

Avaya Solution & Interoperability Test Lab

Application Notes for Configuring QuesCom 400 IP/GSM Gateway with Avaya Communication Manager using H.323 Trunks – Issue 1.0

Abstract

These Application Notes describe the configuration steps required for the QuesCom IP/GSM 400 to successfully interoperate with Avaya Communication Manager using H.323 trunks. The QuesCom 400 IP/GSM is an IP-GSM-gateway, supporting outgoing and incoming GSM calls. All GSM calls made from Avaya Communications Manager will be routed to the QuesCom 400 IP/GSM gateway to the GSM network. The QuesCom 400 IP/GSM can also receive calls from the GSM network and route the calls to Avaya Communications Manager. The QuesCom 400 IP/GSM can provide a backup route for the PSTN and also be backed up by the PSTN.

Information in these Application Notes has been obtained through compliance testing and additional technical discussions. Testing was conducted via the Developer *Connection* Program at the Avaya Solution and Interoperability Test Lab.

1. Introduction

These Application Notes describe a compliance-tested configuration consisting of a QuesCom 400 IP/GSM gateway networked with Avaya Communication Manager 3.1 using H.323 trunks.

The QuesCom 400 IP/GSM is an IP-GSM-gateway, supporting outgoing and incoming GSM calls. All calls made from Avaya Communications Manager destined for the GSM network will be routed to the QuesCom 400 IP/GSM gateway. The QuesCom 400 IP/GSM can also receive calls from the GSM network and route the calls to Avaya Communications Manager. The QuesCom 400 IP/GSM can provide a backup route for the PSTN and also be backed up by the PSTN. This can be configured in Avaya Communication Manager using Automatic Route Selection (ARS). These Application Notes focus on a configuration where an H.323 IP trunk connects Avaya Communication Manager and the QuesCom 400 IP/GSM.

In **Figure 1**, Avaya Communication Manager runs on the Avaya S8500 Media Server. The solution described herein is also extensible to other Avaya Media Servers and Media Gateways. The Avaya G650 Media Gateway is connected to the PSTN via an E1 ISDN-PRI line and to the QuesCom 400 IP/GSM via an H.323 IP trunk. The QuesCom in turn connects to the GSM network via Subscriber Identity Module (SIM) cards that reside on GSM boards inserted in the QuesCom 400 IP/GSM.

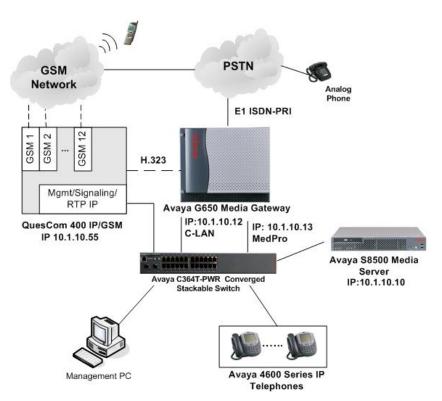


Figure 1: Avaya Communication Manager with QuesCom IP/GSM 400

2. Equipment and Software Validated

Equipment	Software
Avaya S8500 Media Server – Avaya Communication	3.1.2 (03.1-01.0.632.1)
Manager	
Avaya C364T-PWR Converged Stackable Switch	4.3.12
Avaya 4620SW IP Telephones	2.2.3
Avaya G650 Media Gateway	-
TN2312BP IP Server Interface	30
TN799DP C-LAN Interface	17
TN2302AP IP Media Processor	110
TN464CP DS1 Interface	18
QuesCom 400 IP/GSM	IAD04.20 B029 P006
Additional patch needed during compliance testing	ProxyH323.dll version
	4.20.017

3. Configure Avaya Communication Manager

Initial configuration of Avaya Communication Manager is beyond the scope of these Application Notes. See Section 9 for Avaya documentation references. This section illustrates the configuration of the H.323 trunk groups and signalling groups, dial plan, ARS analysis, and route patterns used in the compliance-tested configuration. The steps are performed from the System Access Terminal (SAT) interface.

3.1. Avaya Communication System parameters Special Applications

Step	Description				
1.	Display the system-parameters special-applications command and ensemble H245 Support With Other Vendors ? is set to "y". If not, contact your representative to enable this feature. Note: This feature only needs to be enabled on the QuesCom 400 IP/GSM.	Avaya	auth	oriz	ed sales
	display system-parameters special-applications SPECIAL APPLICATIONS	Page	4	of	6
	(SA8481) - Replace Calling Party Number with ASAI ANI? n (SA8500) - Expanded UUI Display Information? n (SA8506) - Altura Interoperability (FIPN)? n (SA8507) - H245 Support With Other Vendors? y (SA8508) - Multiple Emergency Access Codes? n (SA8510) - NTT Mapping of ISDN Called-Party Subaddress IE? n				

3.2. PSTN E1 ISDN-PRI

This section displays the PSTN E1 ISDN-PRI configuration on Avaya Communication Manager in the sample configuration of **Figure 1**. See Section 9 for Avaya documentation references.

