



Avaya Solution & Interoperability Test Lab

Application Notes for Configuring Yealink T-26 SIP Phones to interoperate with Avaya IP Office - Issue 1.0

Abstract

These Application Notes describe the configuration steps required for the Yealink T-26 SIP phone to interoperate with Avaya IP Office.

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 for Yealink T-26 SIP phone to interoperate with Avaya IP Office.

The Yealink T-26 is an advanced SIP phone with an extra large LCD screen designed for the office environment. It supports three VoIP accounts and up to thirteen programmable function keys providing many call features such as call hold, transfer, conference, call waiting etc. It has high definition (HD) voice quality and power over Ethernet (PoE). It supports up to six expansion modules.

In the compliance testing, the Yealink T-26 was set up as a SIP extension on IP Office and underwent testing of various call scenarios with other Yealink phones and Avaya phones as specified in **Table 1**.

1.1. Interoperability Compliance Testing

The interoperability compliance test included feature and serviceability testing. The feature testing evaluated the ability of Yealink T-26 phone to interoperate with Avaya IP Office to place and receive various types of calls and to verify that good audio quality was sent and received. The calls included intra-switch calls between Yealink phones and Avaya phones on the Avaya IP Office and calls to/from the PSTN. Testing of call functions such as call hold, transfer, conference, call forwarding etc was also performed between the Yealink T-26 and various Avaya phones on the Avaya IP Office.

The serviceability testing focused on verifying the ability of the Yealink T-26 SIP phone to recover from disconnection and reconnection of the Yealink phone and of the Avaya IP Office from the network.

1.2. Support

Technical support from Yealink can be obtained through the following:

- Phone: + 44-161-763-2023
- E-mail: support@yealink.co.uk
- Web: <http://www.yealink.co.uk>

2. Reference Configuration

Figure 1 illustrates a sample configuration that was used to compliance test the interoperability of Yealink SIP Phones and Avaya IP Office. The configuration consists of an IP Office 500 connected to a Layer 2 switch to which the Yealink T-26 phone is connected. This system has connections to the following: Avaya 1600 Series IP Phones, Avaya Digital Phones and a PRI trunk to the PSTN. The phones connected to the system will be used to generate call traffic to the IP Office. These phones will be used to generate intra-switch calls and outbound/inbound calls to/from the PSTN.

