

Avaya Solution & Interoperability Test Lab

Application Notes for Configuring Yealink VP530 SIP Video Phone Version 23.70.0.40 with Avaya IP Office 500 V2 Release 8.1 – Issue 1.0

Abstract

These Application Notes describe the procedures for configuring Yealink VP530 SIP Video Phone which was compliance tested with Avaya IP Office 500 V2 Release 8.1. The overall objective of the interoperability compliance testing is to verify Yealink P530 SIP Video Phone functionalities in an environment comprised of Avaya IP Office and various Avaya H.323, Avaya SIP, Avaya Digital and analog telephones.

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 procedures for configuring Yealink VP530 SIP Video phone which was compliance tested with Avaya IP Office 500 V2 Release 8.1. The Yealink VP530 SIP Video phone provides a powerful and flexible IP communication solution for Ethernet TCP/IP networks, delivering excellent voice quality. The compliance test used the Avaya IP Office 500 V2 Release 8.1 to test with Yealink VP530 SIP Video phone.

Application Notes assume that Avaya IP Office is already installed and basic configuration steps have been performed. Only steps relevant to this compliance test will be described in this document. For further details on configuration steps not covered in this document, consult **Section 9**.

2. General Test Approach and Test Results

The general test approach was to place calls to and from Yealink VP530 SIP Video phone and exercise basic telephone operations. The main objectives were to verify the following:

- Registration.
- Codec Negotiation (G.711MU, G711MA, G722, G729, and G723).
- Basic video call between VP530 phones and between VP530 and Avaya Flare® and IP Office Softphone.
- Telephone functions: Inbound and outbound, hold/resume, call park/pickup, blind and attended transfers, call waiting and conference calls.
- Call termination (origination/destination).
- DTMF method: RFC2833 and INBAND.
- Voicemail and Messaging Waiting Indicator.
- T1/ISDN PSTN Calls.
- Serviceability.

DevConnect Compliance Testing is conducted jointly by Avaya and DevConnect members. The jointly-defined test plan focuses on exercising APIs and standards-based interfaces pertinent to the interoperability of the tested products and their functionalities. DevConnect Compliance Testing is not intended to substitute a full product performance or feature testing performed by third party vendors, nor is it to be construed as an endorsement by Avaya of the suitability or completeness of a third party solution.

2.1. Interoperability Compliance Testing

All test cases were performed manually. The general approach was to place various types of calls to and from Yealink VP530 SIP Video phone and Yealink VP530 SIP Video phone's operations such as inbound calls, outbound calls, hold, transfer, forward, conference and VP530 interactions with Avaya IP Office, Avaya SIP, H.323 and Digital telephones were verified. For serviceability testing, failures such as cable pulls and resets were applied.

2.2. Test Results

All test cases were passed and there is one note as described below.

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• Avaya IP Office does not support local call forward on third party SIP phone therefore the call forward feature needs to be provisioned in the IP Office Manager.

2.3. Support

Technical support for Yealink VP530 SIP Video phone can be obtained by contacting Yealink as below.

Yealink Network Technology Co.,Ltd. Address: 4th-5th Floor, South Building, No. 63 Wanghai Road, 2nd Software Park, Xiamen, China (361008). Phone: +86-592-5702000 Email: <u>support@yealink.com</u> Website: <u>http://www.yealink.com</u>

3. Reference Configuration

Figure 1 illustrates a sample configuration consisting of an Avaya IP Office system and Yealink VP530 SIP Video phone. The IP office has SIP trunk to Avaya Communication Server 1000 to exercise test cases over SIP trunk and has T1/ISDN trunk to Avaya Aura® Communication Manager to exercise test cases over PRI trunk. The testing used Voice Mail Pro which is a feature on the IP Office acting as a voice mail system.



