



Avaya Solution & Interoperability Test Lab

Application Notes for Lyrinx PeopleFind with Avaya Communication Manager using E1 and T1 ISDN-PRI QSIG – Issue 1.0

Abstract

These Application Notes describe the procedures for configuring E1 and T1 ISDN-PRI QSIG integration between Lyrinx PeopleFind and Avaya Communication Manager. Lyrinx PeopleFind is a speech-enabled communications appliance that allows people, customers, teams, and management to find each other without knowledge of their locations or numbers.

During compliance testing, Lyrinx PeopleFind successfully transferred calls to the appropriate Avaya Communication Manager extension, and Lyrinx Voice Messaging successfully provided typical voice messaging functionality, including Message Waiting Indicator.

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

1. Introduction

These Application Notes describe a compliance-tested solution comprised of Avaya Communication Manager Release 3.1.2, Lyrix PeopleFind 4.0.104-94. Lyrix PeopleFind consists of two applications, Lyrix Auto Attendant (Lyrix PeopleFind) and Lyrix Voice Messaging. Lyrix PeopleFind is a speech-enabled communications appliance that allows people, customers, teams, and management to find each other without knowledge of their locations or numbers. The user provisions the number they can be reached at via the Administration menu. A caller speaks the user's name, and Lyrix PeopleFind locates and connects the user to the caller.

Figure 1 illustrates the network configuration used to verify the Lyrix solution. The configuration focuses on the interface between Avaya Communication Manager and the Lyrix PeopleFind server. Site A is comprised of a pair of Avaya S8700 Media Servers, an Avaya G650 Media Gateway, the Lyrix PeopleFind Server, Avaya 4600 Series IP Telephones, an Avaya 9630 IP Telephone, an Avaya 6402 Digital Telephone, and an ISDN-PRI trunk to the PSTN. Site B was added to test a trunk between Avaya Communication Managers. Site B is comprised of an Avaya S8300 Media Server with an Avaya G700 Media Gateway, Avaya 4600 Series IP Telephones and an Avaya Digital Telephone. The solution described herein is also extensible to other Avaya Media Servers and Media Gateways. The Lyrix PeopleFind server is connected to the Avaya G650 Media Gateway by an E1 or T1 ISDN-PRI QSIG trunk. An IP trunk connects the two Avaya Communication Manager systems between Site A and Site B.

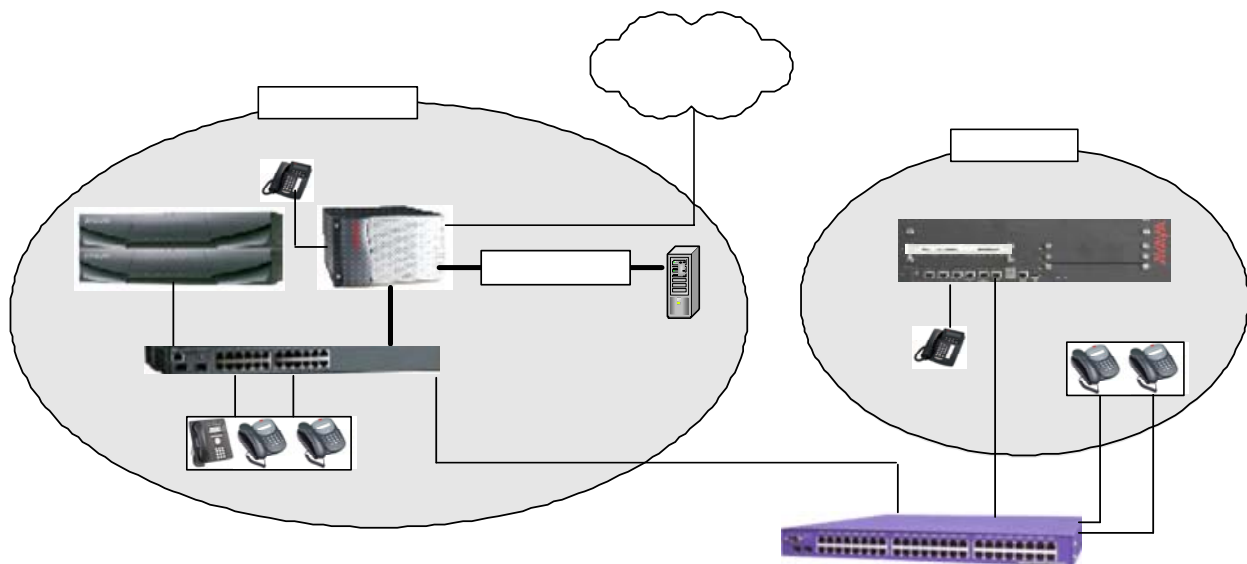


Figure 1: Sample configuration.

2. Equipment and Software Validated

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

Equipment		Software/Firmware
Avaya S8700 Media Servers		Avaya Communication Manager 3.1.2 (R013x.01.2.632.1)
Avaya G650 Media Gateway		-
	TN2312BP IP Server Interface	HW11 FW030
	TN799DP C-LAN Interface	HW1 FW 17
	TN2302AP IP Media Processor	HW20 FW108
	TN464F DS1	000018
Avaya S8300 Media Server with Avaya G700 Media Gateway		Avaya Communication Manager 3.1.2 (R013x.01.2.632.1)
Avaya 4600 Series IP Telephones		
	4620SW (H.323)	2.6
	4625SW (H.323)	2.5
Avaya 9600 Series IP Telephone (H.323)		1.1
Avaya 6400 Series Digital Telephones		-
Lyrix PeopleFind		4.0.104-94

3. Configure Avaya Communication Manager

This section describes the steps for configuring E1 ISDN-PRI QSIG and T1 ISDN-PRI QSIG integration, call coverage, and call routing on Avaya Communication Manager. The steps are performed from the Avaya Communication Manager System Access Terminal (SAT) interface.

