

Avaya Solution & Interoperability Test Lab

Application Notes for Valcom V-9972 Universal Paging Interface with Avaya IP Office Server Edition using SIP Trunk - Issue 1.0

Abstract

These Application Notes describe the configuration steps required to integrate the Valcom V-9972 Universal Paging Interface with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. Valcom V-9972 Universal Paging Interface provides access to paging systems, such as Valcom VIP-430A IP Wall Speakers, which was used in the compliance test. For this compliance test, Valcom V-9972 Universal Paging Interface connected to Avaya IP Office Server Edition via a SIP trunk. The Valcom V-9972 Universal Paging Interface supports two-way audio intercom (talkback) calls and one-way audio group paging calls.

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 DevConnect Program at the Avaya Solution and Interoperability Test Lab.

1. Introduction

These Application Notes describe the configuration steps required to integrate the Valcom V-9972 Universal Paging Interface with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. Valcom V-9972 Universal Paging Interface provides access to paging systems, such as Valcom VIP-430A IP Wall Speakers, which was used in the compliance test. For this compliance test, Valcom V-9972 Universal Paging Interface connected to Avaya IP Office Server Edition via a SIP trunk. The Valcom V-9972 Universal Paging Interface supports two-way audio intercom (talkback) calls and one-way audio group paging calls.

When a call is routed to the Valcom V-9972 Universal Paging Interface, the V-9972 plays dial tone back to the caller. The caller can then dial a Valcom speaker Dial Code or Group Code to establish an intercom call (two-way audio) with a single Valcom speaker or a group paging call (one-way audio) to one or more Valcom speakers.

Alternatively, the Valcom VIP-430A IP Wall Speaker can establish intercom calls by pressing its call button. Pressing the call button would place a call to the specified destination in the V-9972 configuration. Pressing the call button during an active call, terminates the call.

All calls to/from the VIP-430A IP Wall Speaker go through the V-9972. Communication between V-9972 and VIP-430A IP Wall Speaker uses unicast for intercom (talkback) calls and multicast for paging calls.

Valcom offers Universal Paging Adapters 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 V-9972 Universal Paging Interface with the Valcom VIP-430A IP Wall Speaker, Avaya SIP / H.323 IP Deskphones, and the PSTN. Two-way audio intercom calls and one-way audio group paging calls were exercised. In addition, basic telephony features were exercised from Avaya SIP / H.323 IP Deskphones, such as hold/resume, call transfer, and conference.

The serviceability testing focused on verifying that the Valcom V-9972 Universal Paging Interface came back into service after reconnecting the network connection or 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

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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 Valcom V-9972 Universal Paging Interface used TLS/SRTP encryption features.

2.1. Interoperability Compliance Testing

Interoperability compliance testing covered the following features and functionality:

- Establishing a SIP trunk between V-9972 and IP Office Server Edition and IP Office 500V2 Expansion and verifying the exchange of SIP Options messages.
- Calls between V-9972 and Avaya H.323/SIP endpoints with Direct IP Media (Shuffling) enabled and disabled. Shuffling allows IP endpoints to send audio RTP packets directly to each other without using media resources in Avaya IP Office.
- Establishing two-way audio intercom calls between VIP-430A IP Wall Speaker, via V-9972, Avaya H.323 / SIP Deskphones, and PSTN in both directions.
- Establishing one-way paging calls from Avaya H.323 / SIP Deskphones to VIP-430A IP Wall Speaker via V-9972.
- Verifying that higher priority paging calls take precedence over existing lower priority intercom calls.
- Terminating active calls by pressing the call button on the VIP-430A IP Wall Speaker.
- Support of G.711 mu-law codec.
- Support of TLS/SRTP using mutual TLS authentication.
- Since the VIP-430A IP Wall Speaker does not provide a keypad or feature buttons, basic telephony features, such as hold/resume, call transfer, and conference were performed from Avaya H.323/SIP Deskphones.
- Long duration calls and outbound calls from V-9972 that were rejected due to dialing an invalid number or a busy station.
- Proper system recovery after re-establishing network connectivity to the V-9972 or restarting the V-9972.

