

Avaya Solution & Interoperability Test Lab

Application Notes for DiVitas Mobile Unified Communications with Avaya Modular Messaging – Issue 1.0

Abstract

These Application Notes describe the configuration steps required to provide voicemail coverage to DiVitas Clients registered to DiVitas Mobile Unified Communications using Avaya Modular Messaging. The DiVitas solution provides the seamless convergence of WiFi and cellular networks enabling roaming (back and forth) between the two networks. The DiVitas Server accesses voicemail from Avaya Modular Messaging using the IMAP interface. The DiVitas Clients running on Nokia E- and N-Series mobile handsets can then retrieve, play back, and delete voicemail messages using their visual voicemail application. DiVitas Clients are mapped to desktop phones on Avaya Aura Communication Manager.

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 provide voicemail coverage to DiVitas Clients registered to DiVitas Mobile Unified Communications (Mobile UC) using Avaya Modular Messaging. The DiVitas solution provides the seamless convergence of WiFi and cellular networks enabling roaming (back and forth) between the two networks. The DiVitas Server accesses voicemail from Avaya Modular Messaging using the IMAP interface. The DiVitas Clients running on Nokia E- and N-Series mobile handsets can then retrieve and playback voicemail messages using their visual voicemail application. DiVitas Clients are mapped to desktop phones on Avaya AuraTM Communication Manager.

The focus of these Application Notes is on how to enable voicemail coverage for the DiVitas Clients. This solution requires that the configuration described in [5], *Application Notes for DiVitas Mobile Unified Communications and Avaya Aura* Communication Manager and Avaya Aura Session Manager, is also implemented.

1.1. Interoperability Compliance Testing

The focus of the interoperability compliance test was to verify that the DiVitas Server can access voicemail messages stored in the Message Storage Server (MSS) of Avaya Module Messaging using the IMAP interface, and that the DiVitas Client running on Nokia E51 and N95 mobile handsets can retrieve, play back, and delete the voicemail message on their visual voicemail application. The general test approach was to verify the following functionality:

- Voicemail coverage for the DiVitas Clients in WiFi mode¹.
- Leaving voicemail for DiVitas Clients and verifying through the visual voicemail application that there are new messages.
- Retrieving and playing back the voicemail message on the DiVitas Client.
- Verifying that deleting the voicemail message through the DiVitas Client visual voicemail application also deletes the message from the MSS.
- Verifying that logging into Modular Messaging and deleting voicemail messages also deletes it from the DiVitas Client visual voicemail application.
- Reply/forward messages when logged into Modular Messaging.
- Testing of the Find Me and Call Me features on Modular Messaging with the DiVitas Clients.

1.2. Support

For technical support on the DiVitas Mobile Unified Communications Solution and how to configure dual mode handsets connected to it, consult the support pages at http://www.divitas.com/support.html or contact technical support at:

Telephone: (866) 857-6087E-Mail: support@divitas.com

¹ A DiVitas Client in cellular mode can only retrieve voicemail messages when it uses a Cellular Data Channel (CDC) to communicate with the DiVitas Server. In this configuration, a Cellular Voice Channel (CVC) was used. See the note in Section 2 for more information on CVC.

2. Reference Configuration

Figure 1 illustrates a sample configuration consisting of a pair of Avaya S8730 Servers running Avaya AuraTM Communication Manager, an Avaya G650 Media Gateway, Avaya Modular Messaging, Avaya AuraTM Session Manager, and dual-mode wireless telephones registered with DiVitas Mobile Unified Communications. Avaya Modular Messaging consisted of a Message Application Server (MAS) and a Message Storage Server (MSS). Each DiVitas Client was paired with an H.323 IP telephone on Avaya AuraTM Communication Manager. A SIP trunk was established between the DiVitas Mobile UC Server and AuraTM Session Manager. The Avaya G650 Media Gateway connected to the PSTN via an ISDN-PRI trunk. Avaya AuraTM System Manager was used to configure AuraTM Session Manager.

Note: While a DiVitas Client is in Cellular mode, it communicates with the DiVitas Mobile UC Server through a Cellular Voice Channel (CVC). The DiVitas Client places a call using a PSTN number assigned to the DiVitas Mobile UC Server. CVC enables the client to make and receive voice calls and use voice features such as hold and resume. CVC supports multiple simultaneous calls and is used when the Cellular Data Channel (CDC) is not available, which requires a public IP address assigned to the DiVitas Mobile UC Server. In this configuration, a CVC was used which doesn't allow retrieving voicemail messages when a DiVitas Client in Cellular mode. This requires a CDC.

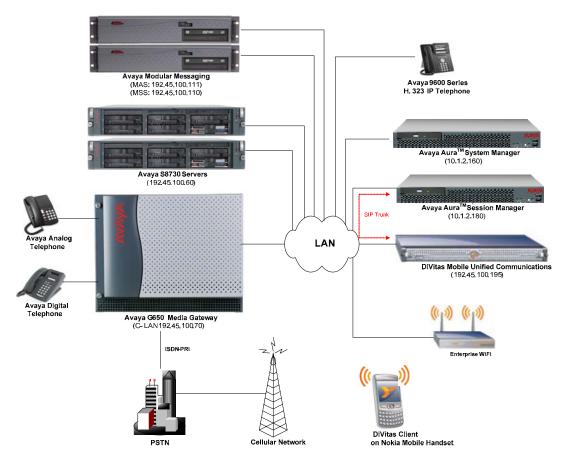


Figure 1: DiVitas Mobile Unified Communications with Avava SIP-based Network

3. Equipment and Software Validated

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

Equipment	Software/Firmware
Avaya Modular Messaging (MAS & MSS)	5.0
Avaya S8730 Server with G650 Media Gateway	Avaya Aura TM Communication Manager 5.2 (R015x.02.0.947.3) with Service Pack 1 (Patch 17294)
Avaya Aura TM Session Manager	1.1.3.1.18022
Avaya Aura TM System Manager	1.0
Avaya 9600 Series IP Telephones	3.0 (H.323)
Avaya 6400 Series Digital Telephones	
Avaya Analog Telephones	
DiVitas Mobile Unified Communications	2.7.2.0 Build 39
DiVitas Client on Nokia E51 and N95	2.7.2.0 Build 39

Table 1: Equipment and Software Validated

4. Configure Avaya AuraTM Communication Manager

This section provides the procedures for configuring Avaya AuraTM Communication Manager. The procedures include the following areas:

- Verify Customer Options for ISDN-PRI and QSIG Trunking
- Configure ISDN-PRI/QSIG Trunk to Avaya Modular Messaging
- Administer QSIG TSC and Extension Length
- Create Hunt Group (Pilot Number) and Coverage Path
- Configure Call Routing to Avaya Modular Messaging
- Create Voicemail Subscribers

Avaya AuraTM Communication Manager was configured using the System Access Terminal (SAT). Refer to [1] for additional details.