Description Step Enter the **display ds1** <board location> to display the PSTN DS1 Circuit Pack configuration. 1. display ds1 01A12 DS1 CIRCUIT PACK Location: 01A12 Name: PRI to BT Bit Rate: 2.048 Line Coding: hdb3 Signaling Mode: isdn-pri Connect: network TN-C7 Long Timers? n Country Protocol: etsi Interworking Message: PROGress Protocol Version: b Interface Companding: alaw CRC? y Idle Code: 01010100 DCP/Analog Bearer Capability: 3.1kHz T303 Timer(sec): 4 Slip Detection? n Near-end CSU Type: other

2. Enter the **display trunk-group** < number> to display the PSTN trunk-group configuration.

```
display trunk-group 19

TRUNK GROUP

Group Number: 19

Group Type: isdn

CDR Reports: y

Group Name: PRI to BT

COR: 1

Direction: two-way

Outgoing Display? n

Carrier Medium: PRI/BRI

Dial Access? y

Busy Threshold: 255

Queue Length: 0

Service Type: public-ntwrk

Far End Test Line No:

TestCall BCC: 4
```

```
display trunk-group 19
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: a Digit Handling (in/out): enbloc/overlap

Trunk Hunt: cyclical QSIG Value-Added? n
Digital Loss Group: 13

Incoming Calling Number - Delete: Insert: Format:
Bit Rate: 1200 Synchronization: async Duplex: full
Disconnect Supervision - In? y Out? n
Answer Supervision Timeout: 0
```

Step	Description
_	-
	display trunk-group 19 Page 3 of 22 TRUNK FEATURES
	ACA Assignment? n Measured: both Wideband Support? n
	Maintenance Tests? y Data Restriction? n NCA-TSC Trunk Member:
	Send Name: n Send Calling Number: y
	Used for DCS? n Send EMU Visitor CPN? n
	Suppress # Outpulsing? y Format: public
	Outgoing Channel ID Encoding: preferred UUI IE Treatment: shared
	Maximum Size of UUI IE Contents: 128
	Replace Restricted Numbers? n
	Replace Unavailable Numbers? n
	Send Connected Number: y Hold/Unhold Notifications? y
	Send UUI IE? y Modify Tandem Calling Number? n
	Send UCID? n BSR Reply-best DISC Cause Value: 31
	Send Codeset 6/7 LAI IE? y Ds1 Echo Cancellation? n
	Apply Local Ringback? n US NI Delayed Calling Name Update? n
	Network (Japan) Needs Connect Before Disconnect? n
	display trunk-group 19 Page 6 of 22
	TRUNK GROUP
	Administered Members (min/max): 1/5
	GROUP MEMBER ASSIGNMENTS Total Administered Members: 5
	Port Code Sfx Name Night Sig Grp
	1: 01A1201 TN2464 C 19 2: 01A1202 TN2464 C 19
	3: 01A1202 IN2464 C 19
	4: 01A1204 TN2464 C 19
	5: 01A1205 TN2464 C 19
3.	Enter the display signaling-group <number> to display the PSTN signaling-group</number>
	configuration.
	display signaling-group 19 Page 1 of 5
	SIGNALING GROUP
	Group Number: 19 Group Type: isdn-pri
	Associated Signaling? y Max number of NCA TSC: 5
	Primary D-Channel: 01A1216 Max number of CA TSC: 5 Trunk Group for NCA TSC: 19
	Trunk Group for Channel Selection: 19 X-Mobility/Wireless Type: NONE
	Supplementary Service Protocol: a

3.3. H.323 Trunks and Signaling Groups

The steps in this section create an H.323 IP trunk group to the QuesCom 400 IP/GSM.

Step	Description									
1.	Enter the change n	ode-names ip command.	Specify the noc	de name and IP address for the						
	QuesCom 400. Observe the node name of the C-LAN card shown in bold. These node names									
	will be used in the configuration of the H.323 signalling group in Step 4.									
	will be used ill the t	Configuration of the 11.32	5 signaming grou	ip iii 5tep 4.						
	-1			Davis 1 a.f. 1						
	change node-name:		DDE NAMES	Page 1 of 1						
	Name	IP Address	Name	IP Address						
	AEServer	10 .1 .10 .20	Name							
	Abacus	10 .1 .10 .31								
	IPO412a_DC1	10 .1 .20 .10								
	Quescom	10 .1 .10 .55								
	S8300a_DC1	10 .1 .30 .10								
	S8500_Val1	10 .1 .10 .14								
	SEServer	10 .1 .10 .22								
	clan1a_DC1	10 .1 .10 .12								
	default	0 .0 .0 .0 10 .1 .10 .13								
	medprola_DC1 procr	10 .1 .10 .13								
2.	Page 1 of the trunk Group Typ Group Nan TAC – ente Carrier Me	a-group form, configure to e – set to "isdn". ne – enter a meaningful nor a Trunk Access Code the dium – set to "H.323". ne – set to "tie".	he following: ame/description	tilable trunk group number. On the provisioned dial plan.						
	add trunk-group	TRUNK (ROUP	Page 1 01 21						
	Group Number: 27 Group Name: H. Direction: two Dial Access? y Queue Length: 0	323 Quescom o-way Outgoing I	cOR: 1 Display? n reshold: 255	CDR Reports: y TN: 1 TAC: 727 Carrier Medium: H.323 Night Service:						