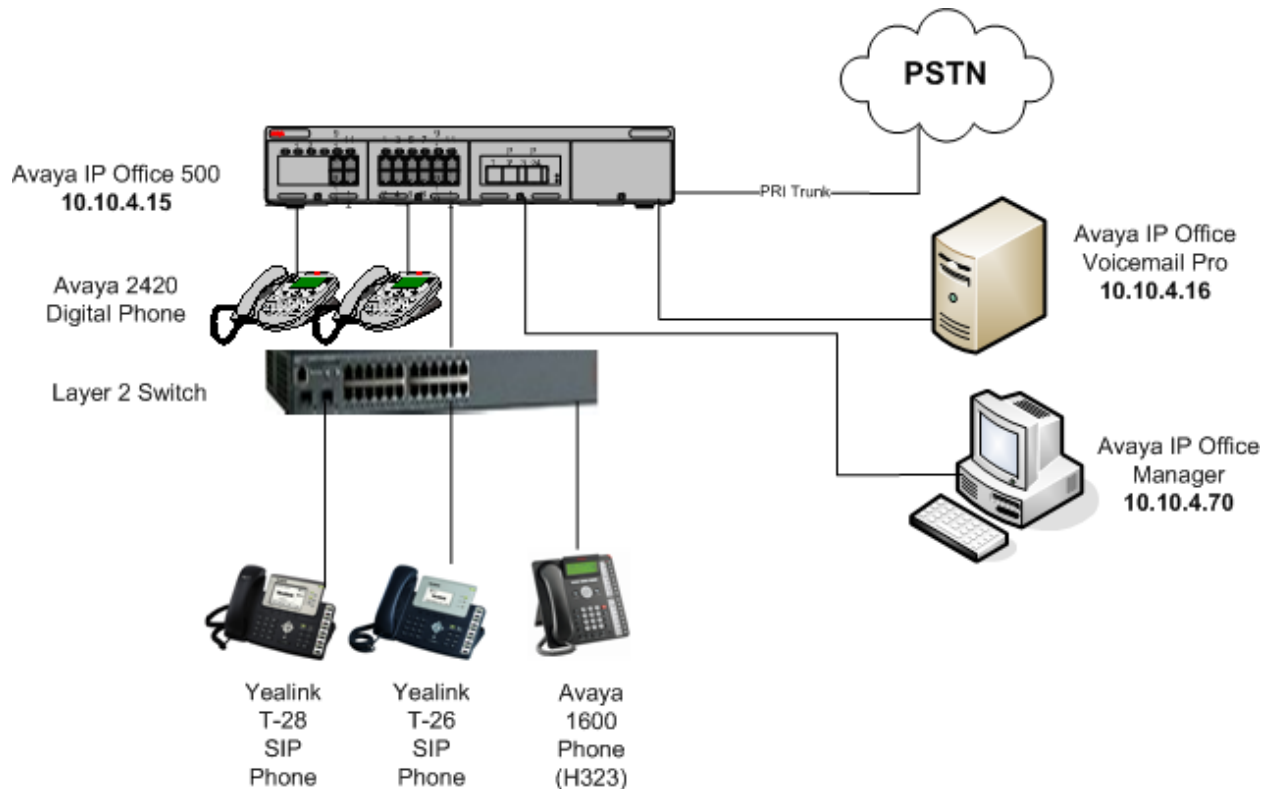


Figure 1: Network Configuration of Yealink SIP Phones with Avaya IP Office

3. Equipment and Software Validated

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

Equipment	Software Version
Avaya IP Office 500	IP Office 6.0 (8)
Avaya 16xx IP Phones (H.323) - 1616	1.22
Avaya 24xx Digital Phones - 2420	-
Yealink T-26 SIP Phone	6.43.23.6

4. Configure the Avaya IP Office

All the configuration changes in this section for IP Office are performed through the IP Office Manager. For more information on configuring IP Office, refer to the Avaya product documentation, **Section 9**, Reference [1].

This section provides the procedures for configuring IP Office. The procedures fall into the following areas:

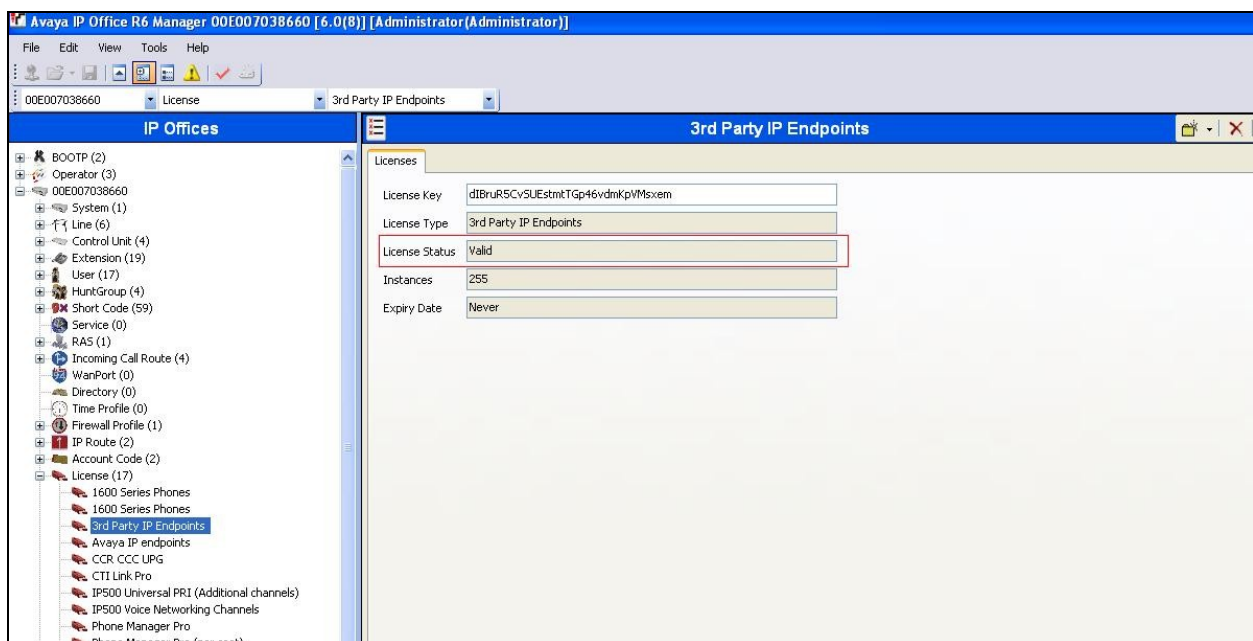
- Verify Avaya IP Office Licensing
- Setting LAN Parameters
- Administer SIP Registrar
- Add SIP Extensions
- Configure SIP User
- Add Shortcode for Voicemail
- Save Configuration

The configuration of the PRI interface to the PSTN is outside the scope of these Application Notes.