4.1. Verify Customer Options for H.323/QSIG Trunking

Avaya Modular Messaging supports various PBX integrations, such as SIP, ISDN/QSIG, and H.323/QSIG. In this configuration, an H.323/QSIG integration was implemented. For this solution, the following software features must be enabled in the **System Parameters Customer Options** form on Communication Manager.

- IP Station Ports
- ISDN-PRI
- Private Networking
- OSIG Features:
 - Basic Call Setup
 - o Basic Supplementary Services
 - o Supplementary Services with Rerouting
 - o Transfer into QSIG Voice Mail
 - o Value-Added (VALU)

4.2. Configure ISDN-PRI/QSIG Trunk to Avaya Modular Messaging

This section covers the configuration of the ISDN-PRI/QSIG trunk between Avaya AuraTM Communication Manager and Avaya Modular Messaging, including the DS1, signaling group, trunk group, and public numbering format.

Install the T1 circuit pack and configure the DS1 interface as shown below.

```
change ds1 2a09
                                                                       1 of
                                                                               2
                                DS1 CIRCUIT PACK
            Location: 02A09
                                                      Name: lzmas1-01
           Bit Rate: 1.544
                                               Line Coding: b8zs
  Line Compensation: 1
                                              Framing Mode: esf
      Signaling Mode: isdn-pri
            Connect: pbx
                                                 Interface: peer-master
  TN-C7 Long Timers? n
                                             Peer Protocol: Q-SIG
Interworking Message: PROGress
                                                      Side: a
                                                       CRC? n
Interface Companding: mulaw
          Idle Code: 11111111
                              DCP/Analog Bearer Capability: 3.1kHz
                                           T303 Timer(sec): 4
                                          Disable Restarts? n
      Slip Detection? n
                                        Near-end CSU Type: other
```

Figure 2: DS1

Configure a signaling group that will be assigned to the DS1 channels. The signaling group should be configured as shown below.

```
change signaling-group 90

SIGNALING GROUP

Group Number: 90

Group Type: isdn-pri

Associated Signaling? y

Page 1 of 1

SIGNALING GROUP

Max number of NCA TSC: 10

Primary D-Channel: 02A0924

Max number of CA TSC: 10

Trunk Group for NCA TSC: 90

Trunk Group for Channel Selection: 90

TSC Supplementary Service Protocol: b

Network Call Transfer? n
```

Figure 3: Signaling Group

Create a trunk group and assign the newly created DS1 channels to it. Configure the **Trunk Group** form as shown in **Figure 4**. Set the **Group Type** field to *isdn*, set the **Carrier Medium** to *PRI/BRI*, and configure the **Service Type** field to *tie*. Configure the other fields in bold and accept the default values for the remaining fields.

```
change trunk-group 90
                                                                       Page
                                                                              1 of 21
                                   TRUNK GROUP
                              Group Type: isdn CDR Reports: y
COR: 1 TN: 1 TAC: 190
Outgoing Display? y Carrier Medium: PRI/BRI
Group Number: 90
 Group Name: LZMAS
  Direction: two-way
                               Busy Threshold: 255 Night Service:
Dial Access? y
Queue Length: 0
Service Type: tie
                                       Auth Code? n
                                                                  TestCall ITC: rest
                         Far End Test Line No:
TestCall BCC: 4
```

Figure 4: Trunk Group – Page 1

On Page 2, set the **Supplementary Service Protocol** field to 'b' to enable QSIG signaling.

```
change trunk-group 90
                                                                Page
                                                                       2 of 21
      Group Type: isdn
TRUNK PARAMETERS
        Codeset to Send Display: 6 Codeset to Send National IEs: 6
        Max Message Size to Send: 260 Charge Advice: none
  Supplementary Service Protocol: b Digit Handling (in/out): enbloc/enbloc
            Trunk Hunt: cyclical
                                                   Digital Loss Group: 13
Incoming Calling Number - Delete: Insert:
Bit Rate: 1200 Synchronization: async
                                                             Format: unk-unk
                                                              Duplex: full
Disconnect Supervision - In? y Out? n
Answer Supervision Timeout: 0
         Administer Timers? n
                                      CONNECT Reliable When Call Leaves ISDN? n
```

Figure 5: Trunk Group – Page 2

On Page 3 of the trunk group form, set the **Format** field to *public*. This field specifies the format of the calling party number sent to the far-end. The **NCA-TSC Trunk Member** field should be set to the highest numbered channel in the trunk group. Configure the other fields in bold as shown below.

```
change trunk-group 90
                                                                                       3 of 21
                                                                              Page
TRUNK FEATURES
                                               Measured: none Widebanu Supper: Maintenance Tests? y
            ACA Assignment? n
                                   Internal Alert? n
Data Restriction? n
Send Name: y
Hop Dgt? n

Maintenance Tests? y
NCA-TSC Trunk Member: 10
Send Calling Number: y
Send EMU Visitor CPN? n
   Used for DCS? n Hop Dg Suppress # Outpulsing? n Format: public
 Outgoing Channel ID Encoding: preferred
                                                     UUI IE Treatment: service-provider
                                                          Replace Restricted Numbers? n
                                                          Replace Unavailable Numbers? n
                                                                  Send Connected Number: y
                                                            Hold/Unhold Notifications? y
                                                         Modify Tandem Calling Number? n
               Send UUI IE? y
                 Send UCID? n
 Send Codeset 6/7 LAI IE? y
                                                               Ds1 Echo Cancellation? n
    Apply Local Ringback? n
 Show ANSWERED BY on Display? y
                                   Network (Japan) Needs Connect Before Disconnect? n
```