2.2. Test Results

All test cases passed.

2.3. Support

For technical support and information on Valcom V-9972 Universal Paging Interface, contact Valcom Technical Support at:

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

3. Reference Configuration

Figure 1 illustrates a sample configuration with an Avaya SIP-based network that includes the following products:

- Avaya IP Office Server Edition and Avaya IP Office 500 V2 Expansion connected via a SCN trunk and configured via Avaya IP Office Manager.
- PSTN connectivity provided by a SIP trunk on Avaya IP Office Server Edition and an ISND-PRI trunk on Avaya IP Office 500 V2 Expansion System.
- Avaya 96x1 Series H.323 Deskphones and Avaya J100 Series SIP Deskphones registered to Avaya IP Office Server Edition and Avaya IP Office 500 V2 Expansion.
- Valcom V-9972 Universal Paging Interface connected to Avaya IP Office Server Edition via a SIP trunk. Valcom VIP-430A IP Wall Speaker serving as the Valcom audio endpoint, and using Valcom VIP-102B IP Solutions Setup Tool to configure Valcom products.



Figure 1: Avaya SIP Network with Valcom V-9972 Universal Paging Interface and Valcom VIP-430A IP Wall Speakers

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	11.1.2.1.0 build 3
Avaya IP Office 500 V2 Expansion	11.1.2.1.0 build 3
Avaya 96x1 Series IP Deskphones	6.8304 (H.323)
Avaya J100 Series IP Deskphones	4.0.7.0.7 (SIP)
Valcom V-9972 Universal Paging Interface, including optional L9972-2 feature license	3.00.14
Valcom VIP-430A IP Wall Speaker	3.23.7
Valcom VIP-102B IP Solutions Setup Tool	8.4.0.0

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 Server Edition

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

- Verify IP Office license
- Obtain LAN IP address
- Enable SIP trunks
- Administer SIP line
- Administer incoming call route
- Administer short code
- Install Valcom V-9972 Universal Paging Interface TLS Certificate

5.1. Verify IP Office License

From a PC with Avaya IP Office Manager installed, select Start \rightarrow Programs \rightarrow IP Office \rightarrow 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 SIP Trunk Channels, and that the Instances value is sufficient.

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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 V-9972.



5.3. Enable SIP Trunks

Select the VoIP sub-tab. Ensure that SIP Trunks Enable is checked as shown below.

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5.4. Administer SIP Line

From the configuration tree in the left pane, right-click on **Line**, and select **New** \rightarrow **SIP Line** from the pop-up list to add a new SIP line (not shown).

Select the **SIP Line** tab. Select an available **Line Number** (e.g., 5) and set **ITSP Domain Name** to the appropriate SIP domain (e.g., *avaya.com*).

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Select the **Transport** tab. For **ITSP Proxy Address**, enter the IP address of V-9972, set the **Layer 4 Protocol** to *TLS* and the **Send Port** and **Listen Port** to the TLS port (e.g., *5061*). Retain the default values for the remaining fields.

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Select the **Call Details** tab, and click **Add** to display the **Call Details** | **SIP URI** section.

In the **Call Details** | **SIP URI** window, set the **Incoming Group** and **Outgoing Group** to a valid number (e.g., *5*) and set the **Max Sessions**.

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The **Call Details** tab is updated as shown below.



Select the **VoIP** tab. For **Codec Selection**, select *Custom* and select G.711 ULAW 64K, which is the codec supported by V-9972 with IP speakers. Enable **Allow Direct Media Path** and **Force direct media with phones** so that audio/RTP flows directly between V-9972 and H.323/SIP endpoints without using media resources in Avaya IP Office Server Edition.

Media Security was enabled for V-9972. Since the V-9972 does not support SDP Capability Negotiation (RFC5939), IP Office Server Edition should only offer SRTP in the SIP SDP. Therefore, the **Media Security** field must be set to *Enforced*. In addition, when SRTP is enabled for V-9972, unencrypted RTCP is required.

Note: The Media Security section shown below was used for Avaya H.323 / SIP Deskphones and the Web Socket SCN trunk between IP Office Server Edition and IP Office 500 V2 Expansion.

