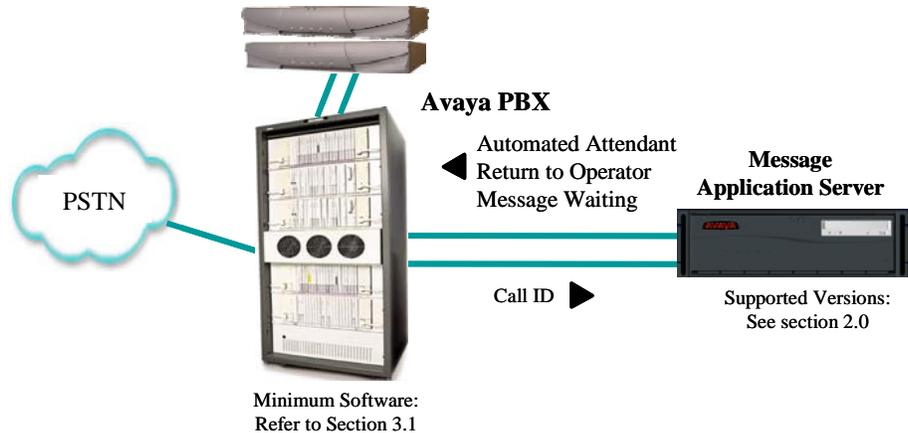


Configuration Note 88502 – Version AI (1/10)

AVAYA Definity G3, Prologix & S87x0/S8xx0 - EMEA E1/QSIG

The PBX and MM are assumed to be collocated. For other configurations please consult with the Switch Integrations group.



Overview

This Configuration Note is intended for Avaya certified Modular Messaging technicians/engineers who are familiar with Modular Messaging procedures and terminology. It also assumes that you are Avaya certified or very familiar with the features and functionality of the Avaya PBXs supported in this Configuration Note and the QSIG protocol.

Use this document in conjunction with [Modular Messaging Installation Guide](#) and the [Avaya PBX Administration Guide](#).

Please [read the entire document](#) before attempting any configuration.

1.0 METHOD OF INTEGRATION

With E1 QSIG integration, one digital pathway between the Avaya™ PBX and the Avaya Message Application Server (MAS) transmits both call information and voice communications. The pathway is provided by an ISDN digital link (QSIG), which provides channels that connect to the Dialogic E1 card. Within the D-Channel, routing information is sent to the MAS containing information regarding the source of the call with reason codes. The MAS processes call information from the supplementary code in the D-Channel, which routes call reasons directly to mailboxes. Message-Waiting indication is set and canceled using the supplementary code service. Voice is carried through the system in digital format.

With E1 QSIG, one digital pathway between the PBX and the Avaya Message Application Server transmits both call information and voice communications

MAS Requirements**¹Release Note:**

Should features of the integration not function optimally when integrated to a PBX or MM that may be operating on an unsupported software release as defined Section 2.0 and 3.1, customers will need to upgrade their PBX and/or MM to a supported software release.

PBX hardware requirements**PBX software requirements****2.0 AVAYA MESSAGE APPLICATION SERVER REQUIREMENTS**

- Dialogic D/600JCT-1E1 or D/300JCT-E1
- CT Bus cable (only required for multiple card installation)
- Software Releases ¹: 1.1, 2.0, 3.0, 3.1, 4.0, 5.0, 5.x

3.0 PBX HARDWARE REQUIREMENTS

- TN464 or MM710 circuits one per each Dialogic E1 Interface Card
- Cables:
-RJ45 on the Dialogic (cable depends on PBX connection)

3.1 PBX SOFTWARE REQUIREMENTS

- Minimum Software ¹:

Definity G3 & Prologix: G3V10.1 Load 43

S8300/S8xx0: MV1.1, CM2.0 / 3.1.x / 4.x / 5.x

Important: Before ordering account teams should check with Avaya Services to determine if there are any applicable patches for customer specific configuration. Specifically if DCS+ or QSIG Networking is used.

- ISDN PRI
- Private Networking
- Basic Call Setup
- Basic Supplementary Services
- Supplementary Services with Rerouting
- DCS+ for Multiple PBX Support (Refer to Section 9.0)
(This option is **only** required for networking (multiple PBXs))

Note: Either DCS+ or QSIG Networks are necessary to support multiple PBXs (Centralized Voice Mail). No other networks are supported. Refer to Section 8.0 (Considerations) for additional info.

