS-2G.



6. Configure Aiphone IX Series Audio Door Station

This section provides steps to configure Aiphone IX-SS-2G.

To configure Aiphone IX-SS-2G, using a web browser, navigate to <https://<IP Address of IX-SS-2G>/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
 Category: Audio Stations Station Type: IX-SS-2G
 Station Information
 Identification
 ID and Password
 Language
 Time
 Expanded System
 Network Settings
 IP Address
 DNS
 SIP
 Audio
 Number 72008 3-5 digits
 Name IX-SS-2G 1-24 alphanumeric characters
 Location 1-24 alphanumeric characters
 Required Settings

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
 Category: Audio Stations Station Type: IX-SS-2G
 Station Information
 Identification
 ID and Password
 Language
 Time
 Expanded System
 Network Settings
 IP Address
 DNS
 SIP
 Audio
 Packet Priority
 NTP
 System Information
 Custom Sound Registry
 Network Settings
 SIP
 SIP Connections
 SIP Signaling Port 5060 1-65535
 User Agent IX-SS-2G 1-36 alphanumeric characters
 SIP Server
 Primary Server
 ID 72008 1-24 alphanumeric characters
 Password ***** 1-24 alphanumeric characters
 IPv4 Address 10.64.110.65 1.0.0.1-223.255.255.254 or hostname(1-64 alphanumeric)
 IPv6 Address ::FF:0:FEFF:FEFF:FEFF:FEFF:FEFF:FEFF:FF ::FF:0:FEFF:FEFF:FEFF:FEFF:FEFF:FEFF:FF
 Port 5060 1-65535

From the left, select **Call Settings** → **Station Settings** and configure as follows:
The numbers configured here will be dialed when the button on the IX-SS-2G 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 Update

Category: Audio Stations Station Type: IX-SS-2G

Call Settings

Station Information

Call Button Function: Call, Answer Call, End Communication

"Cancel Call, End Communication" disabled when using Option Input call.

Called Stations (for Door)

Option Input #: Group 01

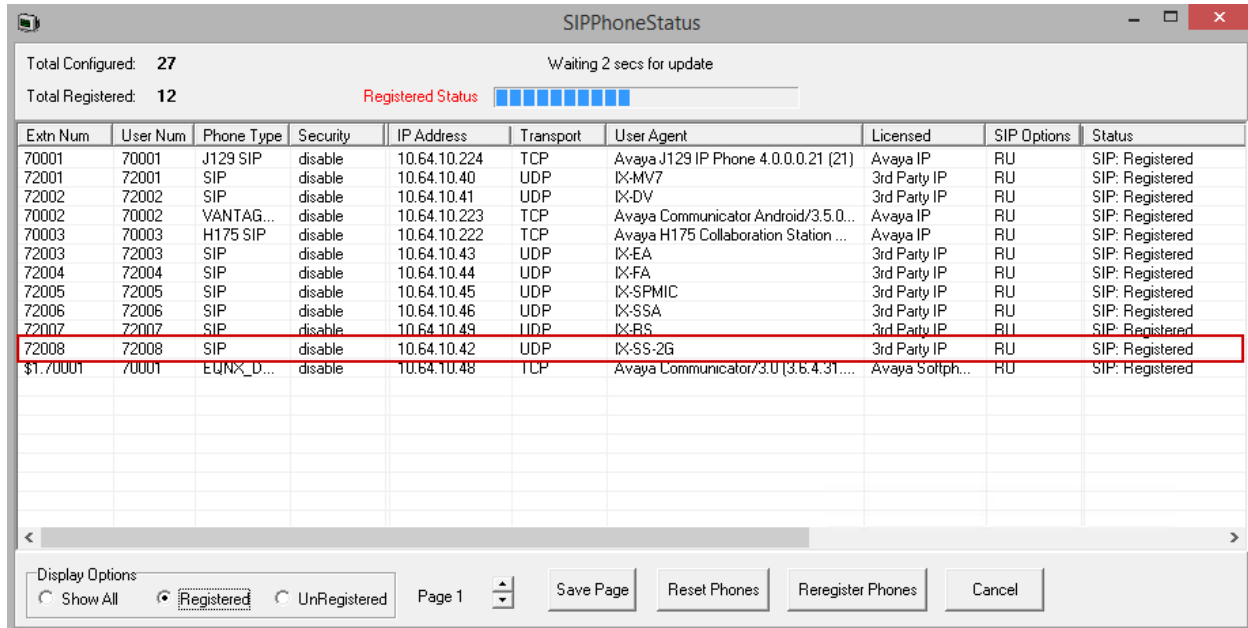
Station Number must be 3-5 digits. (3-32 digits for VoIP Phone)
IPv4 must be 1.0.0.1-223.255.255.254 or hostname(1-64 alphanumeric characters).
IPv6 must be ::FF:0:FEFF:FFFF:FFFF:FFFF:FFFF:FFFF or hostname(1-64 alphanumeric characters).
Enter SIP Primary Server IP address for VoIP Phone, set only one 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	78001	10.64.110.65		VoIP Phone
2				
3				