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5.5. Administer Incoming Call Route

From the configuration tree in the left pane, right-click on **Incoming Call Route**, and select **New** from the pop-up list to add a new route. For **Line Group Id**, select the incoming group number from **Section 5.4**, which corresponds to the SIP line, in this case "5".

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Select the **Destinations** tab. For **Destination**, enter "." to match any dialed number by V-9972.

5.6. Administer Short Code

From the configuration tree in the left pane, right-click on **Short Code** and select **New** from the pop-up list to add a new short code for calls to V-9972. For the compliance test, 41600 was used to route calls to V-9972. Set **Telephone Number** to . so that 41600 would be sent to V-9972 and set **Line Group Id** to the outgoing group number from **Section 5.4**, which corresponds to the SIP line.

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5.7. Install Valcom V-9972 Universal Paging Interface TLS Certificate

From Avaya IP Office Manager for Server Edition, navigate to File \rightarrow Advanced \rightarrow Security Settings. The Security Settings window is displayed. In the left pane, navigate to Security \rightarrow System, and then select the Certificates tab to display the window below.

To enable mutual TLS authentication, install the V-9972 TLS certificate shown below as *TechSupportCA* and set **Received Certificate Checks (Telephony Endpoints)** to *High*.



6. Configure Valcom V-9972 Universal Paging Interface

This section covers the configuration of Valcom V-9972 Universal Paging Interface using the Valcom VIP-102B IP Solutions Setup Tool. The configuration covers the following areas:

- Launch the Valcom VIP-102B IP Solutions Setup Tool
- Configure the Network Settings
- Configure Time
- Install System Manager CA TLS Certificate
- Configure SIP Parameters
- Verify Codec Settings
- Update Universal Paging Interface with the New Configuration

Note: These Application Notes do not cover the configuration of the Valcom VIP-430A IP Wall Speakers, Audio Groups, or the assignment of Dial Codes to Valcom speakers. Refer to [2] and [3] for details.

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.

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2 devices detected, 2 devices loaded	Ethernet : Rea	altek PCIe	GBE Family C	Controller - 1	92.168.10	0.251 .::

6.2. Configure the Network Settings

Click the MAC/hardware address under Universal Page Interface in the left pane and select the **Network** tab. V-9972 must first acquire IP network settings before proceeding with provisioning. These network settings were automatically obtained from a DHCP server as shown below. Alternatively, V-9972 could be configured with static IP addresses, but for the compliance test, DHCP was used.

IVIP-102B IP Solutions Setup Tool			- 🗆 X
<u>File</u> <u>Communications</u> <u>Device</u> Sec <u>u</u> rit	y <u>P</u> rogramming <u>S</u> ystem Co <u>n</u> fli	icts <u>H</u> elp	
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Job Information		_	
Miscellaneous Speaker Plus (TB)	Summary Properties Network Time	System Channels Group Membership SIP	
Host Name: Domain Name: Use DHCP: IP Address: Subnet Mask: Gateway IP Address: Preferred DNS Server SIP SDP NAT: Continuous Beacon: Use Syslog Daemon: Syslog Daemon:	Host Name: Domain Name: Use DHCP: IP Address: Subnet Mask: Gateway IP Address: Preferred DNS Server: Alternate DNS Server: SIP SDP NAT: Continuous Beacon: Use Syslog Daemon: Syslog Daemon:	Eallback 192.168.100.197 255 255 255.0 - (/24) 192.168.100.1 192.168.1.1 □	
Legend ▼ - Status Unknown - Status Normal - Verification Required - Error Status - Update Required - Reset Required - Reset Required - Reset Required - Invalid Password - Firmware Suggested	Local Network	Cgpy Apply 🔀 Çar	hoel
2 devices detected, 2 devices loaded		Ethernet : Realtek PCIe GBE Fa	amily Controller - 192.168.100.251

6.3. Configure the Time

Navigate to the **Time** tab and set the Static NTP Servers to ensure the proper date/time on the device.

