



## **Avaya Solution & Interoperability Test Lab**

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# **Application Note for Configuring the Ascom wireless i75 VoWiFi Handset with an Avaya Aura™ Telephony Infrastructure in a Converged Voice over IP and Data Network - Issue 1.0**

### **Abstract**

These Application Notes describe a solution for supporting wireless interoperability between the Ascom wireless i75 VoWiFi Handsets with an Avaya Aura™ telephony infrastructure consisting of Avaya Aura™ Communication Manager, Avaya Aura™ SIP Enablement Services, Avaya Modular Messaging and Avaya Aura™ Communication Manager Messaging in a converged Voice over IP and Data Network. Emphasis of the testing was placed on verifying good voice quality of calls with Ascom wireless SIP handsets registered to the Avaya Aura™ telephony infrastructure.

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

Implementing wireless telephony requires interoperability between the wireless telephony products and the telephony infrastructure. As IP telephony evolves, potential implementers of this technology look for flexibility and choice when deciding on which particular technology to implement. Regardless of the technology chosen, the telephony infrastructure needs to be flexible enough to support solutions using all available technologies.

These Application Notes describe the configuration process necessary to provide interoperability between Avaya Aura™ Communication Manager, Avaya Aura™ SIP Enablement Services, Avaya Modular Messaging, Avaya Communication Manager Messaging and Ascom wireless i75 VoWiFi SIP Handsets in a Converged Voice over IP and Data Network.

## 1.1. Interoperability Compliance Testing

Testing was conducted via the DevConnect Program at the Avaya Solution and Interoperability Test Lab. Compliance testing verified the integration between Ascom wireless i75 VoWiFi SIP Handsets and an Avaya Aura™ telephony infrastructure. The compliance testing focused on verifying interoperability of the Ascom wireless i75 VoWiFi Handset with Avaya Aura™ Communication Manager, Avaya Aura™ SIP Enablement Services, Avaya Modular Messaging and Avaya Communication Manager Messaging. Additional testing verified proper operation between the Ascom wireless i75 VoWiFi Handset with Avaya 9600 Series SIP & H.323 IP Telephones, and the Avaya 2410 Digital Telephone. Voicemail and MWI using Avaya Modular Messaging and Avaya Communication Manager Messaging was tested and verified to operate correctly. Network level tests included verifying roaming from access point to access point and validating Quality of Service for voice calls in a converged voice and data network configuration.

Avaya's formal testing and Declaration of Conformity is provided only on the headsets/handsets that carry the Avaya brand or logo. Avaya may conduct testing of non-Avaya headset/handset to determine interoperability with Avaya phones. However, Avaya does not conduct the testing of non-Avaya headsets/handsets for: Acoustic Pressure, Safety, Hearing Aid Compliance, EMC regulations, or any other tests to ensure conformity with safety, audio quality, long-term reliability or any regulation requirements or scalability. As a result, Avaya makes no representations whether a particular non-Avaya headset will work with Avaya's telephones or with a different generation of the same Avaya telephone.

Since there is no industry standard for handset interfaces, different manufacturers utilize different handset/headset interfaces with their telephones. Therefore, any claim made by a headset vendor that its product is compatible with Avaya telephones does not equate to a guarantee that the headset will provide adequate safety protection or audio quality.

## 1.2. Support

Technical support for the Ascom wireless i75 VoWiFi handset can be obtained through your local Ascom supplier.

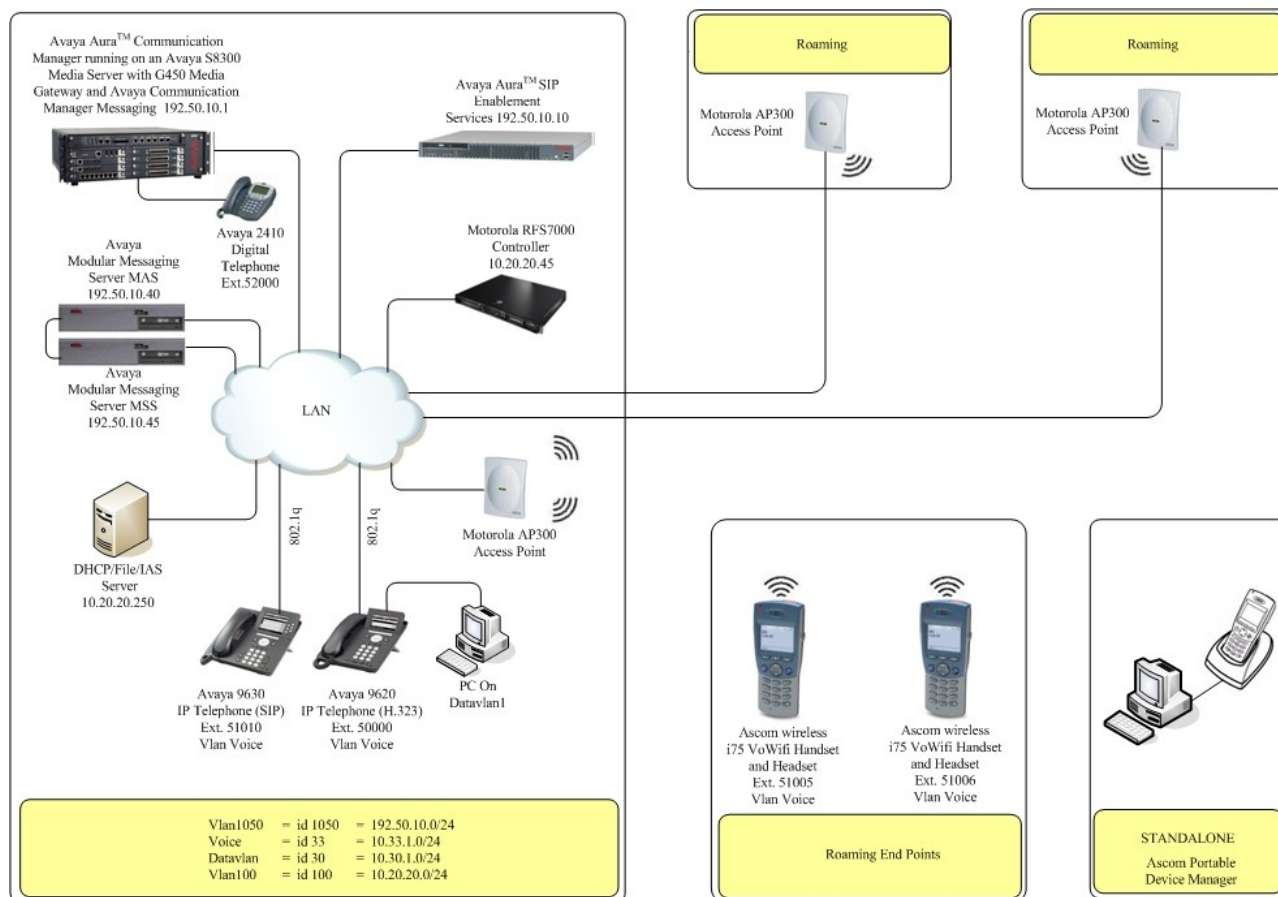
Ascom global technical support:

Phone: +46 31 559450

Email: support@ascom.se

## 2. Reference Configuration

The network diagram shown in **Figure 1** illustrates the testing environment used for compliance testing. The network consists of an Avaya Aura™ Communication Manager running on an Avaya S8300 Server with an Avaya G450 Media Gateway, and Avaya S8500 server running Avaya Aura™ SIP Enablement Services, one Avaya Modular Messaging Application Server, one Avaya Modular Messaging Storage Server, one Avaya 9630 IP Telephone (SIP), one Avaya 9620 IP Telephone (H.323), one Avaya 2420 Digital Telephone, two Ascom wireless i75 VoWiFi SIP Handsets and one Ascom Device Manger (WinPDM). The wireless network consists of one Motorola RFS7000 controller and three Motorola AP300 access points.



