



DevConnect Program

Application Notes for Aiphone IX Series 2 Video Door Station (IX-DVT) 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-DVT) Version 7.00 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500V2 Expansion System 11.1. The Aiphone IX-DVT Video Door Station, which is part of the Aiphone IX Series 2 Video Door Stations, was used for the compliance test. Aiphone IX-DVT Video Door Station is a surface mount, weather resistant video door station. It has one dry contact that can be used to release doors when activated by a phone. Aiphone IX-DVT 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-DVT) Version 7.00 with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500V2 Expansion System 11.1. The Aiphone IX-DVT Video Door Station, which is part of the Aiphone IX Series 2 Video Door Stations, was used for the compliance test. Aiphone IX-DVT Video Door Station is a surface mount, weather resistant video door station. It has one dry contact that can be used to release doors when activated by a phone. Aiphone IX-DVT Video Door Station (IX-DVT) 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-DVT 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-DVT 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-DVT 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-DVT with IP Office Server Edition and IP Office 500V2 Expansion System.
- Audio calls between IX-DVT and Avaya SIP and H.323 deskphones with Direct IP Media (Shuffling) enabled and disabled.
- Audio and video calls between IX-DVT, Workplace, and Vantage K175 with Direct IP Media (Shuffling) enabled and disabled. One-way video from IX-DVT to Workplace and Vantage K175 was verified.
- Audio calls between IX-DVT and the PSTN.
- G.711 codec support.
- UDP transport protocol.
- IX-DVT 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-DVT.

2.2. Test Results

All test cases executed passed successfully.

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-DVT Video Door Station registered to either IP Office Server Edition or IP Office 500 V2 Expansion System (not simultaneously).

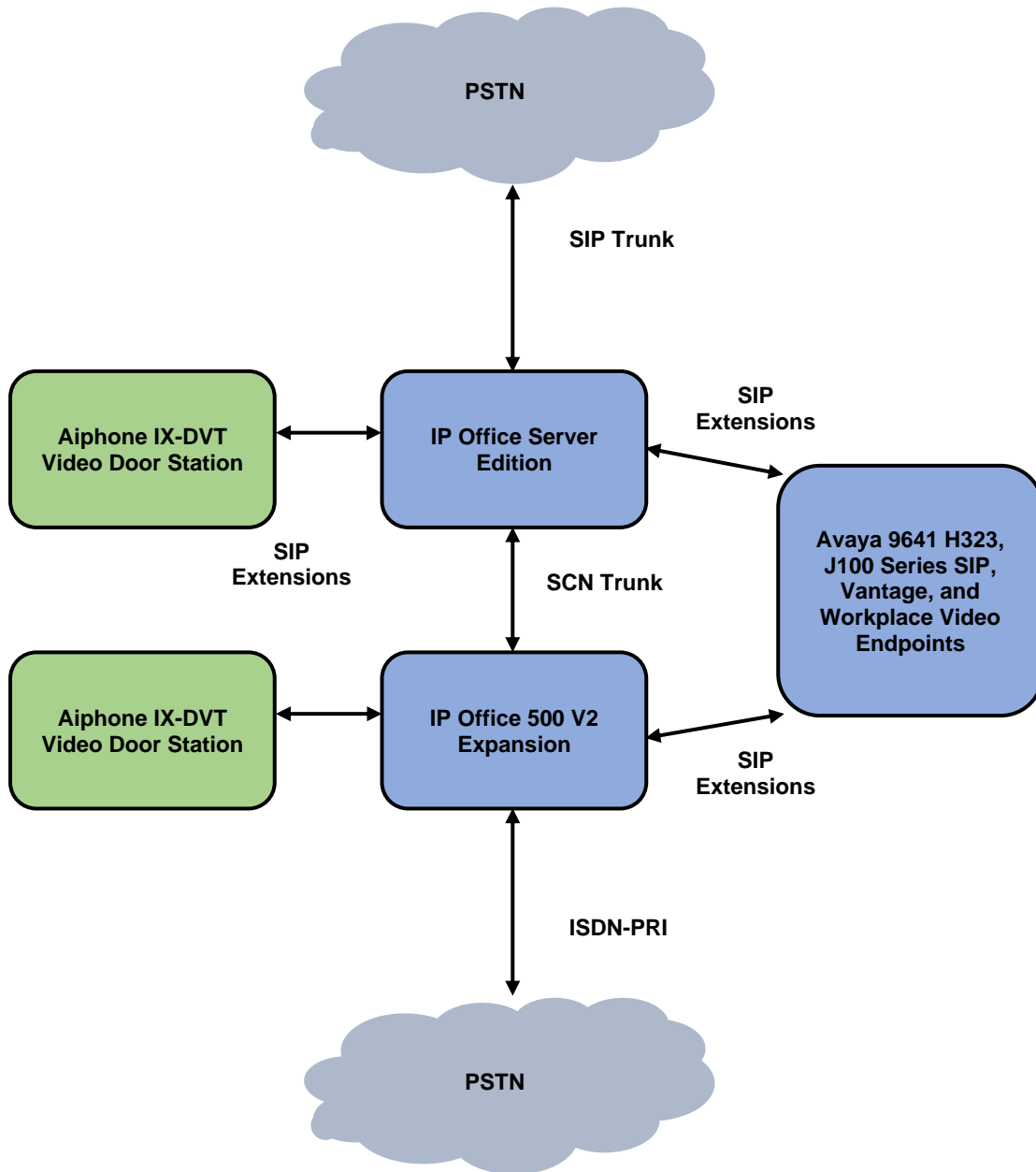


Figure 1: Avaya SIP Telephony Network with Aiphone IX-DVT Video Door Station

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-DVT 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