3.1. E1 Configuration

Before configuring Avaya Communication Manager, the DS1 board must be physically configured for an appropriate mode (T1 or E1). The DS1 board has 24 channels in T1 mode or 32 channels in E1 mode. The default is set to T1 mode. To modify the DS1 board to use it in E1 mode, the dipswitch on the DS1 board must be switched to the 32 channels side.

3.1.1. System Parameters

This section reviews the features that are required for the solution described in these Application Notes. Contact an authorized Avaya account representative to obtain the licenses for required features that are not enabled in the “system-parameters customer-options” form discussed below.

Step	Description
1.	<p>Enter the display system-parameters customer-options command. On Page 3 of the system-parameters features form, verify that the Audible Message Waiting field is set to y. This license allows Avaya Communication Manager telephones to receive stutter dial tone when a message is waiting. With the Audible Message Waiting license enabled, Audible Message Waiting can also be enabled and disabled on a per-telephone basis.</p> <pre>display system-parameters customer-options Page 3 of 11 OPTIONAL FEATURES Abbreviated Dialing Enhanced List? y Audible Message Waiting? y Access Security Gateway (ASG)? n Authorization Codes? n Analog Trunk Incoming Call ID? n Backup Cluster Automatic Takeover? n A/D Grp/Sys List Dialing Start at 01? n CAS Branch? n Answer Supervision by Call Classifier? n CAS Main? n ARS? y Change COR by FAC? n ARS/AAR Partitioning? n Computer Telephony Adjunct Links? n ARS/AAR Dialing without FAC? y Cvg Of Calls Redirected Off-net? n ASAI Link Core Capabilities? y DCS (Basic)? n ASAI Link Plus Capabilities? n DCS Call Coverage? n Async. Transfer Mode (ATM) PNC? n DCS with Rerouting? n Async. Transfer Mode (ATM) Trunking? n ATM WAN Spare Processor? n Digital Loss Plan Modification? n ATMS? n DS1 MSP? n Attendant Vectoring? n DS1 Echo Cancellation? n</pre>

Step	Description
2.	<p>On Page 4 of the system-parameters features form, verify that the ISDN-PRI field is set to y.</p> <pre> display system-parameters customer-options Page 4 of 11 OPTIONAL FEATURES Emergency Access to Attendant? y IP Stations? y Enable 'dadmin' Login? y Internet Protocol (IP) PNC? n Enhanced Conferencing? y ISDN Feature Plus? n Enhanced EC500? y ISDN Network Call Redirection? n Enterprise Survivable Server? n ISDN-BRI Trunks? n Enterprise Wide Licensing? n ISDN-PRI? y ESS Administration? n Local Survivable Processor? n Extended Cvg/Fwd Admin? n Malicious Call Trace? n External Device Alarm Admin? n Media Encryption Over IP? y Five Port Networks Max Per MCC? n Mode Code for Centralized Voice Mail? n Flexible Billing? n Forced Entry of Account Codes? n Multifrequency Signaling? y Global Call Classification? n Multimedia Appl. Server Interface (MASI)? n Hospitality (Basic)? y Multimedia Call Handling (Basic)? n Hospitality (G3V3 Enhancements)? n Multimedia Call Handling (Enhanced)? n IP Trunks? y IP Attendant Consoles? y </pre>
3.	<p>On Page 5 of the system-parameters features form, verify that the Private Networking field is set to y.</p> <pre> display system-parameters customer-options Page 5 of 11 OPTIONAL FEATURES Multinational Locations? n Station and Trunk MSP? n Multiple Level Precedence & Preemption? n Station as Virtual Extension? n Multiple Locations? n System Management Data Transfer? n Personal Station Access (PSA)? n Tenant Partitioning? n Posted Messages? n Terminal Trans. Init. (TTI)? n PNC Duplication? n Time of Day Routing? n Port Network Support? y Uniform Dialing Plan? y Processor and System MSP? n Usage Allocation Enhancements? y Private Networking? y TN2501 VAL Maximum Capacity? y Processor Ethernet? n Wideband Switching? n Wireless? n Remote Office? n Restrict Call Forward Off Net? y Secondary Data Module? y </pre>

Step	Description
4.	<p>On Page 8 of the system-parameters customer-options form, verify that the highlighted fields below are set to y.</p> <pre> display system-parameters customer-options QSIG OPTIONAL FEATURES Basic Call Setup? y Basic Supplementary Services? y Centralized Attendant? n Interworking with DCS? n Supplementary Services with Rerouting? y Transfer into QSIG Voice Mail? y Value-Added (VALU)? y </pre>
5.	<p>Enter the change system-parameters features command. On Page 8 of the system-parameters features form, configure the following:</p> <ul style="list-style-type: none"> • QSIG TSC Extension – enter any unused extension that is valid under the provisioned dial plan. • MWI - Number of Digits Per Voice Mail Subscriber – enter the number of digits used for station extensions. • QSIG Path Replacement Extension – enter any unused extension that is valid under the provisioned dial plan. <pre> change system-parameters features FEATURE-RELATED SYSTEM PARAMETERS ISDN PARAMETERS Send Non-ISDN Trunk Group Name as Connected Name? n Display Connected Name/Number for ISDN DCS Calls? n Send ISDN Trunk Group Name on Tandem Calls? n QSIG TSC Extension: 22228 MWI - Number of Digits Per Voice Mail Subscriber: 5 National CPN Prefix: International CPN Prefix: Pass Prefixed CPN to ASAI? n Unknown Numbers Considered Internal for AUDIX? n USNI Calling Name for Outgoing Calls? n Path Replacement with Measurements? n QSIG Path Replacement Extension: 22444 Path Replace While in Queue/Vectoring? n </pre>