Figure 6: Trunk Group - Page 3

In the trunk member pages, configure the number of channels required. In this example, 10 channels were added to the trunk group.

chang	ge trunk-	group 9	90		Page	5 of	21			
				TRUNK GROUP	TRUNK GROUP					
				Administ	Administered Members (min/max):					
GROUE	MEMBER	ASSIGN	MENTS	Tota	Total Administered Members:					
	Port	Code	Sfx Name	Night	Sig Grp					
1:	02A0901	TN464	F		90					
2:	02A0902	TN464	F		90					
3:	02A0903	TN464	F		90					
4:	02A0904	TN464	F		90					
5:	02A0905	TN464	F		90					
6:	02A0906	TN464	F		90					
7:	02A0907	TN464	F		90					
8:	02A0908	TN464	F		90					
9:	02A0909	TN464	F		90					
10:	02A0910	TN464	F		90					

Figure 7: Trunk Group Members – Page 5

Configure the **Public/Unknown Numbering Format** form to send the calling party number to Avaya Module Messaging. Add an entry so that the 5-digit extension of voicemail subscribers is sent to the voicemail system.

chai	change public-unknown-numbering 2 Page 1										
Ext	Ext	Trk	CPN	CPN							
Len	Code	Grp(s)	Prefix	Len							
					Total Admin	istered	d: 22				
5	2	90		5	Maximum	Entries	s: 999	99			

Figure 8: Public Unknown Numbering Format

4.3. Administer QSIG TSC and Extension Length

In the **Feature-Related System Parameters** form, configure an available extension in the **QSIG/ETSI TSC Extension** field. Specify the length of mailbox extensions in the **MWI** – **Number of Digits Per Voice Mail Subscriber**. In this example, 5-digit extensions are used.

```
change system-parameters features
                                                                Page
                                                                       8 of
                                                                            18
                        FEATURE-RELATED SYSTEM PARAMETERS
ISDN PARAMETERS
                                                         PARAMETERS FOR CREATING
 Send Non-ISDN Trunk Group Name as Connected Name? n
                                                         QSIG SELECTION NUMBERS
Display Connected Name/Number for ISDN DCS Calls? n
                                                         Network Level: 0
      Send ISDN Trunk Group Name on Tandem Calls? n
                                                           Level 2 Code:
                                                           Level 1 Code:
                          QSIG/ETSI TSC Extension: 24155
MWI - Number of Digits Per Voice Mail Subscriber: 5
                                 Feature Plus Ext:
                             National CPN Prefix:
                         International CPN Prefix:
                                Pass Prefixed CPN: ASAI? n VDN/Vector? n
   Unknown Numbers Considered Internal for AUDIX? n
             USNI Calling Name for Outgoing Calls? n
               Path Replacement with Measurements? y
                  QSIG Path Replacement Extension: 24151
   Send QSIG Path Replacement Conf. Event to ASAI? y
            Path Replace While in Queue/Vectoring? n
```

4.4. Create Hunt Group and Coverage Path

Configure a hunt group to be used as the call coverage point for voicemail subscribers. On Page 1 of the hunt group, specify a descriptive name and assign an available extension to the hunt group.

```
change hunt-group 55
                                                                      1 of 60
                                                               Page
                                 HUNT GROUP
           Group Number: 55
                                                          ACD? n
             Group Name: LZMAS
                                                        Queue? n
        Group Extension: 25108
                                                       Vector? n
             Group Type: ucd-mia
                                                Coverage Path:
                     TN: 1
                                    Night Service Destination:
                    COR: 1
                                             MM Early Answer? n
          Security Code:
                                       Local Agent Preference? n
ISDN/SIP Caller Display: mbr-name
```

Figure 9: Hunt Group – Page 1

Configure Page 2 of the hunt group, as shown below. Set the **Message Center** field to *qsig-mwi* and specify the **Voice Mail Number**. In this example, the Modular Messaging pilot number is 25108. Calls to Modular Messaging are routed using AAR as specified by the routing digit of '8' (i.e., AAR feature access code).

```
change hunt-group 55

HUNT GROUP

LWC Reception: none

AUDIX Name:

Message Center: qsig-mwi
Send Reroute Request: n
Voice Mail Number: 25108

Routing Digits (e.g. AAR/ARS Access Code): 8

TSC per MWI Interrogation? n
```

Figure 10: Hunt Group – Page 2

Set up a coverage path for the subscribers' extensions. Increase the number of rings so that calls to DiVitas Clients in Cellular mode will ring before covering to voicemail. In this example, setting the **Number of Rings** field to '5' provided enough time for a DiVitas Client in Cellular mode to answer the call, if available. Set the **Point1** field to the voicemail hunt group configured above.

change coverage path 55	COVERAGE 1		Page 1 of 1	
Cvg Enabled for VDN Ro	e Path Number: 59 oute-To Party? n : Path Number:		fter Coverage? n	
COVERAGE CRITERIA				
Station/Group Status Active? Busy? Don't Answer? All? DND/SAC/Goto Cover? Holiday Coverage?	Inside Call n y y n n y	Outside Call n y y n n y	Number of Rings: 5	
COVERAGE POINTS Terminate to Coverage F Point1: h55 Rr. Point3: Point5:	Pts. with Bridged ng: 2 Point2: Point4: Point6:	d Appearances?	n	

Figure 11: Voicemail Coverage Path

4.5. Call Routing to Avaya Modular Messaging

This section describes how to route calls to Avaya Modular Messaging. When calls cover to voicemail or a subscriber calls the voicemail system using the pilot number (25108), the call is routed using AAR as specified on the second page of the voicemail hunt group configured in Section 4.4 In the **AAR Digit Analysis Table**, add an entry for the voicemail pilot number. In this example, the call is routed over Route Pattern 250.

change aar analysis 25						Page 1 of	2				
	AA	AAR DIGIT ANALYSIS TABLE									
		Location: all Percent Full:									
Dialed	Tota	ıl	Route	Call	Node	ANI					
String	Min	Max	Pattern	Type	Num	Reqd					
25108	5	5	250	aar		n					