Supported integration features

4.0 SUPPORTED INTEGRATION FEATURES

[✓] Items are supported

System Forward to Personal Greeting

All Calls	[✓]
Ring/no answer	[✓]
Busy	[✓]

Station Forward to Personal Greeting

All Calls	[✓]
Ring/no answer	[✓]
Busy	[✓]

Auto Attendant	[✓]
Call Me	[✓]
Direct Call	[✓]
External Call ID (ANI)	[✓]
Fax	[✓]
Find Me	[✓]
Internal Call ID	[✓]
Message Waiting	[✓]
Multiple Call Forward	[✓]
Multiple Greetings	[✓]
N+1	[✓]
Outcalling	[✓]
Queuing	[]
Return to Operator	[✓]

IMPORTANT: PBX options or features not described in this Configuration Note are not supported with this integration. To implement options/features not described in this document, please contact the Avaya Switch Integration product manager.

NOTICE:

The screens in this Config Note are only for illustration purposes.

It is recommended that a qualified technician review the customer's CM QSig programming for accuracy.

***NOTE:**

DCS is only required for networking.

Stand-alone/single PBX do not require DCS, so those options may be set to "N"

****NOTE:**

Cvg of Calls Redirected Off-Net (CCRON) must be set to "y" if you are using *Find Me*.

5.0 CONFIGURING THE PBX TO INTEGRATE

The following programming is intended for certified PBX technicians/engineers. The screens shown in this section are taken from an Avaya PBX administration terminal. Some parameters may not appear on all software releases.

Ensure all required software features are enabled on the PBX. Access the System Parameters Customer Options form. Below is an example of the forms required for QSIG integration, with the required features in **boldface**.

IMPORTANT: Only change the recommended fields. (see note on left)

```

display system-parameters customer-options                               Page 2 of 9
                                OPTIONAL FEATURES

Abbreviated Dialing Enhanced List? y                                Attendant Vectoring? y
Access Security Gateway (ASG)? n                                    Audible Message Waiting? y
Analog Trunk Incoming Call ID? y                                    Authorization Codes? y
A/D Grp/Sys List Dialing Start at 01? n                            CAS Branch? n
Answer Supervision by Call Classifier? y                            CAS Main? n
                                                                    ARS? y                                Change COR by FAC? n
                                                                    ARS/AAR Partitioning? y                Cvg Of Calls Redirected Off-net? y**
ARS/AAR Dialing without FAC? n                                      DCS (Basic)? y*
                                                                    ASAI Interface? y                      DCS Call Coverage? y*
ASAI Proprietary Adjunct Links? y                                  DCS with Rerouting? y*
Async. Transfer Mode (ATM) PNC? n                                    DEFINITY Network Admin? n
Async. Transfer Mode (ATM) Trunking? n                            Digital Loss Plan Modification? n
ATM WAN Spare Processor? n                                          DS1 MSP? y
                                                                    ATMS? y                                DS1 Echo Cancellation? y
    
```

```

display system-parameters customer-options                               Page 3 of 9
                                OPTIONAL FEATURES

Emergency Access to Attendant? y                                    ISDN-BRI Trunks? n
Enhanced EC500? y                                                  ISDN-PRI? y
Extended Cvg/Fwd Admin? y                                          Malicious Call Trace? y
External Device Alarm Admin? 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)? y
Hospitality (G3V3 Enhancements)? n                               Multimedia Call Handling (Enhanced)? y
H.323 Trunks? y                                                    Multiple Locations? n
                                                                    IP Stations? y                          Personal Station Access (PSA)? n
                                                                    ISDN Feature Plus? y
ISDN Network Call Redirection? n
    
```

```

display system-parameters customer-options          Page 4 of 9
                OPTIONAL FEATURES

                PNC Duplication? n                Tenant Partitioning? y
                Processor and System MSP? y        Terminal Trans. Init. (TTI)? y
                Private Networking? y            Time of Day Routing? y
                R9.5 Capabilities? y              Usage Allocation Enhancements? y
                Remote Office? y                  VAL Full 1-Hour Capacity? y
Restrict Call Forward Off Net? y                  Wideband Switching? y
                Secondary Data Module? y          Wireless? y
                Station and Trunk MSP? y
                Station as Virtual Extension? y

(NOTE: You must logoff & login to effect the permission changes.)