Figure 1: Test Configuration of Yealink VP530 SIP Video Phone with Avaya IP Office

4. Equipment and Software Validated

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

Equipment	Software
Avaya IP Office 500 V2	8.1 Build 52
Avaya IP Office Manager	8.1 (52)
Avaya S8800 Communication Manager Server	R016x.00.1.510.1
Avaya Media Gateway G450	31.22.0.1
Avaya Communication Server 1000 CPPM	Release 7.5
Avaya S8800 System Manager Server	6.1 SP6
Avaya S8800 Session Manager Server	6.1 SP6
Avaya Flare® Experience	1.1.0
Avaya IP Office Softphone	3.2.3.20
Avaya IP H323 9650 Phone	3.104S
Avaya IP H323 9641 Phone	6.2.119
Avaya IP H.323 1608L Phone	1.302S
Avaya IP SIP 1140E Phone	4.03.12
Avaya Digital 2420 Phone	-
Avaya Analog Phone	-
YeaLink T28P SIP Phone	2.70.0.60
Yealink VP530 SIP Phone	23.70.0.40

Note: Testing was performed with IP Office 500 v2 R8.1, but it also applies to IP Office Server Edition R8.1 (single site configuration only).

5. Configure Avaya IP Office

This document assumes that Avaya IP Office system is already installed, configured, and operating. For more information on how to configure the Avaya IP Office system please refer to **Section 9**. This section provides the procedures for configuring Avaya IP Office. The procedures include the following areas:

- Verify IP Office license
- Obtain LAN IP address
- Administer SIP registrar
- Administer SIP extensions
- Administer SIP users

These steps are performed from the Avaya IP Office Manager.

5.1. Verify IP Office License

From a PC running the Avaya IP Office Manager application, select **Start > All Programs > IP Office > Manager** (not shown) to launch the Manager application. Select the proper IP Office system if there are more than one IP Office system, and log in with the appropriate credentials.

The Avaya IP Office Manager screen is displayed. From the configuration tree in the left pane, select **License > 3rd Party IP Endpoints** to display the Avaya IP endpoints screen in the right pane. Verify that the License Status field is set to **Valid**.

IP Offices	×××* 	3rd Party IP Endpoints	📸 • 🗙 • < >
BOOTP (1) P→ Operator (3)	Licenses		
DevCon IPO 1	License Key	rUuVz9gudDzey2M2BdM_rNkg9VduGWMC	
DevCon IPO 1	License Type	3rd Party IP Endpoints	
	License Status	Valid	
Extension (36)	Instances	255	
HuntGroup (7)	Expiry Date	Never	
Service (0)			
Incoming Call Route (4)			
Time Profile (0)			
Account Code (1)			
License (64)			
Srd Party IP Endpoints			
CCC Agent Rostering			
Ready			,;;

5.2. Obtain LAN IP Address

From the configuration tree in the left pane, select **System** to display the System screen 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 Yealink VP530 SIP phone.

Note: During the initial configuration of Avaya IP Office, the LAN1 was configured on the private network side and LAN2 was configured on the public network side. Avaya IP Office can support SIP extensions on the LAN1 and/or LAN2 interfaces, but the compliance test used the LAN1 interface. Thus, only the LAN1 configuration will be discussed in these Application Notes.

IP Offices	📴 DevCon IPO 1* 🔤 - 🕹 🖓 🖓 🖓
BOOTP (1) Operator (3) DevCon IPO 1 System (1) System (1) Operator (3) DevCon IPO 1 System (1) Operator (36) DevCon IDO 1 System (1) System (1) Operator (36) System (1) Operator (36) Service (35) Service (0) RAS (1) Operator (0) Directory (0) Time Profile (0) Operator (0) Directory (0) Operator (0) Ope	System LAN1 LAN2 DN5 Voicemail Telephony Directory Services System Events SMTP SMDR () LAN Settings VoIP Network Topology SIP Registrar IP Address 10 10 97 36 IP Mask 255 255 240 Primary Trans. IP Address 0 0 0 RIP Mode None Image: Construction Image: Construction Constructing Constructing Construction Construction Constructing C
CCC Agent Rostering	QK <u>Cancel</u> <u>H</u> elp
Ready	,;;