**Figure 1: Network Diagram**

### 3. Equipment and Software Validated

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

Equipment	Software/Firmware
<b><i>Avaya PBX Products</i></b>	
Avaya S8300 Server running Avaya Aura™ Communication Manager	Avaya Aura™ Communication Manager 5.2
Avaya G450 Media Gateway (Corporate Site) MGP MM712 DCP Media Module	28.22.0 HW9
<b><i>Avaya Aura™ SIP Enablement Services (SES)</i></b>	
Avaya Aura™ SIP Enabled Services (SES) Server	5.2 SP2
<b><i>Avaya Messaging (Voice Mail) Products</i></b>	
Avaya Modular Messaging - Messaging Application Server (MAS)	5.0
Avaya Modular Messaging - Message Storage Server (MSS)	5.0
Avaya Communication Manager Messaging (CMM)	5.2.1-13.0
<b><i>Avaya Telephony Sets</i></b>	
Avaya 9600 Series IP Telephones	Avaya one-X Deskphone Edition 3.0.1
Avaya 9600 Series IP Telephones	Avaya one-X Deskphone SIP 2.4
Avaya 2410 Digital Telephone	5.0
<b><i>Ascom Products</i></b>	
Ascom wireless i75 VoWiFi Handset	1.6.23 (SIP)
Ascom Device Manger (WinPDM)	3.3.5
<b><i>Motorola Products</i></b>	
Motorola RFS7000 controller	1.2.0.0-040R
Motorola AP300 Access Point	01.00-2100r
<b><i>MS Products</i></b>	
Microsoft Windows 2003 Server	Microsoft Windows 2003 Server

## 4. Configure Avaya Aura™ Communication Manager

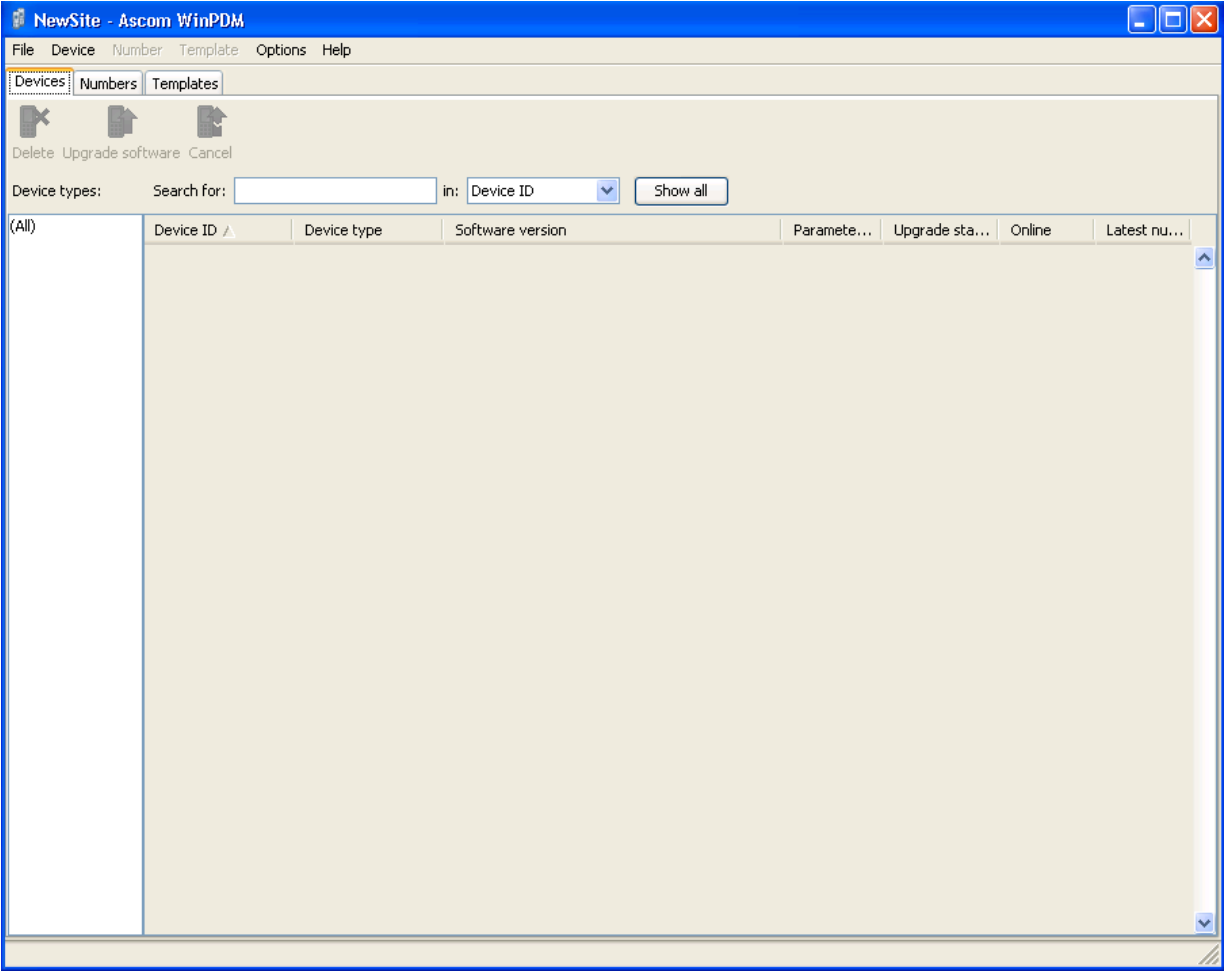
This section describes the steps required for Avaya Aura™ Communication Manager to support the configuration shown in **Figure 1**. The assumption is that the appropriate license and authentication files have been installed on the servers and that login and password credentials are available. It is assumed the Avaya Aura™ Communication Manager and Avaya Aura™ SIP Enabled Services are configured. Refer to [1], [2], and [3] for more information.