Auth Code? n

Service Type: tie

Member Assignment Method: manual

3. On the TRUNK FEATURES screen (Page 3 of the trunk-group form), set the Send Name, Send Calling Number and Send Connected Number to "y" and verify the Format is set to "public" as shown below.

```
add trunk-group 27
                                                                  3 of 21
                                                           Page
TRUNK FEATURES
         ACA Assignment? n
                                     Measured: none
                               Internal Alert? n
                                                       Maintenance Tests? y
                             Data Restriction? n
                                                    NCA-TSC Trunk Member: 1
                                    Send Name: y
                                                     Send Calling Number: y
           Used for DCS? n
                                                     Send EMU Visitor CPN? n
  Suppress # Outpulsing? n
                              Format: public
                                           UUI IE Treatment: service-provider
                                               Replace Restricted Numbers? n
                                              Replace Unavailable Numbers? n
                                                    Send Connected Number: y
                                                Hold/Unhold Notifications? n
```

- 4. Enter the **add signaling-group n** command, where "n" is an unused signalling group number. On Page 1 of the form, configure the following:
 - **Group Type** set to "h.323".
 - **Trunk Group for Channel Selection** enter the number of the trunk group configured in Step 2.
 - Near-end Node Name enter the node name of a local C-LAN board from Step 1.
 - **Near-end Listen Port** specify the local listen port, typically 1720.
 - Far-end Node Name enter the node name of the QuesCom configured in Step 1.
 - **Far-end Listen Port** specify the listen port, typically 1720.
 - **Direct IP-IP Audio Connections** set to "n".

```
add signaling-group 27
                                                           Page
                                                                  1 of
                                                                         5
                               SIGNALING GROUP
 Group Number: 27
                             Group Type: h.323
                          Remote Office? n
                                                   Max number of NCA TSC: 5
                                    SBS? n
                                                   Max number of CA TSC: 5
                               IP Video? n
                                                 Trunk Group for NCA TSC: 27
      Trunk Group for Channel Selection: 27
         Supplementary Service Protocol: a
                        T303 Timer(sec): 10
  Near-end Node Name: clan1a_DC1
                                            Far-end Node Name: Quescom
Near-end Listen Port: 1720
                                         Far-end Listen Port: 1720
                                     Far-end Network Region: 1
        LRQ Required? n
                                       Calls Share IP Signaling Connection? n
        RRQ Required? n
                                             H245 Control Addr On FACility? n
                                            Bypass If IP Threshold Exceeded? n
                                                    H.235 Annex H Required? n
        DTMF over IP: out-of-band
                                             Direct IP-IP Audio Connections? n
                                                      IP Audio Hairpinning? n
                                                Interworking Message: PROGress
```

		Descriptio	11			
Enter the change trunk-group n command, where "n" is the trunk group number configured in						
Step 2. On Page 3 of the	trunk-grou	ıp form, configu	re the following:	-		
• "IP" for Port . Th	ne number o	f ports configure	d should be coordinated v	vith the number of		
	_	•	•			
	io signamig	group comigure.	a m step s for sig sip.			
change trunk-group 27			Page	5 of 21		
		TRUNK GROUP				
		Administ	ered Members (min/max):	1/10		
GROUP MEMBER ASSIGNME	NTS	Total	l Administered Members:	10		
Dank	Nama	Mi wh t	Gi - G			
	Name	NIGHT	0 -	-		
			= 			
			- •			
			= 			
	Step 2. On Page 3 of the "IP" for Port. The SIM cards availa The number of the change trunk-group 27	Step 2. On Page 3 of the trunk-grou "IP" for Port. The number of SIM cards available in the Q The number of the signaling change trunk-group 27 GROUP MEMBER ASSIGNMENTS Port Name 1: IP 2: IP 3: IP 4: IP	Step 2. On Page 3 of the trunk-group form, configure "IP" for Port . The number of ports configure SIM cards available in the QuesCom 400 gate The number of the signaling group configures thange trunk-group 27 TRUNK GROUP Administer GROUP MEMBER ASSIGNMENTS Port Name Night 1: IP 2: IP 3: IP 4: IP	Step 2. On Page 3 of the trunk-group form, configure the following: • "IP" for Port . The number of ports configured should be coordinated voiced SIM cards available in the QuesCom 400 gateway. • The number of the signaling group configured in Step 3 for Sig Grp . Change trunk-group 27 Page TRUNK GROUP Administered Members (min/max): GROUP MEMBER ASSIGNMENTS Total Administered Members: Port Name Night Sig Grp 1: IP 27 2: IP 3: IP 27 4: IP 27		

3.4. ARS Tables and Route Patterns

In the sample configuration described in these Application Notes, when placing outbound calls to the public network, stations on Avaya Communication Manager must first dial the ARS Feature Access Code (FAC) before dialing an external number. The single digit "9" was used as the ARS FAC in the compliance-tested configuration (configuration step not shown).