5.3. Administer SIP Registrar

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

IP Offices	Image: DevCon IPO 1* Image: mail of the second secon
BOOTP (1) Operator (3) DevCon IPO 1 System (1) DevCon IPO 1 Control Unit (5) Extension (36) User (35)	System LAN1 LAN2 DNS Voicemail Telephony Directory Services System Events SMTP SMDR < 3
HuntGroup (7) HuntGroup (7) Group Service (66) HuntGroup (7) HuntGroup (7) H	RTP Port Number Range Port Range (Minimum) 49152
Incoming Call Route (4) WanPort (0) Directory (0)	Port Range (Maximum) 53246
Time Profile (0) Firewall Profile (1)	Instance Extremelle Enable RTCP Monitoring On Port 5005

Select the **SIP Registrar** sub-tab, and either enter a valid Domain Name for SIP endpoints to use for registration with IP Office or leave it blank. In the compliance testing, the **Domain Name** field was left blank. If the **Domain Name** field is left blank, then the SIP endpoints will use the LAN IP address for registration. Keep the **TCP Port** and **UDP Port** fields at default as **5060**.

IP Offices	12	DevCon IPO 1*	initia () × × × × × ×
BOOTP (1) Gradient Control (3) Control IPO 1 Gradient Control IPO 1 Gradient Control IPO 1 Gradient Control Init (5) Gradient Control Init (5) Gradi	System LAN1 LAN2 DNS LAN Settings VoIP Networ Domain Name Layer 4 Protocol TCP Port UDP Port Challenge Expiry Time (secs) Auto-create Extn/User	Voicemail Telephony Directory Services rk Topology SIP Registrar Both TCP & UDP V 5060 ¢ 10 ¢	System Events SMTP SMDR >
🗄 🕞 Incoming Call Route (4)			

5.4. Administer SIP Extensions

From the configuration tree in the left pane, right-click on **Extension** and select **New > SIP Extension** (not shown) from the pop-up list to add a new SIP extension. Enter the desired extension in the **Base Extension** field, e.g. **28257**.

IP Offices	E SIP I	📸 • 🔛 🗙 🗸 < >	
8011 28247	Extn VoIP T38 Fax		
% 8012 28248 % 8013 28249	Extension Id	8017	^
> 8014 28250 > 8022 28253	Base Extension	28257	
	Caller Display Type	On	
8016 28256	Reset Volume After Calls		
	Device Type	Unknown SIP device	≣
🗄 🧌 User (47) 🗄 🎡 HuntGroup (8)	Module	0	
Short Code (70) Service (0)	Port	0	
Bervice (0) ⊡ - 💑 RAS (1)	Force Authorization		

Select the **VoIP** tab, and retain the default values in all fields. Repeat this section to add another new SIP extension for a second Yealink VP530 SIP phone. During the compliance test, extensions **28257** and **28258** were created for two Yealink VP530 SIP phones.

IP Offices		SIP Extension: 8017 28257	📥 • 🗟 🗙 < >
IP Offices 011 28247 011 28247 011 28248 013 28249 014 28250 022 28253 015 28256 015 28256 015 28256 015 28256 019 28251 019 28251 019 28251 019 28251 019 28261 019 2827 019 28261 019 2827 019 2827 019 2827 019 2827 019 2827 019 2827 019	Extn VoIP T38 Fax IP Address Codec Selection	SIP Extension: 8017 28257 0 0 0 System Default ▼ Unused >> Selected G.722 64K G.711 ULAW 64K G.711 ULAW 64K G.729(a) 8K CS-ACELP G.723.1 6K3 MP-MLQ	VoIP Silence Suppression UoIP Silence Suppression Local Hold Music Allow Direct Media Path Re-invite Supported Use Offerer's Preferred Codec Reserve Avaya IP endpoint license Reserve 3rd party IP endpoint licens
Directory (0) Time Profile (0)	Fax Transport Support TDM->IP Gain	None Default	▼
 IP rewall Profile (1) IP Route (2) 	IP->TDM Gain	Default	✓

5.5. Administer SIP Users

In the left navigation pane, right-click on **User** and select **New** from the pop-up list (not shown). Enter desired values in the **Name** and **Full Name** fields. For the **Extension** field, enter the SIP extension created in **Section 5.4**.

