

Avaya Solution & Interoperability Test Lab

Application Notes for Snom M900 Multicell DECT Phones with Avaya IP Office - Issue 1.0

Abstract

These Application Notes describe the configuration steps required to integrate Snom M900 Multicell DECT Phones with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. The Snom M900 DECT base station was connected to the LAN which, in turn, registered M-series DECT phones to Avaya IP Office via SIP. The base station converts IP protocol to DECT protocol and transmits phone calls to and from the M-series DECT phones. For the compliance test, the Snom M65 DECT Handsets were used.

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 Snom M900 Multicell DECT Phones with Avaya IP Office Server Edition 11.1 and Avaya IP Office 500 V2 Expansion System 11.1. The Snom M900 DECT base station was connected to the LAN which, in turn, registered M-series DECT phones to Avaya IP Office via SIP. The base station converts IP protocol to DECT protocol and transmits phone calls to and from the M-series DECT phones.

For the compliance test, the Snom M65 DECT Handsets were used. There are other DECT M-Series handsets that share the same firmware version as the Snom M65 DECT Handset, and therefore the testing also applies to them. See Attachment 1 for additional details.

2. General Test Approach and Test Results

The interoperability compliance test included feature and serviceability testing. The feature testing focused on establishing calls between Snom M65 DECT Handsets and Avaya SIP/H.323 telephones, and exercising basic telephony features, such as hold, mute, and transfer. The M65 handsets gained network access via the M900 DECT base station. Additional telephony features, such as call forward, follow me, call park/unpark, and call pickup were also verified using Avaya IP Office Short Codes.

The serviceability testing focused on verifying that the Snom M900 Multicell Base Station came back into service after re-connecting the Ethernet or rebooting the Snom M65 DECT Handsets.

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 this Application Note, the interface between Avaya systems and Snom M900 Multicell DECT Phones utilized signaling encryption using TLS, but no media encryption (RTP was used).

2.1. Interoperability Compliance Testing

Interoperability compliance testing covered the following features and functionality:

- SIP registration of M65 DECT handsets with IP Office Server Edition and IP Office 500 V2 Expansion System. M900 DECT base station controls the traffic in the air and works as the link between the M65 DECT handsets and IP Office.
- Calls between M65 DECT handsets and Avaya SIP/H.323 deskphones with Direct Media enabled and disabled. Direct Media was verified with M65 DECT handsets and Avaya SIP deskphones only.
- Calls between M65 DECT handsets and the PSTN.
- Calls with TLS enabled.
- TLS using secure PFS cipher of TLS_ECDHE_RSA_WITH_AES_128_GCM_SHA256.
- Support of G.711 and G.722 codecs.
- Proper recognition of DTMF tones.
- Basic telephony features, including hold, mute, redial, multiple calls, blind/attended transfer, and long duration calls.
- Voicemail coverage, MWI support, and logging into voicemail system to retrieve messages.
- Extended telephony features using IP Office short codes for Call Forward, Follow Me, Call Park/Unpark, and Call Pickup.
- Proper system recovery after a restart of M900 Multicell base station and M65 DECT handsets.

2.2. Test Results

All test cases passed with the following observations noted:

- Currently, the Snom M900 Multicell Base Station doesn't support TLS authentication with a Subject Alternate Name (SAN) in the certificate. Therefore, the M900 was configured to accept all certificates by disabling the Use Only Trusted Certificates option under Security in the M900 configuration as described in Section 6.6.
- Secure RTP (SRTP) is not supported. When an Avaya H.323 deskphone originates a call to an M65 DECT handset, places the call on hold, and then resumes the call, there's only one-way audio from the Avaya H.323 deskphone to the M65 DECT handset when SRTP is enabled. The workaround is to disable SRTP. Avaya is investigating this issue.

2.3. Support

For technical support on the Snom M900 Multicell DECT Phones, contact Snom Support via phone, email, or website.

- **Phone:** +1 (339) 227-6160 Option 2
- Web: <u>https://service.snom.com</u>
- Email: <u>supportusa@snom.com</u>

3. Reference Configuration

Figure 1 illustrates a sample configuration consisting of Snom M900 Multicell DECT Phones with Avaya IP Office Server Edition and Avaya IP Office 500 V2 (Expansion System). The Snom M65 DECT Phones registered with Avaya IP Office via SIP through the Snom M900 Multicell Base Station. Avaya Embedded Voicemail served as the voicemail system. Avaya 96x1 Series H.323 Deskphones and an Avaya J100 Series SIP Deskphones were used for placing and receiving calls.