Step	Description
6.	<p>Enter the change system-parameters coverage-forwarding command. Set the Maintain SBA at Principal field to n to ensure that when a call redirects to coverage (i.e., to voicemail), the appearance on the covered station is removed. Removal of the appearance prevents a person at the station from bridging onto the covered call (i.e., prevents a person from listening to the call as a voice message is being left).</p> <pre> change system-parameters coverage-forwarding SYSTEM PARAMETERS CALL COVERAGE / CALL FORWARDING CALL COVERAGE/FORWARDING PARAMETERS Local Cvg Subsequent Redirection/CFWD No Ans Interval (rings): 2 Off-Net Cvg Subsequent Redirection/CFWD No Ans Interval (rings): 2 Coverage - Caller Response Interval (seconds): 4 Threshold for Blocking Off-Net Redirection of Incoming Trunk Calls: 1 COVERAGE Keep Held SBA at Coverage Point? y External Coverage Treatment for Transferred Incoming Trunk Calls? n Immediate Redirection on Receipt of PROGRESS Inband Information? n Maintain SBA At Principal? n QSIG VALU Coverage Overrides QSIG Diversion with Rerouting? n Station Hunt Before Coverage? n FORWARDING Call Forward Override? y Coverage After Forwarding? y </pre>

3.1.2. Dial Plan

Enter the **display dialplan analysis** command to view the provisioned dial plan. Note the following dialed strings are configured in the dial plan below for the test configuration:

- 3-digit dial access codes (indicated with a **Call Type** of **dac**) beginning with the digit 1 – Trunk Access Codes defined for trunk groups must conform to this format.
- 5-digit extensions (indicated with a **Call Type** of **ext**) beginning with the digit 2 or 4 – station, hunt group, QSIG extensions must conform to this format.

display dialplan analysis						Page 1 of 12			
DIAL PLAN ANALYSIS TABLE									
						Percent Full: 2			
Dialed String	Total Length	Call Type	Dialed String	Total Length	Call Type	Dialed String	Total Length	Call Type	
1	3	dac							
2	5	ext							
4	5	ext							

3.1.3. QSIG Trunk

This section describes the steps for configuring Avaya Communication Manager side of the E1 ISDN-PRI QSIG trunk.

Step	Description
1.	<p>Enter the list configuration all command and note the Board Number for the DS1 circuit pack to be configured.</p> <pre>list configuration all</pre> <p style="text-align: right;">Page 3</p> <pre> SYSTEM CONFIGURATION Board Number Board Type Code Vintage Assigned Ports u=unassigned t=tti p=psa 01A10 DS1 INTERFACE TN464F 000018 u u u u u u u u u u u u u u u u u u u u u u u u u u u u u u u u </pre>
2.	<p>Enter the add ds1 x command, where x is the board number of the DS1 circuit pack noted in Step 1. Enter a descriptive Name and set the other highlighted fields below to the values indicated.</p> <pre>add ds1 1a10</pre> <p style="text-align: right;">Page 1 of 1</p> <pre> DS1 CIRCUIT PACK Location: 01A10 Bit Rate: 2.048 Name: PRI QSIG Line Coding: hdb3 Signaling Mode: isdn-pri Connect: pbx Interface: peer-master TN-C7 Long Timers? n Peer Protocol: Q-SIG Interworking Message: PROgress Side: a Interface Companding: alaw CRC? y Idle Code: 11111111 Channel Numbering: timeslot DCP/Analog Bearer Capability: 3.1kHz T303 Timer(sec): 4 Slip Detection? n Near-end CSU Type: other </pre>

Step	Description
3.	<p>Enter the add signaling-group s command, where s is an unused signaling group number. Set the highlighted fields below to the values indicated. Note that the Primary D-Channel field is channel 16 on the DS1 circuit pack for an E1.</p> <pre> add signaling-group 61 Page 1 of 1 SIGNALING GROUP Group Number: 61 Group Type: isdn-pri Associated Signaling? y Primary D-Channel: 01A1016 Max number of NCA TSC: 10 Max number of CA TSC: 10 Trunk Group for NCA TSC: Trunk Group for Channel Selection: Supplementary Service Protocol: b </pre>
4.	<p>Enter the add trunk-group t command, where t is an unused trunk group number. On Page 1 of the trunk-group form, enter a descriptive Group Name and enter a TAC that is valid under the provisioned dial plan in Section 3.1.2. Set the other highlighted fields below to the values indicated.</p> <pre> add trunk-group 61 Page 1 of 21 TRUNK GROUP Group Number: 61 Group Type: isdn CDR Reports: y Group Name: QSIG-E1 COR: 1 TN: 1 TAC: 112 Direction: two-way Outgoing Display? n Carrier Medium: PRI/BRI Dial Access? n Busy Threshold: 255 Night Service: Queue Length: 0 Service Type: tie Auth Code? n TestCall ITC: rest Far End Test Line No: TestCall BCC: 4 </pre>
5.	<p>On Page 2 of the trunk-group form, set the Supplementary Service Protocol field to b to indicate that QSIG supplementary services will be provided on this trunk group.</p> <pre> add trunk-group 61 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 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 </pre>