4.1. Verify Avaya IP Office Licensing

From a PC running the IP Office Manager application, select **Start → Programs → IP Office → Manager** to launch the Manager application. Select the IP Office system, and log in with the appropriate credentials. The **Avaya IP Office R6 Manager** screen is displayed.

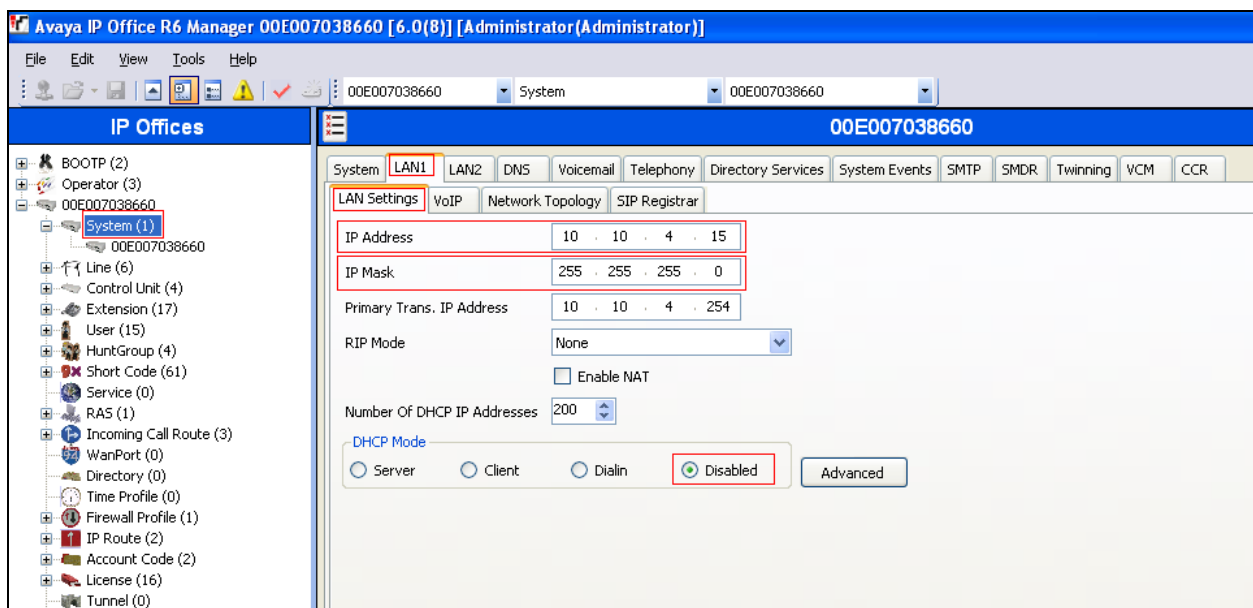
From the configuration tree in the left pane, select **License → 3rd Party IP Endpoints** to display the **3rd Party IP Endpoints** screen in the right pane. Verify that the **License Status** is **Valid**.



4.2. Setting LAN Parameters

In the Avaya IP Office Manager application, LAN parameters including IP Address, IP Mask, and other profile settings can be set.

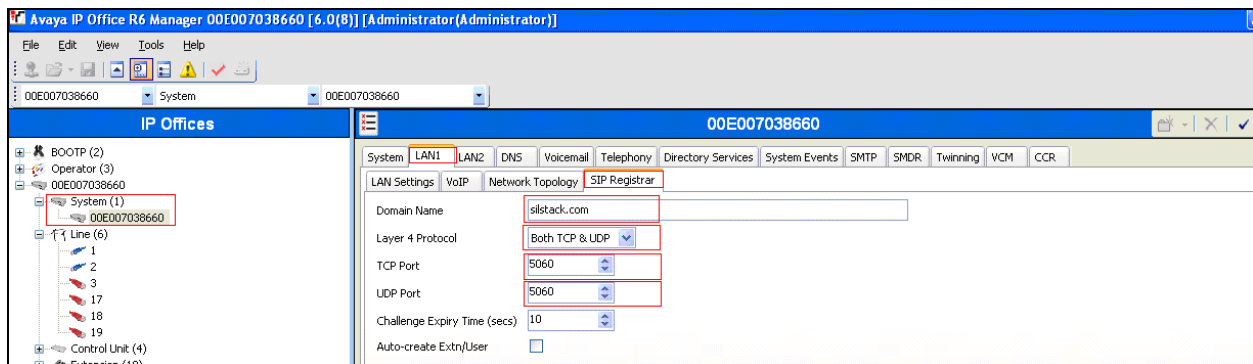
- From the configuration tree in the left pane, select **System**. Access the tab **LAN1** → **LAN Settings** to display the **LAN Settings** screen in the right pane.
- Set the **IP Address**, which is the address of the IP Office.
- Set the **IP Mask** based on the network setup.
- Set the **DHCP Mode** based on your IP Office configuration needs. In this case, the **Disabled** option is chosen since DHCP was not used.
- Other fields can be left blank or at the default settings.