Figure 1: Avaya SIP Network with Snom M900 Multicell DECT Phones

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	11.1.1.0.0 build 209
Avaya IP Office 500 V2 Expansion	11.1.1.0.0 build 209
Avaya 96x1 Series IP Deskphones	6.8304 (H.323)
Avaya J100 Series IP Deskphones	4.0.7.0.7 (SIP)
Snom M900 Multicell Base Station	05.30/B0002
Snom M65 DECT Handsets	05.30/B0002

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

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 M65 handsets
- Administer SIP user for M65 handsets

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 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 M900 Multicell base station.



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5.2. Administer SIP Registrar

Select the **VoIP** sub-tab. Ensure that **SIP Registrar Enable** is checked and enter a valid **Domain Name**. In the compliance testing, the **Domain Name** field was set to *avaya.com*. TLS transport protocol was enabled for the **Layer 4 Protocol**, which was also used by the M900 Multicell base station.

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5.3. Administer SIP Extension for M65 DECT Handsets

From the configuration tree in the left pane, right-click on **Extension** and select New \rightarrow SIP from the pop-up list (not shown) to add a new SIP extension. Enter the desired extension for the **Base Extension** field as shown below. In this example, M65 DECT handset was assigned extension 41510. This is the extension that the M65 DECT handset will use to register the handsets with IP Office Server Edition.

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Select the **VoIP** tab. Configure the codec selection as shown below. 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. Disable **Media Security**.

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5.4. Administer SIP User for M65 DECT Handsets

From the configuration tree in the left pane, right-click on **User** and select **New** from the pop-up list (not shown). Enter desired values for the **Name** and **Full Name** fields. For the **Extension** field, enter the SIP extension created above.

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Solution & Interoperability Test Lab Application Notes ©2021 Avaya Inc. All Rights Reserved. Select the **Voicemail** tab and select **Voicemail On** to enable voicemail for the M65 DECT handset. Specify a **Voicemail Code** to be used when logging into voicemail.

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

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Select the **Supervisor Settings** sub-tab and enter a desired **Login Code**. The **Login Code** is the password that will be used by the M65 DECT handset to register the SIP extension with IP Office Server Edition.

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6. Configure Snom M900 Multicell DECT Phones

This section provides the procedure for configuring the M900. The procedure covers the following areas:

- Open Web User Interface
- Administer Network Settings
- Administer Country/Time Settings
- Administer Servers
- Administer Extensions
- Administer Security

6.1. Open Web User Interface

The Snom M900 Multicell Base Station was configured through the web user interface by using the URL "http://ip-address" in an Internet browser window, where "ip-address" is the IP address of the base station. Log in using the appropriate credentials and then click **OK**.

Sign in			
http://192.16	i8.100.191		
Your connect	tion to this site is not private		
Username			
Password			
		Sign in	Cancel

6.2. Administer Network Settings

To configure network settings, click **Network** in the left pane. The M900 is pre-configured to use DHCP, but a static IP address may be used. For the compliance test, DHCP was used as shown below.

snom	M900				
Home/Status	Network Setting	gs			
Extensions					
Somore	IP Settings			NAT Settings	
Servers	DHCP/Static IP:	DHCP	~	Enable STUN:	Disabled 🗸
Network	IP Address:	192.168.100.19	1	STUN Server:	
Management	Subnet Mask:	255.255.255.0		STUN Bindtime Determine:	Enabled V
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Firmware Update	DNS (Primary):	192.168.1.1		Enable RPORT:	Enabled V
Country	DNS (Secondary):			Keep alive time:	90
	MDNS:	Disabled	~		
Security				SIP/RTP Settings	
Central Directory	VLAN Settings			Use Different SIP Ports:	Disabled 🗸
Multi Coll	ID:	0		RTP Collision Detection:	Disabled 🗸
Multi Cell	User Priority:	0		Always reboot on check-	Disabled 🗸
Dial Plans	Synchronization:	Enabled	~	Outbound Proxy Mode:	Use Always 🗸
Repeaters				Failover SIP Timer B:	5
Alaem	DHCP Options			Failover SIP Timer F:	5
Aldi III	Plug-n-Play:	Enabled	~	Failover Reconnect Timer:	60
Statistics				Local SIP port:	5060
Generic Statistics	TCP Options			SIP ToS/QoS:	0xA0
	TCP Keep Alive Interval:	120		RTP port:	50004
Diagnostics				RTP port range:	254
Configuration	Discovery			RTP ToS/QoS:	0xA0
Syslog	LLDP-MED Send:	Enabled	~	SIP registration mode:	Plug-n-Play 🗸
575109	LLDP-MED Send delay:	30			
SIP Log	VLAN via LLDP-MED:	Enabled	~		
Logout					
	Save and Reboot	t	Save	Cancel	