Figure 12: AAR Digit Analysis Table

Route Pattern 250 routes calls over the H.323 IP trunk to the MAS. As-needed CA-TSCs are enabled for the route preference in the second half of the form.

char	nge 1	cout	e-pat	terr	n 250							I	Page	1 of	3	
					Pattern N	omber (: 25	0 Pat	tern Na	me:	LZMAS					
						SCCAN	1? n	S	ecure S	IP?	n					
	Grp	FRL	NPA	Pfx	Hop Toll	No.	Inse	rted						DCS/	IXC	
	No			Mrk	Lmt List	Del	Digi	ts						QSIG		
						Dgts								Intw		
1:	90	0												n	user	
2:														n	user	
3:														n	user	
4:														n	user	
5:														n	user	
6:														n	user	
								_								
				TSC	CA-TSC	ITC	BCIE	Serv	ice/Fea	ture				_	LAR	
	0 1	2 M	4 W		Request							_	Form	at		
											Suk	paddre	ess			
1:	УУ	УУ	y n	У	as-needed	l rest	;								none	
2:	УУ	УУ	y n	n		rest	:								none	
3:	УУ	УУ	y n	n		rest	;								none	
4:	УУ	УУ	y n	n		rest	;								none	
5:	УУ	УУ	y n	n		rest	;								none	
6:	УУ	УУ	y n	n		rest									none	

Figure 13: Route Pattern for Calls to Modular Messaging

4.6. Create Voicemail Subscribers

This section describes how to map a desktop IP phone on Avaya AuraTM Communication Manager to a DiVitas Client running on a Nokia E51 or N95 handset. This would allow a desktop phone and the DiVitas Client to ring simultaneously when a call is received. The call can then be answered by either the desktop phone or the DiVitas Client.

Configure a station as shown in **Figure 14**. In this example, the station maps to an H.323 IP phone with an extension of 24511. The DiVitas Client has the same extension and is configured in **Figure 34**.

```
add station 24511
                                                            Page 1 of
                                    STATION
                                                                     BCC: 0
Extension: 24511
                                       Lock Messages? n
                                        Security Code: XXXXX
    Type: 9630
                                                                       TN: 1
                                      Coverage Path 1: 55
    Port: IP
                                                                      COR: 1
    Name: DiVitas 24511
                                      Coverage Path 2:
                                                                      COS: 1
                                      Hunt-to Station:
STATION OPTIONS
                                          Time of Day Lock Table:
             Loss Group: 19
                                  Personalized Ringing Pattern: 1
                                               Message Lamp Ext: 24511
       Speakerphone: 2-way
Display Language: english
                                            Mute Button Enabled? y
                                                 Button Modules: 0
Survivable GK Node Name:
         Survivable COR: internal
                                              Media Complex Ext:
  Survivable Trunk Dest? y
                                                    IP SoftPhone? n
                                             Customizable Labels? y
```

Figure 14: Station – Page 1

On Page 2 of the station form, set the **MWI Served User Type** field to *qsig-mwi* to enable MWI.

```
add station 24511
                                                                   2 of
                                                            Page
                                    STATION
FEATURE OPTIONS
          LWC Reception: spe
                                         Auto Select Any Idle Appearance? n
         LWC Activation? y
                                                   Coverage Msg Retrieval? y
 LWC Log External Calls? n
                                                             Auto Answer: none
           CDR Privacy? n
                                                        Data Restriction? n
  Redirect Notification? y
                                              Idle Appearance Preference? n
Per Button Ring Control? n
                                             Bridged Idle Line Preference? n
  Bridged Call Alerting? n
                                                 Restrict Last Appearance? n
 Active Station Ringing: single
                                                        EMU Login Allowed? n
                                  Per Station CPN - Send Calling Number?
       H.320 Conversion? n
      Service Link Mode: as-needed
                                                       EC500 State: disabled
        Multimedia Mode: enhanced
   MWI Served User Type: qsig-mwi
                                              Display Client Redirection? n
                                              Select Last Used Appearance? n
                                                Coverage After Forwarding? s
                                              Direct IP-IP Audio Connections? y
 Emergency Location Ext: 24511
                                       Always Use? n IP Audio Hairpinning? y
```

Figure 15: Station – Page 2

5. Configure Avaya Modular Messaging

This section provides the procedures for configuring Avaya Modular Messaging, including the Message Application Server (MAS) and the Message Storage Server (MSS). The procedures include the following areas:

- Configure the Message Waiting Indicator on the MAS.
- Configure an ISDN-PRI trunk between the MAS and Communication Manager.
- Enable the IMAP interface on the MSS.
- Configure subscribers on the MSS.

For complete coverage of Modular Messaging installation and configuration, refer to [4]. This section only focuses on the items listed above.

5.1. Configure Message Application Server (MAS)

Configuration of the Messaging Application Server (MAS) was performed using the Voice Mail System Configuration application. To launch the application, navigate to Programs \rightarrow Avaya Modular Messaging \rightarrow Voice Mail System Configuration from the MAS. Expand all the fields so that all the applicable options are visible as shown below.

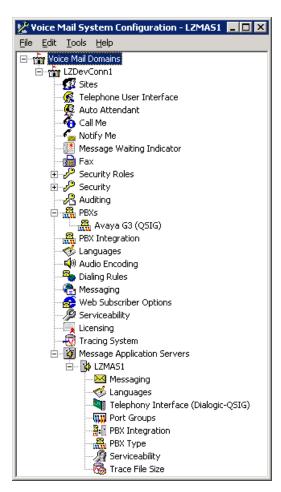


Figure 16: Voice Mail System Configuration

From the **Voice Mail System Configuration** tree-view, select **Telephone User Interface**. In the window displayed below, access the **General** tab and set the **Number of Digits in a Mailbox** field to the appropriate mailbox extension length. In this example, mailbox extensions were 5 digits in length.

