



DevConnect Program

Application Notes for Aiphone IX Series 2 Video Door Station (IX-DB) with Avaya IP Office Server Edition - Issue 1.0

Abstract

These Application Notes describe the configuration steps required to integrate Aiphone IX Series 2 Video Door Station (IX-DB) Version 7.00 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500V2 Expansion System 11.1. The Aiphone IX-DB Video Door Station, which is part of the Aiphone IX Series 2 Video Door Stations, was used for the compliance test. Aiphone IX-DB Video Door Station is a surface mount, weather resistant video door station. Aiphone IX-DB Video Door Station registers with Avaya IP Office as a SIP endpoint.

Readers should pay attention to **Section 2**, in particular the scope of testing as outlined in **Section 2.1** as well as the 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 Avaya DevConnect Program.

1. Introduction

These Application Notes describe the configuration steps required to integrate Aiphone IX Series 2 Video Door Station (IX-DB) Version 7.00 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500V2 Expansion System 11.1. The Aiphone IX-DB Video Door Station, which is part of the Aiphone IX Series 2 Video Door Stations, was used for the compliance test. Aiphone IX-DB Video Door Station is a surface mount, weather resistant video door station. Aiphone IX-DB Video Door Station (IX-DB) registers with Avaya IP Office as a SIP endpoint.

2. General Test Approach and Test Results

The interoperability compliance test included feature and serviceability testing. The feature testing focused on establishing audio and video calls between Aiphone IX-DB Video Door Station, Avaya SIP and H.323 telephones, Avaya Workplace Client for Windows, Avaya Vantage™ K175, and the PSTN, and exercising basic telephony features, such as hold/resume, mute/unmute, transfer, conference, call forwarding, and call coverage from an Avaya IP endpoint. Additional telephony features, such as call forward and call coverage, were also verified.

The serviceability testing focused on verifying that the Aiphone IX-DB Video Door Station comes back into service after re-connecting the Ethernet cable.

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 this DevConnect Application Note 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 this Application Note, the interface between Avaya systems and Aiphone IX-DB Video Door Station did not include use of any specific encryption features as requested by Aiphone.

2.1. Interoperability Compliance Testing

Interoperability compliance testing covered the following features and functionality:

- SIP registration of IX-DB with IP Office Server Edition and IP Office 500V2 Expansion System.
- Audio calls between IX-DB and Avaya SIP and H.323 deskphones with Direct IP Media (Shuffling) enabled and disabled.
- Audio and video calls between IX-DB, Workplace, and Vantage K175 with Direct IP Media (Shuffling) enabled and disabled. One-way video from IX-DB to Workplace and Vantage K175 was verified.
- Audio calls between IX-DB and the PSTN.
- G.711 codec support.
- UDP transport protocol.
- IX-DB placing, answering, and terminating calls.
- Basic telephony features, including hold/resume, mute/unmute, transfer, and 3-way conference, initiated from an Avaya IP endpoint.
- Proper system recovery after re-establishing IP connectivity to IX-DB.

2.2. Test Results

All test cases executed passed successfully with the following observations:

- IX-DB auto answers calls placed to them.
- IX-DB does not support remote door open via DTMF input of Door Release Authorization Authentication Key
- IX-DB only supports G.711 codec.

2.3. Support

For technical support of Aiphone IX Series 2 Video Door Stations, contact Aiphone Technical Support via phone or website.

- Phone: +1 (800) 692-0200
- Web: <https://www.aiphone.com/support/technical-support>

3. Reference Configuration

Figure 1 illustrates a sample configuration with an Avaya SIP-based network. Aiphone IX-DB Video Door Station registered to either IP Office Server Edition or IP Office 500 V2 Expansion System (not simultaneously).