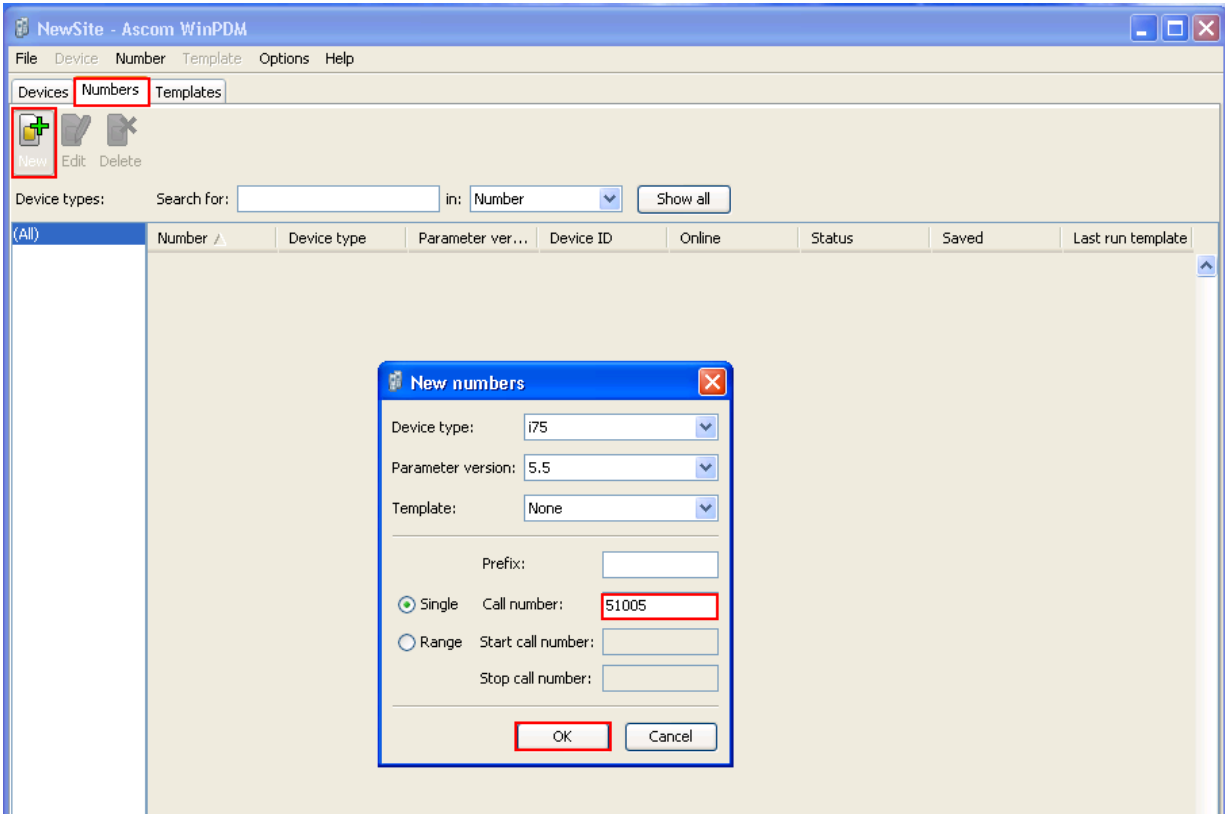
Each Ascom wireless i75 VoWiFi SIP Handset configured in the sample network in **Figure 1** was administered as stations on Communication Manager with the Off-PBX stations option set. For information on how to administer these types of stations refer to [1], [2], and [3].


Step	Description
1.	<p>To enable the features used for testing (Call Park, Call Park Answerback, Call Forwarding and Call Pickup) administer the configuration for Feature-Access-Codes (FAC) on Communication Manager. From the SAT (System Administration Terminal) interface on Communication Manager, use the “<b>change feature-access-codes</b>” command to configure the following parameters on Page 1 and Submit the changes.</p> <pre data-bbox="277 947 1502 1692"> change feature-access-codes                                     Page 1 of 9                                 FEATURE ACCESS CODE (FAC) Abbreviated Dialing List1 Access Code: *600 Abbreviated Dialing List2 Access Code: *601 Abbreviated Dialing List3 Access Code: *602 Abbreviated Dial - Prgm Group List Access Code: Announcement Access Code: *604 Answer Back Access Code: *650 Attendant Access Code: Auto Alternate Routing (AAR) Access Code: 3 Auto Route Selection (ARS) - Access Code 1: 9      Access Code 2: Automatic Callback Activation: *605      Deactivation: *606 Call Forwarding Activation Busy/DA: *607  All: *608  Deactivation: *609 Call Forwarding Enhanced Status:      Act:      Deactivation: Call Park Access Code: *652 Call Pickup Access Code: #6 CAS Remote Hold/Answer Hold-Unhold Access Code: CDR Account Code Access Code: Change COR Access Code: Change Coverage Access Code: Conditional Call Extend Activation:      Deactivation: Contact Closure Open Code:      Close Code: ESC-x=Cancel Esc-e=Submit Esc-p=Prev Pg Esc-n=Next Pg Esc-h=Help </pre>