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**.



The screenshot shows the SIPPhoneStatus application window. At the top, it displays 'Total Configured: 27' and 'Total Registered: 12'. A progress bar indicates 'Registered Status' with 12 blue bars. Below this is a table with columns: Extn Num, User Num, Phone Type, Security, IP Address, Transport, User Agent, Licensed, SIP Options, and Status. The table lists 12 SIP phones, all with a status of 'SIP: Registered'. The row for extension 72008 is highlighted with a red border. At the bottom, there are 'Display Options' (Show All, Registered, UnRegistered), 'Page 1', and buttons for 'Save Page', 'Reset Phones', 'Reregister Phones', and 'Cancel'.

Extn Num	User Num	Phone Type	Security	IP Address	Transport	User Agent	Licensed	SIP Options	Status
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
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
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
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-BS	3rd Party IP	RU	SIP: Registered
72008	72008	SIP	disable	10.64.10.42	UDP	IX-SS-2G	3rd Party IP	RU	SIP: Registered
170001	70001	EQUX_D...	disable	10.64.10.48	TCP	Avaya Communicator/3.0 (3.6.4.31...	Avaya Softph...	RU	SIP: Registered

- Place a call from Aiphone IX-SS-2G to an Avaya endpoint. The state of the call can be viewed on a PC running the **Avaya IP Office System Status** application. Select **Extensions** → Aiphone IX-SS-2G extension.

The screenshot shows the Avaya IP Office System Status application window. The title bar indicates the server edition and IP address. The left sidebar contains a tree view with categories like System, Alarms, Extensions, Trunks, Active Calls, Resources, Voicemail, IP Networking, and Locations. The 'Extensions' category is expanded, showing a list of extensions. Extension 72008 is selected. The main pane displays the 'Extension Status' for 72008, including details like IP address, location, registrar, and current user. Below this, a table shows call details for a specific call (Call Ref 494).

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
494	Connected	00:00:49	78001	Outgoing	Line: 3 SIP avaya.com Group: 1 Ch

8. Conclusion

Aiphone IX-SS-2G was compliance tested with Avaya IP Office. Aiphone IX-SS-2G 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-SS-2G can be found at:

Japan: <https://www.aiphone.co.jp/products/business/ix/>

USA, Canada: <https://www.aiphone.com/home/products/ix-series>

France: <https://www.aiphone.fr/catalogue/interphonie-ip-protocole-sip-ix/>

Australia, New Zealand: <https://www.aiphone.com.au/product/ix/>

Singapore: <http://www.aiphone.com.sg/>

United Kingdom: https://www.aiphone.co.uk/featured_item/ix2/

Appendix A

Following devices are based on the same firmware as IX-SS-2G:

- IX-NVP
- IX-NVP2
- IX-SS-2G
- IX-PR2

The difference in each IX-SS-2G devices is compatible box, attachment and its plate type:

- IX-NVP
 - Compatible with 3-Gang box
- IX-NVP2
 - Compatible with 3-Gang box
 - Emergency call button
- IX-PR2
 - Plate is modified for high security facility
- IX-SS-2G
 - Flush mounting

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