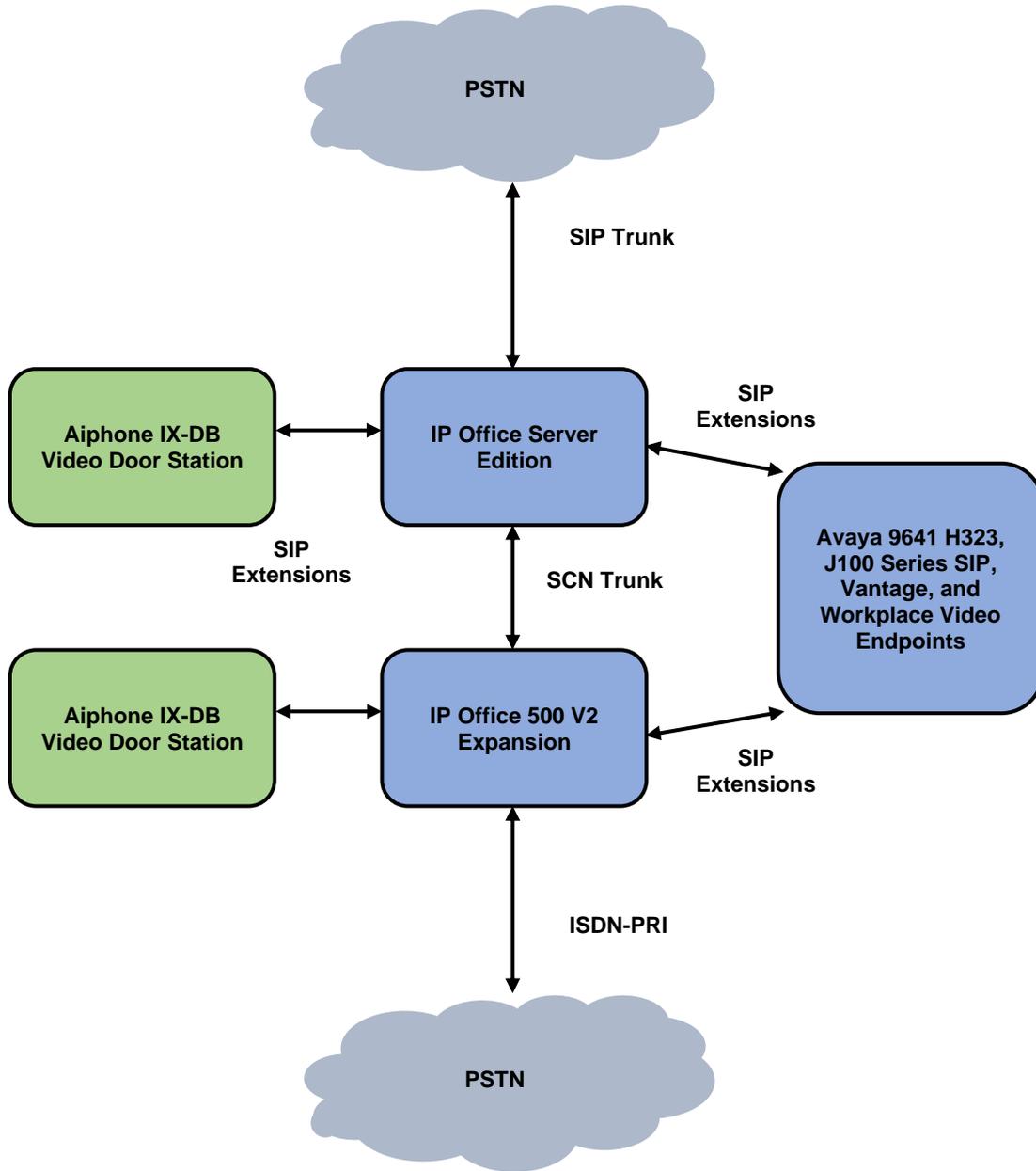


Figure 1: Avaya SIP Telephony Network with Aiphone IX-DB Video Door Station

4. Equipment and Software Validated

The following equipment and software were used for the sample configuration provided:

Equipment/Software	Release/Version
Avaya IP Office Server Edition	11.1.2.4.0 build 18 (FP2 SP4)
Avaya IP Office 500V2 Expansion System	11.1.2.4.0 build 18 (FP2 SP4)
Avaya 96x1 Series IP Deskphones	6.8.5.2.3 (H.323)
Avaya J100 Series IP Phones	4.0.10.3.2 (SIP)
Avaya K175 Vantage Device	3.1.1.2 (bld version 0012)
Avaya Workplace	3.32.0.75
Aiphone IX-DB Video Door Station	7.00

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

5. Configure Avaya IP Office Server Edition

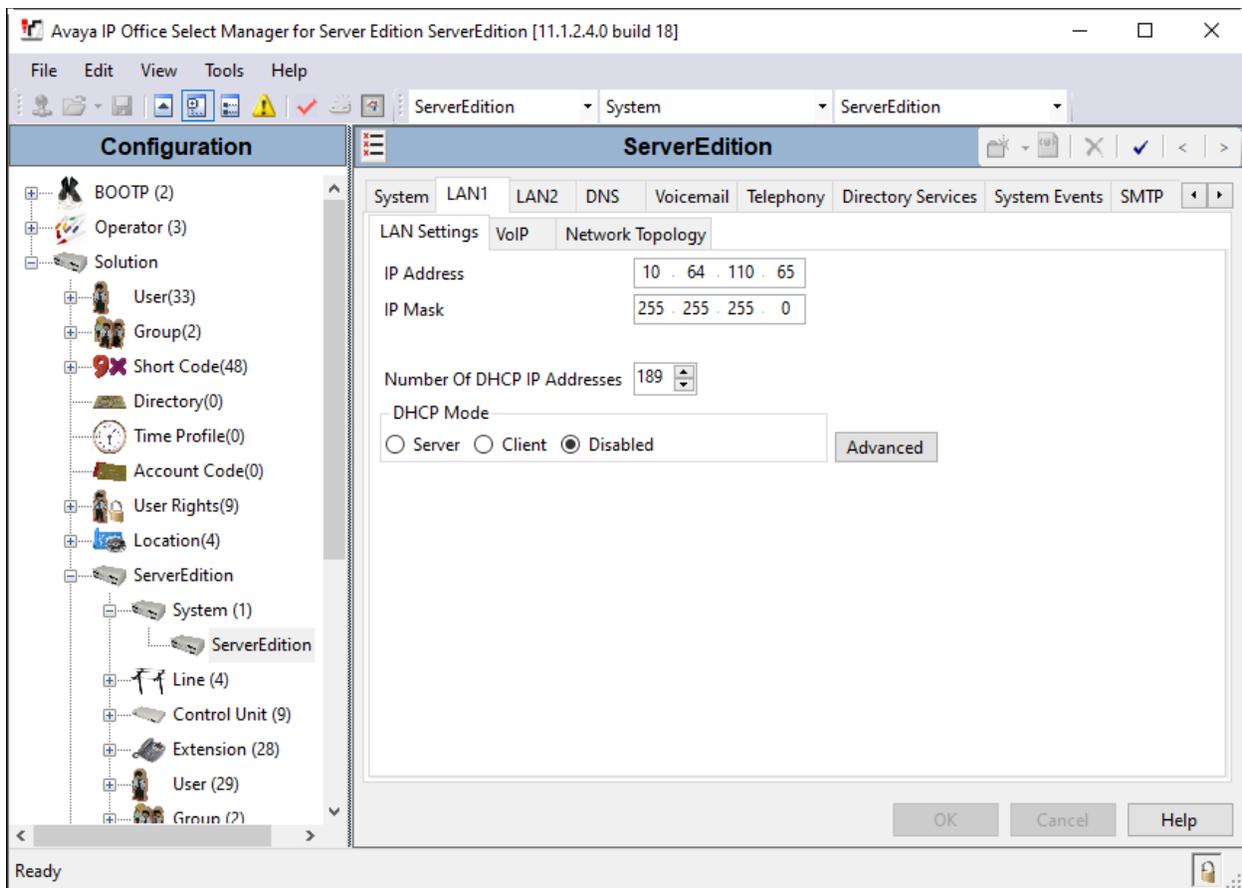
This section provides the procedures for configuring Avaya IP Office Server Edition. The procedures include the following areas:

- Obtain LAN IP Address
- Administer SIP Registrar
- Administer SIP Extension for IX-DB
- Administer SIP User for IX-DB

Note: This section covers the configuration of Avaya IP Office Server Edition, but the configuration is the same for Avaya IP Office 500 V2 Expansion System.

