



## **Application Notes for Valcom One-Way and Talkback IP Speakers with Avaya IP Office Server Edition - Issue 1.0**

### **Abstract**

These Application Notes describe the configuration steps required to integrate Valcom One-Way and Talkback IP Speakers running SW Rev 3.24.5 and SIP Rev sw1.60.38 with Avaya IP Office Server Edition 11.1. For this compliance test, Valcom VIP-130AL-GY IP Secure One-Way Paging Horn and VIP-160A IP Talkback 8" Ceiling Speaker were used. These Valcom IP speakers register with Avaya IP Office Server Edition as SIP endpoints.

Readers should pay attention to **Section 0**, 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 DevConnect Program at the Avaya Solution and Interoperability Test Lab.

# 1. Introduction

These Application Notes describe the configuration steps required to integrate Valcom One-Way and Talkback IP Speakers running SW Rev 3.24.5 and SIP Rev sw1.60.38 with Avaya IP Office Server Edition 11.1. Avaya IP Office consists of IP Office Server Edition running on a virtual platform as the primary server with an IP Office 500 V2 running as an expansion system. For this compliance test, Valcom VIP-130AL-GY IP Secure One-Way Paging Horn and VIP-160A IP Talkback 8" Ceiling Speaker were used. These Valcom IP speakers register with Avaya IP Office Server Edition as SIP endpoints.

The Valcom VIP-130AL-GY IP Secure One-Way Paging Horn is a self-contained paging system which enables paging over an IP network. When a call is placed to the Valcom One-Way IP Speaker, the device automatically answers the call and provides one-way communication to the device.

The Valcom VIP-160A IP Talkback 8" Ceiling Speaker supports both incoming and outgoing pages and hands-free two-way communication. When the call button is pressed on a Valcom Talkback IP Speaker, the device initiates a call to a preconfigured destination that resides on Avaya Aura® Communication Manager.

Valcom offers IP Ceiling Speakers, IP Wall Speakers, and IP Horns as different products/models to accommodate different environments. They share the same SIP stack and firmware version, therefore, this testing also applies to those products, as detailed in **Attachment 1. Section 4** of this document shows the actual products/models and SIP Stack and software versions that were tested. For additional details, contact Valcom Support, as noted in **Section 2.3**.

## 2. General Test Approach and Test Results

The interoperability compliance test included feature and serviceability testing. The feature testing focused on establishing calls between the Valcom One-Way and Talkback IP Speakers, Avaya SIP / H.323 IP Deskphones, and the PSTN. Two-way audio intercom calls with the Talkback IP Speaker and one-way audio paging calls with the One-Way IP Speaker were exercised. The serviceability testing focused on verifying that the Valcom IP speakers came back into service after a reboot.

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 Valcom One-Way and Talkback IP Speakers did not include use of any specific encryption features as requested by Valcom.

## 2.1. Interoperability Compliance Testing

Interoperability compliance testing covered the following features and functionality:

- SIP registration of IP speakers with IP Office Server Edition and IP Office 500 V2 Expansion.
- Establishing one-way audio paging calls from Avaya H.323 / SIP Deskphones and PSTN to the VIP-130AL Paging Horn.
- Establishing two-way audio intercom calls between VIP-160A Ceiling Speaker, Avaya H.323 / SIP Deskphones, and PSTN in both directions.
- Originating and terminating calls through Avaya SIP telephony network.
- Originating calls from the VIP-160A Ceiling Speaker to a predefined number using the call button.
- Terminating active calls by pressing the call button on the VIP-160A Ceiling Speaker.
- Support of G.711 mu-law and G.722 codecs.
- Support of TCP and UDP transport protocols.
- Support of direct IP-to-IP media, which allows IP endpoints to send audio RTP packets directly to each other without using media resources in IP Office.
- Proper recovery after a restart of IP speakers.

## 2.2. Test Results

All test cases passed with the following observations:

- Valcom One-Way and Talkback IP Speakers always advertise G.722 and G.711 (in that order) in the SIP SDP. The codec selection on the IP speaker only enforces the codec when the IP speaker initiates a call to a Valcom gateway (not covered by this compliance test).
- When an outgoing call from an IP speaker fails for whatever reason, such as invalid number, phone busy, or trunk calls blocked, or if the IP speaker doesn't register successfully via SIP, the IP speaker plays, "All circuits are busy at the present time."
- Standalone Valcom IP speakers do not support group paging calls using multicast audio without an optional Valcom controller.

## 2.3. Support

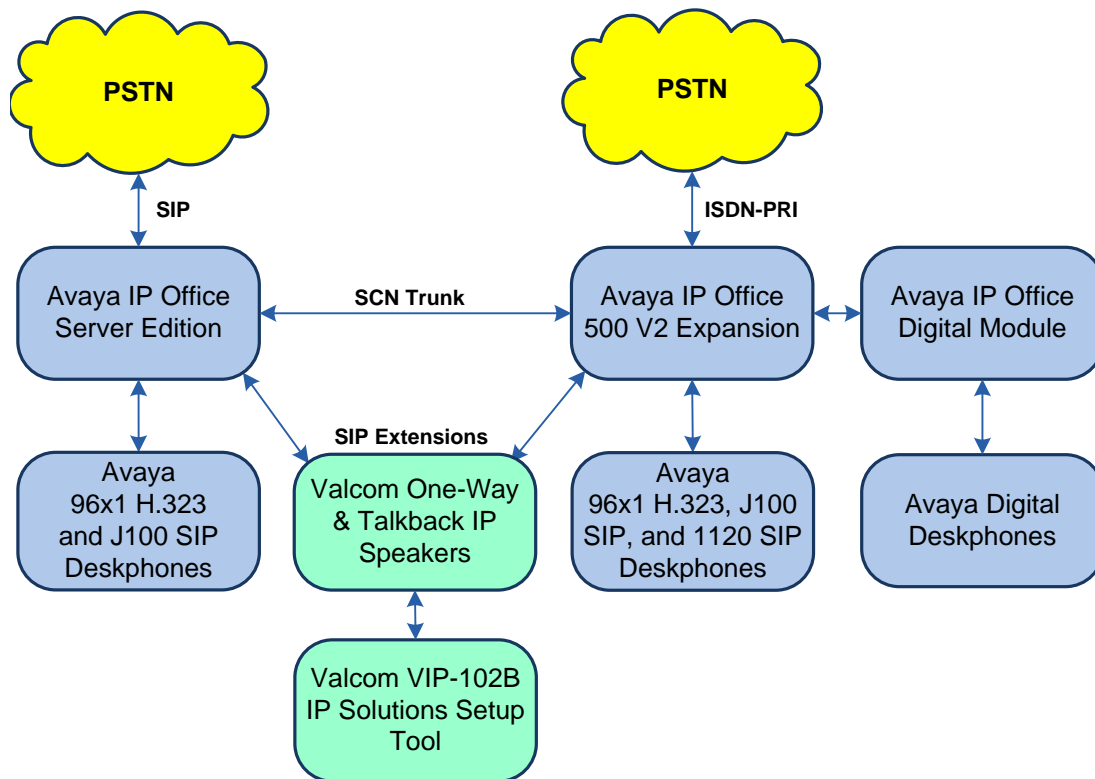
For technical support and information on Valcom One-Way and Talkback IP Speakers, contact Valcom Technical Support at:

- Phone: +1 (800) 825-2661 or +1 (540) 563-2000
- Website: <https://www.valcom.com/Support/techsupport.html>
- Email: [support@valcom.com](mailto:support@valcom.com)

### 3. Reference Configuration

**Figure 1** illustrates a sample configuration with Avaya IP Office that includes the following products:

- Avaya IP Office Server Edition with Avaya IP Office 500 V2 Expansion connected via a SCN trunk.
- Avaya IP Office Server Edition and Avaya IP Office 500 V2 Expansion acting as a Registrar/Proxy for SIP deskphones and Valcom One-Way and Talkback IP Speakers.
- Avaya IP Office Server Edition connected to the PSTN via a SIP trunk.
- Avaya IP Office 500 V2 Expansion connected to the PSTN via an ISDN-PRI trunk.
- Avaya IP Office 500 V2 Expansion with a digital module providing connectivity for digital phones.
- Avaya 96x1 Series H.323 Deskphones, J100 Series SIP Deskphones, and 1100 Series SIP Deskphones.
- Valcom IP speakers, including the Valcom VIP-130AL-GY IP Secure One-Way Paging Horn and the Valcom VIP-160A IP Talkback 8" Ceiling Speaker configured with the ValcomVIP-102B IP Solutions Setup Tool.