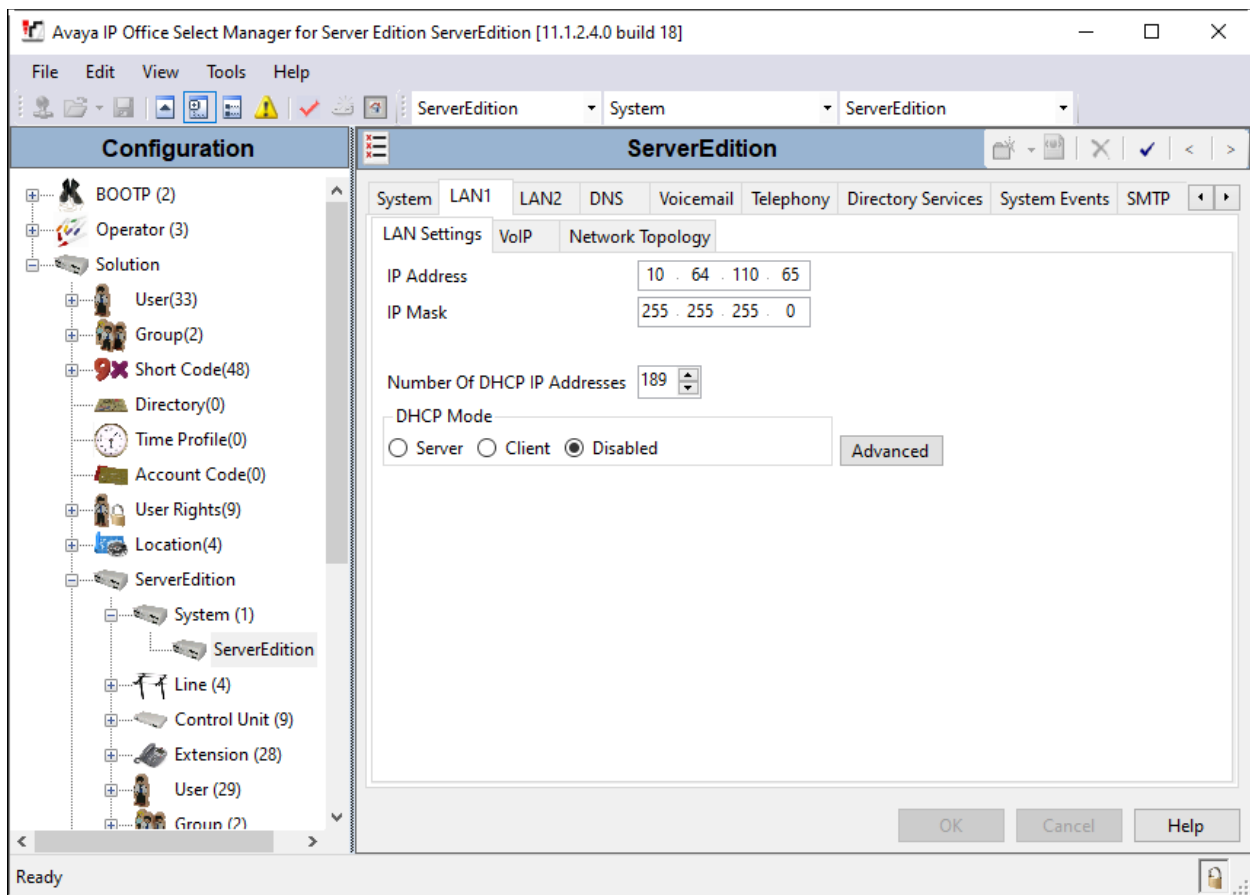
This section provides the procedures for configuring Avaya IP Office Server Edition. The procedures include the following areas:

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

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.