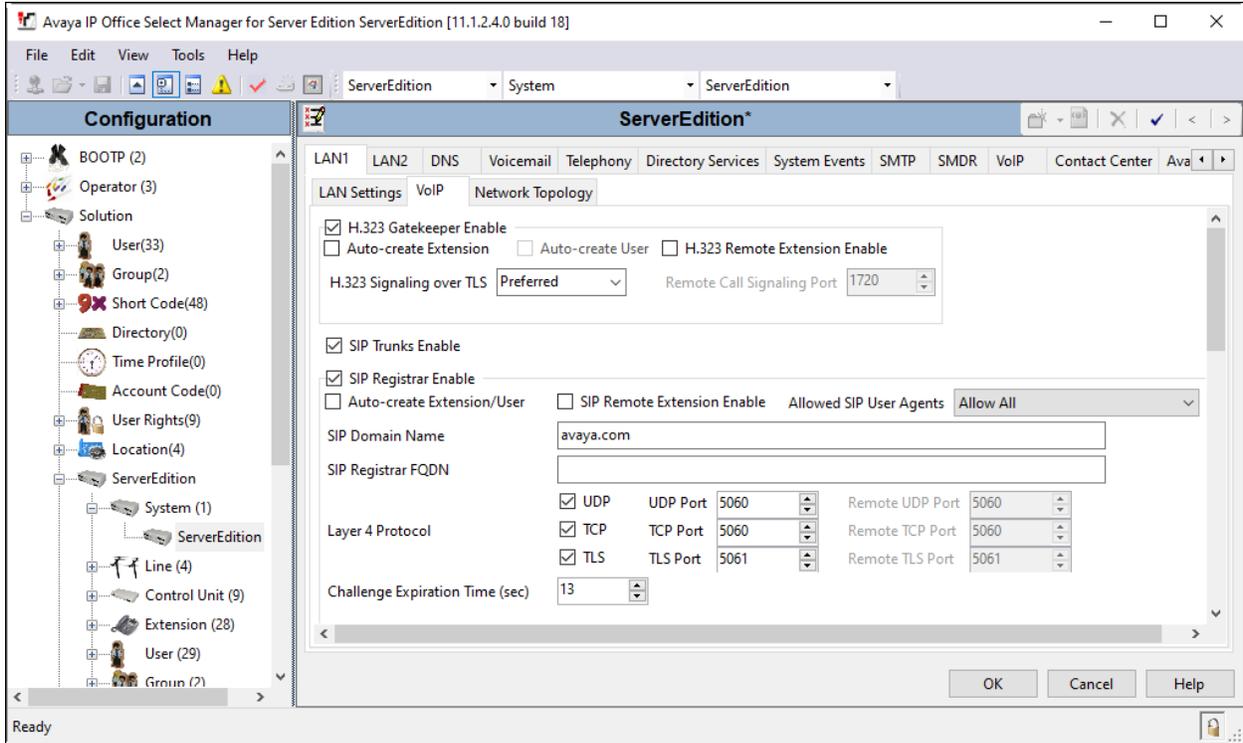
5.1. Obtain LAN IP Address

From a PC running the IP Office Manager application, on the configuration tree in the left pane, select **System** to display the **System** screen for the IP Office Server Edition 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 IX-DB.