Step	Description
6.	<p>On Page 3 of the trunk-group form, set the highlighted fields below to the values indicated.</p> <pre> add trunk-group 61 TRUNK FEATURES ACA Assignment? n Measured: none Wideband Support? n Maintenance Tests? y Data Restriction? n NCA-TSC Trunk Member: Send Name: y Send Calling Number: y Used for DCS? n Send EMU Visitor CPN? n Suppress # Outpulsing? n Format: unk-pvt Outgoing Channel ID Encoding: preferred UII 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 UII IE? y Send UCID? n Send Codeset 6/7 LAI IE? y Dsl Echo Cancellation? n Apply Local Ringback? n Network (Japan) Needs Connect Before Disconnect? n </pre>
7.	<p>On Page 4 of the trunk-group form, set the highlighted fields below to the values indicated.</p> <pre> add trunk-group 61 QSIG TRUNK GROUP OPTIONS Diversion by Reroute? y Path Replacement? y Path Replacement with Retention? n Path Replacement Method: always SBS? n Display Forwarding Party Name? y Character Set for QSIG Name: eurofont </pre>

Step	Description
8.	<p>On Page 5 of the trunk-group form, add trunk members by entering:</p> <ul style="list-style-type: none"> • xxxxxxzz for Port, where xxxxxx is the board number of the DS1 circuit pack configured in Step 2, and zz is a channel in the E1 ISDN-PRI. • the number of the signaling group configured in Step 3 for the Sig Grp field. <p>For the compliance test, channels 1 – 15 and 17 – 31 (not shown) of the E1 ISDN-PRI were added. Channel 16, the signaling channel configured in Step 3, was excluded.</p> <pre> add trunk-group 61 TRUNK GROUP Administered Members (min/max): 0/0 GROUP MEMBER ASSIGNMENTS Total Administered Members: 0 Port Code Sfx Name Night Sig Grp 1: 01A1001 TN464 F 2: 01A1002 TN464 F 3: 01A1003 TN464 F 4: 01A1004 TN464 F 5: 01A1005 TN464 F 6: 01A1006 TN464 F 7: 01A1007 TN464 F 8: 01A1008 TN464 F 9: 01A1009 TN464 F 10: 01A1010 TN464 F 11: 01A1011 TN464 F 12: 01A1012 TN464 F 13: 01A1013 TN464 F 14: 01A1014 TN464 F 15: 01A1015 TN464 F </pre>
9.	<p>Return to Page 3 of the trunk-group form, and set the NCA-TSC Trunk Member field to a trunk member added in Step 8.</p> <pre> add trunk-group 61 TRUNK FEATURES ACA Assignment? n Measured: none Wideband Support? n Maintenance Tests? y Data Restriction? n NCA-TSC Trunk Member: 31 Send Name: y Send Calling Number: y Send EMU Visitor CPN? n Used for DCS? n Suppress # Outpulsing? n Format: unk-pvt Outgoing Channel ID Encoding: preferred UII 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 Dsl Echo Cancellation? n Apply Local Ringback? n Network (Japan) Needs Connect Before Disconnect? n </pre>

Step	Description
10.	Enter the change signaling-group s command, where s is the number of the signaling group configured in Step 3. Set the Trunk Group for NCA TSC and Trunk Group for Channel Selection field values to the trunk group configured in Steps 4 – 9.
	<div>change signaling-group 61<div>Page1 of 1</div></div> <div>SIGNALING GROUP</div> <div>Group Number: 61Group Type: isdn-pri</div> <div>Associated Signaling? yMax number of NCA TSC: 10</div> <div>Primary D-Channel: 01A1016Max number of CA TSC: 10</div> <div>Trunk Group for NCA TSC: 61</div> <div>Trunk Group for Channel Selection: 61</div> <div>Supplementary Service Protocol: b</div>
11.	Enter the change private numbering command. Ensure that the Network Level field is set to 0 and the Level 2 Code and Level 1 Code field values are blank.
	<div>change private-numbering<div>Page1 of 1</div></div> <div>NUMBERING - PRIVATE FORMAT</div> <div>Network Level: 0PBX Identifier:</div> <div>Level 2 Code:Deleted Digits: 0</div> <div>Level 1 Code:</div>
12.	Enter the change public-unknown numbering l command, where l is an extension length defined in the dial plan (see Section 3.1.2). This table defines the Calling Party Number (CPN) on outbound calls sent to specific ISDN-PRI trunk groups, such as the trunk group connected to Lyrix Enterprise Voice Messaging. Add an entry as follows:
	<div><div><div>Ext Len</div><div>Ext Code</div><div>Trk Grp(s)</div><div>CPN Prefix</div><div>Total CPN Len</div></div><div><div>Ext Len</div><div>Ext Code</div><div>Trk Grp(s)</div><div>CPN Prefix</div><div>Total CPN Len</div></div></div> <div><div>54615</div></div> <div><div>change public-unknown-numbering 5<div>Page1 of 2</div></div></div> <div>NUMBERING - PUBLIC/UNKNOWN FORMAT</div>

3.1.4. Routing to Lyrix PeopleFind

This section describes the configuration steps for routing calls to the E1 ISDN-PRI QSIG trunk connected to Lyrix PeopleFind and Lyrix Voice Messaging.