6.3. Administer Country/Time Settings

Navigate to **Country** in the left pane to configure the Time Server and set the correct time.

Note: It is important to use correct date and time of the system when using trusted certificates. In case of undefined time/date, the certificate validation can fail.

snom	M900	
Home/Status	Country/Time Settin	qs
Extensions	Select country:	
Servers	State / Region:	New Jersey 🗸
	Notes:	
Network	Select Language:	English 🗸
Management		
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Country	Time Server:	168.61.215.74
Security	Allow broadcast NTP:	
Security	Refresh time (h):	1
Central Directory	Set timezone by country/region:	
Multi Cell	Timezone:	-5:00 🗸
	Set DST by country/region:	
Dial Plans	Daylight Saving Time (DST):	Automatic 🗸
Repeaters	DST Fixed By Day:	Use Month and Day of Week 🗸
	DST Start Month:	March 🗸
Alarm	DST Start Date:	0
Statistics	DST Start Time:	2
	DST Start Day of Week:	Sunday 🗸
Generic Statistics	DST Start Day of Week Last in Month	Second First In Month 🗸
Diagnostics	DST Stop Month:	November V
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SIP LOg	Save and Reboot	Save
Logout		Garcel

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6.4. Administer Servers

To configure SIP server, click **Servers** in the left pane, and then click **Add Server** (not shown). Configure the following fields:

- Server Alias: Specify a server alias (e.g., *ipose*).
- **Registrar:** Specify the SIP server proxy IP address (e.g., 10.64.102.90).
 - Specifying the port number is optional.
 - **SIP Transport:** Set to *TLS*.
- **Codec Priority:** Specify the codec priority. For the compliance test, G.711 and G.722 were verified.

Note: With the configuration specified above, the M900 will send the IP address in the SIP URI and From/To headers of SIP Invite message. To send the domain instead, configure the domain (e.g., *avaya.com*) in **Registrar** and the SIP server proxy IP address in **Outbound Proxy**.

snom	M900			
Home/Status	Servers			
Extensions		ipose:		
Servers	devcon-sm 10.64.102.117	• Server Alias:	ipose	
Scivers	devcon-shc	NAT Adaption:	Enabled	~
Network	10.64.101.102	Popistrary	10 64 102 00	
Management	ipose:	Outhourd Pressu	10.04.102.50	
	10.64.102.90			
irmware Update	ipo500v2	Conference Server:		
Country	192.168.100.90	Call Log Server:		
	Add Server	Music on Hold Server:		
ecurity	Remove Server	Reregistration time (s):	3600	
Central Directory		Deregister After Failback:	Disabled	~
		SIP Session Timers:	Enabled	~
Multi Cell		Session Timer Value (s):	3600	
Dial Plans		Dial Plan ID:	2	~
		Use SIP as XSI Authentication:	Disabled	~
Repeaters		SIP Transport:	TLS	~
larm		Signal TCP Source Port:	Enabled	~
10111		Use One TCP Connection per SIP Extension:	Disabled	~
Statistics		RTP from own base station:	Disabled	~
Canonia Chatistics		Keep Alive:	Enabled	~
seneric Statistics		Show Extension on Handset Idle Screen:	Enabled	~
Diagnostics		Hold Behaviour:	RFC 3264	~
C		Remote Ring Tone Control:	Enabled	~
computation		Attended Transfer Behaviour:	Hold 2nd Call	~
Syslog		Semi-Attended Transfer Behaviour:	Allow Semi-Attended Transfer	~
		Use Own Codec Priority:	Disabled	~
SIP Log		DTMF Signalling:	RFC 2833	~
Logout		DTMF Payload Type:	101	
		Remote Caller ID Source Priority:	PAI - FROM	~
		Enable Blind Transfer:	Enabled	~
		XSI User Services:	Enabled	~
			G711U	
		Codec Priority:	G711A	
		- Max number of codecs is 5	G720 G722	-

JAO; Reviewed: SPOC 1/14/2022

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• Secure RTP: Set to *Disabled*.