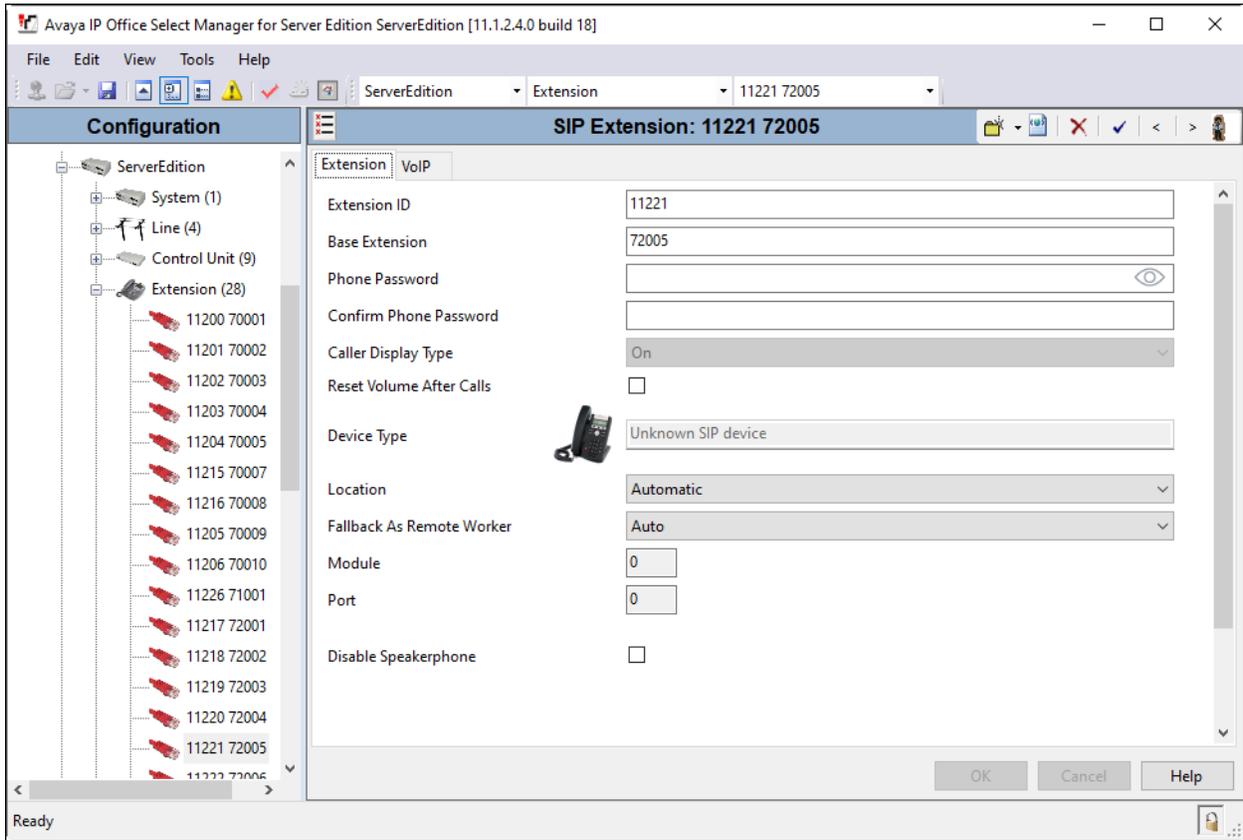
5.2. Administer SIP Registrar

Select the **VoIP** sub-tab. Ensure that **SIP Registrar Enable** is checked and enter a valid **SIP Domain Name**. In the compliance testing, the **SIP Domain Name** field was set to *avaya.com*. UDP transport protocol was enabled for the **Layer 4 Protocol**, which was used by IX-DB.