## 5. Configure the Ascom wireless i75 VoWiFi Handset

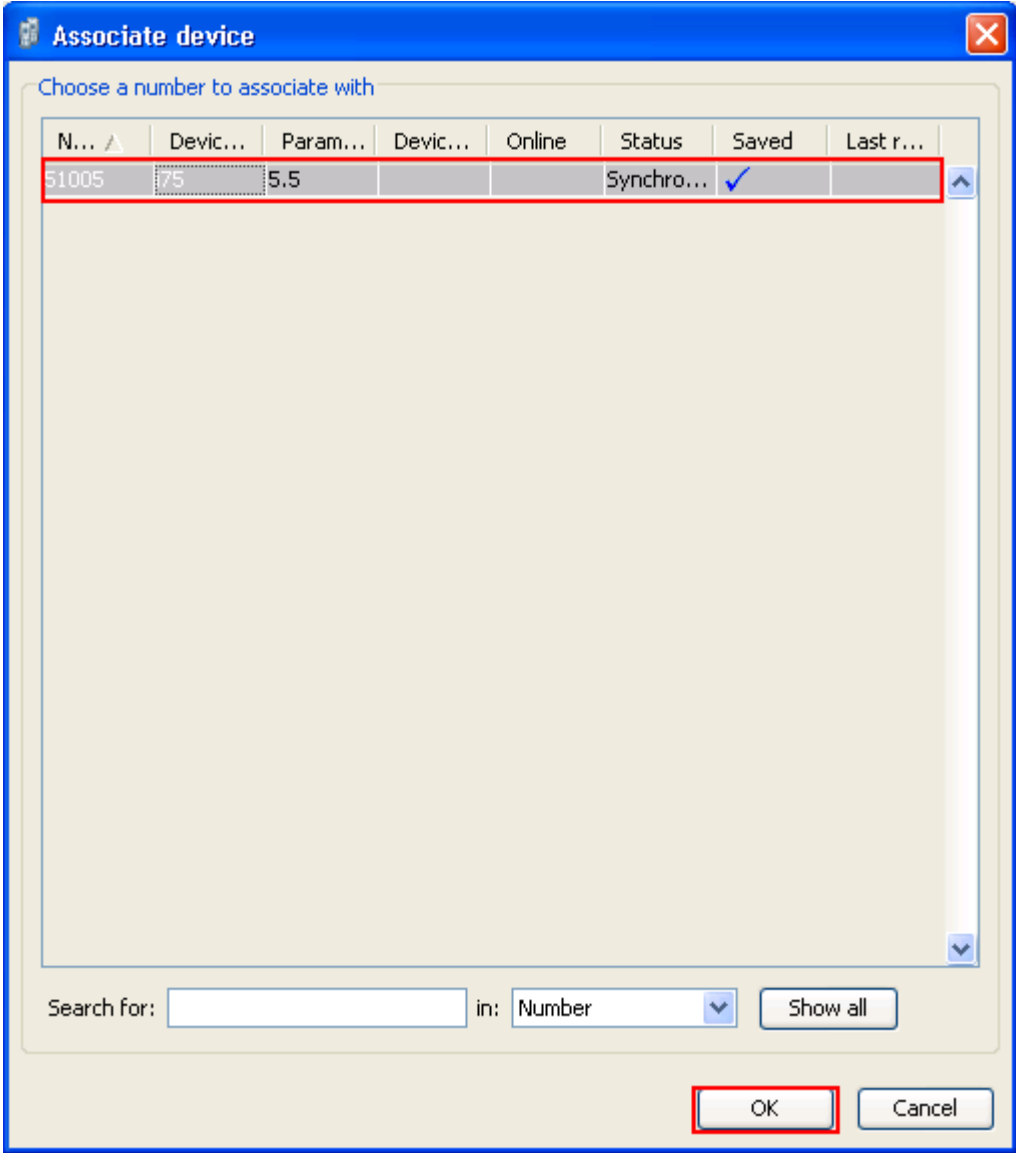
The following steps detail the configuration process for the Ascom wireless i75 VoWiFi Handset using the Ascom Device Manger (WinPDM) Windows-based application. For complete details on all the supported features on the Ascom wireless i75 VoWiFi Handset refer to **Section 9, [8] and [9]**.

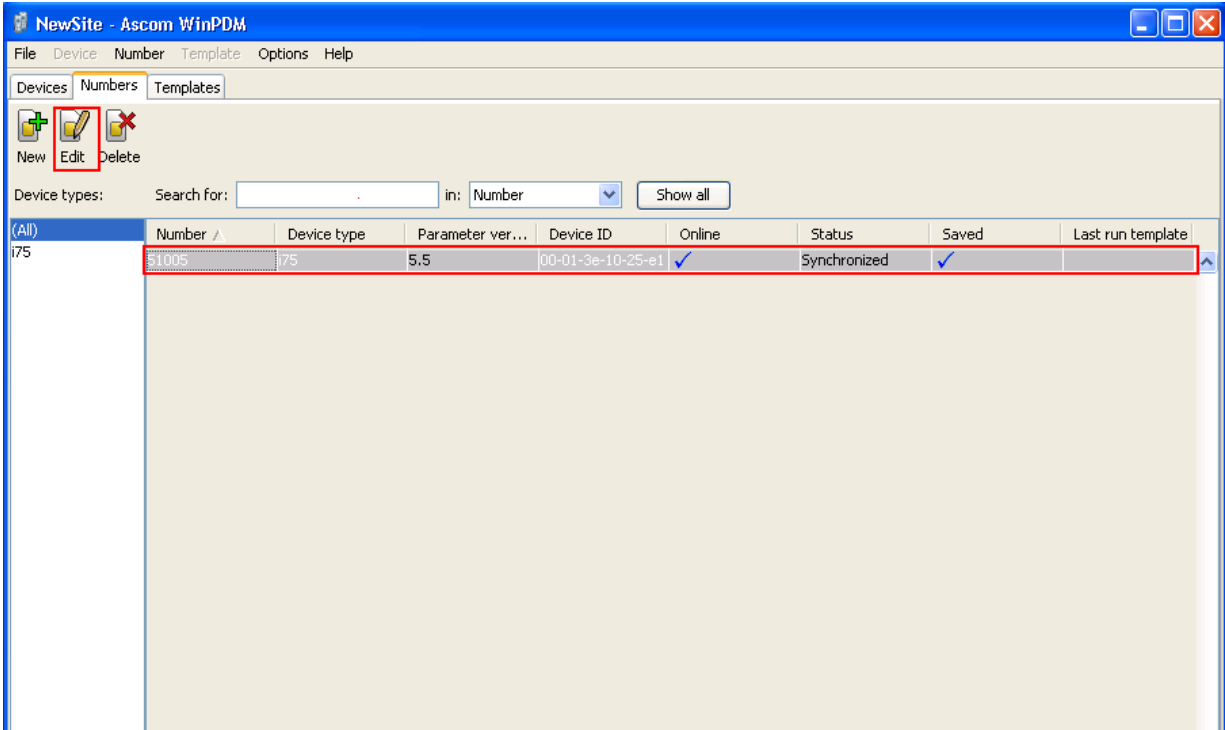
Step	Description
1.	<p>Launch the WinPDM application from the computer that has the application installed and has the WinPDM physically attached via a USB cable. Before the user is presented with the following screen a login is required. See <b>Section 9, [8] and [9]</b> for administration and configuration information on the WinPDM. After the user has logged on to the WinPDM the following screen is displayed which shows the devices found in the database. Since no devices have been plugged into the WinPDM, none are shown at this time.</p> 

Step	Description
2.	<p>Create the extension profiles on the Ascom WinPDM. For this example extension 51005 will be used. From the Ascom WinPDM window, click <b>Numbers</b> → <b>New</b>. The <b>New numbers</b> dialog window appears, Set the following options:</p> <ul style="list-style-type: none"> <li>• <b>Call number = 51005</b></li> </ul> <p>Click <b>OK</b> to continue.</p> 
3.	Repeat step 2 for all Ascom i75 handsets as shown in <b>Figure 1</b> .