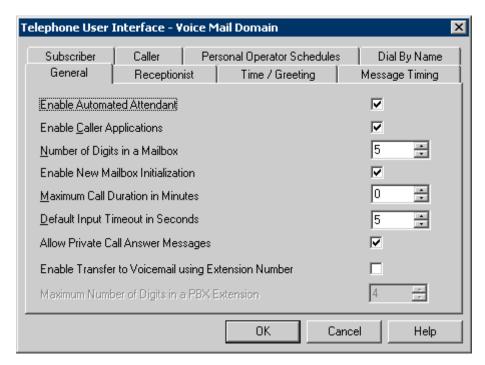


Figure 17: Telephone User Interface – General Tab

From the Voice Mail System Configuration tree-view, expand PBXs and select Avaya G3 (QSIG). In the window displayed below, access the Transfer/Outcall tab and set the Transfer Mode field to "full". Click OK.

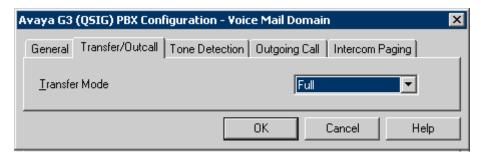


Figure 18: PBX Configuration – Transfer/Outcall Tab

In the **Outgoing Call** tab, set the **Origin Number** to the voice mail pilot number (i.e., 25108). That is, set the field to the number entered in the **Voice Mail Number** field on the Page 2 of the **Hunt Group** form shown in **Figure 10**. Click **OK** to save the changes.

Note: The MAS will prompt to restart the services. This step will be performed later.

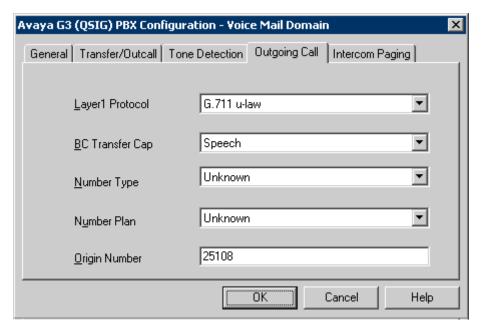


Figure 19: PBX Configuration – Outgoing Call Tab

Access the **Message Waiting Indicator** option and configure the parameters as follows:

- Enable Message Waiting Indicator (MWI) by checking the box.
- Enter the name of the MWI server created during the installation procedure in the MAS MWI server field. In this example, the MAS server, "LZMAS1", was also the MWI server.
- The box entitled Message Application Servers that support MWI should contain the MAS server capable of placing MWI requests. Again, the MWI server is "LZMAS1".
- Click **OK** to save the changes.

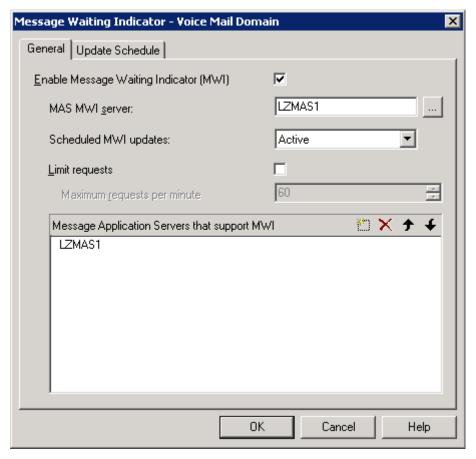


Figure 20: Message Waiting Indicator

Access the **Telephony Interface** (**Dialogic-QSIG**) option and set the **Maximum Concurrent Calls** field to the number of ports supported for the ISDN-PRI trunk. In this example, 10 ports were used. By default the ports are enabled. Click **OK** to save the changes. Restart the MAS Service when prompted and then continue with the step below.

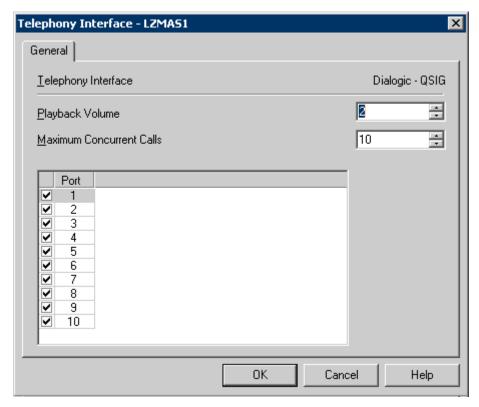


Figure 21: Telephony Interface

Access the **Port Groups** options under the MAS name in **Figure 16** and click the **Add Group** button. Provide a name for the **Port Group** and only enable "Outgoing" for the **Port Group Usage** field. Uncheck all of the ports, except the ports that will be used for MWI. In this example, port 10 is being used for MWI. Click **OK**.

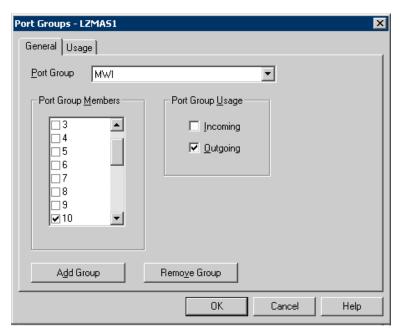


Figure 22: Port Groups - MWI

Next, select the "Default Group" in the **Port Group** field and ensure that it is configured to meet the customer's need for "Incoming" and "Outgoing" under **Port Group Usage** by checking all ports, except for the port used for MWI. Click **OK** to save the changes.

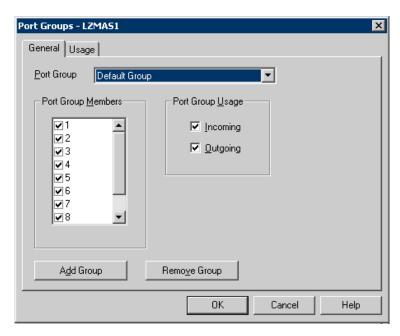


Figure 23: Port Groups – Default Group

Next, access **PBX Integration** and within the **General** tab select "QSIG" for the **Integration Type**.

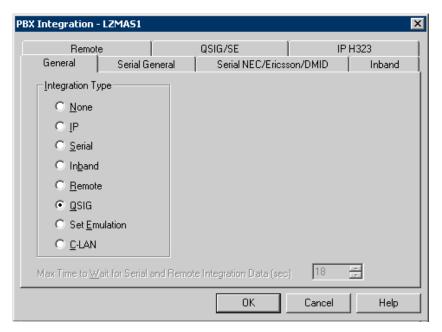


Figure 24: PBX Integration – General Tab

Next, configure the **QSIG/SE** tab with the following settings:

- Specify the MWI Port Group.
- Set the Max MWI Sessions the appropriate value. In this example, one session was used.