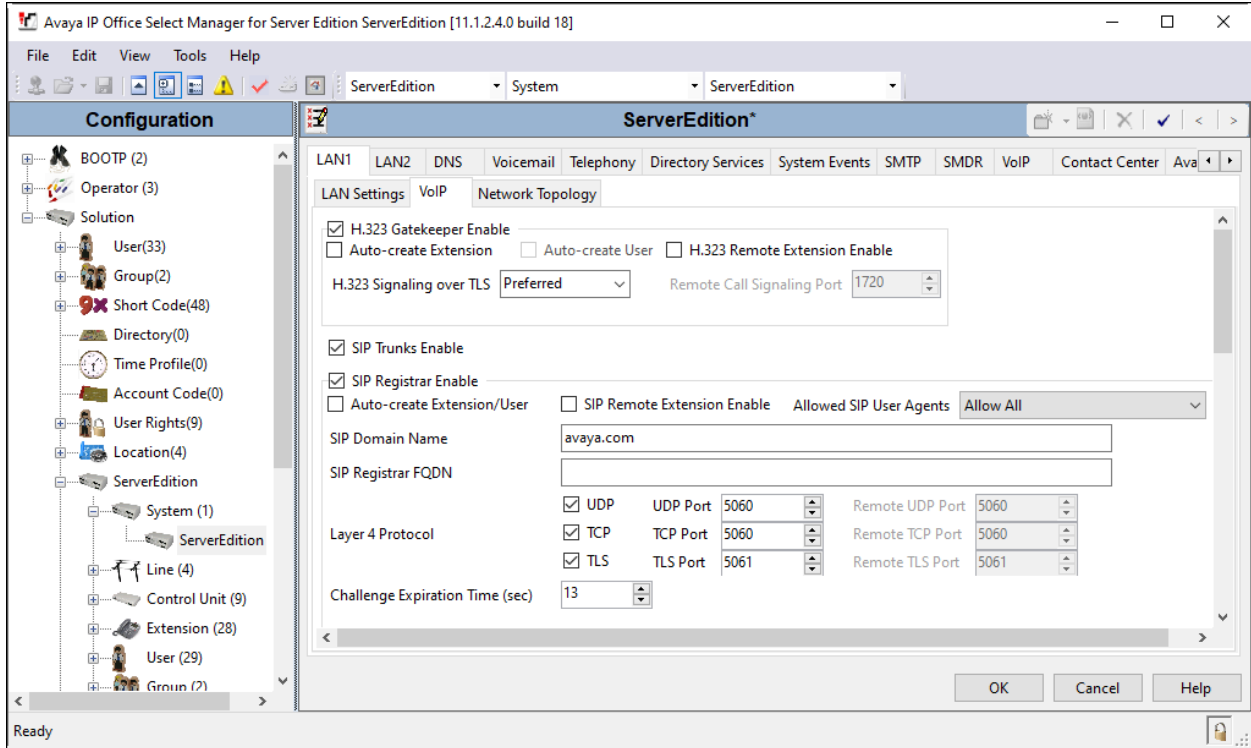
5.1. Obtain LAN IP Address

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-DVT.



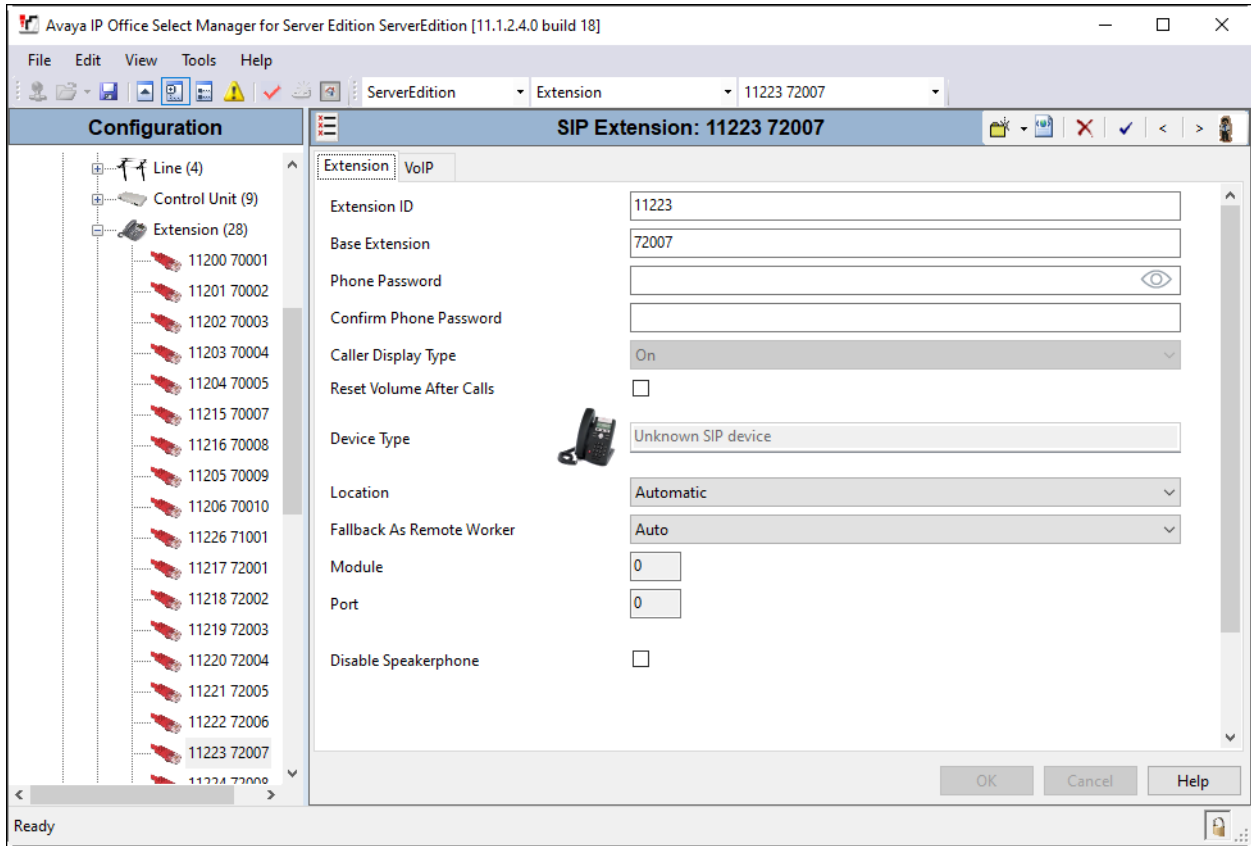
5.2. Administer SIP Registrar

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-DVT.