Accept the default values for the remaining fields. Restart the M900 after saving the changes to Servers.

PTD Dacket Cize:	20 mc		~
NTP PAUKEL DIZE.	20 1115		•
Secure RTP:	Disabled		~
Secure RTP Auth:	Enabled		~
	AES_CM_128_HMA AES_CM_128_HMA	C_SHA1_32 C_SHA1_80	*
SRTP Crypto Suites:			-
	Up	Down	
Media Security:	Disabled		~
Media Security only for TLS:	Disabled		\sim
Client Initiated Connections (RFC5626):	Disabled		~

6.5. Administer Extensions

To create an extension for an M65 handset, click **Extensions** in the left pane to display the **Extensions** page below. Click **Add extension.**

snom	M9	M900											
Home/Status	Ext	Extensions											
Extensions	AC:	AC: 0000											
Servers		Sav	e	Cancel									
Network	Add e	xtensi	on										
Management		<u>Idx</u>	<u>IPEI</u>	<u>Handset</u> State	<u>Handset</u> <u>Type</u>	<u>FWU</u> Prograss		<u>VoIP</u> Idv	<u>Extension</u>	<u>Display Name</u>	<u>Server</u>	<u>Server Alias</u>	<u>State</u>
Firmware Update				State	<u>FW Info</u>	FTOGICSS		100					
Country		1	0328DCAF32	Present@RPN00	M65 530.2	Complete		1	<u>41510</u>		10.64.102.90	ipose	SIP Registered@RPN00
Security		2	0328DCAF77	Present@RPN00	M65 530.2	Complete		2	<u>41511</u>		10.64.102.90	ipose	SIP Registered@RPN00
Central Directory		3	0328DCAF5A	Present@RPN00	M65 530.2	Complete		3	<u>41512</u>		10.64.102.90	ipose	SIP Registered@RPN00
Multi Cell	Check All /					Check All Extensions /							
Dial Plans	Uncheck All						Uncheck All Extensions						
Popostors	With selected: Delete Handset(s) Register Handset(s) Dereoister Handset(s), Start SIP Registration(s), SIP Delete Extension(s)												

In the Add Extension page, configure the following fields:

- Line name:
- Extension:
- Authentication User Name:
- Authentication Password:
- Mailbox Name:
- Mailbox Number:
- Server:

Specify a line name for extension (e.g., 41510).
Enter SIP extension (e.g., 41510).
Specify the user name (e.g., 41510) used to register with Session Manager.
Specify the password used to register with Session Manager.
Specify the mailbox number for the SIP user (e.g., 41510).
Specify the voicemail number (e.g., *17).
Specify the SIP server proxy configured in Section 6.4.

snom **M900**

Edit extension

Extensions					
LACCIDIOID	Line name:	41510			
Servers	Handset:	Handset Idx 1 🗸			
Network	Push-To-Talk:	Disabled V			
	Extension:	41510			
Management	Authentication User Name:	41510			
Firmware Update	Authentication Password:	•••••			
	Display Name:				
Country	XSI Username:				
Security	XSI Password:	•••••			
a	PIN:				
Central Directory	Mailbox Name:	41510			
Multi Cell	Mailbox Number:	*17			
Dial Planc	Server:	ipose: 10.64.102.90 🗸			
	Call waiting feature:		Enabled V		
Repeaters	BroadWorks Shared Call Appearance:		Disabled 🗸		
Alarm	BroadWorks Feature Event Package:		Disabled 🗸		
	UaCSTA:		Disabled 🗸		
Statistics	Forwarding Unconditional Number:		Disabled 🗸		
Generic Statistics	Forwarding No Answer Number:		Disabled 🗸	90	s
	Forwarding on Busy Number:		Disabled 🗸		
Diagnostics					
Configuration	Save Cancel				
Syslog					

JAO; Reviewed: SPOC 1/14/2022

Logout

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6.6. Administer Security

Navigate to Security in the left pane to disable **Use Only Trusted Certificates** as shown below. This will allow all certificates received from Session Manager to be accepted. This setting must be disabled, because the M900 currently doesn't support a SAN in the certificate as mentioned in **Section** Error! Reference source not found. Since the **Use Only Trusted Certificates** option is disabled, there's no need to download TLS certificates to the M900.