4.3. Administer SIP Registrar

Select **SIP Registrar** sub-tab in the right pane and enter the following values:

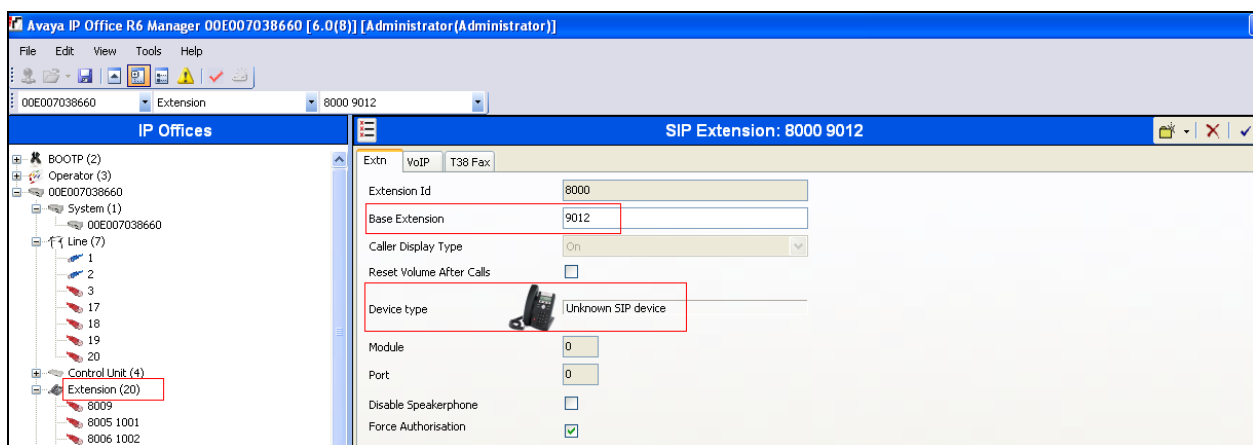
- **Domain Name** Enter a valid Domain Name, in this case **silstack.com** is used.
- **Layer 4 Protocol** Select **Both TCP & UDP**.
- **TCP Port** Select **5060**
- **UDP Port** Select **5060**



Click **OK** (not shown).

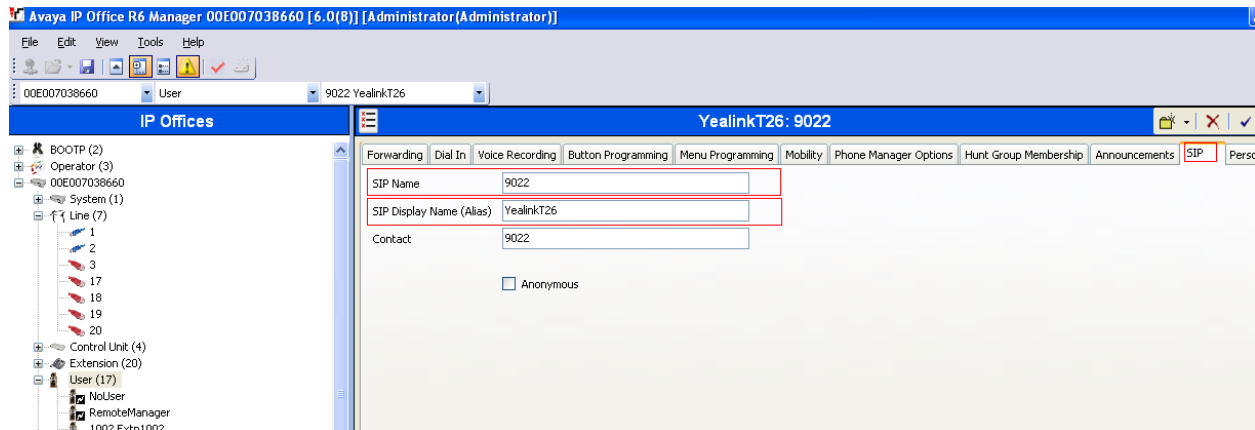
4.4. Add SIP Extensions

Add a SIP Extension by selecting **Extension** from the left pane. Right-click and choose **New** and **SIP Extension** (not shown). The **Extension Id** is automatically created i.e., **8000** in this case. Set the **Base Extension** to **9012**. Note that the **Device type** is **Unknown SIP device**. Click **OK** at bottom of screen (not shown).