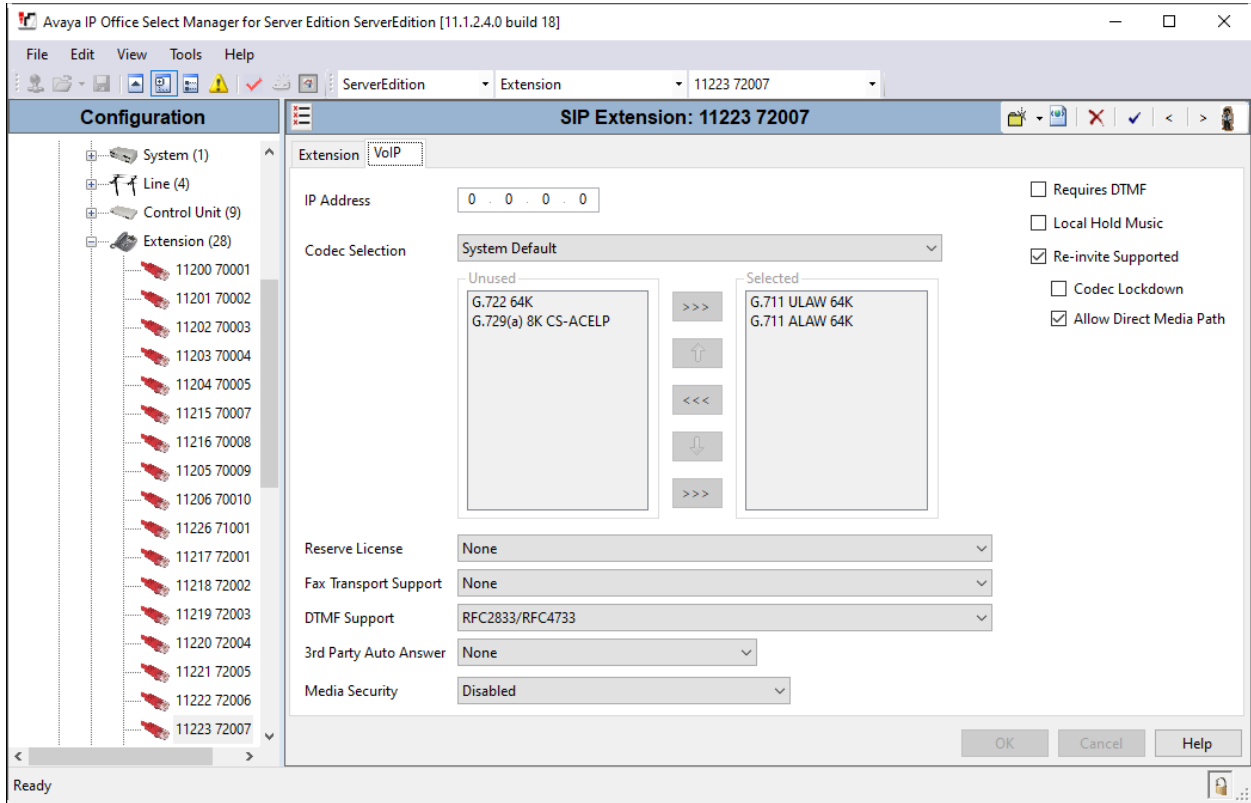


5.3. Administer SIP Extension for IX-DVT

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-DVT was assigned extension 72007. This is the extension that IX-DVT will use to register with IP Office Server Edition.