**Figure 1: Valcom One-Way and Talkback IP Speakers with Avaya IP Office Server Edition**

## 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	11.1.0.0.0 build 237
Avaya IP Office 500 V2 Expansion	11.1.0.0.0 build 237
Avaya 96x1 Series IP Deskphones	6.8304 (H.323)
Avaya J100 Series IP Deskphones	4.0.3.1.4 (SIP)
Avaya 1100/1200 Series IP Deskphones	04.04.26.00 (SIP)
Valcom VIP-130AL-GY IP Secure One-Way Paging Horn	Software Rev: 3.24.5 SIP Rev: sw1.60.38
Valcom VIP-160A IP Talkback 8" Ceiling Speaker	Software Rev: 3.24.5 SIP Rev: sw1.60.38
Valcom VIP-102B IP Solutions Setup Tool	8.1.0.0

## 5. Configure Avaya IP Office Server Edition

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

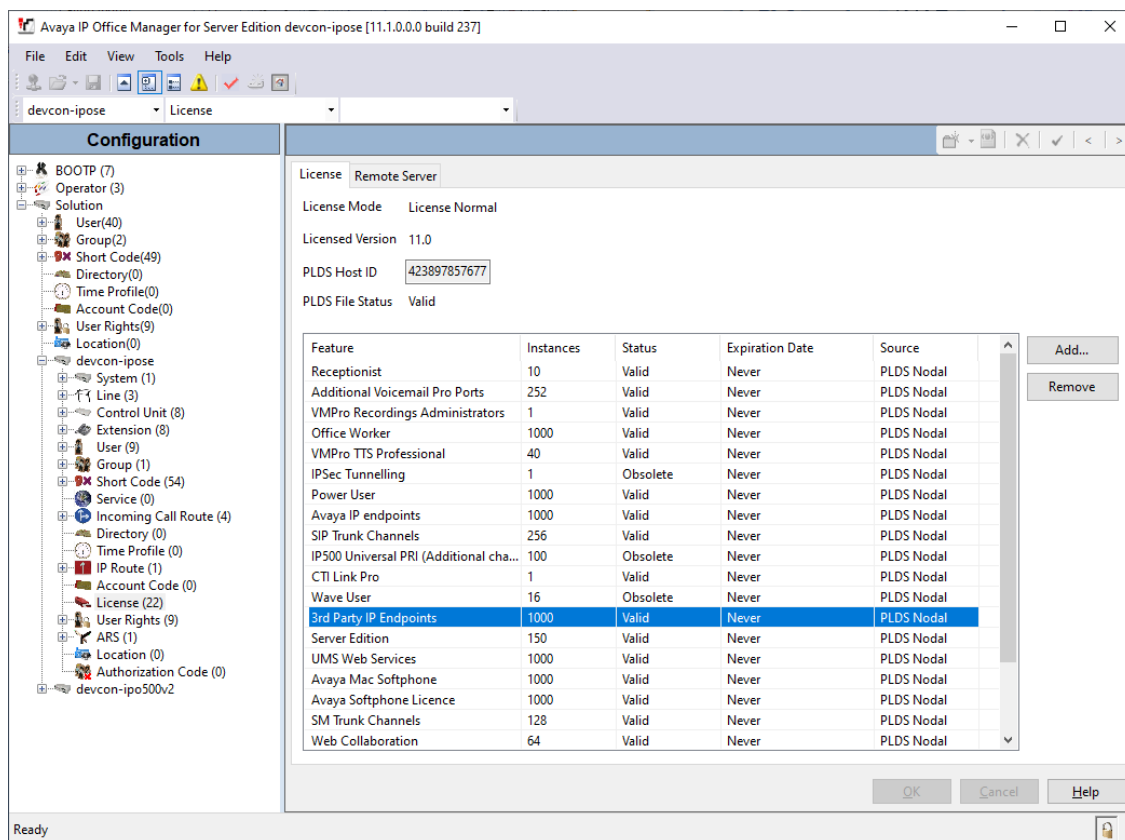
- Verify IP Office License
- Obtain LAN IP Address
- Administer SIP Registrar
- Administer SIP Extension
- Administer SIP User

**Note:** Integration of IP Office 500 V2 Expansion and call routing to the PSTN are outside the scope of these Application Notes.

### 5.1. Verify IP Office License

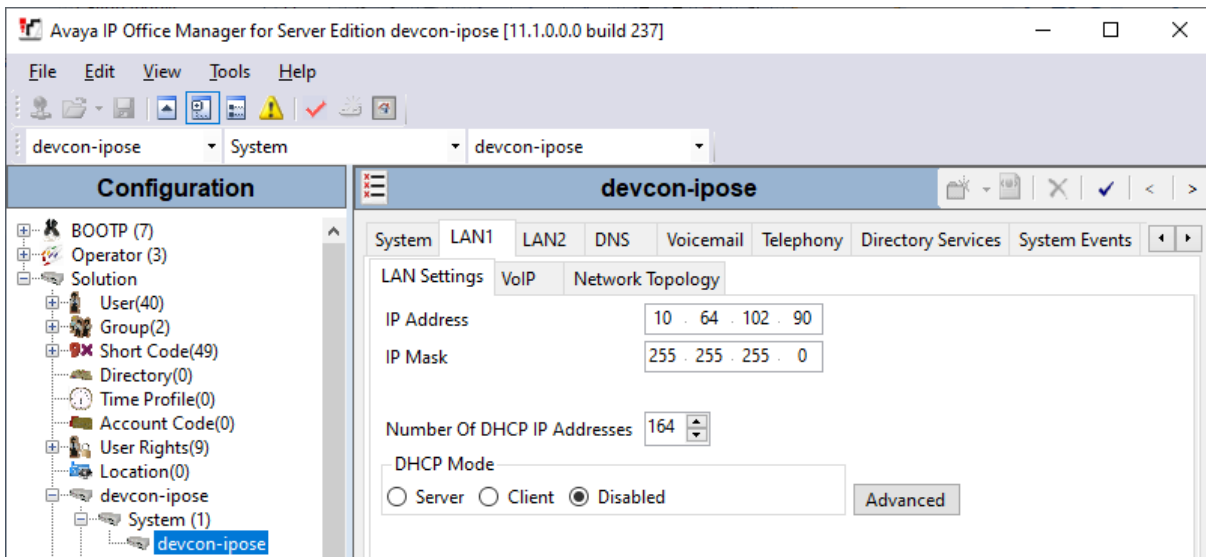
From a PC with Avaya IP Office Manager installed, select **Start** → **Programs** → **IP Office** → **Manager** to launch the Manager application. Select the required IP Office system and log in with the appropriate credentials.

The **Avaya IP Office Manager for Server Edition** screen is displayed. From the configuration tree in the left pane, select **License** to display the license screen in the right pane. Verify that the **License Status** is “Valid” for **3rd Party IP Endpoints**.