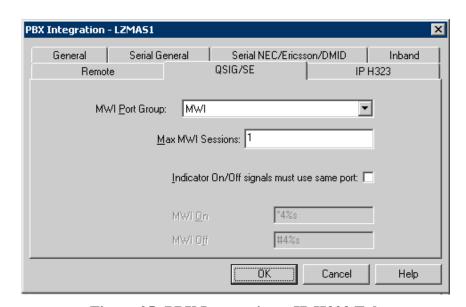


Figure 25: PBX Integration – IP H323 Tab

Finally, access **PBX Type** and verify that the **Telephony Type** field is set to "Dialogic QSIG" and that "Avaya G3 (QSIG)" is selected under **PBXs**. Click **OK** to save the changes.

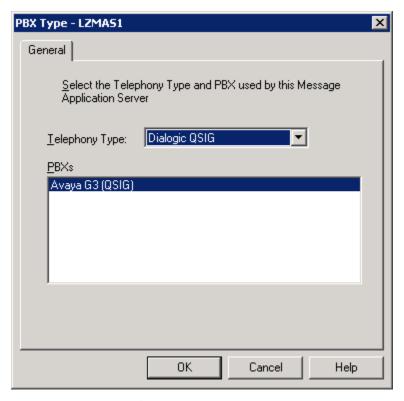


Figure 26: PBX Type

After configuring the voicemail system, restart the Message Application Server (MAS) services to apply these changes.

5.2. Configure the Message Storage Server (MSS)

This section describes the procedure for enabling the IMAP interface and configuring a voicemail subscriber. The Message Storage Server (MSS) is configured from a web interface. Launch a web browser and enter the MSS IP address in the URL field. The **Messaging Administration** webpage is displayed as shown below.

For complete coverage of MSS configuration, refer to [4]. This section only covers enabling the IMAP interface and adding a subscriber.

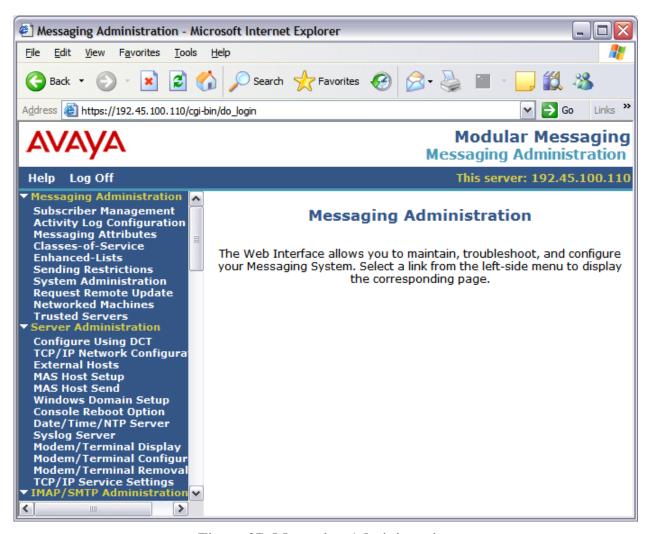


Figure 27: Messaging Administration

From the left pane of the **Messaging Administration** web page, select **System Administration** to enable the IMAP interface. **Figure 28** shows the bottom half of the webpage with the **System TCP/IP Ports** section. Enable the **IMAP4 Port** associated with port 143. Click **Save**.

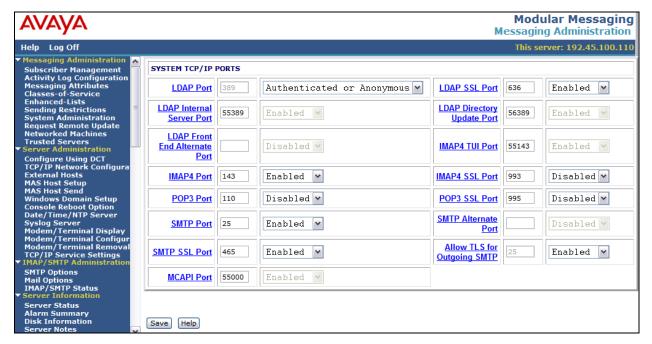


Figure 28: System Administration – System TCP/IP Ports Section

To add a subscriber, select **Subscriber Management** in the left pane and then enter the subscriber extension in the **Local Subscriber Mailbox Number** field on the webpage that appears (not shown). The **Add Local Subscriber** webpage is displayed as shown in **Figure 29**. Configure the **Last Name** and **First Name** fields corresponding to the subscriber. Next, enter the extension assigned to the desktop phone configured on Avaya Aura TM Communication Manager in the **Mailbox Number** and **Numeric Address** fields. Assign an appropriate **Class of Service** to the subscriber. In this example, extension 24511 is added as a subscriber and the **Class of Service** specified uses the AUDIX TUI. At the bottom of the webpage, click the **Save** button (not shown).

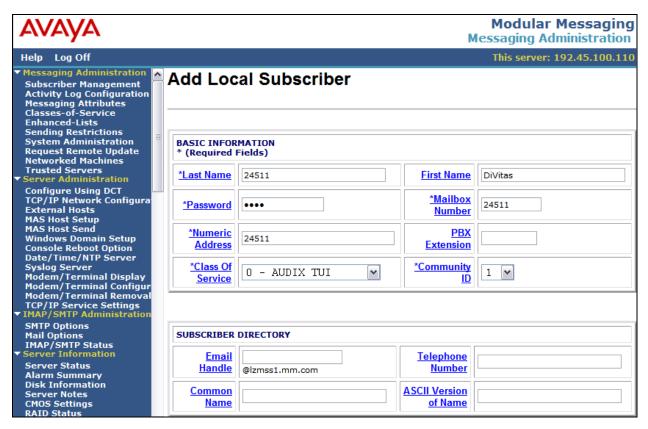


Figure 29: Add Local Subscriber

6. Configure DiVitas Mobile Unified Communications

This section describes the steps for enabling the IMAP interface on the DiVitas Server and enabling voicemail on the DiVitas Clients. Refer to [6] for additional configuration information.