IP Offices	1 2			Ŷ	'ealink VP53	0 1: 282	57*		C	* - 🖻 🗙 🗸	< >
	User	Voicemail	DND	ShortCodes	Source Numbers	Telephony	Forwarding	Dial In	Voice Recording	Button Programming	Me
	Name			Yealink	VP530 1						^
28244 Extn28244	Passw	ord									
	Confirr	m Password		I							
	Full Na	me		Yealink	VP530 Ext28257						=
	Extens	sion		28257							
28253 Extn28253	Locale								1	¥	
	Priority	/		5					•	*	
	System	n Phone Rig	hts	None					*		
	Profile			Basic U	lser				*		
				📃 Rec	eptionist						
28256 Yealink 1282 2				Ena	ble Softphone						
📲 28258 Yealink VP530				Ena	ble one-X Portal Se	rvices					
Short Code (70)				Ena	ble one-X TeleCom	muter					~
Service (0)											1 lele
Incoming Call Route (3)									QK		Help

Select the **Telephony** tab, followed by the **Call Settings** sub-tab. Check the **Call Waiting On** field, as shown below. Note that the **Call Waiting On** must be enabled so that the call waiting, call transfer and conference on Yealink VP530 phone works properly.

Select the **Supervisor Settings** sub-tab, and enter a desired **Login Code**, e.g. **"1234"**. Repeat this section for each SIP extension from **Section 5.4**. This Login Code will be used when Yealink VP530 phone registers to the SIP user **28257** on the IP Office.

IP Offices	Yealink VP530 1: 28257*	📸 🗕 🔛 🗙 🖌 < >
	User Voicemail DND ShortCodes Source Numbers Telephony For Call Settings Supervisor Settings Multi-line Options Call Log	warding Dial In Voice Recording Button Programming
	Login Code **** Login Idle Period (secs)	Force Login Force Account Code
	Monitor Group	Force Authorization Code
	Coverage Group <none> Status on No-Answer Logged On (No change)</none>	🗌 Outgoing Call Bar 📃
28235 IVR 28235	Reset Longest Idle Time	Inhibit Off-Switch Forward/Transfer
	All Calls External Incoming	☐ Can Intrude ✓ Cannot be Intruded
		Can Trace Calls
	After Call Work Time (secs) System Default (10)	Automatic After Call Work
🕀 🎇 HuntGroup (8)		

6. Configure Yealink VP530 SIP Video Phone

This section only provides steps to configure Yealink VP530 SIP Video phone to interface with Avaya IP Office. From a PC, launch webpage of Yealink VP530 phone by entering its IP address in to the address box of a browser as shown below. Note that the IP address of the phone can be found by pressing OK button on the physical phone.

6.1. Register VP530 SIP phone to Avaya IP Office

To register the VP530 to IP Office, login to the VP530 phone, the login page is displayed with password field, enter appropriate credentials of the VP530 phone and click the **OK** button to log in.

	E S
The server 10.33.5.4 requires a username Warning: This server password be sent in a without a secure con	14 at Enterprise IP phone SIP-T28P and password. is requesting that your username and an insecure manner (basic authentication nection).
<u>U</u> ser name:	🔮 admin 🛛 👻
Password:	••••
	Remember my password
	OK Cancel

The **Status** page of Yealink VP530 phone is displayed. This displays the current status and information of the phone.

Veglink			<u>Loqout</u>
tealink	Status Account	Network DSS Key	Phone Directory Security
Status	Version 🕜		NOTE
Status	Firmware Version	23.70.0.40	Version:
	Hardware Version	23.5.1.201.18.0.8	It shows the version of firmware.
	Network 🕜		
	WAN Port Type	DHCP	Network: It shows the information of
	WAN IP Address	10.33.5.68	WAN port and LAN port.
	Subnet Mask	255.255.255.0	
	Gateway	10.33.5.1	
	Primary DNS	135.10.98.60	
	Secondary DNS	255.255.255.255	
	MAC Address	00:15:65:36:B3:49	
	Link Status	Connected	
	Device Type	Bridge	
	Account Status		
	Account1	Disabled	
	Account2	Disabled	