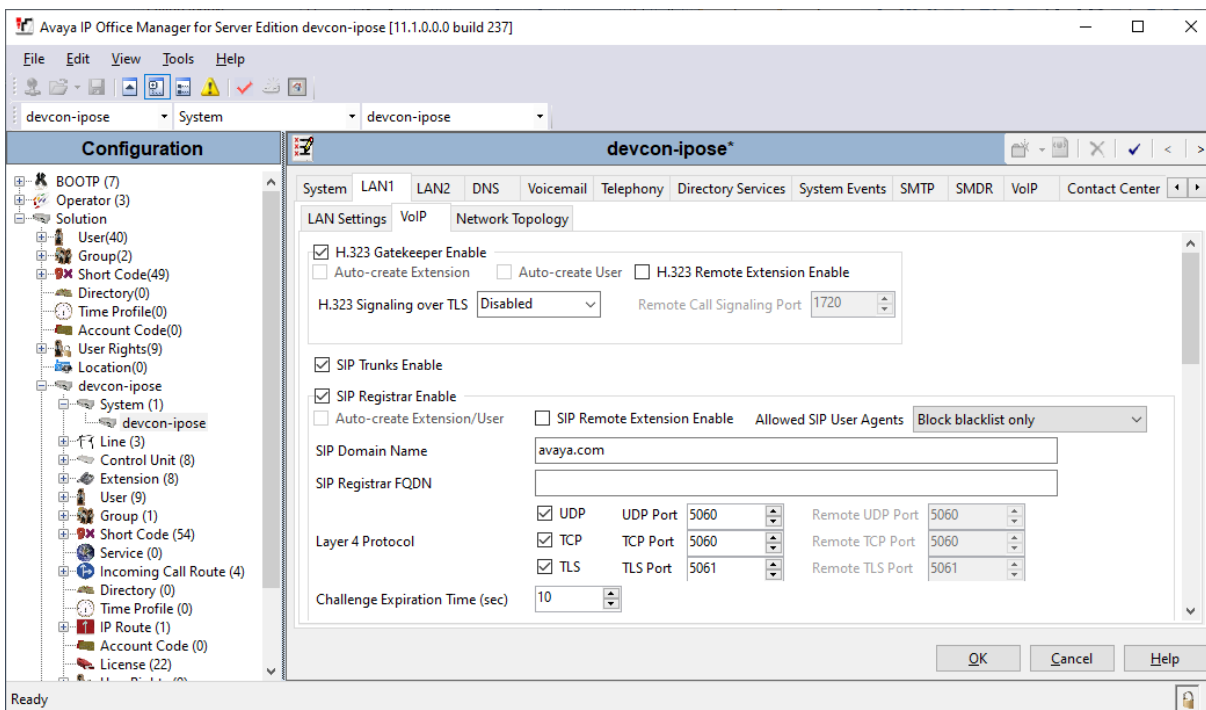
## 5.2. Obtain LAN IP Address

From 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 the Valcom IP speaker.



## 5.3. Administer SIP Registrar

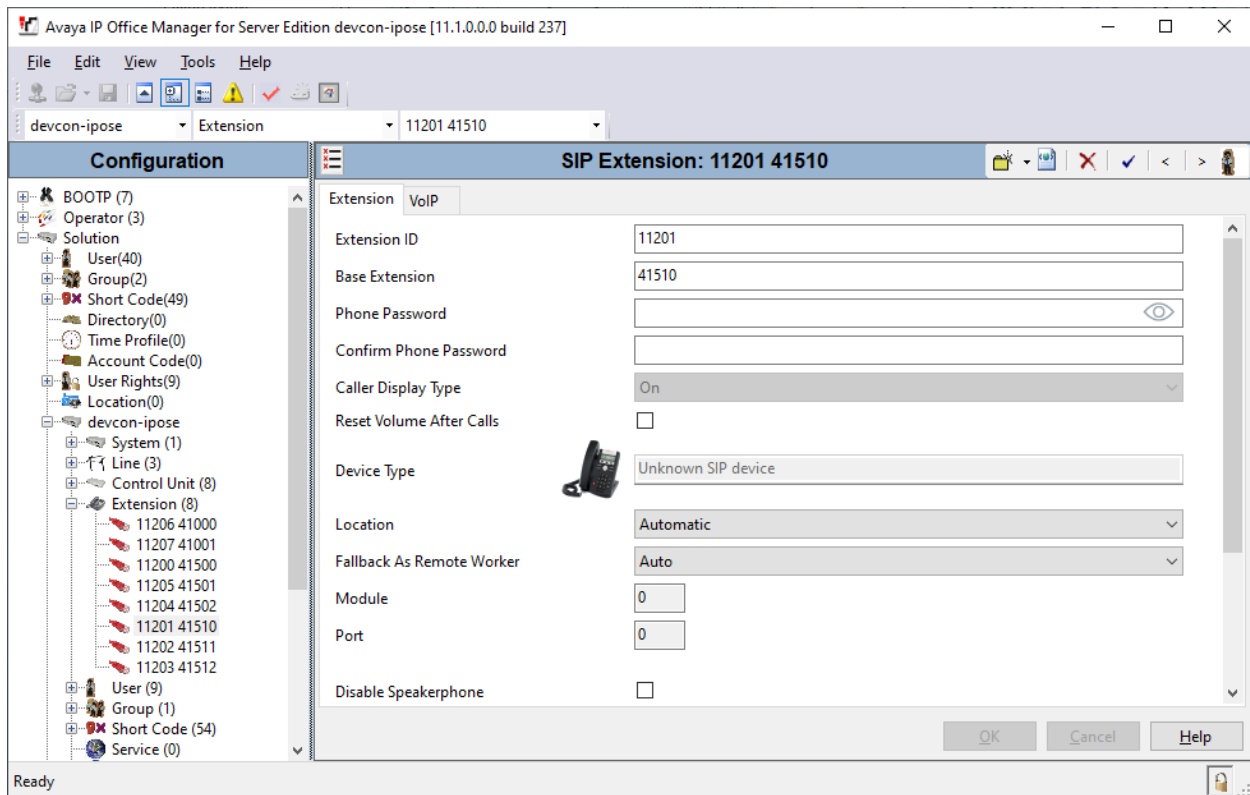
Select the **VoIP** sub-tab. Ensure that **SIP Registrar Enable** and that UDP and/or TCP transport is selected, which will be used by the IP speaker. Also, enter a valid **Domain Name**.