Step	Description
4.	<p>Place an Ascom wireless i75 Handset into the WinPDM, Once an Ascom wireless i75 Handset is placed into the cradle, the WinPDM recognizes the telephone. Click the radio button labeled <b>Associate with number</b> and then click <b>Next</b>.</p> 



Step	Description
5.	<p>The <b>Associate device</b> dialogue window appears, select the extension that the Ascom wireless i75 Handset is associating to and select <b>OK</b>.</p> 

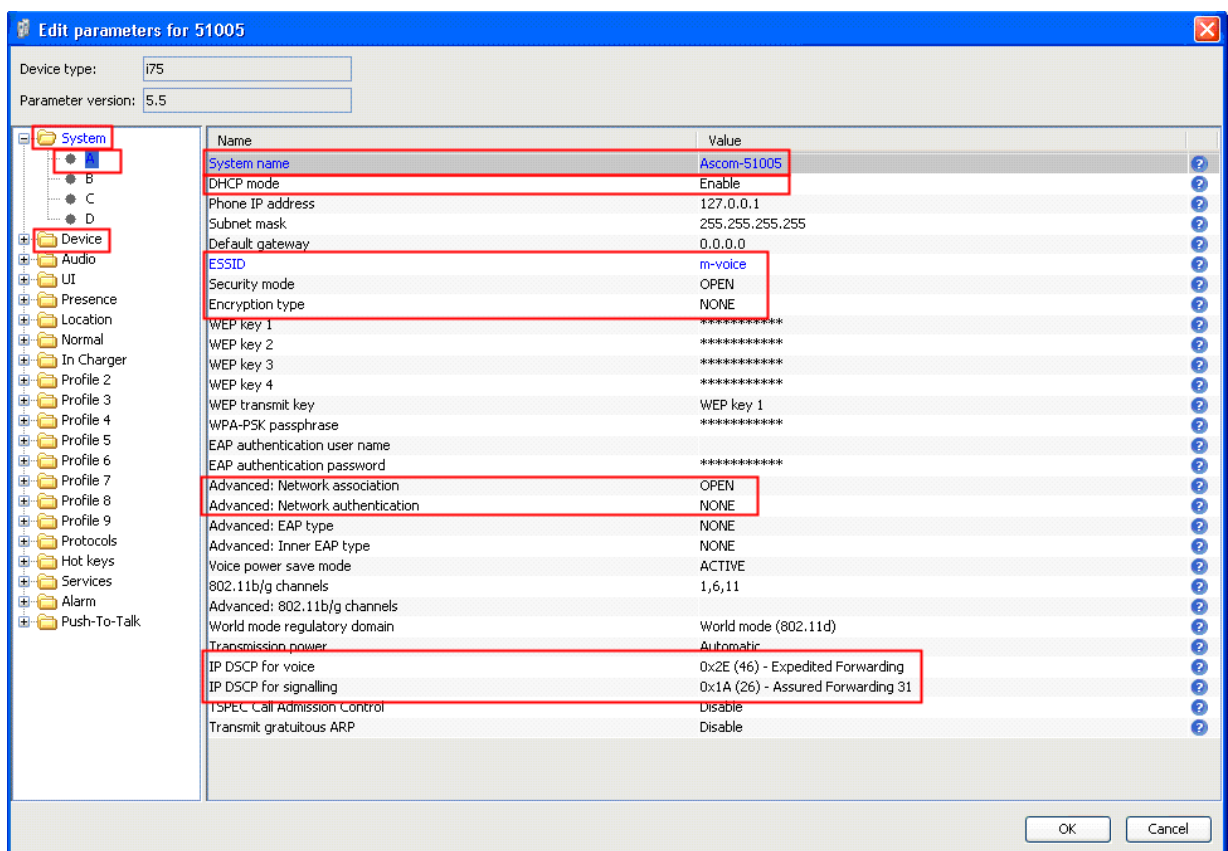
Step	Description																		
6.	<p>After entering OK, the new extension is created. Highlight the extension and select <b>Edit</b> tab.</p>  <p>The screenshot shows the 'NewSite - Ascom WinPDM' application window. The 'Numbers' tab is selected. In the toolbar, the 'Edit' button (represented by a pencil icon) is highlighted with a red box. Below the toolbar is a search area with 'Device types:' and 'Search for:' fields. A table displays the following data:</p> <table border="1"> <thead> <tr> <th>(All)</th> <th>Number</th> <th>Device type</th> <th>Parameter ver...</th> <th>Device ID</th> <th>Online</th> <th>Status</th> <th>Saved</th> <th>Last run template</th> </tr> </thead> <tbody> <tr> <td>i75</td> <td>1005</td> <td>i75</td> <td>5.5</td> <td>00-01-3e-10-25-e1</td> <td>✓</td> <td>Synchronized</td> <td>✓</td> <td></td> </tr> </tbody> </table>	(All)	Number	Device type	Parameter ver...	Device ID	Online	Status	Saved	Last run template	i75	1005	i75	5.5	00-01-3e-10-25-e1	✓	Synchronized	✓	
(All)	Number	Device type	Parameter ver...	Device ID	Online	Status	Saved	Last run template											
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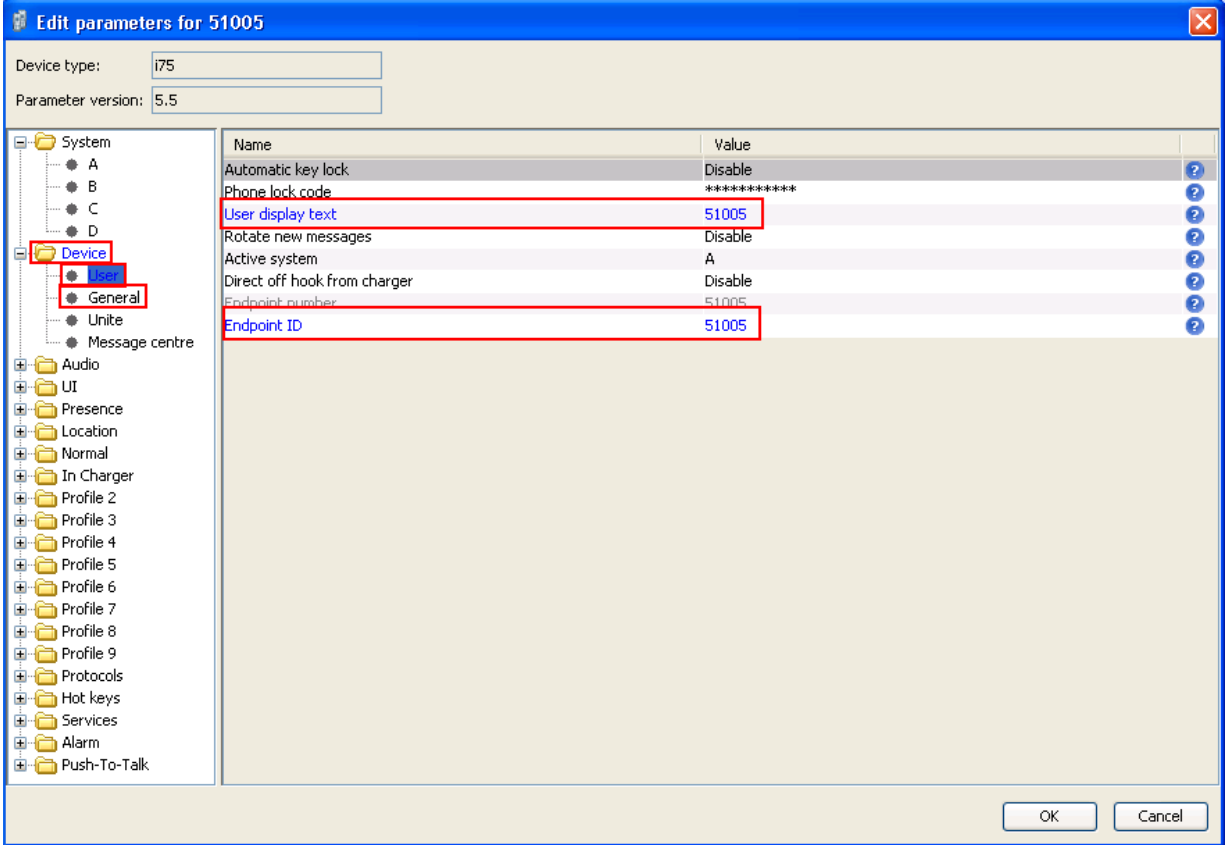
Step	Description
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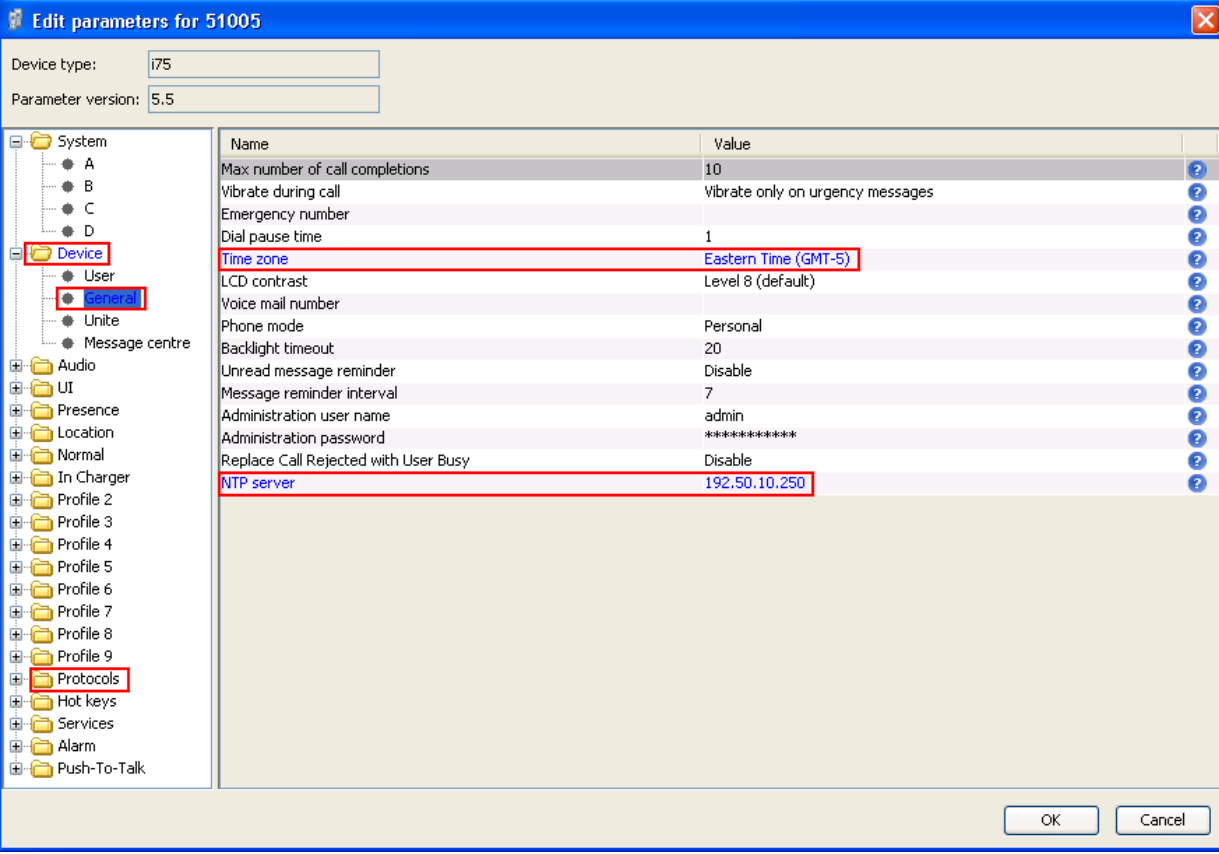
7. The **Edit parameters for 51005** dialogue window appears. Navigate to the System A configuration page by clicking **SYSTEM** and then **A**. Verify and Configure the parameters that are listed below, click **Device** → **User** to continue.