Solution & Interoperability Test Lab Application Notes ©2013 Avaya Inc. All Rights Reserved. Click on the **Account** tab to display **Account** page. Select **Account1** in the **Account** dropdown list. Under the **Basic** section, enter the following information as shown below:

- Account Active: select Enabled to enable Account1.
- Display Name: enter a descriptive name, e.g. "Ext 28258".
- **Register Name**: enter the extension **28258** as configured in **Section 5.4**.
- User Name: enter the extension 28258 as configured in Section 5.4.
- **Password**: enter the Login Code "1234" as configured in Section 5.5.
- **SIP Server**: enter the LAN1 IP address "10.10.97.36" as defined in Section 5.2 and leave port 5060 as default.
- Enable Outbound Proxy Server: select Enabled in the dropdown list.
- **Outbound Proxy Server**: enter the LAN1 IP address "10.10.97.36" as above.
- **Transport**: leave the default port 5060.

Click **Confirm** (not shown) button in the bottom of the Account page to save the change and apply configuration to the VP530 phone.

			<u>Logout</u>	
Yealink	Status Account	Network DSS Key Phone	Directory Security	
	Account	Account 1	NOTE	
Basic	Register Status	Disabled		
Codec	Account Active	Enabled	SIP service subscriber's name	
Advanced	Label		display.	
	Display Name	Ext 28258	Register Name SIP service subscriber's ID used	
	Register Name	28258	for authentication.	
	User Name	28258	User Name User account, provided by VoIP	
	Password	sword ••••		
	SIP Server	10.10.97.36 Port 5060	NAT Traversal Defines the STUN server will be	
	Enable Outbound Proxy Server	Enabled 🕑 🕐	active or not.	
	Outbound Proxy Server	10.10.97.36 Port 5060	A special parameter just for Nortal server. If you login to	
	Transport	UDP 🕑 🕐	Nortel server, the value should he: com nortelnetworks firewall	
	Backup Outbound Proxy Server	Port 5060	Codecs	
	NAT Traversal	Disabled 💌	Choose the codecs you want to use.	
	STUN Server	Port 3478	Advanced	
	Voice Mail		The Advanced parameters for administrator.	

The screen below shows the VP530 phone registers successfully to SIP user 28258 on the IP Office.

Yealink	Status	Network DSS Key Phone	Logout
Pacie	Account	Account 1	NOTE
Codec	Register Status Account Active	Registered Enabled	Display Name SIP service subscriber's name which will be used for Caller ID
Advanced	Label Display Name	28258 ? Ext 28258 ?	display. Register Name STB service subscriber's TD used

6.2. Configure Voice Mail

Configure the VP530 phone to subscribe MWI to IP Office and voicemail number so that if any new voice message is left for the phone the MWI light is turned on and user is able to press the voice mail key on the physical phone to check voicemail.

In the **Account** page and under the **Basic** section, enter *17 as the short code in the **Voice Mail** field to access to voicemail pro application on the IP Office.

	Account	Account 1	NOTE
Basic	Register Status	Registered	Display Mana
Codec	Account Active	Enabled	SIP service subscriber's name
Advanced	Label	28258	display.
	Display Name	Ext 28258	Register Name SIP service subscriber's ID used
	Register Name	28258	for authentication.
	User Name	28258	User Name User account, provided by VoIP
	Password	•••••	service provider.
	SIP Server	10.10.97.36 Port 5060	NAT Traversal Defines the STUN server will be
	Enable Outbound Proxy Server	Enabled 🕑 💡	active or not.
	Outbound Proxy Server	10.10.97.36 Port 5060	A special parameter just for Nortel server. If you login to
	Transport	UDP 🕜	Nortel server, the value should be: com.nortelnetworks.firewall
	Backup Outbound Proxy Server	Port 5060	Codecs
	NAT Traversal	Disabled	Choose the codecs you want to use.
	STUN Server	Port 3478 🧿	Advanced
	Voice Mail	*17	administrator.
	Proxy Require		

Expand the **Advance** section. Select **Enabled** in the **Subscribe for MWI** dropdown list and keep the **MWI Subscription Period** (seconds) field at default. Click **Confirm** (not shown) button at the bottom of the **Account** page to save the changes.