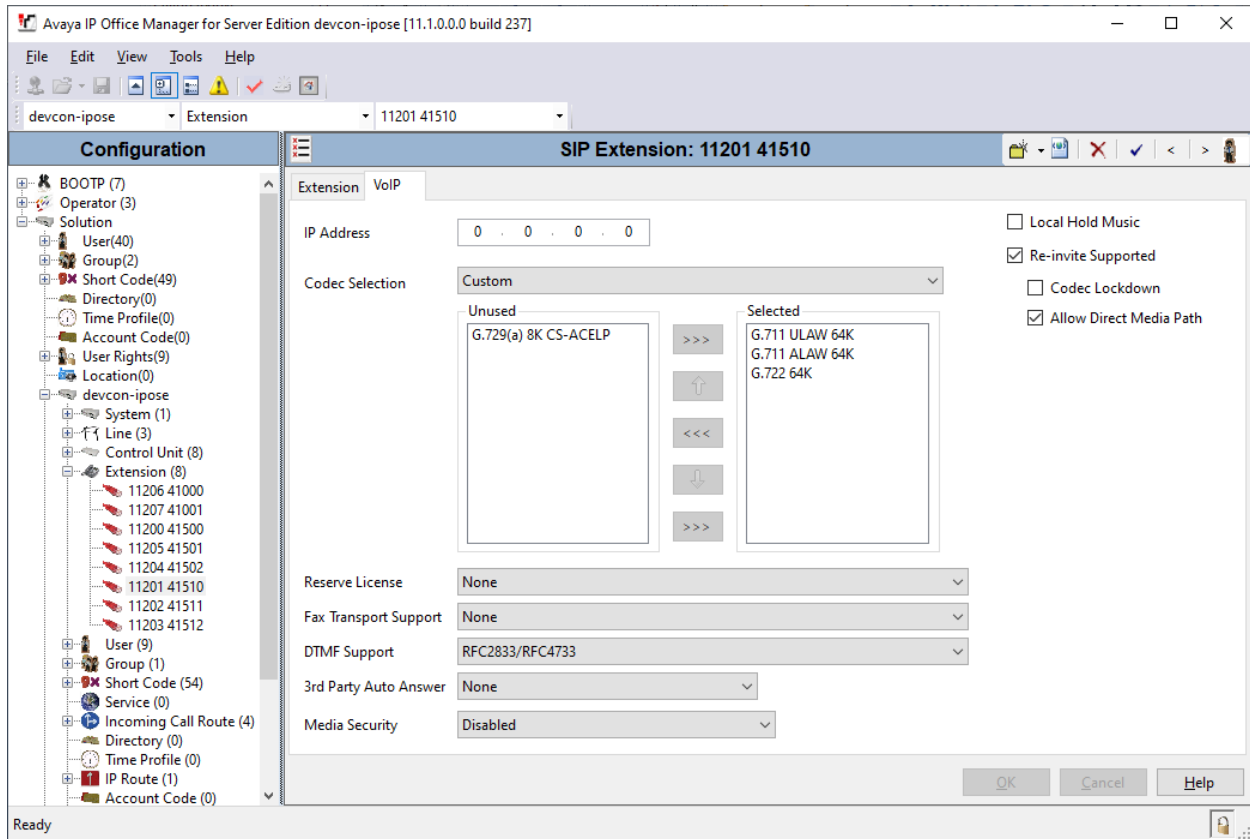


## 5.4. Administer SIP Extension

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 (not shown). Enter the desired extension for the **Base Extension** field as shown below. In this example, the IP speaker was assigned extension *41510*.



Select the **VoIP** tab and retain the default values in the all fields. The IP speaker supports G.711 and G.722 codecs.



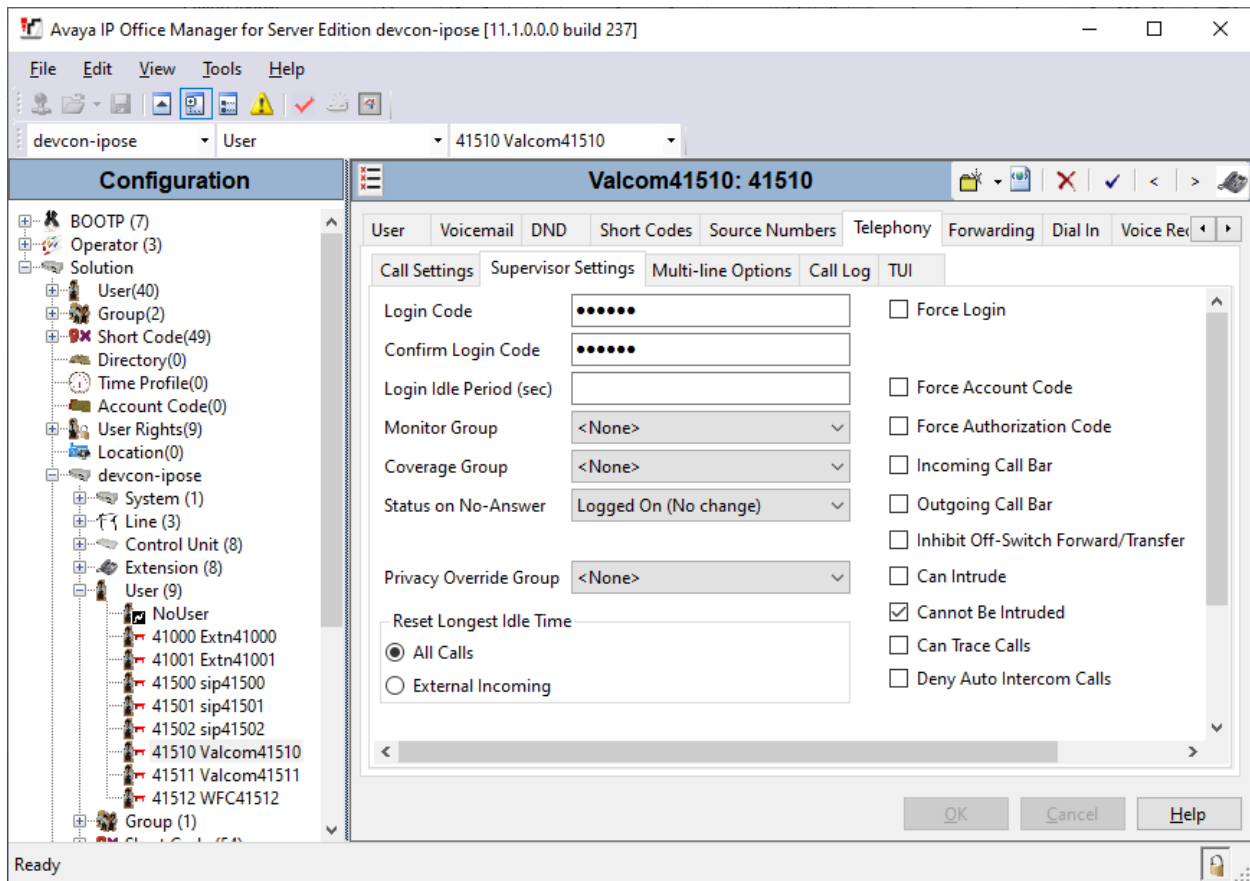
## 5.5. Administer SIP User

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 created in **Section 5.4**. The **Extension** field specifies the username that will be used by the IP speaker to register with IP Office Server Edition. Note that the SIP authentication password was configured in the **Telephony** tab below instead of in the **Password** field shown below.

The screenshot shows the Avaya IP Office Manager for Server Edition configuration window. The title bar indicates the version is 11.1.0.0.0 build 237. The left pane shows a configuration tree with 'User' selected under 'Solution'. The right pane shows the configuration for 'Valcom41510: 41510'. The 'User' tab is active, showing fields for Name, Password, Confirm Password, Unique Identity, Conference PIN, Confirm Audio, Conference PIN, Account Status (Enabled), Full Name (Valcom), Extension (41510), Email Address, Locale, Priority (5), System Phone Rights (None), and Profile (Basic User). Below the profile section, there are several checkboxes for enabling various features: Receptionist, Enable Softphone, Enable one-X Portal Services, Enable one-X TeleCommuter, Enable Remote Worker, Enable Desktop/Tablet VoIP client, Enable Mobile VoIP Client, Send Mobility Email, and Web Collaboration. The status bar at the bottom shows 'Ready'.