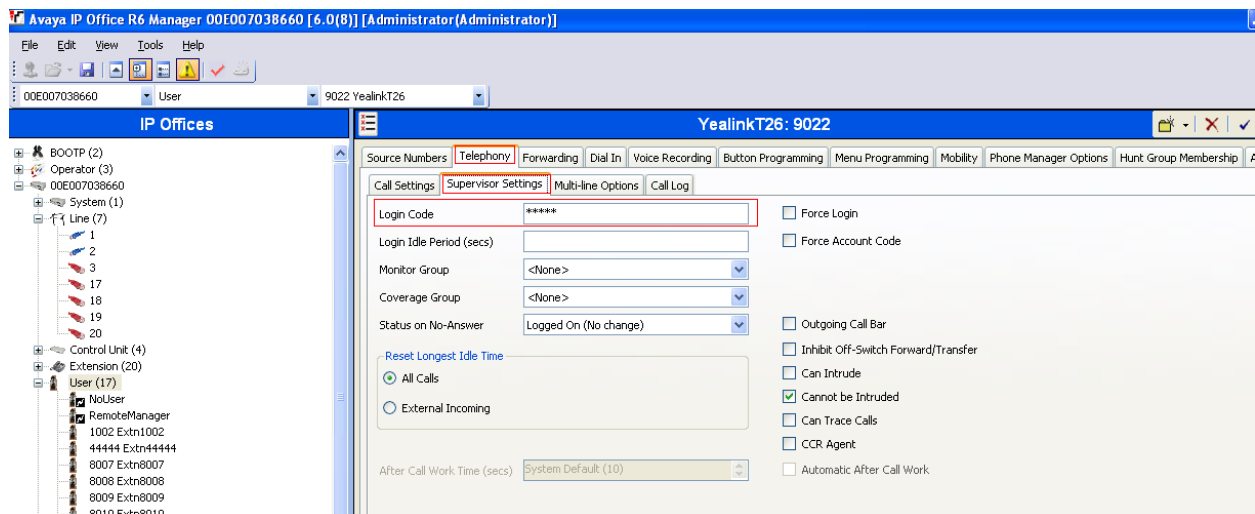


4.5. Configure SIP User

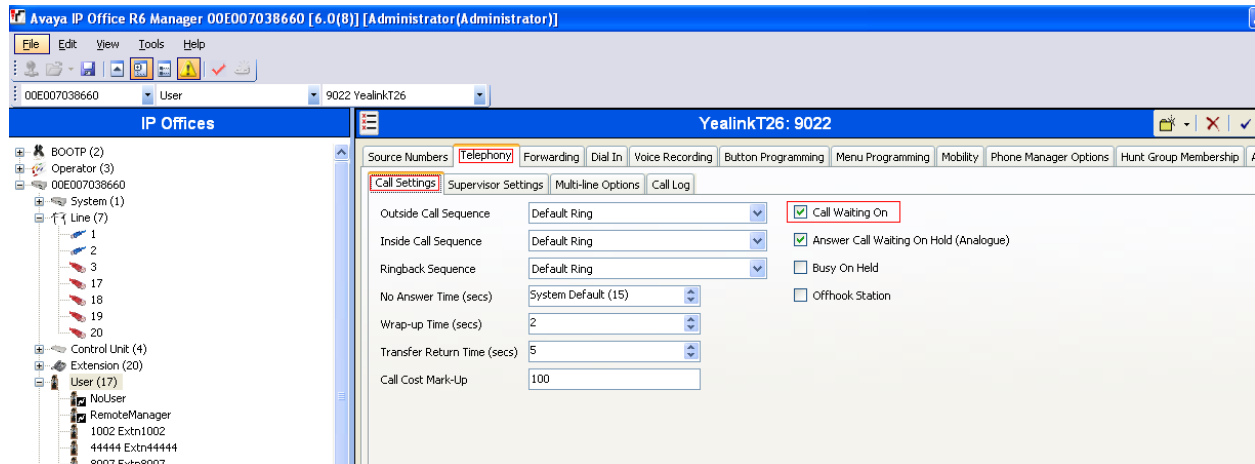
From the left pane, select a **User** and in the right-hand pane, select **SIP** tab. Modify the **SIP Name** to be the same as the user's extension number, in this case, **9022**. Set the **SIP Display Name** as required, in this case **YealinkT26**. The other fields can be left as default. Click **OK** (not shown). The completed user should be displayed as shown below. Repeat this for all users.