Step	Description
1.	<p>Enter the add hunt-group h command, where h is an unused hunt group number. On Page 1 of the hunt-group form, enter a descriptive Group Name and assign a Group Extension that is valid under the provisioned dial plan.</p> <pre> add hunt-group 2 Page 1 of 60 HUNT GROUP Group Number: 2 ACD? n Group Name: Voicemail Queue? n Group Extension: 22999 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: grp-name </pre>
2.	<p>On Page 2 of the hunt-group form, configure the Message Center field to qsig-mwi, the Send Reroute Request field to y and the Voice Mail Number field to a number to be used for routing calls to Lyrix Voice Messaging. The Lyrix Voice Messaging extension for the compliance test was configured as x44445.</p> <p>Calls placed to the Group Extension of this hunt group (i.e., for retrieval of voice messages or management of voice mailboxes) will be routed via AAR. Calls placed to Avaya Communication Manager stations and covered to this hunt group will also be routed via AAR. AAR will use the Voice Mail Number to select a route pattern containing the QSIG trunk group to Lyrix Enterprise Voice Messaging.</p> <pre> add hunt-group 2 Page 2 of 60 HUNT GROUP LWC Reception: none AUDIX Name: Message Center: qsig-mwi Send Reroute Request: y Voice Mail Number: 44445 Routing Digits (e.g. AAR/ARS Access Code): Provide Ringback? n TSC per MWI Interrogation? n </pre>

Step	Description
3.	<p>Enter the change aar analysis d command, where d is any digit that is valid under the provisioned dial plan. Enter the whole or a partial Voice Mail Number configured in Step 2 for the Dialed String field. Enter the number of an unused route pattern for the Route Pattern field. The route pattern will be defined in Step 5. The Call Type field is set to aar.</p> <pre> change aar analysis 444 Page 1 of 2 AAR DIGIT ANALYSIS TABLE Percent Full: 2 Dialed Total Route Call Node ANI String Min Max Pattern Type Num Req'd 444 5 5 61 aar n </pre>
4.	<p>Enter the change uniform-dialplan d command, where d is any digit that is valid under the provisioned dial plan. Enter the whole or a partial Voice Mail Number configured in Step 2 for the Dialed String field. Enter the Voice Mail Number's first digit (or first few digits) for the Matching Pattern field. Enter the length of the Voice Mail Number extension for the Len field. The Del field set to 0, and the Net field is set to aar.</p> <pre> change uniform-dialplan 4 Page 1 of 2 UNIFORM DIAL PLAN TABLE Percent Full: 0 Matching Len Del Insert Node Pattern Len Del Digits Net Conv Num 4 5 0 aar aar n n n n </pre>

Step	Description
5.	<p>Enter the change route-pattern r command, where r is the number of the route pattern specified in Step 3. Enter the number of the trunk group configured in Section 3.1.3 Steps 4 – 9 for the Grp No field. Assign a Facility Restriction Level to this routing preference for the FRL field. The FRL value 0 is the least restrictive.</p> <p>During compliance testing, the extension, 44444, was configured on the Lyrix PeopleFind, and the extension, 44445, was configured on Lyrix Voice Messaging as the voicemail access number.</p> <pre> change route-pattern 61 Pattern Number: 61 Pattern Name: Lyrix SCCAN? n Secure SIP? n Grp FRL NPA Pfx Hop Toll No. Inserted DCS/ IXC No Mrk Lmt List Del Digits QSIG Intw 1: 61 0 2: 3: 4: 5: 6: n user n user n user n user n user n user BCC VALUE TSC CA-TSC ITC BCIE Service/Feature PARM No. Numbering LAR 0 1 2 3 4 W Request Dgts Format Subaddress 1: y y y y y n n rest none 2: y y y y y n n rest none 3: y y y y y n n rest none 4: y y y y y n n rest none 5: y y y y y n n rest none 6: y y y y y n n rest none </pre>
6.	<p>To allow external/PSTN callers to access Lyrix Voice Messaging (i.e., to retrieve voice messages) ensure that the proper digit treatment is applied to incoming trunk calls. For the compliance test, the incoming called number needs be manipulated to match the hunt group extension. This can be accomplished by using the change inc-call-handling-trmt trunk-group x, where x is the incoming calls trunk group number.</p>

3.1.5. Coverage Path

This section describes the steps for configuring a coverage path and assigning the coverage path to Avaya Communication Manager stations.

Step	Description
1.	<p>Enter the add coverage path c command, where c is the number of an unused coverage path, and set the Point1 field to the hunt group configured in Section 3.1.4 Steps 1 – 2. The value for the Point1 field is set to h2, which means utilizing Hunt Group 2.</p> <pre> add coverage path 99 Page 1 of 1 COVERAGE PATH Coverage Path Number: 99 Next Path Number: Hunt after Coverage? n Linkage COVERAGE CRITERIA Station/Group Status Inside Call Outside Call Active? n n Busy? Y Y Don't Answer? Y Y Number of Rings: 3 All? n n DND/SAC/Goto Cover? Y Y Holiday Coverage? n n COVERAGE POINTS Terminate to Coverage Pts. with Bridged Appearances? n Point1: h2 Rng: Point2: Point3: Point4: Point5: Point6: </pre>