Field	Value
Name	Valcom41510
Password	
Confirm Password	
Unique Identity	
Conference PIN	
Confirm Audio	
Conference PIN	
Account Status	Enabled
Full Name	Valcom
Extension	41510
Email Address	
Locale	
Priority	5
System Phone Rights	None
Profile	Basic User
Receptionist	<input type="checkbox"/>
Enable Softphone	<input type="checkbox"/>
Enable one-X Portal Services	<input type="checkbox"/>
Enable one-X TeleCommuter	<input type="checkbox"/>
Enable Remote Worker	<input type="checkbox"/>
Enable Desktop/Tablet VoIP client	<input type="checkbox"/>
Enable Mobile VoIP Client	<input type="checkbox"/>
Send Mobility Email	<input type="checkbox"/>
Web Collaboration	<input type="checkbox"/>

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



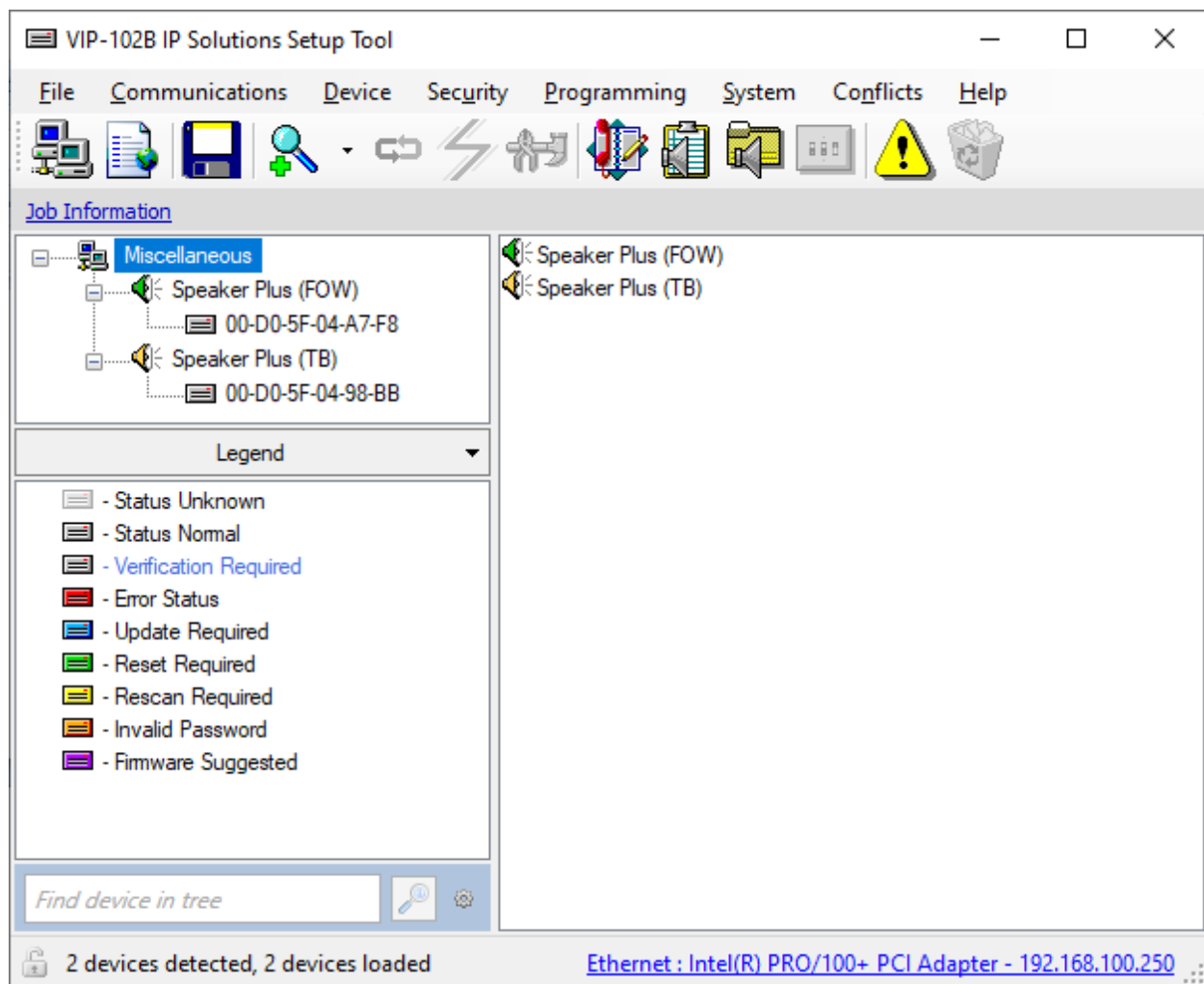
## 6. Configure Valcom One-Way and Talkback IP Speakers

This section covers the configuration of the VIP-160A IP Talkback 8” Ceiling Speaker using the Valcom VIP-102B IP Solutions Setup Tool. The configuration of the VIP-130AL-GY IP Secure One-Way Paging Horn is similar, unless otherwise specified. The configuration covers the following areas:

- Launch the Valcom VIP-102B IP Solutions Setup Tool
- Configure the Network Settings of Valcom One-Way and Talkback IP Speakers
- Configure SIP Parameters of Valcom One-Way and Talkback IP Speakers
- Verify Codec Settings
- Specify Call Destination
- Update SIP Intercom Controller with the New Configuration

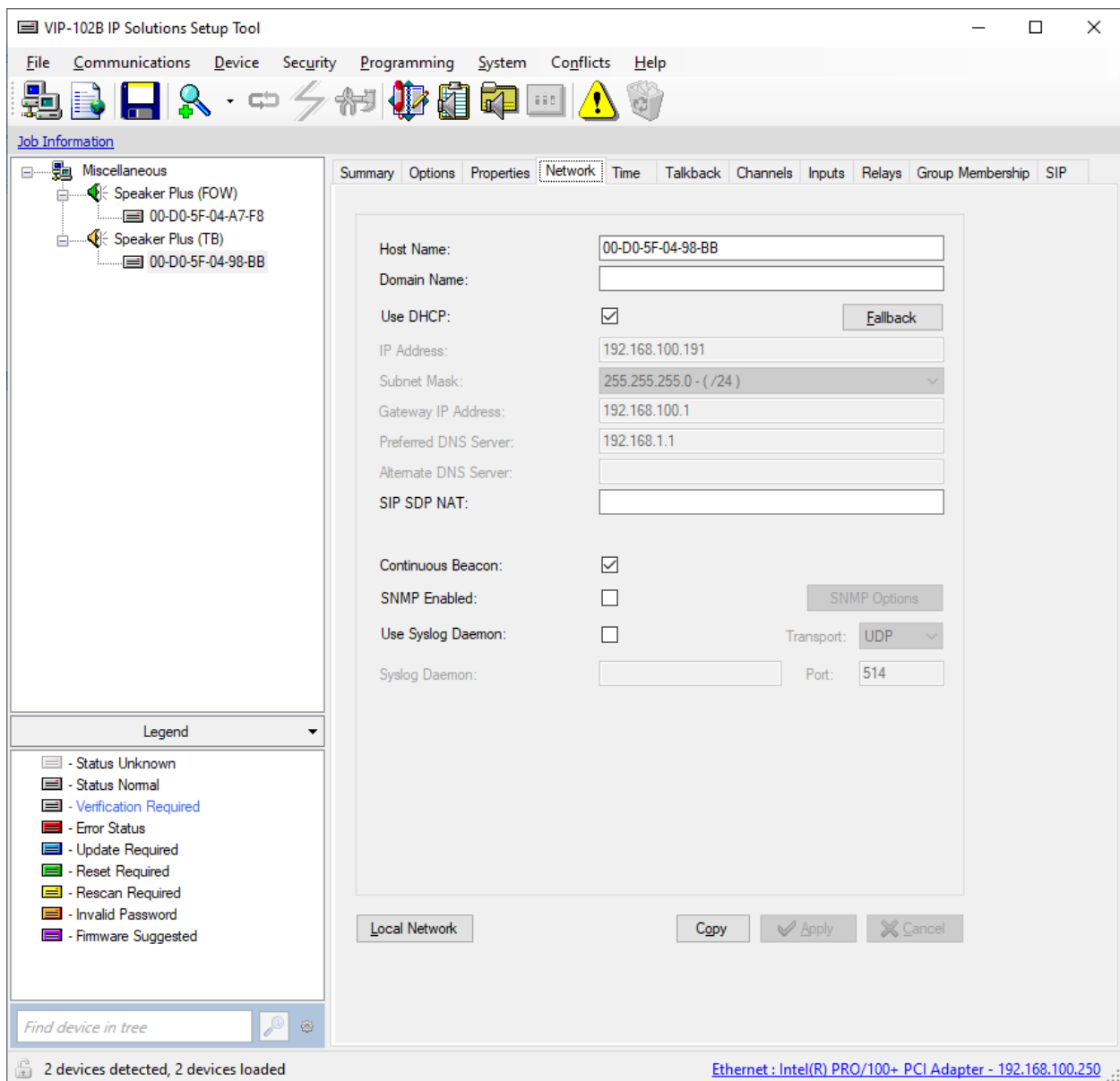
### 6.1. Launch Valcom VIP-102B IP Solutions Setup Tool

Launch the **VIP-102B IP Solutions Setup Tool** and follow the prompts. The main window is displayed as shown below.