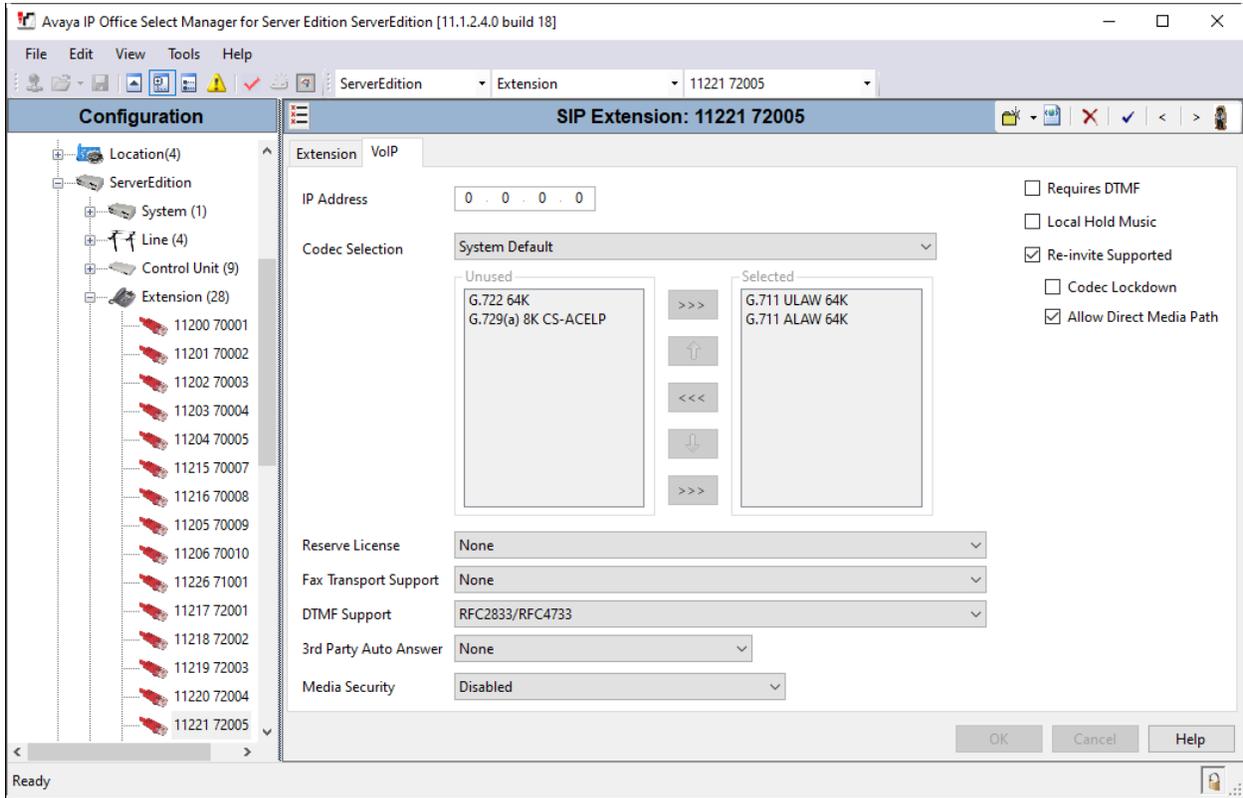


5.3. Administer SIP Extension for IX-DB

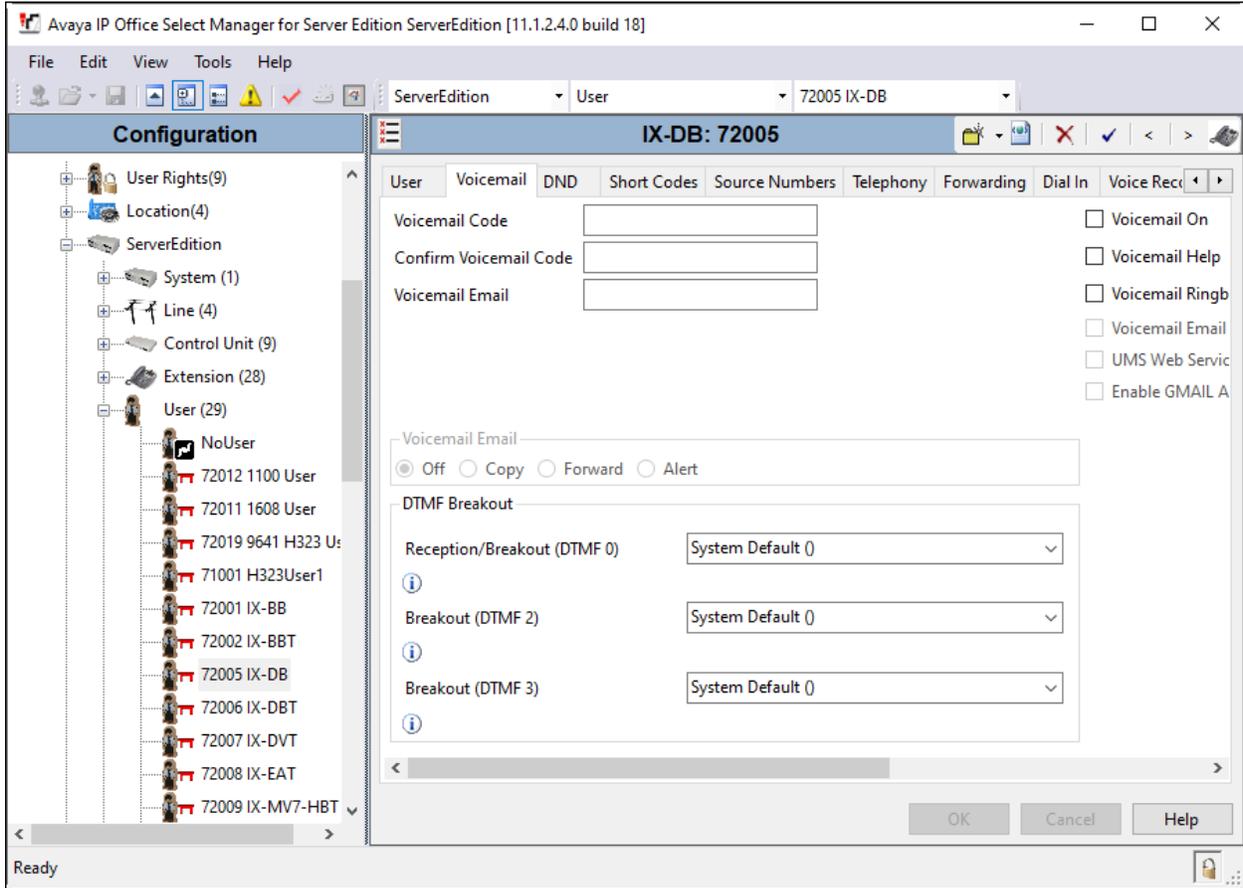
From the configuration tree in the left pane, right-click on **Extension** and select **New → SIP** from the pop-up list to add a new SIP extension. Enter the desired extension for the **Base Extension** field as shown below. In this example, IX-DB was assigned extension 72005. This is the extension that IX-DB will use to register with IP Office Server Edition.



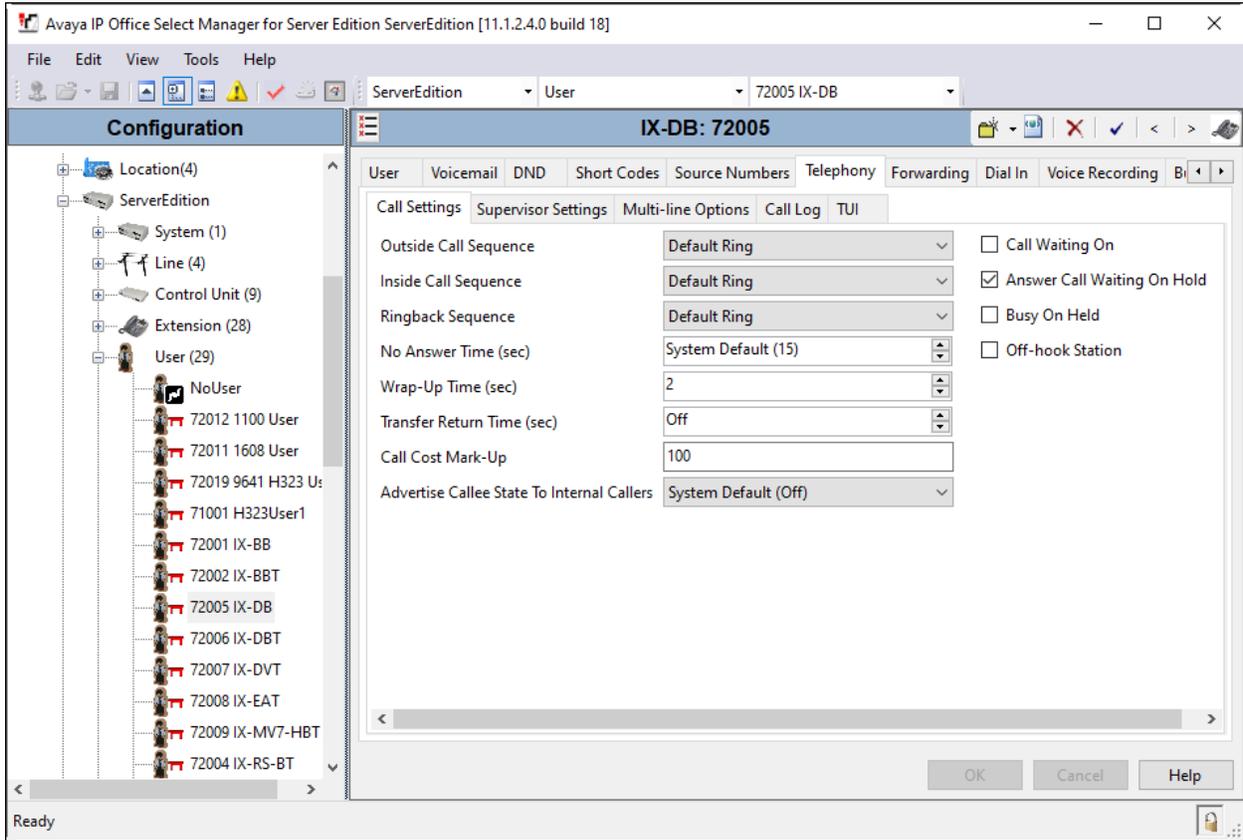
Select the **VoIP** tab and retain the default values. During the compliance test, IX-DB was tested with *G.711 ULaw* codec. Enable **Allow Direct Media Path** so that audio/RTP flows directly between two SIP endpoints without using media resources in Avaya IP Office Server Edition. **Media Security** was *disabled* for IX-DB.