VIP-102B IP Solutions Setup Tool			_		×
Car Vie 1020 in Solutions Setup 1001					~
<u>File</u> <u>Communications</u> <u>Device</u> Sec <u>u</u> rit	ty <u>P</u> rogramming <u>S</u> ystem Co <u>i</u>	<u>n</u> flicts <u>H</u> elp			
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II 📑 🖬 🔸 👘 🥂	n - 👬 🕅 🖬 📖				
Job Information					
Miscellaneous	Summary Properties Network Tim	System Channels Group Membership SIP			
Universal Page Interface	NTP Selection Order:	DHCP Only			
	Static NTP Servers:	10.64.102.117			
	NTP Rate (hh:mm:ss):	01 : 00 : 00 NTP Error Rate: 00 : 00 : 30			
	Adjustment Method:	Manual O POSIX String			
	Hours Offset From UTC:	(UTC -05:00) : Eastern Time ~			
	Additional Minutes Offset:	0 ~			
	Perform DST Correction:				
	Defaults				
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 : Realtek PCIe GBE Family Cont	roller - 1	92.168.10	0.251

6.4. Install the System ManagerCA TLS Certificate

Navigate to the **Properties** tab to install the System Manager CA certificate. Note that the V-9972 has a device certificate (*V-9972-Avaya-Priv-Key-and-Cert.pem*) signed by a different CA other than the System Manager. Click on **Certificates**.

Note: Avaya Aura® System Manager was used as the Certificate Authority.

INVIP-102B IP Solutions Setup Tool		-		×
<u>File</u> <u>Communications</u> <u>Device</u> Sec <u>u</u> rity	<u>P</u> rogramming <u>Sy</u> stem Co <u>n</u> flicts <u>H</u> elp			
♣	** 🕸 🛱 📼 🔝 🎲			
Job Information				
Job Information Image: Miscellaneous Image: Speaker Plus (TB) Image: D0-D0-SF-05-SE4-1B Image: D0-D0-SF-05-CB-C5	Summary Properties Network Time System Channels Group Membership SIP Properties Device Name: Log Level: Active Device Certificate: V-9972-Avaya-Priv-Key-and-Cert.pem Comments			
Legend	Certificates Web Interface Cgpy Apply Cancel			
Find device in tree 	Ethernet : Realtek PCIe GBE Family Contr	oller - 192	2.168.10	0.251

In the **Certificate** dialog box, add the System Manager CA TLS certificate. Note that the certificate has already been imported as shown below. In addition, the V-9972 root certificate (*techsupportca.crt*) is also installed. This certificate must be installed on IP Office to support mutual TLS authentication as shown in **Section 5.7**.

Certificates				×
File Type: Source File: Description:	CA Certificates			Add Update
File techsupportca.crt SystemManagerCA.pem		Size (Bytes) 1200 1224	Description DevConnect SystemManager CA Cert	Delete
				Refresh Close

6.5. Configure SIP Parameters

From the **VIP-102B IP Solutions Setup Tool**, navigate to the **SIP** tab of the Universal Page Interface and configure the parameters as follows.

•	Transport:	Set to Accept: TLS, Originate: TLS.
•	Phone Number:	Set to number that will be routed to V-9972 (e.g., 41600).
•	Description:	Provide optional description.
•	Authentication Name:	Leave blank.
	Secret:	Leave blank.
	Realm:	Set to SIP domain (e.g., avaya.com).
•	Validate Remote	
	Certificate:	Enable this option so that V-9972 validates the remote
		TLS certificate installed in Section 6.4.
•	Primary Server:	Set to IP Office IP address (i.e., 10.64.102.90).
•	Port:	Set to TLS port (e.g., 5061).
•	Register:	Disable this option.
•	Max Calls:	Specify maximum number of calls (e.g., 4). For example,
		V-9972 could establish an intercom call to the IP speaker
		and then a higher priority paging call to the same IP
		speaker. In addition, V-9972 could establish up to four
		calls to four different IP speakers (not tested).
•	SRTP:	Enable SRTP and then select Media Encryption
		Mandatory.
•	Auto Destination:	Set to the number that should be dialed when the call
		button on the VIP-430A IP Wall Speaker is pressed.

Accept the values in the remaining fields and click **Apply**.