On the **Telephony** tab select the **Supervisor Settings** tab. Set the **Login Code**. This will be required to configure the Yealink T-26 phone as referenced in **Section 5.2**.

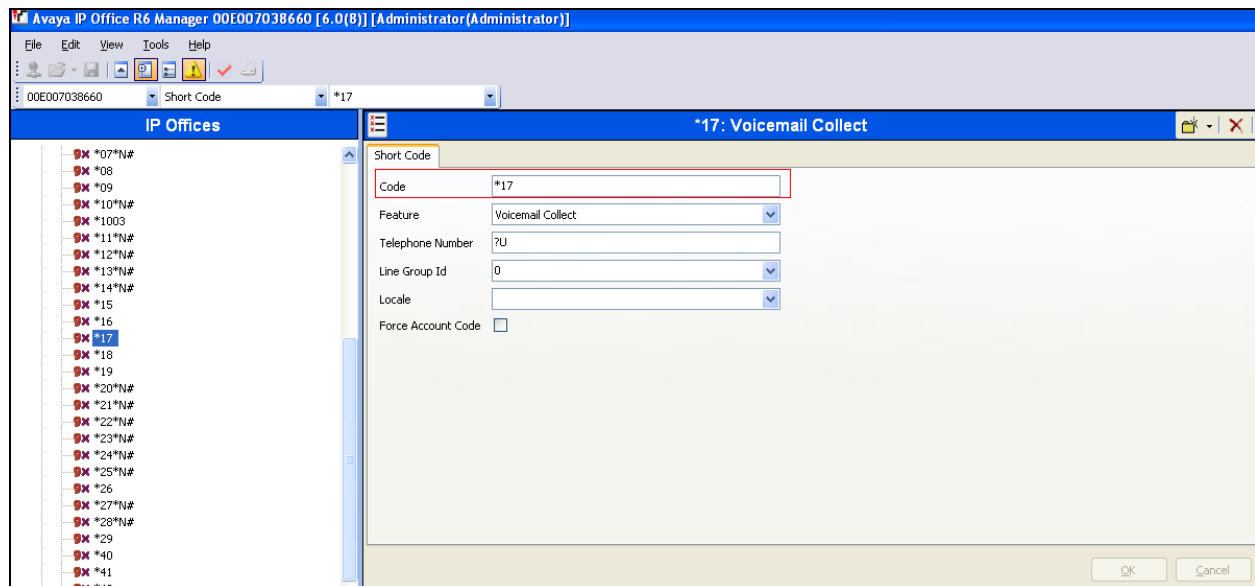


If call transfer is required for the Yealink SIP T-26 ensure that the **Call Waiting On** parameter is ticked on the **Telephony -> Call Settings** tab as shown below.



4.6. Add Shortcode for Voicemail

Voicemail is already set up on IPO and *17 is used as the shortcode as shown in the screen below. This value is referenced by Yealink in **Section 5.2**.



4.7. Save Configuration

Select **File -> Save Configuration** to save and send the configuration to the IP Office server.

5. Configure Yealink SIP Phones

This section provides the procedures to configure Yealink T-26 SIP phone with IP Office.

5.1. Configure SIP Phone Settings

If the DHCP is not enabled on the IP Office set a static IP address on the SIP phone. On the T-26 SIP phone choose **Menu → Settings → Advanced Settings**. The user is prompted for a password – the default is ‘admin’. Choose **Network → WAN Port → Select Static IP Client**. On this screen enter in the IP address, Subnet Mask and Gateway chosen. The phone restarts automatically.

5.2. Configure SIP Account Parameters

Open up a web browser on the PC and enter in the URL field the IP address of the SIP Phone used in **Section 5.1**. An authentication screen is displayed (not shown) and user name and password is required. Enter in the default User Name and Password as ‘admin’.

A new screen is displayed as seen below. The **Status** tab appears. Ensure that the **Link Status** is **Connected**. Note the **Firmware Version** on the same screen.