Step	Description								
1.	Enter the change ars analysis 0 command. Configure Dialed String entries according to								
	customer requirements. In t	customer requirements. In the example below, the entries match dialed numbers as follows:							
	• The "078" Dialed St		-	*					
	routes calls to Route	_		_					*
			111 /0	. Poi exam	pie, a u	iaicu iii		091 900	126 Would
	be matched by this e	ntry.							
	change ars analysis 0	70.	DG D1		TO MADI		Page	1 of	2
		A	RS DI	GIT ANALYS Location:		ı Ei	Percent	₽.,11·	0
				Location.	all		Percent	ruii.	U
	Dialed	Tot	al	Route	Call	Node	ANI		
	String	Min	Max	Pattern	Type	Num	Reqd		
	01	11	11	9	pubu		n		
	078	11	11	78	pubu		n		
	079	11	11	79	pubu		n		
	123	3	3	9	pubu		n		
							n		
1									

- 2. Enter the **change route-pattern n** command, where "n" is the route pattern that processes dialed numbers configured in Step 1. Add two routing preference entries as follows:
 - 1) First Routing Preference H.323 IP trunk to QuesCom 400
 - **Grp No** enter the trunk group number routed to the QuesCom 400 gateway (see Section 3.3, Step 2)
 - **FRL** assign a Facility Restriction Level to this routing preference.
 - LAR set Look Ahead Routing to "next" to rehunt within the next routing preference if calls are rejected. LAR allows Avaya Communication Manager to re-attempt the call on another channel if the call is rejected with certain cause values.
 - 2) Second Routing Preference PSTN E1 ISDN-PRI
 - **Grp No** enter the trunk group that contains trunk members from the PSTN E1 ISDN-PRI (see Section 3.2 Step 2).
 - **FRL** assign a Facility Restriction Level to this routing preference.

na	nge :	rout	e-pa	tteri		<u> </u>	3
					Pattern 1	Number: 78 Pattern Name: Quescom H.323	
						SCCAN? n Secure SIP? n	
	_	FRL	NPA		-		/ IXC
	No			Mrk	Lmt List	Del Digits QSI	3
						Dgts Into	V
_	27	0				n	user
	19	0				n	user
3:						n	user
4:						n	user
5:						n	user
6:						n	user
						ITC BCIE Service/Feature PARM No. Numbering	LAR
	0 1	2 3	4 W		Request	Dgts Format	
						Subaddress	
1:	УУ	УУ	y n	n		rest	next
			y 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	v v	v n	n		rest	none

3.5. Avaya Extension to Cellular(EC500) Configuration

Avaya Extension to Cellular allows a cell phone to be treated as if it were an extension defined in Avaya Communication Manager. This is accomplished by mapping the user's main office phone to the cellular telephone number. All other types of calls, such as direct calls to and from the published cell phone number, are unaffected by Extension to Cellular.

```
Step
                                               Description
 1.
      Verify that the Avaya Communication Manager license has permissions for EC500 extensions.
      Enter the command display system-parameters customer-options. On the OPTIONAL
      FEATURES screen, verify that the Maximum Off-PBX Telephones - EC500 is sufficient.
       display system-parameters customer-options
                                                                             Page
                                                                                    1 of 11
                                         OPTIONAL FEATURES
            G3 Version: V13
               Location: 2
                                                        RFA System ID (SID): 1
               Platform: 12
                                                        RFA Module ID (MID): 1
                                                                         USED
                                         Platform Maximum Ports: 3200 259
                                               Maximum Stations: 2400 209
                                       Maximum XMOBILE Stations: 2400 0
                            Maximum Off-PBX Telephones - EC500: 2400 1
                             Maximum Off-PBX Telephones - OPS: 2400 32
                             Maximum Off-PBX Telephones - SCCAN: 2400 0
      Enter the display system-parameters customer-options command. On the OPTIONAL
      FEATURES screen, ensure that Enhanced EC500? is set to "y" as shown below.
       display system-parameters customer-options
                                                                                    4 of 11
                                                                             Page
                                         OPTIONAL FEATURES
          Emergency Access to Attendant? y
                                                                             IP Stations? y
                   Enable 'dadmin' Login? y

Enhanced Conferencing? y

Enhanced EC500? y

Internet Protocol (IP) PNC? n

ISDN Feature Plus? n

ISDN Network Call Redirection? n
           Enterprise Survivable Server? n
                                                                        TSDN-BRI Trunks? n
               Enterprise Wide Licensing? n
                                                                                ISDN-PRI? y
                      ESS Administration? n
                                                            Local Survivable Processor? n
                  Extended Cvg/Fwd Admin? y
                                                                   Malicious Call Trace? n
             External Device Alarm Admin? n
                                                               Media Encryption Over IP? n
```

Five Port Networks Max Per MCC? n Mode Code for Centralized Voice Mail? n

Global Call Classification? y Multimedia Appl. Server Interface (MASI)? n

Hospitality (Basic)? y Multimedia Call Handling (Basic)? n

Flexible Billing? n

IP Trunks? y

Forced Entry of Account Codes? n

Hospitality (G3V3 Enhancements)? n

Multifrequency Signaling? y

Multimedia Call Handling (Enhanced)? n

3. Enter the **change off-pbx-telephone configuration-set n**, command, where "n" is an available configuration set number. Enter a descriptive name for the **Configuration Set Description**. The rest of the parameters can be left with default values.

```
Configuration Set Description: H323

Calling Number Style: network
CDR for Origination: phone-number

CDR for Calls to EC500 Destination: y
Fast Connect on Origination: n
Post Connect Dialing Options: dtmf
Cellular Voice Mail Detection: none
Barge-in Tone: n
Calling Number Verification: y
Identity When Bridging: principal
```