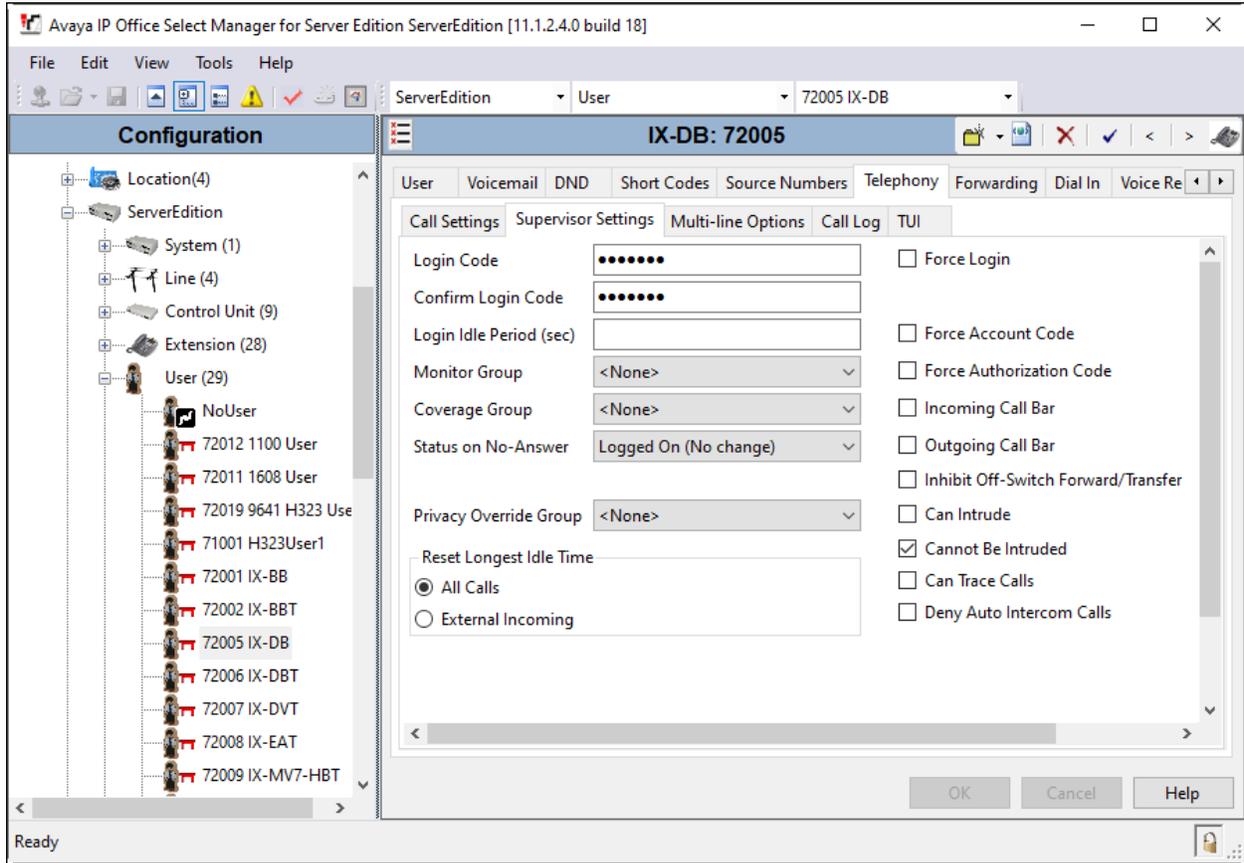
Select the **Voicemail** tab and disable voicemail for IX-DB.



Select the **Telephony** tab followed by the **Call Settings** sub-tab. Note the settings below for the user.



Select the **Supervisor Settings** sub-tab and enter a desired **Login Code**. The **Login Code** is the password that will be used by IX-DB to register with IP Office Server Edition.



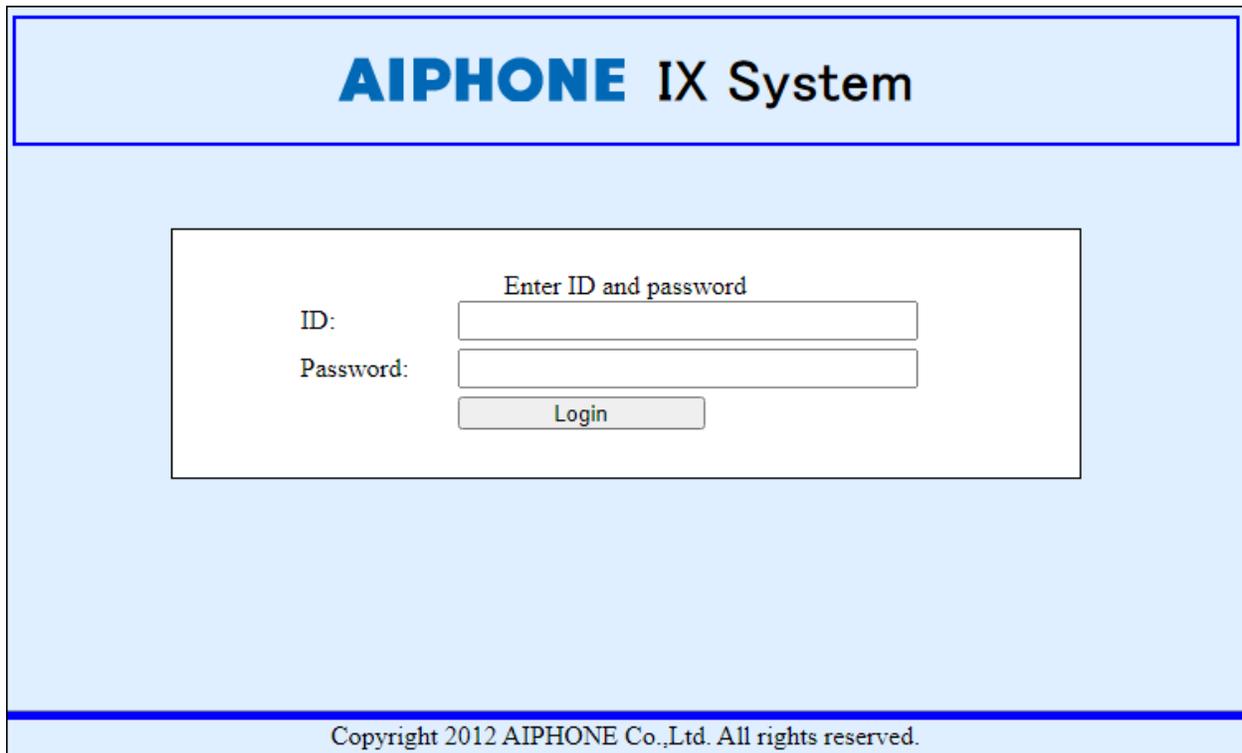
6. Configure Aiphone IX-DB Video Door Station