## 6.2. Configure the Network Settings of Valcom One-Way and Talkback IP Speakers

Click the MAC/hardware address of the corresponding IP speaker in the left pane and select the **Network** tab. The IP speaker must first acquire IP network settings before proceeding with provisioning. These network settings were automatically obtained from a DHCP server as shown below. Alternatively, the IP speaker could be configured with static IP addresses, but for the compliance test, a DHCP server was used.



### 6.3. Configure SIP Parameters of Valcom One-Way and Talkback IP Speakers

From the **VIP-102B IP Solutions Setup Tool**, navigate to the **SIP** tab of the IP speaker. For **Transport**, select UDP or TCP transport. Set the **Phone Number** and **Authentication** to the SIP extension (e.g., *41510*) and **Secret** to the SIP password used to register with IP Office. Select the **Register** checkbox and set the **Primary Server** to the IP Office LAN1 IP address (e.g., *10.64.102.90*). Leave all other fields at their default values. Click **Apply**.

The screenshot displays the 'VIP-102B IP Solutions Setup Tool' interface. The 'SIP' tab is selected, showing various configuration fields. The 'Transport' dropdown is set to 'Accept: TCP + UDP, Originate: UDP'. The 'Phone Number' is '41510', 'Description' is 'VIP-160A', 'Authentication Name' is '41510', 'Secret' is '123456', and 'Realm' is 'avaya.com'. The 'SIP Servers' table lists a primary server at 10.64.102.90:5060 and three backup servers. The 'Register' checkbox is checked. The 'Busy Message' dropdown is set to 'Busy'. The 'Call Fwd Busy (302)' field is empty. The 'Outbound Proxy' field is empty, and the 'Outbound Port' is '5060'. The 'Keep Alive Timer (secs)' is '3600', 'Options Timer (secs)' is '0', 'SIP Port' is '5060', 'Idle Timeout (secs)' is '0', 'RTP Port' is '20000', and 'Max Call Timer (secs)' is '0'. The 'CID Number' is '41510' and the 'CID Name' is 'VIP-160A'. The 'Legend' section on the left lists various status icons. The bottom status bar shows '2 devices detected, 2 devices loaded' and the network interface 'Ethernet : Intel(R) PRO/100+ PCI Adapter - 192.168.100.250'.

**VIP-102B IP Solutions Setup Tool**

File Communications Device Security Programming System Conflicts Help

Job Information

Miscellaneous

- Speaker Plus (FOW) 00-D0-5F-04-A7-F8
- Speaker Plus (TB) 00-D0-5F-04-98-BB

Summary Options Properties Network Time Talkback Channels Inputs Relays Group Membership SIP

Transport: Accept: TCP + UDP, Originate: UDP

1

Phone Number: 41510

Description: VIP-160A

Authentication Name: 41510

Secret: 123456

Realm: avaya.com

SIP Servers:

	Server	Port
Primary	10.64.102.90	5060
Backup 1		5060
Backup 2		5060
Backup 3		5060

Register: ☒

DNS SRV: ☐

Busy Message: Busy

Call Fwd Busy (302):

Outbound Proxy: Outbound Port: 5060

Keep Alive Timer (secs): 3600 Options Timer (secs): 0

SIP Port: 5060 Idle Timeout (secs): 0

RTP Port: 20000 Max Call Timer (secs): 0

CID Number: 41510

CID Name: VIP-160A

Legend

- Status Unknown
- Status Normal
- Verification Required
- Error Status
- Update Required
- Reset Required
- Rescan Required
- Invalid Password
- Firmware Suggested