- **4.** Enter the **change off-pbx-telephone station-mapping n**, command where "n" is the station that will be mapped to the cell phone. In the example below station "10000" is mapped to cell number "07891900128"
 - **Station Extension** set to the office extension that will be mapped to the cell number.
 - **Application** enter "EC500".
 - **Phone Number** enter the cell phone number.
 - **Trunk Selection** specify the outgoing trunk selection to "ars".
 - Configuration Set enter the number of the configuration set configured in Step 3.

```
change off-pbx-telephone station-mapping 10000
                                                            Page
                                                                   1 of
                                                                         2
                STATIONS WITH OFF-PBX TELEPHONE INTEGRATION
 Station
           Application Dial
                              Phone Number
                                                            Configuration
                                                 Trunk
 Extension
                       Prefix
                                                Selection
                                                            Set
 10000
              EC500
                              - 07891900128
```

5. Outside callers may use the QuesCom 400 to reach Avaya Communication Manager extensions by first calling a SIM card number on the QuesCom 400. The QuesCom 400 may be configured to directly route incoming calls from the SIM card to a specific extension on Avaya Communication Manager. If the extension is a Vector Directory Number (VDN), the vector associated with the VDN may then prompt and collect digits from the caller. Alternatively, the QuesCom 400 may be configured to prompt the caller to enter digits. The QuesCom 400 then forwards the call to Avaya Communication Manager with the Called Party Number set to the entered digits.

Step		Description							
6.	Enter the change off-pbx-telephone feature-name-extensions and enter extensions to be								
	associated with Avaya Communication Manager features. For example, Idle Appearance								
	Select can be used by an EC500 cell phone to make a call from an idle call appearance on the								
	*	phone to make a can from an role can appearance on the							
	mapped office extension.								
	change off-pbx-telephone feature-	9							
	EXTENSIONS TO CALL WHICH ACT	CTIVATE FEATURES BY NAME							
	Active Appearance Select: 10	10600 Idle Appearance Select: 10617							
	Automatic Call Back: 10								
	Automatic Call-Back Cancel: 10								
	Call Forward All: 10	10603 Malicious Call Trace Cancel:							
	Call Forward Busy/No Answer: 10	10604 Off-Pbx Call Enable: 10621							
	Call Forward Cancel: 10	10605 Off-Pbx Call Disable: 10622							
	Call Park: 10	10606 Priority Call: 10623							
	Call Park Answer Back: 10	10607 Send All Calls: 10624							
	Call Pick-Up: 10	10608 Send All Calls Cancel: 10625							
	Conference on Answer: 10								
	Calling Number Block: 10								
	Calling Number Unblock: 10								
	Directed Call Pick-Up: 10								
	Drop Last Added Party: 10								
	Exclusion (Toggle On/Off): 10	10614							
	Extended Group Call Pickup:								
	Held Appearance Select: 10	10616							

4. Configure the QuesCom 400 IP/GSM

This section describes the steps for configuring the QuesCom 400 gateway. The steps are provided for illustration only; users should consult with Quescom for specific instructions.

4.1. QuesCom Server Configuration

Step	Description
1.	After the initial installation of the QuesCom server, telnet into the QuesCom server from the management PC shown in figure 1, using the default IP address "192.168.1.1.". Log in using the appropriate username and password.
	C:\> telnet 192.168.1.1 login: administrator Password: *******
	Q400 IP/GSM Series, Serial# Q400-B4-00010381, Version IAD04.20B029P006 Security Patch SP001 Copyright (c) 1998-2005 QuesCom S.A.
	At the prompt, type the following command gwconfig /setup. X:\>gwconfig /setup
	Application has been registered to the QCFGSvc QCFGSvc Version 4.20.000.012 Copyright (c) 1998-2006 QuesCom S.A.
	Enter "1" for English.
	Enter the SmartIAD Administration language [1]: 1 English 2 French 3 German
	> 1 GWconfig language: English
	Enter a name for the QuesCom 400 gateway. Setting up SmartIAD components
	Enter the SmartIAD network name [Q400]:Q400 SmartIAD Network Name: Q400
	Enter IP address for the QuesCom gateway. Enter the SmartIAD IP address [192.168.1.1]: 10.1.10.55 The SmartIAD IP address: 10.1.10.55