VIP-102B IP Solutions Setup Tool						_		×
<u>File</u> <u>Communications</u> <u>Device</u> Sec <u>u</u> ri	ity <u>P</u> rogramming <u>S</u> ystem Co <u>r</u>	flicts <u>H</u> elp						
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Job Information	1							
i ⊡·····∰in Miscellaneous i ⊡······€ Speaker Plus (TB)	Summary Properties Network Time	e System Chan	nels Group Membership SIP					
00-D0-5F-05-B4-1B	Transport: Accept: TLS, Origin	nate: TLS	~					
□ Oliversal Page Interface	1 2 3 4							
	Phone Number:	41600						
	Description:	VIP-430A						
	Authentication Name:							
	Secret:							
	Realm:	avaya.com		Validate Ren	note Certificate:			
	SIP Servers:		Server		Port			
		Primary	10.64.102.90		5061			
	Register:	Backup 1			5061	_		
	DNS SRV:	Backup 2			5061	_		
		P Backup 5			5001	-		
	Max Calls:	4	SRTP: 🗹 🛛	Media Encryptio	n Mandatory	~		
	Busy Message:					~		
	Call Fwd Busy (302):		Ring Timeout	(secs): None		~		
	Outbound Proxy:		Outbound	d Port: 5061				
	Keep Alive Timer (secs):	600	Options Timer	(secs): 40				
	SIP Port:	5061	Idle Timeout	(secs): 0		~		
	RTP Port:	20000	Max Call Timer	(secs): 0		~		
	Night Ring:	Night	Ring Group:			\sim		
locard -	CID Number:	41600						
Legend +	CID Name:	VIP-430A						
E - Status Nomal	Auto Destination:	41501						
- Verification Required								
 Error Status Update Required 	Channel Priority:	Medium	\sim					
- Reset Required								
 Rescan Required Invalid Password 								
🖃 - Firmware Suggested								
	Defaults Statue			Conv	Apply	& Cancel		
Find device in tree				C TON	A Unda A			
2 devices detected, 2 devices loaded			Ethern	et : Realtek PC	le GBE Family Co	ontroller - 1	92.168.10	0.251 _{.::}

6.6. Verify Codec Settings

Navigate to the **Channels** tab shown below. The Codec Type should be set G.711, currently the only option supported with VIP-430A IP Wall Speaker.

VIP-102B IP Solutions Setup Tool			—	
<u>File</u> <u>Communications</u> <u>Device</u> Sec <u>u</u> rity	<u>P</u> rogramming <u>S</u> ystem Co	nflicts <u>H</u> elp		
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Job Information				
Miscellaneous M	Summary Properties Network Tin	e System Channels Group Membership	SIP	
	Dial Code:	802		
	Description:			-
	CID Number:	802		\exists
		002		\exists
	CID Name:			
	Auto Destination:			
				_
	Codec Type:	G.711		~
	Call Fwd Busy:			
	Call Fwd No Answer:		4 Rings	\sim
Legend 👻				
- Status Unknown - Status Normal				
- Verification Required				
Error Status	Group Membership Sar	nple Page	Copy 🖉 Apply 💥 🤇	ancel
 Opdate Required Reset Required 				
E - Rescan Required				
 Invalid Password Firmware Suggested 				
Find device in tree				
2 devices detected, 2 devices loaded		Etherne	t : Realtek PCIe GBE Family Controller - 19	2.168.100.251