This section provides the procedure for configuring IX-DB to provide SIP connectivity to IP Office. Configuration of IX-DB is performed via Aiphone IX System web interface. The following configuration is covered:

- Log into Aiphone IX System Web Interface
- Administer Station Information
- Administer SIP Parameters
- Administer Video SIP Channel
- Administer Audio Settings
- Administer Call Settings

6.1. Log into Aiphone IX System Web Interface

Access the Aiphone IX System Web Interface by using the URL <https://<ip-address>/webset.cgi?login> in an Internet browser, where <ip-address> is the IX-DB IP address. Select language (not shown) and log in using the appropriate credentials.



AIPHONE IX System

Enter ID and password

ID:

Password:

Login

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6.2. Administer Station Information

Navigate to **Station Information** → **Identification** and set the **Number** to the IX-DB SIP extension (e.g., 72005). Input an appropriate **Name**.

The screenshot displays the AIPHONE IX System Setting web interface for a Video Door Station. The page is titled "Station Information" and features a sidebar on the left with navigation links: "Station Information", "Identification", "ID and Password", "Language", "Time", "Network Settings", "IP Address", "DNS", and "SIP". The main content area is titled "Station Information" and contains a sub-section for "Identification". The "Identification" section includes three fields: "Number" (set to 72005, with a red note "3-5 digits"), "Name" (set to IX-DB, with a red note "1-24 alphanumeric characters"), and "Location" (empty, with a red note "1-24 alphanumeric characters"). An "Update" button is located in the top right corner of the interface.

6.3. Administer SIP Parameters

Navigate to **Network Settings** → **SIP** from the left pane and configure the following parameters:

- **SIP Signaling Port:** Set to *5060*.
- **User Agent:** Enter desired value (e.g., *IX-DB*).
- **ID:** Set to SIP extension (e.g., *72005*) from **Section 5.3**.
- **Password:** Enter SIP password from **Section 5.4**.
- **IPv4 Address:** Set to signaling IP address of IP Office (e.g., *10.64.110.65*).
- **Port:** Set to *5060*.

Click **Update** to save changes.

The screenshot displays the 'AIPHONE IX System Setting' interface. The top navigation bar includes a keypad and an 'Update' button. The left sidebar lists various settings categories: Station Information, Network Settings, System Information, and Contact / Audio Output Settings. The main content area is titled 'Network Settings' and contains a 'SIP' section. Under 'SIP Connections', the 'SIP Signaling Port' is set to 5060 and the 'User Agent' is IX-DB. The 'SIP Server' section shows 'SIP Compatibility Mode' set to Standard Mode. The 'Primary Server' section includes fields for ID (72005), Password (masked), IPv4 Address (10.64.110.65), IPv6 Address (empty), and Port (5060). Red text next to several fields indicates character limits or constraints.

Parameter	Value	Constraint
SIP Signaling Port	5060	1-65535
User Agent	IX-DB	1-36 alphanumeric characters
SIP Compatibility Mode	Standard Mode	
Primary Server ID	72005	1-24 alphanumeric characters
Primary Server Password	*****	1-24 alphanumeric characters
Primary Server IPv4 Address	10.64.110.65	1.0.0.0-223.255.255.255
Primary Server IPv6 Address		::FF:0-FFFF:FFFF:FFFF:FFFF:FFFF:FFFF:FFFF
Primary Server Port	5060	1-65535

6.4. Administer Video SIP Channel

Navigate to **Network Settings** → **Video** in the left pane and configure the video settings as shown below.

The screenshot displays the AIPHONE IX System Setting interface. At the top, it shows 'Station Type: Video Door Station' and an 'Update' button. The main content area is titled 'Network Settings' and contains a 'Video' section. This section includes a warning about coding systems and RTP ports, followed by a 'SIP Channel' configuration table.

SIP Channel		
Coding System:	<input type="text" value="H.264 / AVC"/>	
Resolution:	<input type="text" value="320x240 (QVGA)"/>	
Frame Rate [fps]:	<input type="text" value="15"/>	
Select Profile [H.264 / AVC]:	<input type="text" value="High"/>	
I-picture interval [H.264 / AVC]: ♦	<input type="text" value="15"/>	1-100
Bit rate [kbps] [H.264 / AVC]:	<input type="text" value="512"/>	
Select Quality [Motion-JPEG]:	<input type="text" value="6"/>	
RTP Start Port: ♦	<input type="text" value="30000"/>	1-65534
RTP End Port: ♦	<input type="text" value="31000"/>	1-65535

6.5. Administer Audio Settings

Navigate to **Network Settings** → **Audio** in the left pane and set **Audio Codec** to select *G.711 (u-law)*.