Step	Description
2.	<p>Enter the change station e command, where e is the extension of a station that is a Lyrrix Voice Messaging subscriber. On Page 1 of the station form, set Coverage Path 1 to the number of the coverage path configured in the previous step.</p> <pre> change station 22001 Page 1 of 4 STATION Extension: 22001 Lock Messages? n BCC: 0 Type: 4620 Security Code: ***** TN: 1 Port: S00003 Coverage Path 1: 99 COR: 1 Name: STA-22001 Coverage Path 2: COS: 1 Hunt-to Station: STATION OPTIONS Loss Group: 19 Personalized Ringing Pattern: 1 Message Lamp Ext: 50001 Speakerphone: 2-way Mute Button Enabled? y Display Language: english Survivable GK Node Name: Survivable COR: internal Media Complex Ext: Survivable Trunk Dest? y IP SoftPhone? y IP Video Softphone? y Customizable Labels? y </pre>
3.	<p>On Page 2 of the station form, set the MWI Served User Type field to qsig-mwi. If the station does not have a MWI, for example if the station is an analog telephone, then it may be desirable to set the Audible Message Waiting field to y.</p> <pre> change station 22001 Page 2 of 4 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? y Active Station Ringing: single Conf/Trans on Primary Appearance? n EMU Login Allowed? n H.320 Conversion? n Per Station CPN - Send Calling Number? Service Link Mode: as-needed Multimedia Mode: enhanced Audible Message Waiting? n MWI Served User Type: qsig-mwi Display Client Redirection? n Select Last Used Appearance? n Coverage After Forwarding? s Remote Softphone Emergency Calls: as-on-local Direct IP-IP Audio Connections? y Emergency Location Ext: 50001 Always Use? n IP Audio Hairpinning? y </pre>
4.	<p>Enter the save translation command to save the changes made on Avaya Communication Manager.</p>

3.2. T1 Configuration

Before configuring Avaya Communication Manager, the DS1 board must be physically configured for an appropriate mode (T1 or E1). The DS1 board has 24 channels in T1 mode or 32 channels in E1 mode. The default is set to T1 mode. To modify the DS1 board to use it in T1 mode, the dipswitch on the DS1 board must be switched to the 24 channels side. Steps for the T1 configuration are mostly the same as for the E1 configuration, previously discussed in Section 3.1. This section only describes the steps unique to the T1 configuration.

3.2.1. System Parameters

Refer to Section 3.1.1.

3.2.2. Dial Plan

Refer to Section 3.1.2.

3.2.3. QSIG Trunk

This section describes the steps for configuring the Avaya Communication Manager side of the T1 ISDN-PRI QSIG trunk.

Step	Description																																																												
1.	Enter the list configuration all command and note the Board Number of the DS1 circuit pack to be configured.																																																												
	list configuration all Page 3																																																												
	SYSTEM CONFIGURATION																																																												
	<table><tr><th>Board Number</th><th>Board Type</th><th>Code</th><th>Vintage</th><th colspan="8">Assigned Ports u=unassigned t=tti p=psa</th></tr><tr><td>01A10</td><td>DS1 INTERFACE</td><td>TN464F</td><td>000018</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td></tr><tr><td></td><td></td><td></td><td></td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td></tr><tr><td></td><td></td><td></td><td></td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td></tr><tr><td></td><td></td><td></td><td></td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td><td>u</td></tr></table>	Board Number	Board Type	Code	Vintage	Assigned Ports u=unassigned t=tti p=psa								01A10	DS1 INTERFACE	TN464F	000018	u	u	u	u	u	u	u	u					u	u	u	u	u	u	u	u					u	u	u	u	u	u	u	u					u	u	u	u	u	u	u	u
	Board Number	Board Type	Code	Vintage	Assigned Ports u=unassigned t=tti p=psa																																																								
01A10	DS1 INTERFACE	TN464F	000018	u	u	u	u	u	u	u	u																																																		
				u	u	u	u	u	u	u	u																																																		
				u	u	u	u	u	u	u	u																																																		
				u	u	u	u	u	u	u	u																																																		

Step	Description
2.	<p>Enter the add ds1 xxxxx command, where xxxxx is the board number of the DS1 circuit pack noted in Step 1. Enter a descriptive Name and set the other highlighted fields below to the values indicated.</p> <pre> display ds1 1a10 DS1 CIRCUIT PACK Page 1 of 2 Location: 01A10 Name: PRI QSIG 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 Interface Companding: alaw CRC? n Idle Code: 11111111 DCP/Analog Bearer Capability: 3.1kHz T303 Timer(sec): 4 Slip Detection? n Near-end CSU Type: other </pre>
3.	<p>Enter the add signaling-group s command, where s is an unused signaling group number. Set the highlighted fields below to the values indicated. Note that the Primary D-Channel field is set to channel 24 of the DS1 circuit pack for the T1 signaling-group.</p> <pre> add signaling-group 61 SIGNALING GROUP Page 1 of 1 Group Number: 61 Group Type: isdn-pri Associated Signaling? y Max number of NCA TSC: 10 Primary D-Channel: 01A1024 Max number of CA TSC: 10 Trunk Group for NCA TSC: Trunk Group for Channel Selection: Supplementary Service Protocol: b </pre>
4.	<p>Enter the add trunk-group t command, where t is an unused trunk group number. On Page 1 of the trunk-group form, enter a descriptive Group Name and enter a TAC that is valid under the provisioned dial plan. Set the other highlighted fields below to the values indicated.</p> <pre> add trunk-group 61 TRUNK GROUP Page 1 of 21 Group Number: 61 Group Type: isdn CDR Reports: y Group Name: QSIG-T1 COR: 1 TN: 1 TAC: 112 Direction: two-way Outgoing Display? n Carrier Medium: PRI/BRI Dial Access? n Busy Threshold: 255 Night Service: Queue Length: 0 Service Type: tie Auth Code? n TestCall ITC: rest Far End Test Line No: TestCall BCC: 4 </pre>