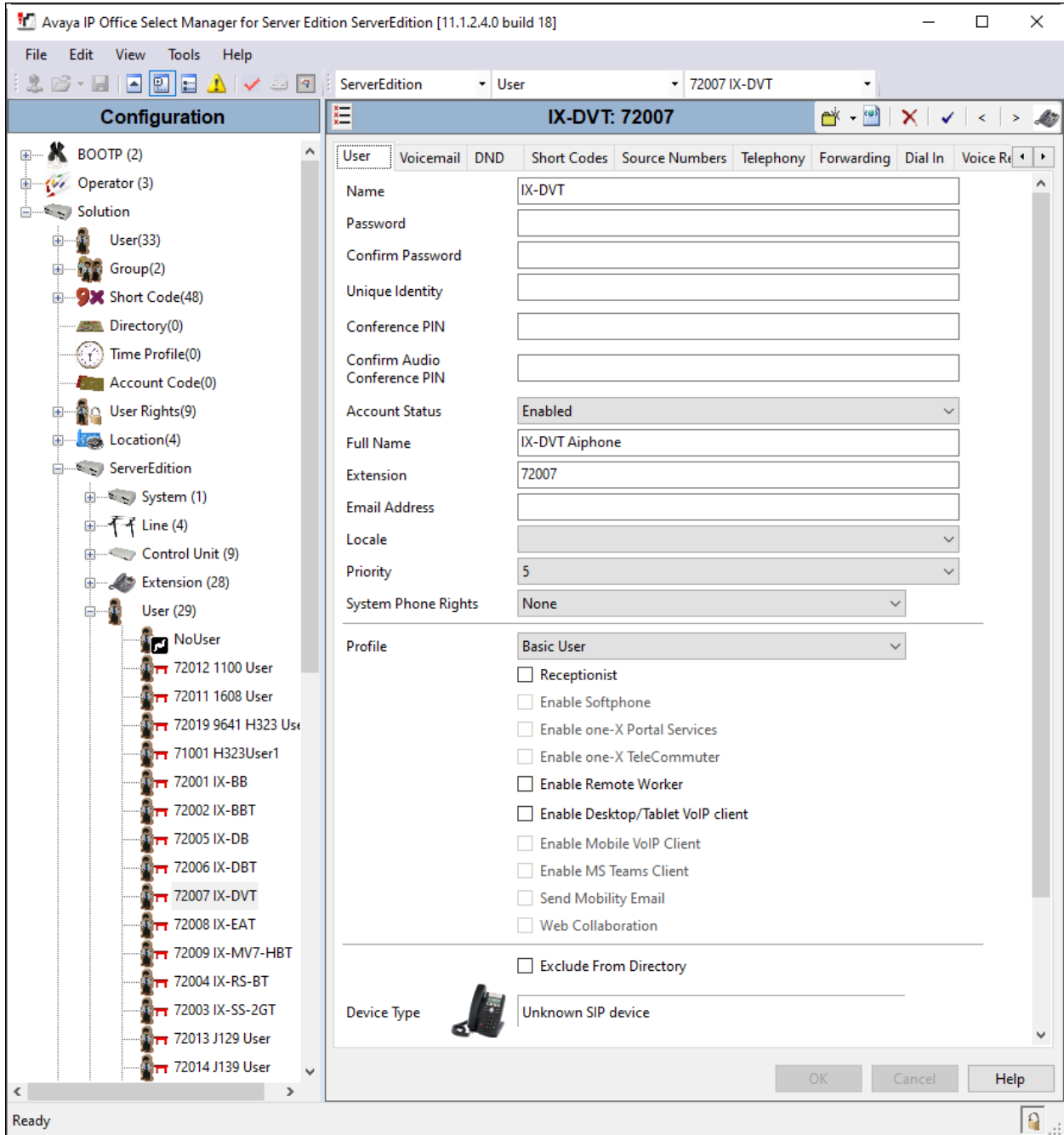


Select the **VoIP** tab and retain the default values. During the compliance test, IX-DVT was tested with *G.711 ULaw* codec. Enable **Allow Direct Media Path** so that audio/RTP flows directly between two SIP endpoints without using media resources in Avaya IP Office Server Edition. **Media Security**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