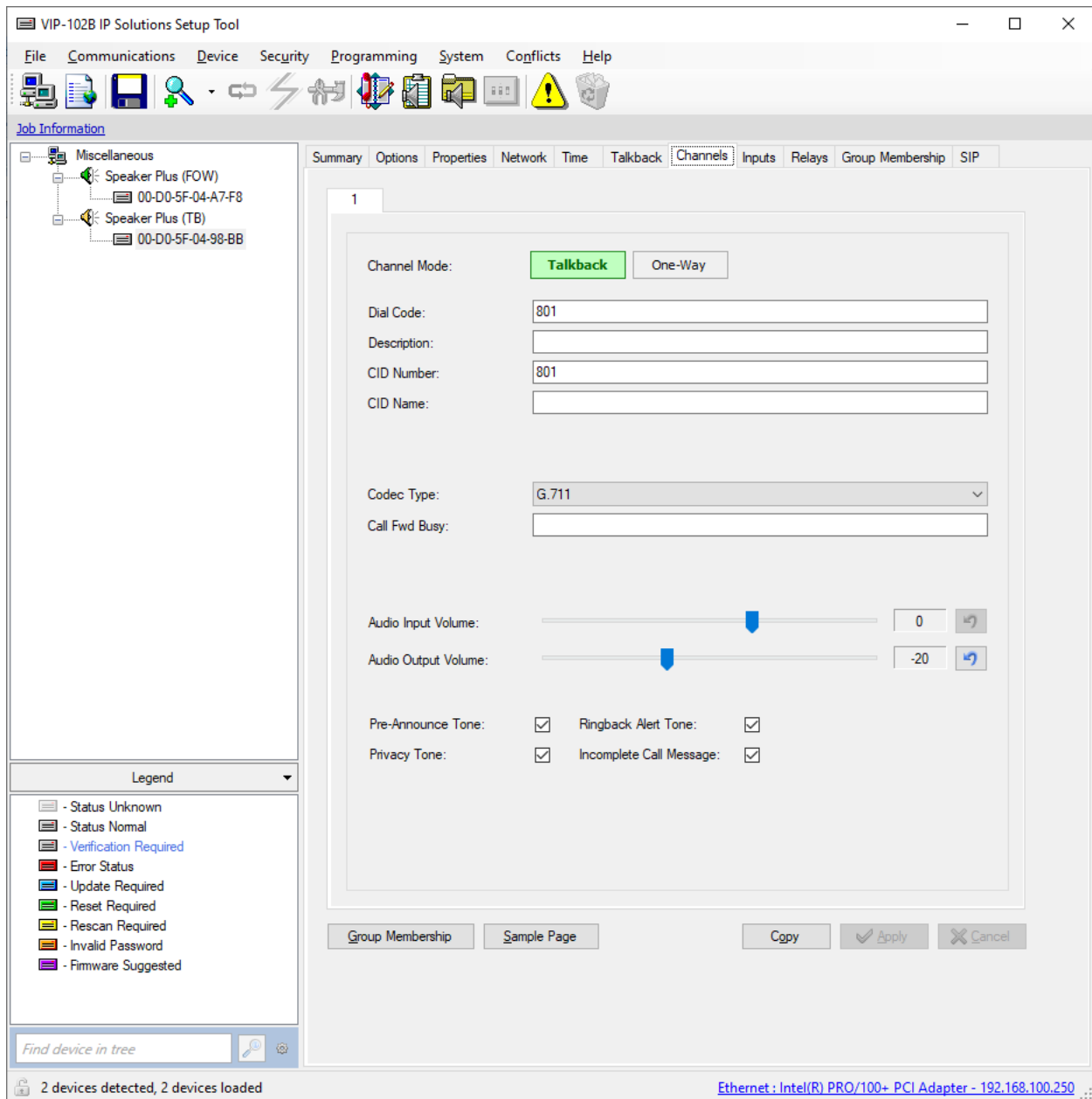
Find device in tree

2 devices detected, 2 devices loaded

Ethernet : Intel(R) PRO/100+ PCI Adapter - 192.168.100.250

## 6.4. Verify Codec Settings

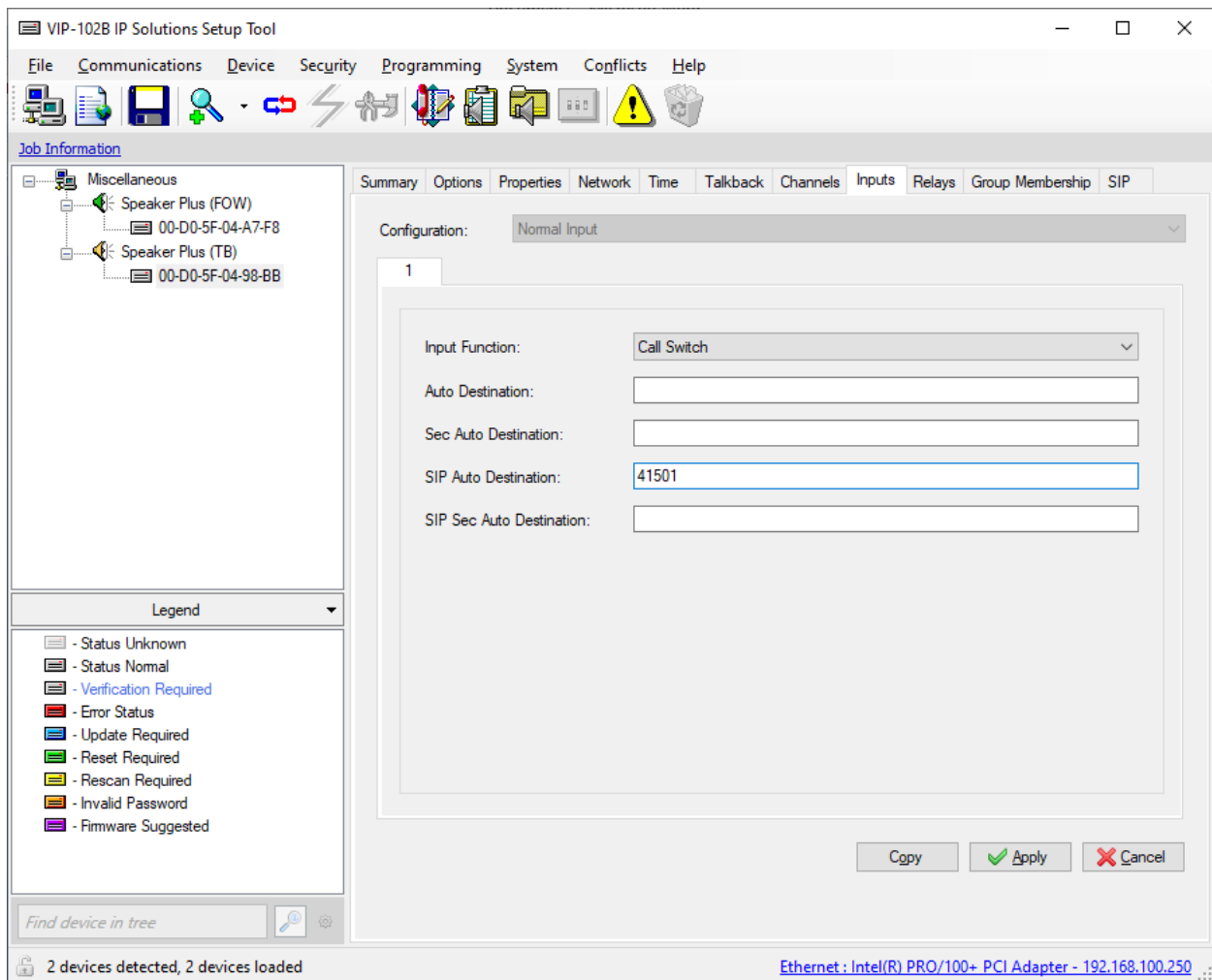
Navigate to the **Channels** tab shown below. For this solution, the IP speaker will always advertise G.722 and G.711 in the SDP section of the SIP INVITE message. The setting of the **Codec Type** will not impact the codecs supported by the IP speaker. The **Codec Type** may be left at the default value and both codecs will still be supported.





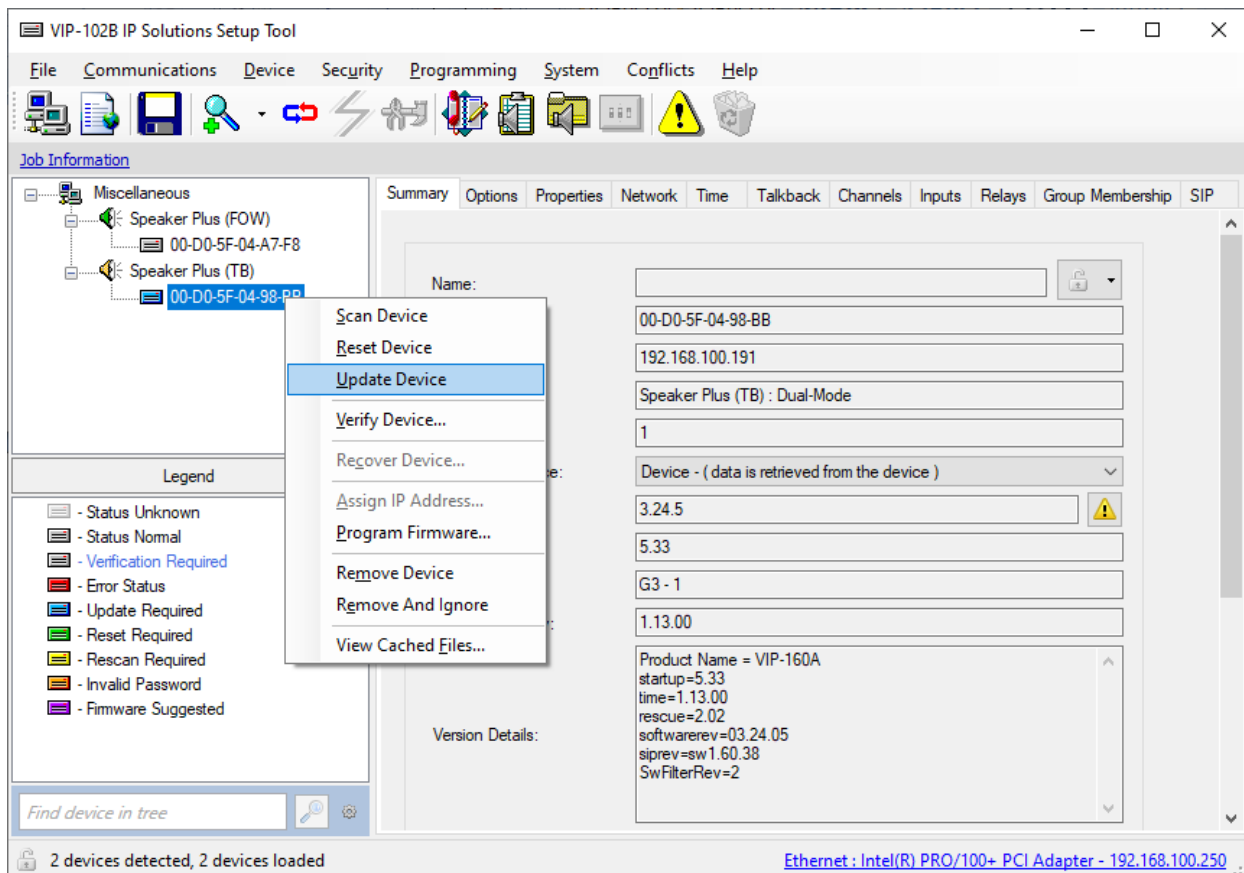
## 6.5. Specify Call Destination

For Talkback IP speakers with a call switch button only, a **SIP Auto Destination** may be configured to specify the number that should be dialed when the call switch button is pressed. In the following example, when the call switch button is pressed, the IP speaker will dial *41501*. After the call is established, the IP speaker can terminate the call by pressing the call switch button again.