Step	Description
5.	On Page 2 of the trunk-group form, set the Supplementary Service Protocol field to b to indicate that QSIG supplementary services will be provided on this trunk group.
	<div>add trunk-group 61<div>Group Type: isdn</div></div> <div>Page2 of 21</div> <div>TRUNK PARAMETERS<div><div>Codeset to Send Display: 6</div><div>Codeset to Send National IEs: 6</div><div>Max Message Size to Send: 260</div><div>Charge Advice: none</div><div>Supplementary Service Protocol: b</div><div>Digit Handling (in/out): enbloc/enbloc</div><div>Trunk Hunt: cyclical</div><div>QSIG Value-Added? n</div><div>Digital Loss Group: 13</div><div>Incoming Calling Number - Delete: Insert: Format:</div><div>Bit Rate: 1200</div><div>Synchronization: async</div><div>Duplex: full</div><div>Disconnect Supervision - In? y Out? n</div><div>Answer Supervision Timeout: 0</div></div></div>
6.	On Page 3 of the trunk-group form, set the highlighted fields below to the values indicated.
	<div>add trunk-group 61<div>TRUNK FEATURES</div></div> <div>Page3 of 21</div> <div><div>ACA Assignment? n</div><div>Measured: none</div><div>Wideband Support? n</div><div>Maintenance Tests? y</div><div>Data Restriction? n</div><div>NCA-TSC Trunk Member:</div><div>Send Name: y</div><div>Send Calling Number: y</div><div>Send EMU Visitor CPN? n</div><div>Used for DCS? n</div><div>Suppress # Outpulsing? n</div><div>Format: unk-pvt</div><div>Outgoing Channel ID Encoding: preferred</div><div>UII IE Treatment: service-provider</div><div>Replace Restricted Numbers? n</div><div>Replace Unavailable Numbers? n</div><div>Send Connected Number: y</div><div>Hold/Unhold Notifications? y</div><div>Modify Tandem Calling Number? n</div><div>Send UII IE? y</div><div>Send UCID? n</div><div>Send Codeset 6/7 LAI IE? y</div><div>Dsl Echo Cancellation? n</div><div>Apply Local Ringback? n</div><div>Network (Japan) Needs Connect Before Disconnect? n</div></div>

Step	Description
7.	<p>On Page 4 of the trunk-group form, set the highlighted fields below to the values indicated.</p> <pre> add trunk-group 61 Page 4 of 21 QSIG TRUNK GROUP OPTIONS Diversion by Reroute? y Path Replacement? y Path Replacement with Retention? n Path Replacement Method: always SBS? n Display Forwarding Party Name? y Character Set for QSIG Name: eurofont </pre>
8.	<p>On Page 5 of the trunk-group form, add trunk members by entering:</p> <ul style="list-style-type: none"> • xxxxxxzz for Port, where xxxxxx is the board number of the DS1 circuit pack configured in Step 2, and zz is a channel in the T1 ISDN-PRI. • the number of the signaling group configured in Step 3 for the Sig Grp field. <p>For the compliance test, channels 1 – 23 of the T1 ISDN-PRI were added (channel 24, the signaling channel configured in Step 3, was excluded). The following screen only shows a partial trunk group member (1 through 15).</p> <pre> add trunk-group 61 Page 5 of 21 TRUNK GROUP Administered Members (min/max): 0/0 GROUP MEMBER ASSIGNMENTS Total Administered Members: 0 Port Code Sfx Name Night Sig Grp 1: 01A1001 TN464 F 61 2: 01A1002 TN464 F 61 3: 01A1003 TN464 F 61 4: 01A1004 TN464 F 61 5: 01A1005 TN464 F 61 6: 01A1006 TN464 F 61 7: 01A1007 TN464 F 61 8: 01A1008 TN464 F 61 9: 01A1009 TN464 F 61 10: 01A1010 TN464 F 61 11: 01A1011 TN464 F 61 12: 01A1012 TN464 F 61 13: 01A1013 TN464 F 61 14: 01A1014 TN464 F 61 15: 01A1015 TN464 F 61 </pre>

Step	Description
9.	<p>Return to Page 3 of the trunk-group form, and set the NCA-TSC Trunk Member field to a trunk member added in Step 8.</p> <pre> add trunk-group 61 Page 3 of 21 TRUNK FEATURES ACA Assignment? n Measured: none Wideband Support? n Maintenance Tests? y Data Restriction? n NCA-TSC Trunk Member: 23 Send Name: y Send Calling Number: y Send EMU Visitor CPN? n Used for DCS? n Suppress # Outpulsing? n Format: unk-pvt Outgoing Channel ID Encoding: preferred UII 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 UII IE? y Send UCID? n Send Codeset 6/7 LAI IE? y Dsl Echo Cancellation? n Apply Local Ringback? n Network (Japan) Needs Connect Before Disconnect? n </pre>
10.	<p>Enter the change signaling-group s command, where s is the number of the signaling group configured in Step 3. Set the Trunk Group for NCA TSC and Trunk Group for Channel Selection field values to the trunk group configured in Steps 4 – 9.</p> <pre> change signaling-group 61 Page 1 of 1 SIGNALING GROUP Group Number: 61 Group Type: isdn-pri Associated Signaling? y Max number of NCA TSC: 10 Primary D-Channel: 01A1023 Max number of CA TSC: 10 Trunk Group for NCA TSC: 61 Trunk Group for Channel Selection: 61 Supplementary Service Protocol: b </pre>
11.	<p>Enter the change private numbering command. Ensure that Network Level is set to 0 and the Level 2 Code and Level 1 Code field values are blank.</p> <pre> change private-numbering Page 1 of 1 NUMBERING - PRIVATE FORMAT Network Level: 0 PBX Identifier: Level 2 Code: Deleted Digits: 0 Level 1 Code: </pre>