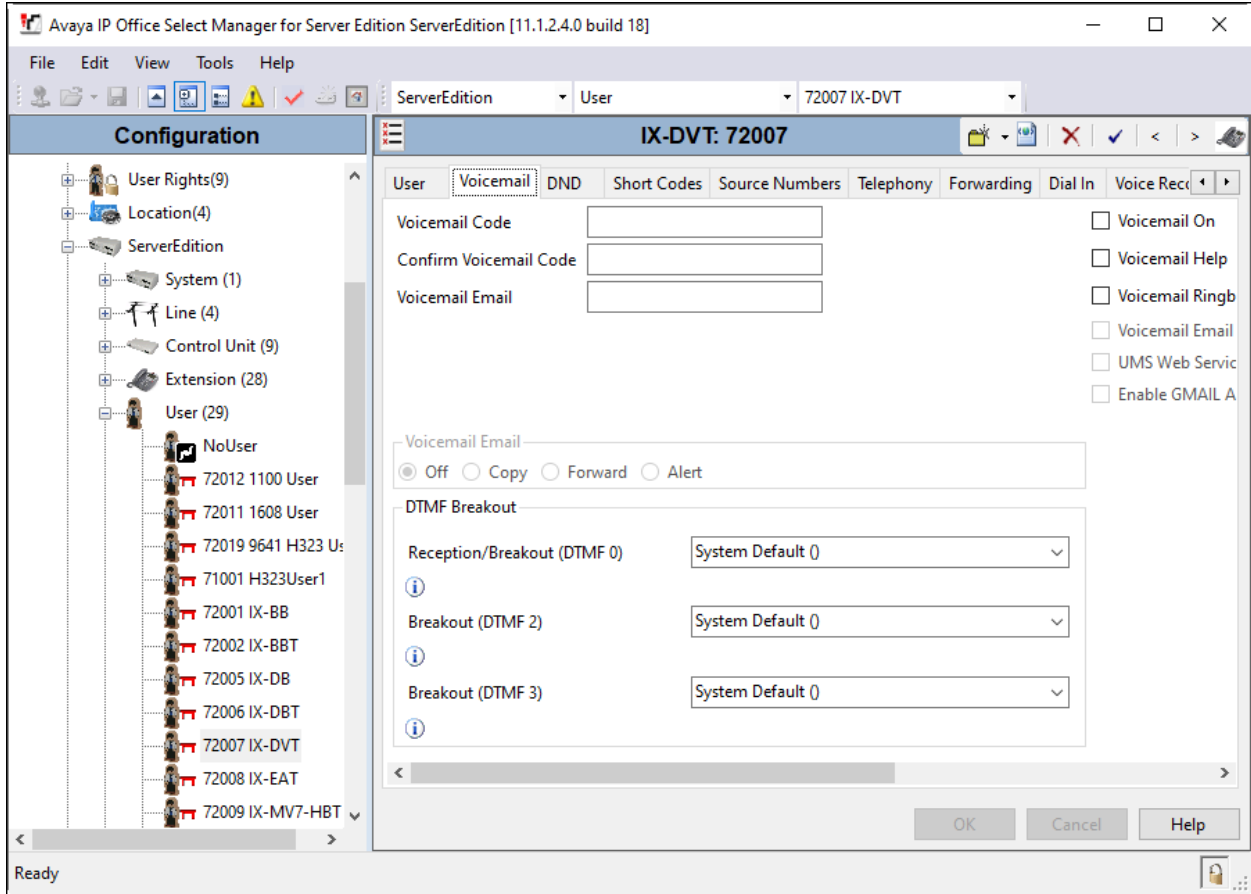


5.4. Administer SIP User for IX-DVT

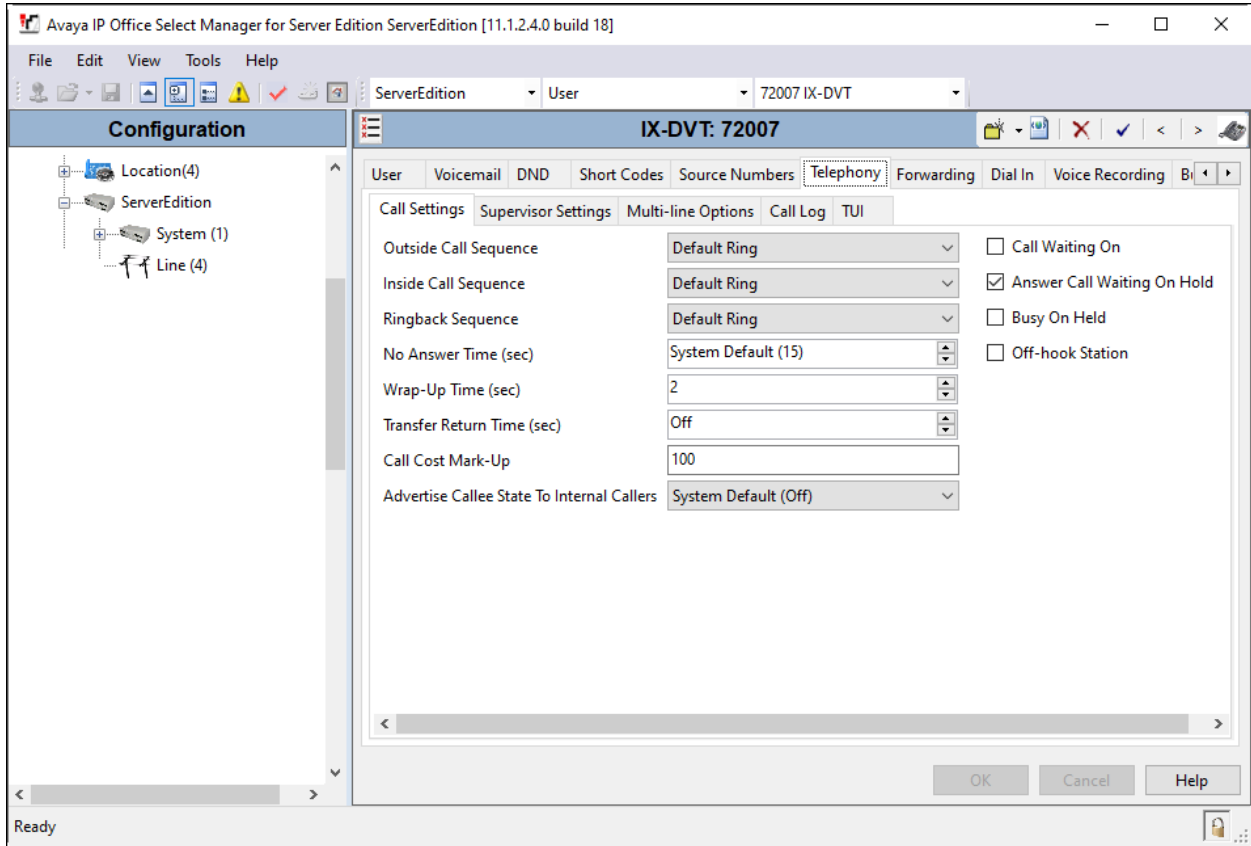
From the configuration tree in the left pane, right-click on **User** and select **New** from the pop-up list. Enter desired values for the **Name** and **Full Name** fields. For the **Extension** field, enter the SIP extension from **Section 5.3** (e.g., 72007).