Step	Description
	Enter subnet mask or press enter to choose default.
	Enter the SmartIAD subnet mask [255.255.255.0]: The SmartIAD subnet mask: 255.255.25.0
	Enter default Gateway IP address. Enter the SmartIAD default Gateway [192.168.10.1]: 10.1.10.1
	The SmartIAD default Gateway: 10.1.10.1
	<pre>Enter "2" for United Kingdom Enter the SmartIAD country code (ISDN, Tones, Numbering plan, Emails) [1]:</pre>
	IVR language country: ENG - English Country Tones: United Kingdom Country Numbering: United Kingdom Network Operator: EuroISDN Enter "O" for the server to operate in Stand-Alone mode.
	Enter the 'Call Server' mode [0]: 0 Stand-Alone mode 1 Relay mode > 0 Call Server mode: Stand-Alone
	Enter Company Name. This can be any alphanumeric name. Enter Company Name []: Avaya
	Enter "0" to select the H.323 protocol. Select the VoIP Protocol to use[0]: 0 H.323 1 SIP > 0 VoIP Protocol: H.323
	Enter "N" as the QuesCom 400 IP/GSM does not need to register to a GateKeeper in this configuration. Does the QuesCom IP/GSM need to register to a GateKeeper [Y/N]: N

```
Step
                                         Description
      Enter the name for Avaya Communication Manager C-LAN board.
            Enter the name of the H.323 Gateway: CM
            H.323 Gateway name: CM
      Enter the IP address for Avaya Communication Manager C-LAN board.
            Enter the IP Address of the VoIP Gateway: 10.1.10.12
            VoIP Gateway IP Address: 10.1.10.12
      Follow the instruction and press any key to continue.
             Selected parameters for Quick setup mode are:
             SmartIAD Network Name: Q400
             The SmartIAD IP address: 10.1.10.55
            The SmartIAD subnet mask: 255.255.255.0
            The SmartIAD default Gateway: 10.1.10.1
            Press any key to continue..
      Enter "1" to confirm the setup.
             SmartIAD's serial number: Q400-B4-00010381
             IVR language country: ENG - English
            Email language country: ENG - English
            Country Tones: United Kingdom
            Country Numbering: United Kingdom
            Call Server mode: Stand-Alone
            Company Name: Avaya
            VoIP Protocol: H.323
            H.323 Gateway name = CM
            H.323 Gateway IP Address = 10.1.10.12
            Do you confirm this setup [1]:
                   0 No (to exit, and GWconfig /setup command can be re-entered)
                   1 Yes(to continue the setup and restart the QuesCom Q400)
             Setup is confirmed.
      Wait for 3 minutes for the QuesCom 400 gateway to reboot.
             Setting up SmartIAD System Configuration...
             Setting up Gateway Application...
             Please wait...
             Setting up Call Server Application...
             Setting up QuesCom QGsm Application...
             Setting up QuesCom Web Server Application...
             Setting up QuesCom ODBC Socket Server Application...
             Setting up QPortal Application...
            Please wait...
             Setting up NTPClient Application...
             Setting up Pilot Application...
             Setting up GeoPort Application...
            Rebooting system...
             Warning: Do not restart the SmartIAD, update process in progress...
            Please, wait up to 3 minutes.
```

4.2. QuesCom Routing Configuration

1. Open a web browser from the management PC and enter the following URL http://<QuesCom gateway IPaddress:8000>. For this configuration "http://10.1.10.55:8000" was entered. Log in using the appropriate user name and password.



2. On the left hand side of the screen under the QuesCom 400 menu. Click on **Licenses and Profiles**→ VoIP Profile. A default entry is created with Name "NO_RAS_H323" due to the initial configuration in Section 4.1. Click on the pencil(edit) button next to the "NO_RAS_H323" record.



3. The following screen is presented for illustration. Default values may be retained on the VoIP Profile screen below.



4. On the left hand side of the screen under the QuesCom 400 menu, click on **Objects** → **Foreign Gatekeeper**. An entry with the **ID** "CM" and the IP address of Avaya Communication Manager

C-LAN board is created due to the initial configuration in Section 4.1. Click on the pencil (edit) button next to the "CM" record.



Step Description Verify the **VoIP Profile** is set to "NO_RAS_H323". No other changes need to be made to the 5. default values on the Foreign Gatekeeper screen below. Ques Com HOME FILTERS LOGOUT SAVE RESET CANCEL QuesCom 400 Foreign Gatekeeper Settings ▼ Objects SIP registration and authentication • VoIP Device ID CM Foreign Gatekeepe Gatekeeper Type **⊙** H323 **○** SIP • SmartIAD® Name CM Device Group • CTI Application IP Address 10.1.10.12 ServicesCosts Localisation Default International Prefix Licenses and Profiles No Resp. delay (ms) Country Code SIM Management Logs Default SmartIAD® Q400 (SmartIAD) Area Code VoIP Profile NO RAS H323 National Prefix Supported prefixes Prefix | Number Number Add Remove Clear All @ QuesCom All Rights reserved

6. Click on Services → Service. Four entries are present by default. ID "3" is created by default and is routing for calls from Avaya Communication Manager to the QuesCom 400 gateway. ID "4" is routing of outbound calls from the QuesCom 400 gateway to the GSM network. Click on the pencil (edit) button next to ID "3" record.

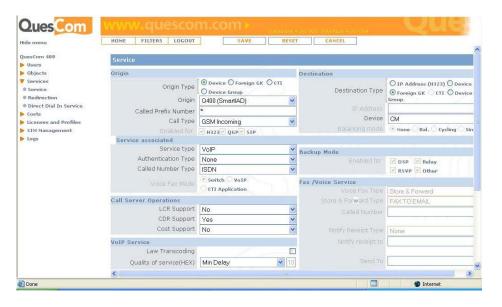


Description Step Verify the Call Type is set to "Foreign Gatekeeper". No other changes need to be made to the 7. default values for the record **ID** 3 on the Service screen below. Ques Com HOME FILTERS LOGOUT SAVE RESET CANCEL QuesCom 400 ▶ Users ▼ Objects Destination Origin O Device O Foreign GK O CTI IP Address (H323) O Device Origin Type Foreign Gatekeeper O Foreign GK OCTI O Device O Device Group ● SmartIAD® Origin CM Device Group Called Prefix Number © CTI Application Device GSM POOL (SmartIAD) Call Type Foreign Gatekeeper ▼ Services Balancing mode O None O Bal. O Cycling Service Enabled for H323 QGP SIP Service associated Direct Dial In Service Service type VolP Backup Mode Costs Authentication Type None Licenses and Profiles Enabled for DSP Relay SIM Management Called Number Type Dialled Number RSVP Other ▶ Logs Switch VolP Fax /Voice Service CTI Application Call Server Operations LCR Support No CDR Support Yes Cost Support No VoIP Service Law Transcoding **v** 10 Quality of service(HEX) Min Delay Internet From the screen shown in Step 6, click the pencil (edit) button next to ID "4". Verify the Call 8. **Type** is set to "VoIP Outgoing". No other changes need to be made to the default values for the record **ID** 4 on the Service screen below.