Note: It is important to use correct date and time of the system when using trusted certificates. In case of undefined time/date, the certificate validation can fail.

Alarm	Trustee	l Root Certificates						
Alum	Idx	Issued To	Issued By	Valid Until				
Statistics	0	Avaya	Avaya	23/03 08:59:21 2040				
Generic Statistics	1	Chambers of Commerce Root	Chambers of Commerce Root	30/09 16:13:44 2037				
Diagnostics	2	Chambers of Commerce Root - 2008	Chambers of Commerce Root - 2008	31/07 12:29:50 2038				
Diagnostics	3	Global Chambersign Root	Global Chambersign Root	30/09 16:14:18 2037				
Configuration	4	Global Chambersign Root - 2008	Global Chambersign Root - 2008	31/07 12:31:40 2038				
Suclea	5	Actalis Authentication Root CA	Actalis Authentication Root CA	22/09 11:22:02 2030				
Sysiog	6	Amazon Root CA 1	Amazon Root CA 1	17/01 00:00:00 2038				
SIP Log	7	Amazon Root CA 2	Amazon Root CA 2	26/05 00:00:00 2040				
	8	Amazon Root CA 3	Amazon Root CA 3	26/05 00:00:00 2040				
Logout	9	Amazon Root CA 4	Amazon Root CA 4	26/05 00:00:00 2040				
	10	Starfield Services Root Certificate Authority - G2	Starfield Services Root Certificate Authority - G2	31/12 23:59:59 2037				
	11	IdenTrust Public Sector Root CA 1	IdenTrust Public Sector Root CA 1	16/01 17:53:32 2034				
	12	ISRG Root X1	ISRG Root X1	04/06 11:04:38 2035				
	Check All /Uncheck All With selected: Delete Certificate(s) Import Root Certificate: Filename: Choose File No file chosen Use Only Trusted Certificates: Disabled Save Cancel							
	Secure Web Server: HTTPS: Disabled ~							
	Password:							
	Username	admin						
	Current Pa	ssword:						
	New Password:							
	Confirm Pa	assword:						
	Sa	ve Clear						

7. Verification Steps

This section provides the tests that can be performed to verify proper configuration of Avaya IP Office and Snom M900 Multicell DECT Phones.

1. Verify that M65 handsets have successfully registered with IP Office. In **IP Office System Status**, navigate to the SIP extension and verify **Media Stream** is set to *RTP*, **Layer 4 Protocol** is set to *TLS*, and **Current State** is shown as *Idle*.



2. Alternatively, the SIP registration status may be verified by navigating to **Extensions** in the M900 web user interface.

รกิงก	M9	M900											
Home/Status	Ext	Extensions											
Extensions	AC:	AC: 0000											
Servers		Save Cancel											
Network	Add e	extensi	on										
Management		<u>Idx</u>	<u>IPEI</u>	<u>Handset</u>	<u>Handset</u> <u>Type</u>	<u>FWU</u>		<u>VoIP</u>	Extension	<u>Display Name</u>	<u>Server</u>	<u>Server Alias</u>	<u>State</u>
Firmware Update				State	<u>FW Info</u>	Progress		10x					
Country		1	0328DCAF32	Present@RPN00	M65 530.2	Complete		1	<u>41510</u>		10.64.102.90	ipose	SIP Registered@RPN00
Security		2	0328DCAF77	Present@RPN00	M65 530.2	Complete		2	<u>41511</u>		10.64.102.90	ipose	SIP Registered@RPN00
Central Directory		3	0328DCAF5A	Present@RPN00	M65 530.2	Complete		3	<u>41512</u>		10.64.102.90	ipose	SIP Registered@RPN00
Multi Cell	Check All / Check All Extensions /												
Dial Plans	Uncheck All						Uncheck All Extensions						
Repeaters	With selected: Delete Handset(s) Register Handset(s) Deregister Handset(s) Start SIP Registration(s) SIP Delete Extension(s)												

3. Establish a call between M65 handset and a local Avaya SIP deskphone. In **IP Office System Status**, navigate to the SIP extension and verify that the **Connection Type** is *Direct Media* as shown below, if SRTP is disabled on the Avaya SIP deskphone. If SRTP is enabled on the Avaya SIP deskphone, the **Connection Type** would be set to *VCM (SRTP)*.