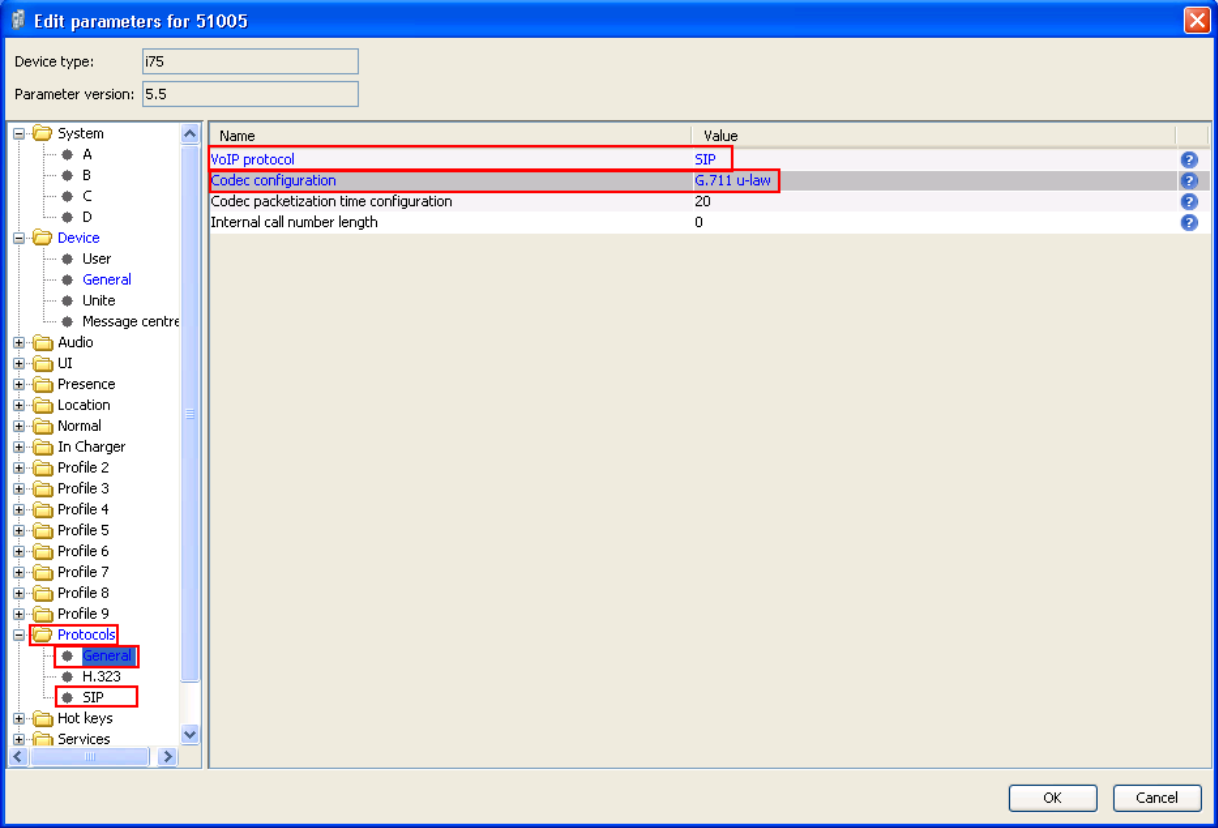
Two security schemas were tested: None/Open, and WPA2- AES-CCMP. Only OPEN will be shown in this document. For complete details on how to configure these parameters using the WinPDM refer to **Section 9, [8] and [9]**.