Advanced	Login Expire (seconds)	3600	0
	Local SIP Port	5062	0
	RPort	Disabled 💌	0
	SIP Session Timer (seconds) T1	0.5	0
	SIP Session Timer (seconds)T2	4	
	SIP Session Timer (seconds)T4	5	
	Subscribe Period(seconds)	1800	0
	DTMF Type	RFC2833	0
	How to INFO DTMF	DTMF-Relay	
	DTMF Payload (seconds)	101	
	100 reliable retransmission	Enabled 💌	0
	Enable Precondition	Disabled 💌	0
	Subscribe Register	Disabled 💌	0
	Subscribe for MWI	Enabled 💌	0
	MWI Subscription Period (Scope:0~84600) (seconds)	3600	
	Caller ID Header	FROM 💌	0

6.3. Administer Codec

Expand the **Codecs** section in the **Account** page. Add desired voice and video codecs that the VP530 supports in the two **Enable Codecs** table. Note that the voice codec used to establish call in the IP Office is decided by the codec setting in the IP Office system and not by list of codes in the **Enable Codes** table on the VP530 phone. Click **Confirm** (not shown) button to save change in the Codec section.

Yealink	Status	Network DSS Key Phon	e Directory Security
Basic	Account	Account 1	NOTE
Codec	Voice codecs	Enable orders	
Advanced		← PCMU PCMA G729 G723 ← ↓	
	Video codecs Disable codecs	Enable codecs	
		→ H264 H263 mp4v-es ↑ ↓	

7. Verification Steps

The following steps may be used to verify the configuration:

 From a PC which the Avaya IP Office Monitor application installed, select Start > All Programs > IP Office > Monitor to launch the application. The Avaya IP Office SysMonitor window is displayed (screen not shown) and then select Status > SIP Phone Status from the top menu. The SIPPhoneStatus window is displayed as below.

🗊 SIPPhoneStatus						
Total Configured: 20 Waiting 3 secs for update						
Total Registe	Total Registered: 5 Registered Status					
Extn Num	IP Address	Transport	User Agent	SIP Options	SIP Events	Status
28244	10.33.5.30	TCP	Avaya IP Phone 1140E (SIP1140e.04	RU	ТН	SIP: Registered
28255	10.33.5.44	UDP	Yealink SIP-128P 2.70.0.60 Yealink SIP-T28P 2.70.0.60	RU		SIP: Registered SIP: Registered
28257	10.33.5.67	UDP	VP530P 23.70.0.40	U	TH	SIP: Registered
28258	128258 10.33.5.58 UDP VP530P 23.70.0.40 RU TH SIP: Registered					
Display Options C Show All © Registered © UnRegistered Print Reset Phones Cancel						

- Verify that there is an entry for each Yealink VP530 SIP extensions from Section 5.4 and the Status is SIP: Registered.
- Place video calls from and to Yealink VP530 SIP phone and verify that the calls are successfully established with two-way video and talk path.

8. Conclusion

During compliance testing, Yealink VP530 SIP Video phone successfully registered with Avaya IP Office 500 V2 Release 8.1, placed and received calls to and from SIP and non-SIP telephones, and executed other telephony features like three-way conference, transfers, hold, etc. Yealink VP530 SIP Video phone was compliant with Avaya IP Office Release 8.1.

9. Additional References

The following Avaya product documentation can be found at http://support.avaya.com
[1] IP Office 8.1GA Knowledge Base Documentation, December 17, 2012.
[2] Avaya IP Office Application Server Installation and Management, Release 8.1, Oct 2012, Document number 15-601011.

The following Yealink VP530 SIP phone documents can be found at <u>http://www.yealink.com</u> [1] Yealink VP530 Quick Installation Guide.pdf

[2] Yealink VP530 User Guide.pdf

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