6.7. Update Universal Page Interface with the New Configuration

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

			- · · · · · · ·			
IVIP-102B IP Solutions Setup Too	bl			_		\times
<u>File</u> <u>Communications</u> <u>D</u> evic	e Sec <u>u</u> rity <u>P</u> rogramming	<u>S</u> ystem	Co <u>n</u> flicts <u>H</u> elp			
🔚 📄 🔲 🔍 🕠	co /4 der 🌆 🗖	1 🛜 🗔				
· 38 📑 🛄 🔹	// 0.2					
Job Information						
Miscellaneous	Summary Propertie	es Network	Time System Channels Group Membership	SIP		
	в					
Universal Page Interface	e			(<u> </u>	
	Scan Device	1				
	<u>R</u> eset Device		00-D0-5F-05-CB-C5			
	<u>U</u> pdate Device		192.168.100.197			
	Verify Device		Universal Page Interface			
	Recover Device		1			
	Assign ID Address	urce:	Device - (data is retrieved from the device)		`	/
	Program Firmware		3.00.14			
	Eemove Device		1.0.1			
	Remove And Ignore		DSPG - 1 : 9970			
Legend	View Cached Files		-			
Status Unknown Status Normal	view Cached <u>Files</u>	1	Product Name = V-9972			
- Verification Required			startup=1.0.1			
Error Status	Version Deta	ile ·	softwarerev=3.0.14			
Reset Required	Verbiert Deta		siprev=sw1.70.12			
- Rescan Required						
 Firmware Suggested 					~	
Find during in two						
rina device in tree						
🔒 2 devices detected, 2 devices lo	oaded		Ethernet : Realtek PCIe GBE Family Cont	troller - 19	92.168.10	0.251

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

Updating Device		
	Updating device	
	<u>C</u> ancel	

Solution & Interoperability Test Lab Application Notes ©2022 Avaya Inc. All Rights Reserved. 28 of 32 V9972-IPO11-TRK 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 completed, the window will disappear.

🖃 Wai	iting For Reset To Complete			×
	Waiting for devices to	finish resetting	Cancel Wait	Never Wait
	Name 00-D0-5F-05-CB-C5	MAC Address 00-D0-5F-05-CB-C5	Type Universal Page Inter	face
Reset co	omplete for 0 of 1 devices			

7. Verification Steps

This section provides the tests that may be performed to verify proper configuration of Valcom V-9972 Universal Paging Interface with Avaya IP Office Server Edition.

1. From **IP Office System Status**, expand **Trunks** in the left pane and select the SIP line from **Section 5.4**, in this case "5".

Verify that the **SIP Trunk Summary** screen shows **Media Stream** is set to *SRTP*, **Layer 4 Protocol** is set to *TLS*, and **Current State** of channels is set to *Idle*.



- 1. Place a call to the V-9972 and at the dial tone, enter the dial code for the IP speaker to establish an intercom call from an Avaya IP deskphone to a Valcom speaker. Verify two-way audio. Terminate the call from the Avaya IP deskphone or by pressing the call button on the IP speaker.
- 2. Place a call to the V-9972 and at the dial tone, enter the dial code a group page code to establish a one-way paging call from an Avaya IP deskphone to IP speaker(s). Verify one-way audio. Terminate the call from the Avaya IP deskphone.

JAO; Reviewed:
SPOC 5/20/2022

Solution & Interoperability Test Lab Application Notes ©2022 Avaya Inc. All Rights Reserved. 3. Place an intercom call by pressing the call button on the IP speaker. Verify two-way audio to the call destination. Terminate the call.

8. Conclusion

These Application Notes described the configuration steps required to integrate Valcom V-9972 Universal Paging Interface with Avaya IP Office Server Edition using a SIP trunk. Intercom and group paging calls were established with Valcom V-9972 Universal Paging Interface, Valcom VIP-430A IP Wall Speakers, Avaya H.323 / SIP Deskphones, and the PSTN. All feature and serviceability test cases were completed successfully.

9. References

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

- [1] Administering Avaya IP Office Platform using Manager, Release 11.1, August 16, 2021.
- [2] Valcom VIP-102B IP Solutions Setup Tool Version 8.4.0.0 Reference Manual, Revision 17 3/16/22, available at https://www.valcom.com/resources/documents-manuals.
- [3] *Valcom V-9972 Universal Page Interface Configuration Guide*, Rev. 3.1, available at <u>https://www.valcom.com/resources/documents-manuals</u>.

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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 <u>devconnect@avaya.com</u>.

ATTACHMENT 1



Declaration of Conformance

May 20, 2022

Jeff Gartner Senior Manager DevConnect Program Avaya

Dear Jeff Gartner:

We, Valcom Inc, declare under sole responsibility that product series named Universal Paging Adapter, including product models V-9972, V-9972-2 or VRCPA 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