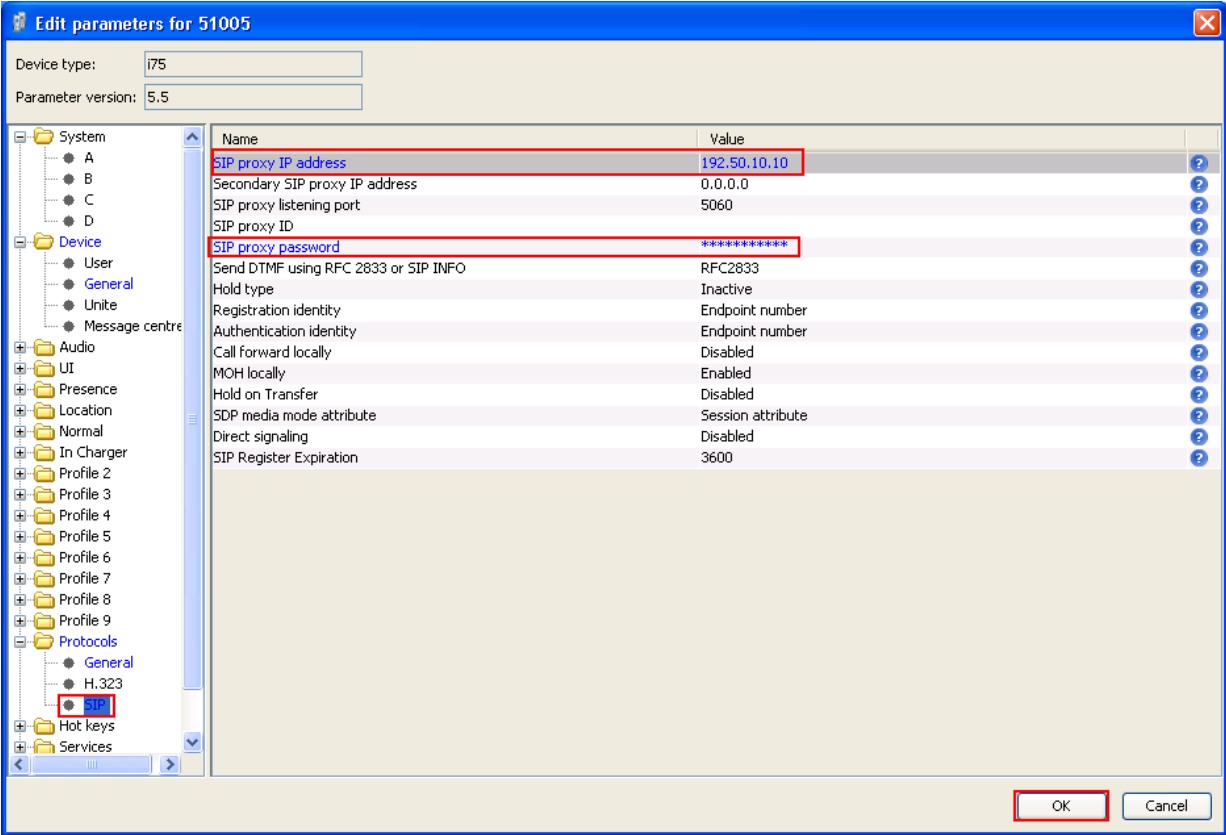
<b>System Name</b>	<b>“Ascom-51001”</b>
<b>DHCP mode</b>	<b>“Enable”</b>
<b>ESSID</b>	<b>“m-voice”</b>
<b>Security mode</b>	<b>“Open”</b>
<b>Encryption type</b>	<b>“NONE”</b>
<b>Advanced Network association</b>	<b>“OPEN”</b>
<b>Advanced Network authentication</b>	<b>“NONE”</b>
<b>IP DSCP for voice</b>	<b>“0x2E (46) – Expedited Forwarding”</b>
<b>IP DSCP for signaling</b>	<b>“0x1A (26) – Assured Forwarding 31”</b>



Step	Description
8.	<p>Navigate to the <b>USER</b> configuration page by clicking <b>DEVICE</b> and then <b>USER</b>. Verify and Configure the parameters that are listed below, click <b>General</b> to continue.</p> <p><b>User display text</b>      “51005”  <b>Endpoint ID</b>            “51005”</p> 

Step	Description																																
9.	<p>Ensure that the <b>Time zone</b> and <b>NTP server</b> values are set. Click <b>Protocols</b> to continue.</p>  <p><b>Edit parameters for 51005</b></p> <p>Device type: <input type="text" value="i75"/></p> <p>Parameter version: <input type="text" value="5.5"/></p> <table border="1"> <thead> <tr> <th>Name</th> <th>Value</th> </tr> </thead> <tbody> <tr><td>Max number of call completions</td><td>10</td></tr> <tr><td>Vibrate during call</td><td>Vibrate only on urgency messages</td></tr> <tr><td>Emergency number</td><td></td></tr> <tr><td>Dial pause time</td><td>1</td></tr> <tr><td>Time zone</td><td>Eastern Time (GMT-5)</td></tr> <tr><td>LCD contrast</td><td>Level 8 (default)</td></tr> <tr><td>Voice mail number</td><td></td></tr> <tr><td>Phone mode</td><td>Personal</td></tr> <tr><td>Backlight timeout</td><td>20</td></tr> <tr><td>Unread message reminder</td><td>Disable</td></tr> <tr><td>Message reminder interval</td><td>7</td></tr> <tr><td>Administration user name</td><td>admin</td></tr> <tr><td>Administration password</td><td>*****</td></tr> <tr><td>Replace Call Rejected with User Busy</td><td>Disable</td></tr> <tr><td>NTP server</td><td>192.50.10.250</td></tr> </tbody> </table> <p>OK Cancel</p>	Name	Value	Max number of call completions	10	Vibrate during call	Vibrate only on urgency messages	Emergency number		Dial pause time	1	Time zone	Eastern Time (GMT-5)	LCD contrast	Level 8 (default)	Voice mail number		Phone mode	Personal	Backlight timeout	20	Unread message reminder	Disable	Message reminder interval	7	Administration user name	admin	Administration password	*****	Replace Call Rejected with User Busy	Disable	NTP server	192.50.10.250
Name	Value																																
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Dial pause time	1																																
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LCD contrast	Level 8 (default)																																
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NTP server	192.50.10.250																																