- **9.** Routing of inbound calls to the QuesCom 400 gateway from the GSM network is created by clicking on the **ADD RECORD** button on the main Service screen shown in Step 6. On the Service screen, configure the following as shown below.
 - **Origin Type** select radio button "Device".
 - **Origin** select "Q400(SmartAD)"
 - Called Prefix Number enter "*"
 - Call Type select "GSM Incoming"
 - **Service type** select "VoIP"
 - **Destination Type** select radio button "Foreign GK"
 - **Device** select "CM"

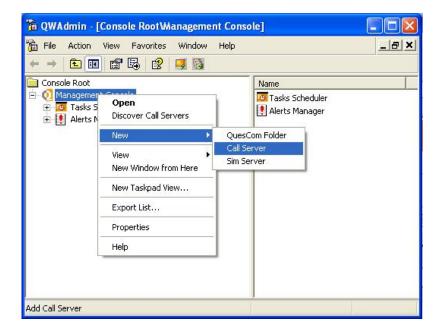
The other parameters can be left with default values. Click on Save.



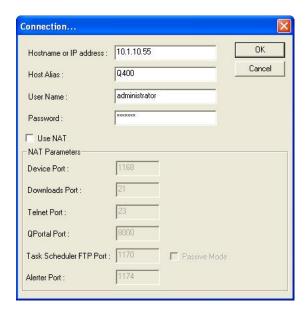
10. The inbound call route pattern added in Step 9 can be displayed on the main Service screen by clicking on Services \rightarrow Service.



11. From the management PC shown in **Figure 1**, launch the QuesCom 400 QWA management console by clicking **Start** → **Programs** → **QuesCom** → **QuesCom** Management Console. Right click on Management Console and click **New** → **Call Server**.

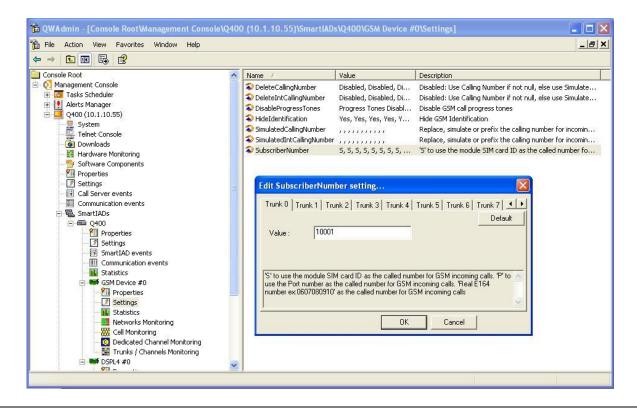


- 12. In the Connection dialog, configure the following, and click **OK**:
 - **Hostname or IP address** enter the IP address of the QuesCom 400 gateway
 - **Host Alias** enter a descriptive name for the QuesCom 400 gateway
 - User Name and Password



13. Expand the Management Console tree by clicking on Q400(10.1.10.55) → SmartIADs → Q400 → GSM Device #0 → Settings → SubscriberNumber. In the Edit SubscriberNumber setting dialog box, click on the Trunk 0 tab and enter the extension that incoming calls will be routed to in the Value field. Replicate this field for all Trunk tabs 1 to 12 and click OK.

Right click on **Q400** under **SmartIADs** and click on **Save** configuration, then right click back on **Q400** and click on **Stop**. Right click Q400 and click on **Start** and wait for the SIM cards to register.



5. Interoperability Compliance Testing

The interoperability compliance testing focused on verifying the routing of inbound/outbound calls to/from the QuesCom 400.

5.1. General Test Approach

The general approach was to place inbound and outbound calls through the QuesCom 400 and verify successful call completion. The main objectives were to verify that:

- When internal extensions place outbound calls to GSM numbers, the calls are routed to the QuesCom 400, and the QuesCom 400 decides on the least cost routing and routes the call to the GSM network.
- When the landline is out of service, all outbound calls can successfully be routed via the QuesCom 400 if need be.
- If the landline is operational, then Avaya Communication Manager will successfully reroute calls rejected by the QuesCom 400 to the landline. This can be configured to occur for various reasons, such as no more free minutes left on the SIM cards.
- Inbound calls from the GSM network to the QuesCom 400 are successfully forwarded to Avaya Communication Manager using both direct routing (mapping of a SIM card phone number to an Avaya Communication Manager extension) and post-dialing (SIM card answers an inbound call and upon a prompt, the external caller enters an Avaya Communication Manager extension).
- Transfers and conferences between Avaya Communication Manager stations complete properly on outbound and inbound calls routed through the QuesCom 400.