The screenshot shows the Yealink web interface with the 'Status' tab selected. The 'Version' section displays 'Firmware Version' as 6.43.23.6 and 'Hardware Version' as 4.0.0.1. The 'Network' section displays 'WAN Port Type' as Static IP, 'WAN IP Address' as 10.10.4.21, 'Subnet Mask' as 255.255.255.0, 'MAC Address' as 00-15-65-11-41-6c, 'Link Status' as Connected, 'PC IP Address' as 0.0.0.0, 'Device Type' as Bridge, and 'DHCP Server Status(PC)' as Disabled. A 'NOTE' section on the right explains the 'Version' and 'Network' information.

Version	
Firmware Version	6.43.23.6
Hardware Version	4.0.0.1

Network	
WAN Port Type	Static IP
WAN IP Address	10.10.4.21
Subnet Mask	255.255.255.0
MAC Address	00-15-65-11-41-6c
Link Status	Connected
PC IP Address	0.0.0.0
Device Type	Bridge
DHCP Server Status(PC)	Disabled

NOTE

Version
It shows the version of firmware.

Network
It shows the information of WAN port and LAN port.

Select the **Network** tab. Note the details of **IP Address**, **Subnet Mask** and **Default Gateway** already assigned to the SIP phone.

The screenshot shows the Yealink web interface with the 'Network' tab selected. The 'Internet Port (WAN)' section is active, showing 'Static IP Address' configuration. The 'IP Address' is 10.10.4.21, 'Subnet Mask' is 255.255.255.0, and 'Default Gateway' is 10.10.4.1. Other fields include 'Primary DNS' (212.23.3.100) and 'Secondary DNS' (212.23.6.100). The 'DHCP' option is unselected, and the 'PPPoE' section is also unselected. A 'NOTE' section on the right explains the 'DHCP' and 'Static IP Address' configurations.

☐ DHCP

☒ Static IP Address

IP Address	10.10.4.21
Subnet Mask	255.255.255.0
Default Gateway	10.10.4.1
Primary DNS	212.23.3.100
Secondary DNS	212.23.6.100

☐ PPPoE

User:

Password:

NOTE

DHCP
The network configuration will be acquired from DHCP server.

Static IP Address
Specify the IP address, Subnet Mask, Default Gateway, Primary DNS, Secondary DNS fields manually.

PPPoE
Contact your ISP if it should be used.

Select the **Account** tab. Enter in the parameters as follows:

- **Account Active** Set this value to **On**.
- **Label** A description can be entered that will be displayed against the line key on the SIP phone display.
- **Display Name** This is the name or number displayed on the SIP phone screen. It is usually set to the extension number.
- **Register Name** This is the User name for the extension used on the IP Office.
- **User Name** This is the extension number for the SIP phone on IP Office.
- **Password** This matches the Login Code of the user on the IP Office as referenced in **Section 4.5**.
- **SIP Server** Enter in the IP address of IP Office, **10.10.4.15**.
- **Port** Keep to the default of **5060** which must correspond to the IP Office TCP and UDP Port in LAN1\SIP Registrar tab as referenced in **Section 4.3**.
- **Voice Mail** This value is set up on IP Office, ***17** is used. See **Section 4.6**.

Yealink
SIP VoIP

Status Account Network Phone Contacts Upgrade Security

Account Account 1

Basic >>

Register Status	Unknown
Account Active	<input checked="" type="radio"/> On <input type="radio"/> Off
Label	9022
Display Name	9022
Register Name	YealinkT26
User Name	9022
Password	•••••
SIP Server	10.10.4.15 Port 5060
Enable Outbound Proxy Server	Disabled
Outbound Proxy Server	Port 5060
Transport	UDP
Backup Outbound Proxy Server	Port 5060
NAT Traversal	Disabled
STUN Server	Port 3478
Voice Mail	*17

NOTE

Display Name
SIP service subscriber's name which will be used for Caller ID display.

Register Name
SIP service subscriber's ID used for authentication.

User Name
User account, provided by VoIP service provider.

NAT Traversal
Defines the STUN server will be active or not.

Proxy Require
A special parameter just for Nortel server. If you login to Nortel server, the value should be: com.nortelnetworks.firewall

Codecs
Choose the codecs you want to use.

Press **Confirm** at the bottom of the screen (not shown) and the final screen should display as shown below with the **Register Status** as **Registered**.