IJ Avaya IP Office System	Status - devcon-ipose (10.64.102.90) - IP (Office Linux PC 11.1.1.0.0 build 209	– 🗆 X
AVAYA		IP Office System Status	
Help Snapshot LogOff Ex	it About		
E System Alarms (30) Extensions (7) 41000 41001 41501 41502 ▲41510 41514	Extension Number: IP address: Standard Location: Registrar: Telephone Type:	Extension Status 41510 192.168.100.191 None Primary Unknown SIP Device	
41511 41512 Trunks (3) Active Calls Resources Voicemail IP Networking Locations	User-Agent SIP header: Media Stream: Layer 4 Protocol: Current User Extension Number: Current User Name: Forwarding: Twinning: Do Not Disturb: Message Waiting: Number of New Messages: Phone Manager Type: SIP Device Features: License Reserved:	snomM900/05.30.0002 (MAC=000413B66E9B; SER= 00000; HW=2) RTP TLS 41510 Snom41510 Off Off Off Off 0 None REFER,UPDATE No	
	Last Date and Time License Allocated: Packet Loss Fraction: Jitter: Round Trip Delay: Call Ref Current State 208 Connected	10/11/2021 12:26:05 PM Connection Type: Codec: Remote Media Address: Time in State Calling Number or Called Direction Number 00:00:30 Outgoing	Direct Media G711 Mu 192. 168. 100. 195 Other Party on Call Extn 41501, sip41501
	Trace Trace All Pause	Ping Call Details Print Save As	12:42:05 PM Online

4. While the call is active, basic telephony features can be exercised to verify proper operation.

8. Conclusion

These Application Notes describe the configuration steps required to integrate Snom M900 Multicell DECT Phones with Avaya IP Office. The Snom M900 Multicell DECT Phones registered to Avaya IP Office Server Edition or Avaya IP Office 500 V2 Expansion System. Calls were then established to H.323 / SIP deskphones and the PSTN with TLS. In addition, basic telephony features were verified. All feature and serviceability test cases were completed successfully with observations noted in **Section 2.2**.

9. References

This section references the Avaya documentation relevant to these Application Notes. The Avaya product documentation is available at <u>http://support.avaya.com</u> and the Snom product documentation is available at <u>https://service.snom.com/display/wiki/M900</u>.

- [1] Administering Avaya IP Office Platform with Manager, Release 11.1.1, Issue 28.1.1, June 2021.
- [2] Snom M900 and M900 Outdoor Base Station Admin and Installation Guide v1.03.

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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



VTech Technologies Canada Ltd.

Date: November 15, 2021

Declaration of Conformance

We, VTech Technologies Canada LTD., declare under sole responsibility that product series DECT M-Series handsets all share the same firmware version. Therefore; the products are expected to behave in the same manner. The differences between the different models in the series are detailed in the table below.

Model	Description
M25	DECT Office Handset, color display, and 3.5 mm headset jack
M65	DECT Professional Handset, Wideband speakerphone
M70	DECT Ruggedized Office Handset, HD Audio, Color LCD, Bluetooth, Alarm
M80	DECT M80 Ruggedized Handset, IP65 Rating, Bluetooth, Alarm
M85	DECT Industrial Handset, IP65 Rating, Bluetooth, Alarm
M90	Antibacterial DECT Handset, JIS-Z 2801 tested, MIL-STD-810g 516.6 tested, IP65
	Rating, Bluetooth, Alarm

Please do not hesitate to contact should you require further information.

Thank you,

Ralph Tischler Director of Engineering Vtech Technologies Canada Ltd 604-233-5203

222 - 13888 Wireless Way Richmond, BC V6V 0A3 Tel: 604-273-5131 Fax: 604-273-1425