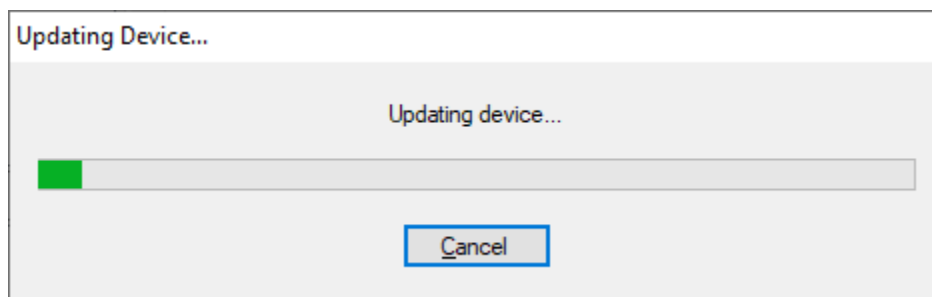


## 6.6. Update SIP Intercom Controller with the New Configuration

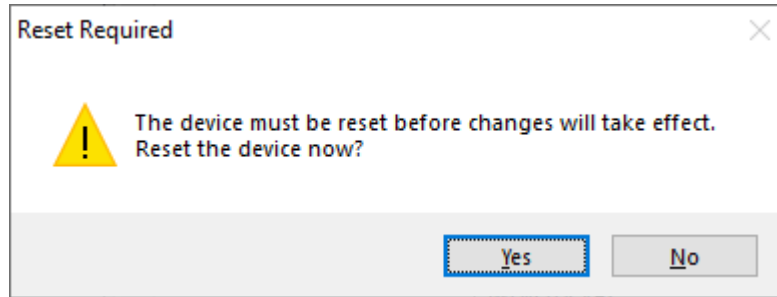
From the **VIP-102B IP Solutions Setup Tool**, right-mouse click on the MAC/hardware address of the SIP Intercom Controller and select **Update Device** from the pop-up menu as shown below.



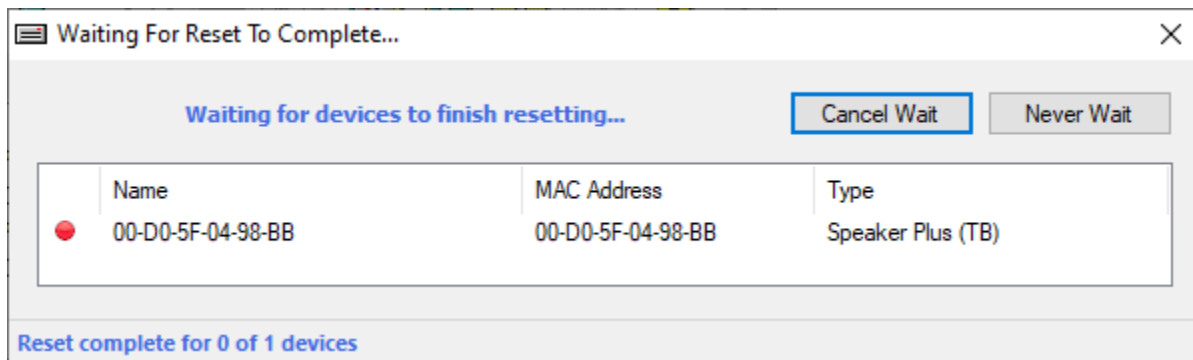
The following window is displayed indicating that the device is being updated.



A device reset is required so respond with **Yes** when prompted.



The following window will be displayed while the device is being reset. When the reset is complete, the window will disappear.



## 7. Verification Steps

This section provides the tests that may be performed to verify proper configuration of Valcom One-Way and Talkback IP Speakers with Avaya IP Office Server Edition.

1. Verify that the IP speaker has successfully registered with IP Office Server Edition. Launch **IP Office System Status** and navigate to **Extensions** → **<SIP Extension>**, where **<SIP Extension>** is the IP speaker 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 - devcon-ipose (10.64.102.90) - IP Office Linux PC 11.1.0.0.0 build 237". The main window has a menu bar with "Help", "Snapshot", "LogOff", "Exit", and "About". On the left is a navigation pane with a tree view containing "System", "Alarms (9)", "Extensions (4)", "Trunks (3)", "Active Calls", "Resources", "Voicemail", "IP Networking", and "Locations". The "Extensions (4)" folder is expanded, showing "41000", "41001", "41502", and "41510" (which is selected). The main area displays the "Extension Status" for extension 41510. 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" as the current state and "00:00:31" as the time in state. At the bottom of the window are buttons for "Trace", "Trace All", "Pause", "Ping", "Call Details", "Print...", and "Save As...". The status bar at the bottom shows the time "11:29:57 AM" and the status "Online".

Call Ref	Current State	Time in State	Calling Number or Called Number	Direction	Other Party on Call
	Idle	00:00:31			

2. Place a call to a Valcom IP speaker. Verify two-way audio for Talkback IP speakers and one-way audio for One-Way IP speakers. Terminate the call from the Avaya IP Deskphone or by pressing the call button on the speaker.
3. Place an intercom call by pressing the call button on a Talkback IP speaker. Verify two-way audio to the call destination. Terminate the call from the IP speaker by pressing the call button.

## 8. Conclusion

These Application Notes described the configuration steps required to integrate Valcom One-Way and Talkback IP Speakers with Avaya IP Office Server Edition. Intercom and paging calls were established with Valcom VIP-130AL-GY IP Secure One-Way Paging Horn, VIP-160A IP Talkback 8” Ceiling Speaker, Avaya H.323 / SIP Deskphones, and the PSTN. All feature and serviceability test cases were completed successfully with observations noted in **Section 2.2**.

## 9. References

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

- [1] *Administering Avaya IP Office Platform with Manager*, Release 11.1 SP1, Issue 26, July 2020, available at <http://support.avaya.com>.
- [2] *Valcom VIP-102B IP Solutions Setup Tool Version 8.1.0.0 Reference Manual*, Revision 12 – 7/10/20, available at <http://www.valcom.com/vipsetuptool>.

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Please e-mail any questions or comments pertaining to these Application Notes along with the full title name and filename, located in the lower right corner, directly to the Avaya DevConnect Program at [devconnect@avaya.com](mailto:devconnect@avaya.com).



## Declaration of Conformance

**December 1, 2020**

Jeff Gartner  
Senior Manager  
DevConnect Program  
Avaya

**Dear Jeff Gartner:**

We, Valcom Inc, declare under sole responsibility that product series named IP Ceiling Speakers, IP Wall Speakers and IP Horns share the same hardware circuitry, software, SIP stack and firmware version. Therefore, the products are expected to behave in the same manner. The differences between the different models in each series are generally cosmetic in nature, such as enclosure shape or color, mounting arrangement, etc.

Sincerely,

**/s/ David Ellison**

**David Ellison**  
**Technical Support Manager**  
**Valcom Inc**  
**dellison@valcom.com**