Step	Description																				
12.	<p>Enter the change public-unknown numbering l command, where l is an extension length defined in the dial plan (see Section 3.1.2). This table defines the Calling Party Number (CPN) on outbound calls sent to specific ISDN-PRI trunk groups, such as the trunk group connected to Lyrix Enterprise Voice Messaging. Add an entry as follows:</p> <ul style="list-style-type: none">• Ext Len and Ext Code – set to the length and one or more digits consistent with dial plan.• Trk Grp(s) – enter the number of the trunk group configured in Steps 4 – 9.• CPN Prefix – enter any digits to prepend to the extension of the calling station. In the example below, no additional digits are prepended.• CPN Len – enter the total CPN length, comprised of the Ext Len and the CPN Prefix.																				
	<div><div>change public-unknown-numbering 5</div><div>Page 1 of 2</div></div> <div>NUMBERING - PUBLIC/UNKNOWN FORMAT</div> <div>Total</div> <div>Total</div> <table><tr><th>Ext Len</th><th>Ext Code</th><th>Trk Grp(s)</th><th>CPN Prefix</th><th>CPN Len</th><th>Ext Len</th><th>Ext Code</th><th>Trk Grp(s)</th><th>CPN Prefix</th><th>CPN Len</th></tr><tr><td>5</td><td>4</td><td>61</td><td></td><td>5</td><td></td><td></td><td></td><td></td><td></td></tr></table>	Ext Len	Ext Code	Trk Grp(s)	CPN Prefix	CPN Len	Ext Len	Ext Code	Trk Grp(s)	CPN Prefix	CPN Len	5	4	61		5					
Ext Len	Ext Code	Trk Grp(s)	CPN Prefix	CPN Len	Ext Len	Ext Code	Trk Grp(s)	CPN Prefix	CPN Len												
5	4	61		5																	

3.2.4. Routing to Lyrix PeopleFind

Refer to Section 3.1.4

3.2.5. Coverage Path

Refer to Section 3.1.5.

4. Configuring Lyrix PeopleFind

Lyrix configures the PeopleFind application for their end customers. Ensure that the PeopleFind configuration is consistent with the corresponding Avaya Communication Manager configurations described in Section 3.

5. Interoperability Compliance Testing

The interoperability compliance testing focused on verifying E1 and T1 ISDN-PRI QSIG integration between Avaya Communication Manager and Lyrix PeopleFind.

5.1. General Test Approach

The general test approach was to place calls to Lyrix PeopleFind and provide the employee's name. The Lyrix PeopleFind locates the employee by dialing the number at which they can be reached. The employee previously provisioned the number in Lyrix PeopleFind using the Administration menu. The main objectives were to verify that:

- Calls from internal and external callers placed to Lyrix PeopleFind are successfully transferred to the extension selected or entered by the caller.
- Internal and external callers are able to leave voice messages on the Lyrix Voice Messaging server for the correct subscribers.

- Subscribers are able to retrieve their voice messages from the Lyrix Enterprise Voice Messaging server from their own stations, other stations, and external telephones.
- Lyrix Voice Messaging properly turns the Message Waiting Indicator (MWI) of subscriber stations on and off.
- Lyrix PeopleFind components function properly after recovering from failures such as cable disconnects, maintenance activities (busyout/release and reset) on the Avaya TN464F DS1 circuit pack, reset of the Lyrix PeopleFind server, and reset of Avaya Communication Manager.
- Lyrix PeopleFind successfully performs QSIG path replacement after transferring a caller to an Avaya Communication Manager extension.

5.2. Test Results

The feature and functionality test cases passed.

6. Verification Steps

The following steps may be used to verify the configuration:

- From the SAT, enter the command **status signaling-group s**, where **s** is the number of the signaling group configured in Section 3, and verify that the Group State is “in-service”.
- From the SAT, enter the command **status trunk-group s**, where **s** is the number of the trunk group configured in Section 3, and verify that the Service States of all trunks are either “in-service/idle” or “in-service/active”.
- Place a call to the Lyrix PeopleFind number and provide the employee name. Verify the call is successfully transferred and that the trunks between Avaya Communication Manager and the Lyrix server are released due to QSIG path replacement.
- Place a call to the Lyrix PeopleFind number and verify the Administration Menu.

7. Support

For technical support on Lyrix PeopleFind, contact Lyrix support at:

- Phone: 1-800-982-9900
- E-mail: lyrix@lyrix.com

8. Conclusion

These Application Notes described the procedures for configuring E1 and T1 ISDN-PRI QSIG integration between a Lyrix PeopleFind server and Avaya Communication Manager Release 3.1.2. During compliance testing, Lyrix PeopleFind successfully transferred calls to the appropriate Avaya Communication Manager extension, and provided typical voice messaging functionality, including Message Waiting Indicator.

9. Additional References

Product documentation for Avaya products may be found at <http://support.avaya.com>.

[1] *Administrator Guide for Avaya Communication Manager*, Issue 2.1, May 2006, Document Number 03-300509

Product documentation for Lyrix products may be requested at <http://www.lyrix.com>.

[2] *Lyrix PeopleFind Administration Tool*, Version 3.0

[3] *Lyrix QSIG User's Guide*, February 7, 2007

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