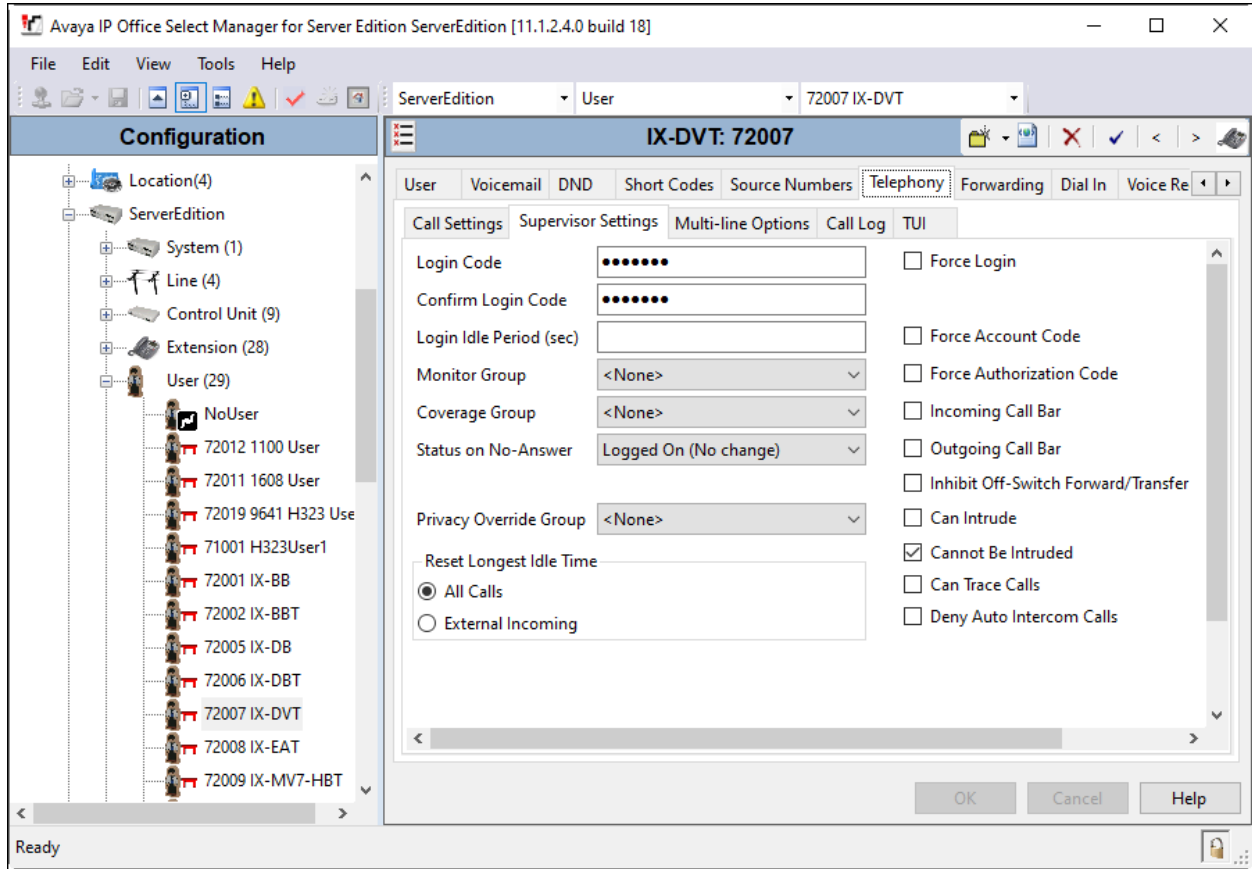
Select the **Voicemail** tab and disable voicemail for IX-DVT.



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-DVT to register with IP Office Server Edition.



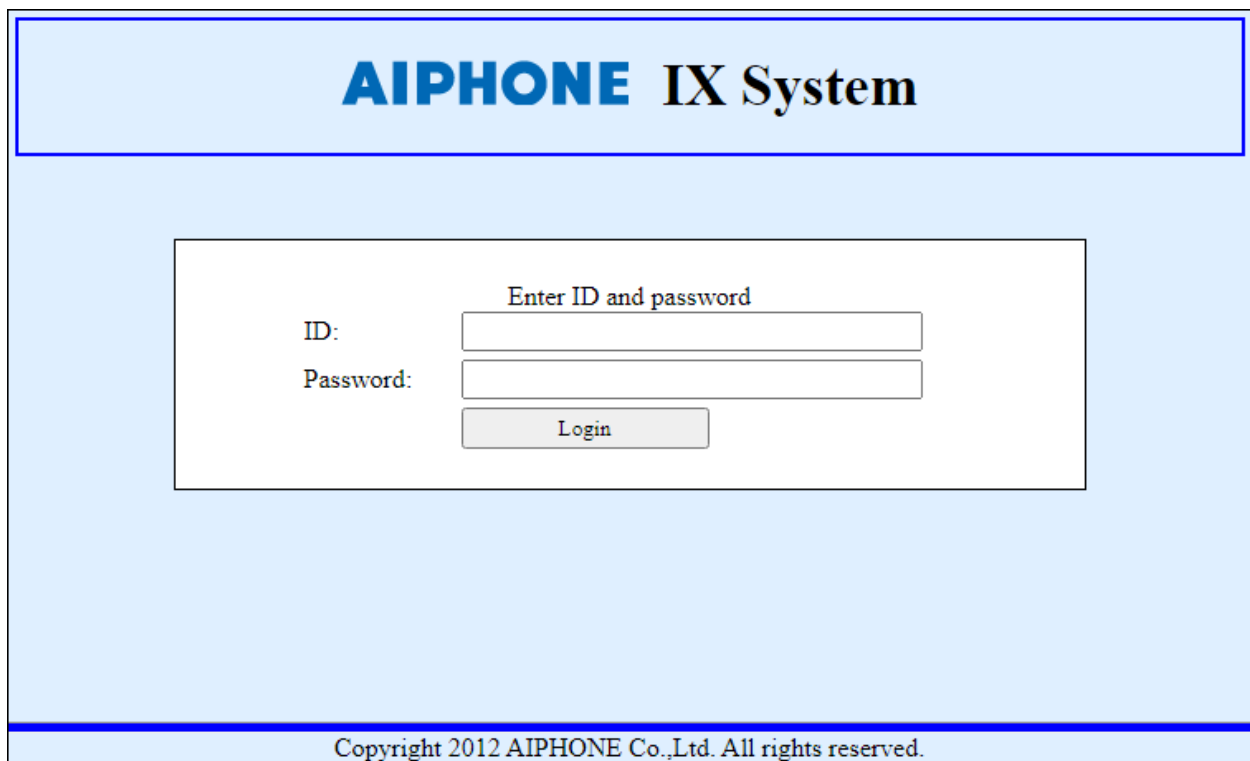
6. Configure Aiphone IX-DVT Video Door Station

This section provides the procedure for configuring IX-DVT to provide SIP connectivity to IP Office. Configuration of IX-DVT 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-DVT 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-DVT SIP extension (e.g., 72007). Input an appropriate **Name**.