Step	Description										
10.	<p>Click <b>GENERAL</b>. Verify and configure the parameters that are listed below. Ensure that the codec chosen matches whatever is used on Communication Manager. Click <b>SIP</b> to continue.</p> <p style="text-align: center;"><b>VoIP protocol “SIP”</b> <b>Codec configuration “G.711 u-law”</b></p>  <table border="1" data-bbox="495 573 1490 688"> <thead> <tr> <th>Name</th> <th>Value</th> </tr> </thead> <tbody> <tr> <td>VoIP protocol</td> <td>SIP</td> </tr> <tr> <td>Codec configuration</td> <td>G.711 u-law</td> </tr> <tr> <td>Codec packetization time configuration</td> <td>20</td> </tr> <tr> <td>Internal call number length</td> <td>0</td> </tr> </tbody> </table>	Name	Value	VoIP protocol	SIP	Codec configuration	G.711 u-law	Codec packetization time configuration	20	Internal call number length	0
Name	Value										
VoIP protocol	SIP										
Codec configuration	G.711 u-law										
Codec packetization time configuration	20										
Internal call number length	0										

Step	Description
11.	<p>Verify and Configure the parameters that are listed below. Ensure that the codec chosen matches whatever is used on Communication Manager Branch.</p> <p>The <b>SIP proxy password</b> field must match the user password configured on SES. Once the information has been configured, the WinPDM reports the information as ****. After clicking <b>OK</b>, pick up the i75 handset from the WinPDM in order to reboot the handset and activate the new configuration.</p> <p><b>SIP proxy IP address</b> “192.50.10.10”  <b>SIP proxy password</b> “123456”</p> 
12.	Repeat <b>Steps 1 – 11</b> for each Ascom wireless i75 VoWiFi Handset being provisioned, but modify the appropriate extension fields to avoid duplication.

## 6. General Test Approach and Test Results

### 6.1. General Test Approach

All feature functionality test cases were performed manually. The general test approach entailed verifying the following:

- Registration, re-registration of Ascom i75 VoWiFi Portable Handsets with Avaya Aura™ SIP Enablement Services.
- Verify Message Waiting Indicator and message retrieval from Avaya Modular Messaging Server & Avaya Communication Manager Messaging
- VoIP calls between Ascom and Avaya Digital Telephones, Avaya SIP and Avaya H.323 IP Telephones.
- Inter-office calls using SIP, G.711 codec, shuffling, conferencing, voicemail, DTMF and sending low priority data traffic over the LAN.
- Wireless Roaming, Wireless Security, Wireless Authentication and Wireless Quality of Service.
- Verifying that QoS directed the voice signaling and voice media to the higher priority queue based on WMM QoS.

### 6.2. Test Results

The Ascom wireless i75 VoWiFi Handset passed all test cases. Ascom wireless i75 VoWiFi Handsets were verified to successfully register with Avaya Aura™ Communication Manager and Avaya Aura™ SIP Enablement Services. The compliance testing also focused on verifying WMM Quality of Service for voice traffic while low priority wireless background traffic was competing for bandwidth. The Ascom wireless i75 VoWiFi Handset was verified to roam successfully between access points while maintaining voice calls. Multiple security schemas, OPEN and WPA2-AES-CCMP and codecs, G.711MU were used for testing. Telephone calls were verified to operate correctly with the media path direct between the telephones (shuffling enabled) and with the media path centralized through Avaya Aura™ Communication Manager (shuffling disabled). Calls were maintained for durations over one minute without degradation to voice quality. The telephony features verified to operate correctly included attended/unattended transfer, conference call participation, conference call add/drop, multiple call appearances, caller ID operation, call forwarding unconditional, call forwarding on busy, call forwarding clear, pick groups, call pickup, bridged appearance alerting, voicemail using Avaya Modular Messaging & Avaya Communication Manager Messaging, MWI, hold and return from hold.



## 7. Verification Steps

The following steps can be used to verify proper operation of the Ascom wireless i75 VoWiFi Handset.

- Ensure that the **ESSID** value of the wireless network matches the **ESSID** field value configured in **Section 5 Step 7** on the Ascom wireless i75 VoWiFi Handset.
- Ensure that the **VoIP Protocol** and **Codec configuration** field values are set correctly, see **Section 5, Step 10**.
- Ensure that the **SIP proxy IP address** and **SIP proxy password** field values are set correctly, see **Section 5, Step 11**.
- Ensure that the Ascom wireless i75 VoWiFi Handset was removed from the Device Manager after completing the configuration to apply the changes and reboot the handset.
- Place calls from the Ascom wireless i75 VoWiFi Handset and verify two-way audio.
- Place a call to the Ascom wireless i75 VoWiFi Handset, allow the call to be directed to voicemail, leave a voicemail message and verify the MWI message is received.
- Using the Ascom wireless i75 VoWiFi Handset that received the voicemail, connect to the voicemail system to retrieve the voicemail and verify the MWI message clears.
- Place calls to the Ascom wireless i75 VoWiFi Handset and exercise calling features such as transfer, conference and hold.

## 8. Conclusion

These Application Notes illustrate the procedures necessary for configuring the Ascom wireless i75 VoWiFi Handset with an Avaya Aura™ telephony infrastructure using Avaya Aura™ Communication Manager, Avaya Aura™ SIP Enablement Services, Avaya Modular Messaging and Avaya Communication Manager Messaging. All feature functionality test cases described in **Section 6.1** passed.

## 9. Additional References

Avaya documentation was obtained from <http://support.avaya.com>.

- [1] *Administering Avaya Aura™ Communication Manager*, May 2009 , Issue 5.0, Document Number 03-300509..
- [2] *Administering Avaya Aura™ SIP Enablement Services*, May 2009, Issue 2.1, Document 03-602508.
- [3] *Avaya Aura™ SIP Enablement Services (SES) Implementation Guide*, May 2009, Issue 6, Document 16-300140.
- [4] *Avaya one-X Deskphone Edition for 9600 Series IP Telephones Administrator Guide Release 3.0*, Document Number 16-300698.
- [5] *Avaya one-X Deskphone SIP for 9600 Series IP Telephones Administrator Guide, Release 2.0*, Document Number 16-601944.
- [6] *Modular Messaging, Release 5.0 with the Avaya MSS Messaging Application Server (MAS) Administration Guide*, January 2009.
- [7] *Avaya Communication Manager Messaging Application Release 5.1 Administering. Communication Manager Servers to Work with IA 770*, June 2008.

The Ascom wireless documentation was obtained from <http://www.Ascom wireless.com>.

- [8] *Installation and Operation Manual – Device Manager (WinPDM), Windows version*, December 2006, Version C, Document Number TD 92325GB
- [9] *User Manual Ascom i75 VoWiFi Handset*, September 2006, Version B, Document Number TD 92319GB

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