Yealink
EASY VOP

Status Account Network Phone Contacts Upgrade Security

Account Account 1

Basic >>

Register Status Registered

Account Active ☒ On ☐ Off

Label 9022

Display Name 9022

Register Name YealinkT26

User Name 9022

Password

SIP Server 10.10.4.15 Port 5060

Enable Outbound Proxy Server Disabled

Outbound Proxy Server Port 5060

Transport UDP

Backup Outbound Proxy Server Port 5060

NAT Traversal Disabled

STUN Server Port 3478

NOTE

Display Name
SIP service subscriber's name which will be used for Caller ID display.

Register Name
SIP service subscriber's ID used for authentication.

User Name
User account, provided by VoIP service provider.

NAT Traversal
Defines the STUN server will be active or not.

Proxy Require
A special parameter just for Nortel server. If you login to Nortel server, the value should be: com.nortelnetworks.firewall

Codecs
Choose the codecs you

6. General Test Approach and Test Results

The general test approach was to place intra-switch calls and inbound and outbound PSTN trunk calls to and from the Yealink T-26 SIP phone connected to the IP Office. Different call scenarios were used including hold, conference, call forwarding etc. During serviceability testing, the Yealink phone recovered successfully from disconnection and re-connection as did the IP Office. All executed test cases were passed successfully.

7. Verification Steps

This section provides the tests that can be performed to verify correct configuration of the Yealink / Avaya solution.

7.1. Verify Avaya IP Office

The following steps can ensure that there is communication between IP Office and the Yealink T-26. From a PC running the Avaya IP Office Monitor application, select **Start → Programs → IP Office → Monitor** to launch the application. Choose the **Status** menu and select **SIP Phone Status**. This will display a table of the SIP phones and indicate those registered.

7.2. Verify Yealink T-26 SIP Phone

Ensure that the Yealink T-26 phone has been registered successfully to IP Office by checking the **Account** tab on the Yealink phone's configuration page to ensure that the **Register Status** displays **Registered**.

The screenshot shows the Yealink T-26 SIP Phone configuration page, specifically the **Account** tab. The page has a green header with the Yealink logo and a navigation bar with tabs: Status, Account (selected), Network, Phone, Contacts, Upgrade, and Security. Below the navigation bar, there's a section for 'Account' with a dropdown menu set to 'Account 1'. Under the 'Basic >>' section, the 'Register Status' is highlighted with a red box and shows 'Registered'. Other fields include 'Account Active' (On), 'Label' (9022), 'Display Name' (9022), 'Register Name' (YealinkT26), 'User Name' (9022), 'Password' (masked), 'SIP Server' (10.10.4.15, Port 5060), 'Enable Outbound Proxy Server' (Disabled), 'Outbound Proxy Server' (empty, Port 5060), 'Transport' (UDP), 'Backup Outbound Proxy Server' (empty, Port 5060), 'NAT Traversal' (Disabled), and 'STUN Server' (empty, Port 3478). On the right side, there's a 'NOTE' section with four items: 'Display Name' (SIP service subscriber's name), 'Register Name' (SIP service subscriber's ID), 'User Name' (User account, provided by VoIP service provider), and 'NAT Traversal' (Defines the STUN server will be active or not). Below these is 'Proxy Require' (A special parameter just for Nortel server) and 'Codecs' (Choose the codecs you).

Account	
Account	Account 1
Basic >>	
Register Status	Registered
Account Active	<input checked="" type="radio"/> On <input type="radio"/> Off
Label	9022
Display Name	9022
Register Name	YealinkT26
User Name	9022
Password
SIP Server	10.10.4.15 Port:5060
Enable Outbound Proxy Server	Disabled
Outbound Proxy Server	Port:5060
Transport	UDP
Backup Outbound Proxy Server	Port:5060
NAT Traversal	Disabled
STUN Server	Port:3478

NOTE

- Display Name**
SIP service subscriber's name which will be used for Caller ID display.
- Register Name**
SIP service subscriber's ID used for authentication.
- User Name**
User account, provided by VoIP service provider.
- NAT Traversal**
Defines the STUN server will be active or not.
- Proxy Require**
A special parameter just for Nortel server. If you login to Nortel server, the value should be: com.nortelnetworks.firewall
- Codecs**
Choose the codecs you

8. Conclusion

These Application Notes describe the configuration steps required for configuring Yealink SIP T-26 phone to interoperate with Avaya IP Office. All feature and serviceability tests were passed successfully.

9. Additional References

This section references the Avaya and Yealink product documentation that are relevant to these Application Notes.

The following Avaya product documentation can be found at <http://support.avaya.com>.

[1] *IP Office 6.0 Documentation CD*, February 2010, available at <http://support.avaya.com>.

The Yealink documentation can be found at <http://www.yealink.co.uk/downloads>

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