The screenshot shows the AIPHONE IX System Setting web interface. At the top, it says "AIPHONE IX System Setting" with an "Update" button. Below that, it indicates "Category: Video Stations" and "Station Type: IX-DVT". The main content area is titled "Station Information" and has a sub-section for "Identification".

Number	<input type="text" value="72007"/>	3-5 digits
Name	<input type="text" value="IX-DVT"/>	1-24 alphanumeric characters(*1)
Location	<input type="text"/>	1-24 alphanumeric characters(*1)

(*1)Certain characters may not be displayed correctly on IX-MV.

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. The top header shows 'AIPHONE IX System Setting' and an 'Update' button. Below the header, the 'Category' is 'Video Stations' and the 'Station Type' is 'IX-DVT'. The left sidebar contains a menu with 'Station Information' (including Identification, ID and Password, Language, Time, and Expanded System) and 'Network Settings' (including IP Address, DNS, SIP, Multicast Address, Video, Audio, Packet Priority, and NTP). The main content area is titled 'Network Settings' and features a 'Video' section. Under 'SIP Channel', there is a warning: 'The "SIP Channel" RTP End Port should be greater than 90 digits from the RTP Start Port.' The configuration options are: Resolution (radio buttons for 320x240(QVGA) and 640x480(VGA), with 640x480(VGA) selected), Frame Rate [fps] (dropdown menu set to 15), Select Profile (dropdown menu set to High), I-picture interval (input field set to 15, with a range of 1-100), Bit rate [kbps] (dropdown menu set to 1024), RTP Start Port (input field set to 30000, with a range of 1-65534), and RTP End Port (input field set to 31000, with a range of 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 main content area is titled "Network Settings" and is divided into several sections:

- Audio**: This section contains three radio buttons for selecting the audio codec: G.711(u-law), G.711(A-law), and G.722. Below these are two input fields: "Audio RTP Transmission Interval [msec]" set to 20 and "RTP Idle Detection Time [sec]" set to 10. A red note states: "This setting is ignored when transmitting to multiple stations (paging, etc.) 10-180 sec".
- SIP Channel**: This section has two rows of input fields. The first row is for "RTP Start Port" (20000) and "RTP End Port" (21000), with a range of 1-65534. The second row is for "RTP Start Port" (21000) and "RTP End Port" (22000), also with a range of 1-65534.
- ONVIF Transmit Channel**: This section has two rows of input fields. The first row is for "RTP Start Port" (22000) and "RTP End Port" (23000), with a range of 1-65534. The second row is for "RTP Start Port" (23000) and "RTP End Port" (24000), also with a range of 1-65534.
- Audio Buffer**: This section contains two input fields: "Packets Buffered at Audio Start" set to 1 and "Maximum Packets Buffered" set to 3. A red note states: "Maximum Packet Buffer must be larger than Audio Start Buffer".

The left sidebar contains a navigation menu with the following categories and links:

- Station Information**: Identification, ID and Password, Language, Time, Expanded System
- Network Settings**: IP Address, DNS, SIP, Multicast Address, Video, Audio, Packet Priority, NTP
- Call Settings**: Station Settings, Called Stations (for Door), Call Origination, Incoming Call
- Option Input / Relay**: Output Settings
- Function Settings**: Paging Settings, Email

At the top right of the interface, there is an "Update" button.

6.6. Administer Call Settings

Navigate to **Call Settings** in the left pane and set the **Call Button Function** to *Call, Answer Call, End Communication* in the **Station Information** section.

In the **Called Stations (for Door)** 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 **Station Type** to *VoIP Phone*. Only one VoIP phone may be specified.

AIPHONE IX System Setting
 Category: Video Stations Station Type: IX-DVT Update

Call Settings Required Settings

Station Information

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

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

#	Station Number	IPv4 Address	IPv6 Address	Station Type
1	72015	10.64.110.65		VoIP Phone
2				
3				

7. Verification Steps

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

1. Verify that IX-DVT has successfully registered with with IP Office. Launch **IP Office System Status** and navigate to **Extensions** → **<SIP Extension>**, where **<SIP Extension>** is the IX-DVT 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 "72007" is selected. The main content area displays the "Extension Status" for 72007. The status is "Idle". Below the status 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 "00:29:51" time. At the bottom of the window, there are buttons for "Trace", "Trace All", "Pause", "Ping", "Call Details", "Print...", and "Save As...". The bottom right corner shows the time "3:46:36 PM" and the status "Online".

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
	Idle	00:29:51			

2. Establish inbound and outbound video calls to IX-DVT with Avaya Workplace 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-DVT) with Avaya IP Office Server Edition. The Aiphone IX-DVT 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 no observations noted.

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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