All DiVitas Mobile UC Server configuration and management features are accessed from a Webbased interface. From an Internet browser, enter the IP address of the DiVitas Mobile UC Server in the URL field and log in using the appropriate credentials. The screen shown in **Figure 30** is displayed.

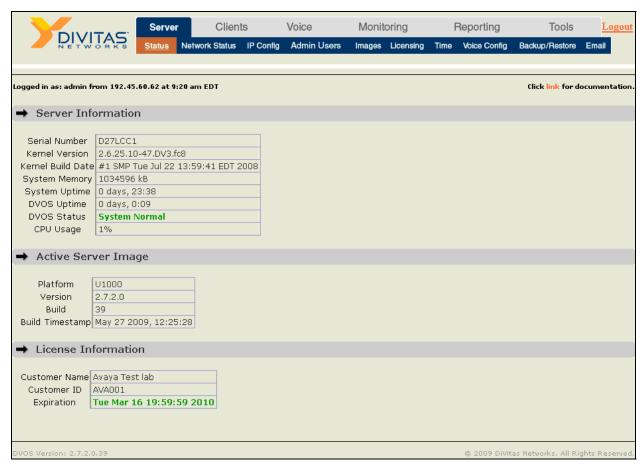


Figure 30: DiVitas Mobile UC Server Web Interface

In the **Server** → **IP Config** webpage, configure the IP network parameters of the DiVitas Server corresponding to the customer's network as shown in **Figure 31**. The remaining fields on this webpage (not shown) may be left at the default values. Click **Submit**.

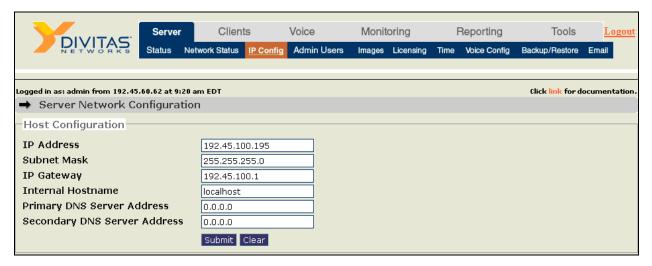


Figure 31: Server IP Configuration

In the **Voice > Voicemail** webpage, configure the **IMAP Global Configuration** parameters as follows:

- Enable IMAP Voicemail Support.
- Set the **IMAP Server Address** to the IP Address of the MSS (e.g., 192.45.100.110).
- Set the **IMAP Server Port** to "143".
- Set the IMAP Voicemail Host to the "Avaya MM".
- Set the **IMAP User Polling Interval** to an appropriate value. In this example, the DiVitas Server was configured to check for new messages every 15 seconds.
- Configure the other parameters as shown in Figure 32.
- Click Submit.

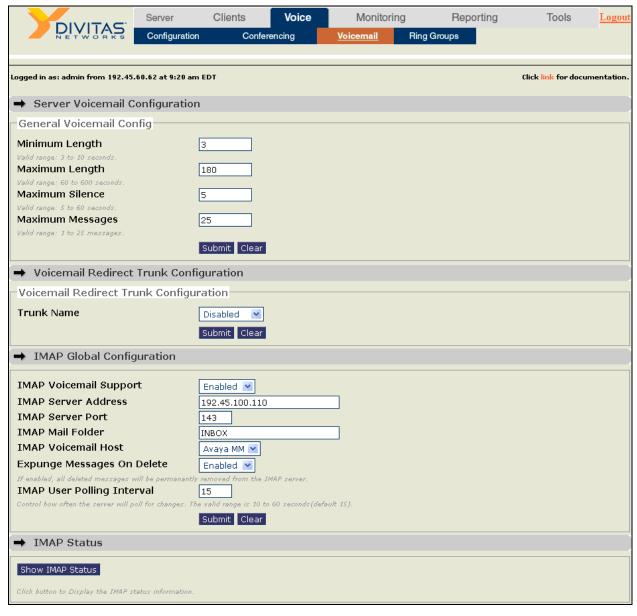


Figure 32: IMAP Global Configuration

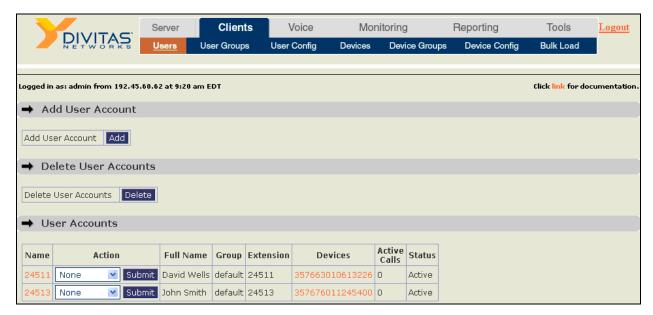


Figure 33: User Accounts

When adding a **User**, specify the user's **Full Name** and **Extension** as shown in **Figure 34**. The figure below shows the user account after it has been configured. The **Add User Account** webpage will appear slightly different, but contain similar fields. To enable voicemail with Avaya Modular Messaging, configure the **IMAP Configuration** section. Enable IMAP and specify the appropriate username and password configured for this subscriber in the MSS (see **Figure 29**). Click **Submit**.

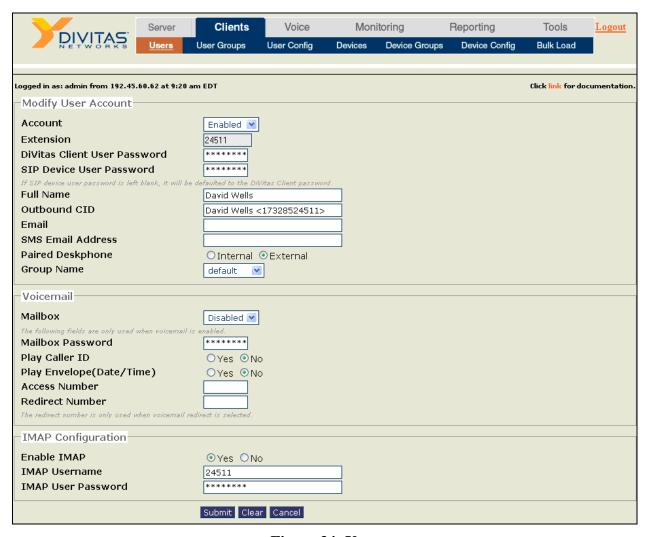


Figure 34: User

In the Clients Device Config webpage, increase the Cell No Answer Timeout field to allow a DiVitas Client in Cellular mode to ring before the call covers to voicemail. Click Submit.