The screenshot displays the 'AIPHONE IX System Setting' web interface. The top navigation bar includes a keypad and an 'Update' button. The left sidebar contains a menu with categories: Station Information, Network Settings, System Information, Contact / Audio Output Settings, Call Settings, and Function Settings. The 'Audio' option under 'Network Settings' is selected. The main content area, titled 'Network Settings', shows the 'Audio' configuration page. It includes a warning at the top: 'The 'Sip Channel' RTP End Port should be greater than 210 digits from the RTP Start Port. The 'ONVIF Transmit Channel RTP End Port should be greater than 10 digits from the RTP Start Port.' The 'Audio CODEC' is set to 'G.711(μ-law)'. Other settings include 'Audio RTP Transmission Interval [msec]' (20), 'RTP Idle Detection Time [sec]' (10), 'SIP Channel' (RTP Start Port: 20000, RTP End Port: 21000), 'ONVIF Transmit Channel' (RTP Start Port: 22000, RTP End Port: 23000), and 'Audio Buffer' (Packets Buffered at Audio Start: 1, Maximum Packets Buffered: 3).

Station Type: Video Door Station

Network Settings

- Audio**

The 'Sip Channel' RTP End Port should be greater than 210 digits from the RTP Start Port.
The 'ONVIF Transmit Channel RTP End Port should be greater than 10 digits from the RTP Start Port.

Audio CODEC: G.711(μ-law) G.711(A-law)

Audio RTP Transmission Interval [msec]: This setting is ignored when transmitting to multiple stations (paging, etc.)

RTP Idle Detection Time [sec]: 10-180sec

SIP Channel

RTP Start Port: 1-65534

RTP End Port: 1-65535

ONVIF Transmit Channel

RTP Start Port: 1-65534

RTP End Port: 1-65535

Audio Buffer

Packets Buffered at Audio Start:

Maximum Packets Buffered: Maximum Packet Buffer must be larger than Audio Start Buffer.

6.6. Administer Call Settings

Navigate to **Call Settings** → **Called Stations** in the left pane. In the **Called Stations** section, add an entry that specifies the number that should be dialed when the call button is pressed. Set the **Station Number** to the called number (e.g., 72015), set the **IPv4 Address** to the signaling IP address of IP Office (e.g., 10.64.110.65), and set **Protocol** to *U*.

Station Information

- Identification
 - ID and Password
 - Language
 - Time
- Network Settings**
 - IP Address
 - DNS
 - SIP
 - Multicast Address
 - Video
 - Audio
 - Packet Priority
 - NTP
- System Information**
 - Custom Sound Registry
- Contact / Audio**
 - Contact Input
 - Output Specifications
- Call Settings**
 - Called Stations

Call Settings

•Called Stations

Call Button Assignment

U = Unicast, M = Multicast
If designating "M", multicast IP addresses must be configured for the station(s).

	Number 3-32 digits	IPv4 Address 1.0.0.0-223.255.255.255	IPv6 Address 2000::0- 3FFF:FFFF:FFFF:FFFF:FFFF:FFFF or FD00::0- FDFE:FFFF:FFFF:FFFF:FFFF:FFFF	Protocol
1	72015	10 64 110 65		U ▼
2				▼
3				▼
4				▼
5				▼
6				▼
7				▼
8				▼
9				▼

7. Verification Steps

This section provides the tests that can be performed to verify proper configuration of IP Office and Aiphone IX-DB Video Door Station.

1. Verify that IX-DB has successfully registered with with IP Office. Launch **IP Office System Status** and navigate to **Extensions** → **<SIP Extension>**, where **<SIP Extension>** is the IX-DB extension. Verify that the **Current State** is **Idle** as shown below.

The screenshot shows the Avaya IP Office System Status application window. The title bar reads "Avaya IP Office System Status - ServerEdition (10.64.110.65) - IP Office Linux PC 11.1.2.4.0 build 18". The main window has a blue header with the AVAYA logo and the title "IP Office System Status". Below the header is a menu bar with "Help", "Snapshot", "LogOff", "Exit", and "About". A left-hand navigation pane contains a tree view with categories: System, Alarms (2), Extensions (8), Trunks (4), Active Calls, Resources, Voicemail, IP Networking, and Locations. The "Extensions (8)" category is expanded, and extension "72005" is selected. The main content area displays the "Extension Status" for 72005. The status is "Idle". Below the status information is a table with columns: Call Ref, Current State, Time in State, Calling Number or Called Number, Direction, and Other Party on Call. The table shows one row with "Idle" state and "00:28:28" time in state. At the bottom of the window, there are buttons for "Trace", "Trace All", "Pause", "Ping", "Call Details", "Print...", and "Save As...". The system time is "3:44:55 PM" and the status is "Online".

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
	Idle	00:28:28			

2. Establish inbound and outbound video calls to IX-DB with Workplace Client and/or Vantage endpoints and verify two-way audio and one-way video.

8. Conclusion

These Application Notes describe the administration steps required to integrate Aiphone IX Series 2 Video Door Stations (IX-DB) with Avaya IP Office Server Edition. The Aiphone IX-DB Video Door Station successfully registered with IP Office as a SIP endpoint and audio and video calls were verified. All test cases executed passed with observations as noted in **Section 2.2**.

9. References

This section references the Avaya and Aiphone documentation relevant to these Application Notes.

Avaya product documentation is available at <https://support.avaya.com>.

[1] *Administering Avaya IP Office using Manager*, Release 11.1, available at <http://support.avaya.com> as an HTML document.

Aiphone product documentation is available at <https://www.aiphone.com>.

[2] *Aiphone IX Door Stations Web Setting Manual*, Software version 6.00 or later, available from Aiphone.

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