5.2. Test Results

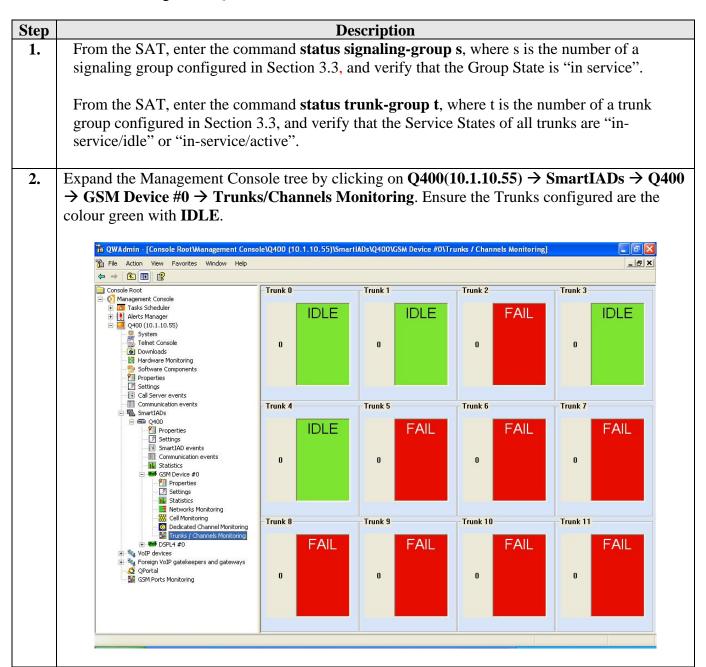
The test objectives of Section 5.1 were verified. For serviceability testing, outbound and inbound calls routed through the QuesCom 400 complete successfully after recovering from failures such as Ethernet cable disconnects, and resets of Avaya Communication Manager and the QuesCom 400.

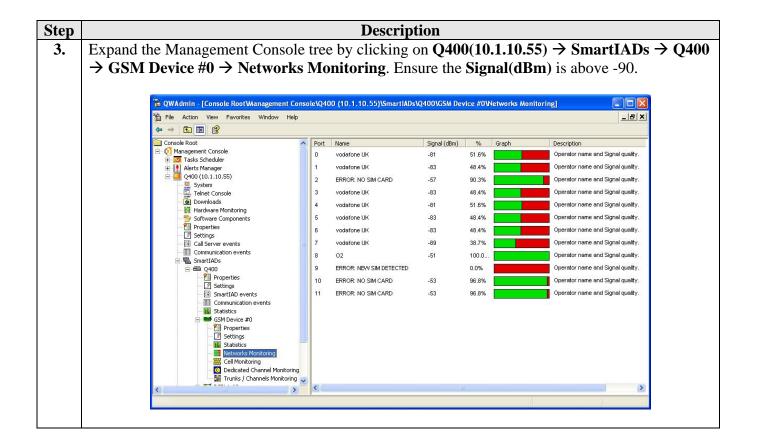
The following are observations obtained from testing Avaya Extension to Cellular (EC500) with QuesCom 400:

- Calls placed to EC500-enabled telephones on Avaya Communication Manager were successfully extended to EC500-mapped external wireless telephones through the QuesCom 400.
- EC500-mapped external wireless telephones successfully placed calls to Avaya Communication Manager telephones through the QuesCom 400, and the displays of the answering telephones showed the name and extensions of the corresponding EC500-enabled telephones as the calling party.
- Avaya Extension to Cellular Feature Name Extensions were verified. For example, EC500-mapped external wireless telephone callers successfully activated the Exclusion, Idle Appearance Select, and Transfer on Hangup EC500 features through the QuesCom 400 by dialing the corresponding EC500 Feature Name Extensions.

6. Verification Steps

This section provides the tests that can be performed to verify proper configuration of Avaya Communication Manager and QuesCom 400.





7. Support

Technical support from QuesCom can be requested in any of the following three ways.

- The corporate QuesCom Reporting Tool (QRT) account on the QuesCom web site at http://support.quescom.com and follow instructions.
- The Support Line number. +33 820203846 (France) Voice Message is available during off days and non-working time.
- Sending an email to support@quescom.com

8. Conclusion

These Application Notes describe the configuration steps required for QuesCom IP/GSM 400 to successfully interoperate with Avaya Communication Manager 3.1 using H.323 trunks. All feature functionality, performance and serviceability test cases were completed successfully.

9. Additional References

This section references the Avaya and QuesCom IP/GSM 400 product documentation that are relevant to these Application Notes.

The following Avaya Documents are available at http://support.avaya.com

- Administrator Guide for Avaya Communication Manager, Document ID 03-300509, Issue 2, Feb 2006.
- Administration for Network Connectivity for Avaya Communication Manager, Document ID 555-233-504, Issue 11, Feb 2006.

The following documents can be obtained from QuesCom.

- Getting Started with QuesCom 400 IP/GSM: GS-Q400IPGSM400-V01.pdf
- QuesCom 400 IP/GSM Administrator Guide: AG-Q400IPGSM400-V01.pdf
- How to configure an IP-GSM linked with an external H.323 gateway: Configuration of a H323 IP-GSM.pdf
- How to configure GSM Incoming calls to a remote Gatekeeper: Configuring GSM incoming calls.pdf

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