Figure 35: Client Device Timer Configuration

7. General Test Approach and Test Results

The focus of the interoperability compliance test was to verify that the DiVitas Server can access voicemail messages stored in the Message Storage Server (MSS) of Avaya Module Messaging using the IMAP interface, and that the DiVitas Client running on Nokia E51 and N95 mobile handsets can retrieve, play back, and delete the voicemail message on their visual voicemail application.

All tests passed; however, a few observations were noted.

- While the DiVitas Client was in Cellular mode and logged into Modular Messaging, the DiVitas Client could not delete a message (*D) or transfer out of voicemail (*T). Note that these are options available when using the AUDIX TUI. Activating other options that didn't begin with an asterisk successful. This behavior was not observed when the DiVitas Client was in WiFi mode.
- When a subscriber replies/forwards a voicemail message with a comment from the Modular Messaging menu to a DiVitas Client, the DiVitas Client can retrieve the message from the visual voicemail application, but only the original message is heard. The comment added by the subscriber is not heard.
- Replying and forwarding voicemail messages from the DiVitas Client's visual voicemail application is not yet supported.
- To retrieve a voicemail message from a DiVitas Client in Cellular mode requires a Cellular Data Channel (CDC) a Cellular Voice Channel (CVC) is not sufficient for this purpose. This is expected behavior.

8. Verification Steps

This section provides the verification steps that may be performed to verify voicemail coverage for the DiVitas Clients and that the DiVitas Client could retrieve voicemail messages from their visual voicemail application.

- 1. From the Avaya AuraTM Communication Manager SAT, verify that the ISDN-PRI signaling group and trunk group to Avaya Modular Message are in-service using the **status signaling-group** and **status trunk** commands, respectively.
- 2. From the MAS, open the **Port Monitor** application by navigating to **Programs** Avaya Modular Messaging Port Monitor. Select the appropriate MAS from the list when prompted and ensure that the ISDN-PRI trunk ports are in the "idle" state as shown below.

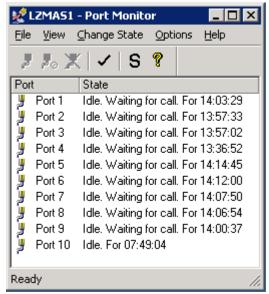


Figure 36: Port Monitor

3. From the DiVitas web interface, navigate to the **Voice Voicemail** webpage and verify that the **IMAP Server Status** is "Active" and that the DiVitas Clients can log into Modular Messaging as indicated by the **User State** being set to "Authenticated".

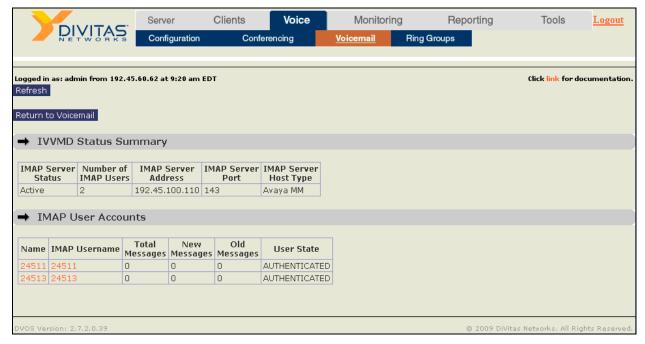


Figure 37: User's Voicemail Status

- 4. Place a call to a DiVitas Client and allow the call to cover to voicemail. Leave a message.
- 5. After leaving a new voicemail, verify that the visual voicemail application running on the DiVitas Client shows a new message as seen in **Figure 38**.



Figure 38: Visual Voicemail Application with New Message

In the list of voicemail messages as shown below, listen to the message on the DiVitas Client.

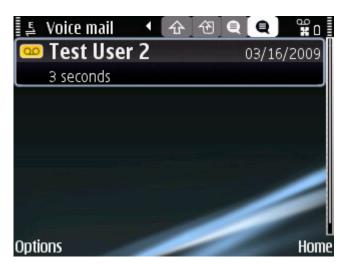


Figure 39: Voicemail Messages

9. Conclusion

These Application Notes describe the configuration steps required to provide voicemail coverage to DiVitas Clients registered to DiVitas Mobile Unified Communications using Avaya Modular Messaging. The DiVitas Server was able to access voicemail from Avaya Modular Messaging using the IMAP interface and the DiVitas Clients running on Nokia E- and N-Series mobile handsets were able to retrieve, play back, and delete messages using their visual voicemail application.

10. Additional References

This section references the product documentation that is relevant to these Application Notes.

- [1] Administrator Guide for Avaya AuraTM Communication Manager, Document 03-300509, Issue 5, May 2009, available at http://support.avaya.com.
- [2] Avaya AuraTM Communication Manager Feature Description and Implementation, Document 555-245-205, Issue 7, May 2009, available at http://support.avaya.com.
- [3] SIP Support in Avaya AuraTM Communication Manager Running on Avaya S8xxx Servers, Issue 9, May 2009, Document Number 555-245-206, available at http://support.avaya.com.
- [4] Modular Messaging for the Avaya Message Storage Server (MSS) Configuration Release 5.0 Installation and Upgrades, February 2009, available at http://support.avaya.com.
- [5] Application Notes for DiVitas Mobile Unified Communications with Avaya Communication Manager and Avaya SIP Enablement Services, Issue 1.0, available at http://support.avaya.com.
- [6] DiVitas Server Administration Guide, Version 2.7, Part Number: DOC-DVOS-AG-206.

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