```

```

display system-parameters customer-options          Page 7 of 9
                QSIG OPTIONAL FEATURES

                Basic Call Setup? y
                Basic Supplementary Services? y
                Centralized Attendant? y
                Interworking with DCS? n*
                Supplementary Services with Rerouting? y
                Transfer into QSIG Voice Mail? y
                Value-Added (VALU)? y

```

***Note:** This parameter would be set to “y” if you are using DCS to network Avaya PBXs.

Define the QSIG TSC Extension and MWI Number of Digits per Voice Mail Subscriber. The following is an example of the form, with the required features in **boldface**.

```

change system-parameters features          Page 7 of 12
                FEATURE-RELATED SYSTEM PARAMETERS

                ISDN PARAMETERS

                Send Non-ISDN Trunk Group Name as Connected Name? n
                Display Connected Name/Number for ISDN DCS Calls? y
                Send ISDN Trunk Group Name on Tandem Calls? n
                Send Custom Messages Through QSIG? y

                QSIG TSC Extension: 2998
                MWI - Number of Digits Per Voice Mail Subscriber: 4 (*see note below)
                Feature Plus Ext:
                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? y
                Path Replacement with Measurements? y
                QSIG Path Replacement Extension: 2798 (vacant extension)
                Path Replace While in Queue/Vectoring? n

```

NOTE: This parameter must match the number of digits used for mailbox/extension length. For **multiple length extensions** leave this field blank. **Important:** The option to leave this field blank requires Avaya CM 2.1 and later.

Also access the System Parameter Coverage-Forwarding and change or ensure the following parameters are set appropriately:

Maintain SBA At Principal? **n*** (see note on left)
 Coverage Of Calls Redirected Off-Net Enabled? **n**** (see note on left)

***NOTE:**

Maintain SBA at principal set to “n” ensures that when the call covers to voice messaging the appearance on the station is removed to ensure privacy. Doing this prevents someone from listening to the call as it is being recorded by the voice messaging system.

****NOTE:**

Coverage of Calls Redirected Off-Net (CCRON) must be set to “y” if you are using *Find Me*.

```
display system-parameters coverage-forwarding          Page 1 of 2
                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): 2
        Threshold for Blocking Off-Net Redirection of Incoming Trunk Calls: 3

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? n
                Coverage After Forwarding? n
```

```
display system-parameters coverage-forwarding          Page 2 of 2
                SYSTEM PARAMETERS CALL COVERAGE / CALL FORWARDING

COVERAGE OF CALLS REDIRECTED OFF-NET (CCRON)

                Coverage Of Calls Redirected Off-Net Enabled? n**
        Activate Answer Detection (Preserves SBA) On Final CCRON Cvg Point? n
                Ignore Network Answer Supervision? n
        Disable call classifier for CCRON over ISDN trunks? n
```

Change features and assign your private network access code, in this example we assigned 107

```
change feature-access-codes                            Page 1 of 6
                FEATURE ACCESS CODE (FAC)
        Abbreviated Dialing List1 Access Code: 101
        Abbreviated Dialing List2 Access Code: 102
        Abbreviated Dialing List3 Access Code: 103
        Abbreviated Dial - Prgm Group List Access Code:
        Announcement Access Code: 104
        Answer Back Access Code:
        Auto Alternate Routing (AAR) Access Code: 107
        Auto Route Selection (ARS) - Access Code 1: 9      Access Code 2:
        Automatic Callback Activation:                    Deactivation:
        Call Forwarding Activation Busy/DA: 190    All: *9    Deactivation: #9
                Call Park Access Code: *6
                Call Pickup Access Code:
        CAS Remote Hold/Answer Hold-Unhold Access Code:
        CDR Account Code Access Code: 188
                Change COR Access Code:
                Change Coverage Access Code:

        Data Origination Access Code:
                Data Privacy Access Code:
        Directed Call Pickup Access Code:
```

- Install the E1 circuit pack, making sure they are “strapped” for 30-channel operation (E-1). Configure the DS1 circuits as follows:

```

display ds1 01b17
                                DS1 CIRCUIT PACK

                                Location: 01B17                Name: E1 QSIG
                                Bit Rate: 2.048                Line Coding: hdb3

                                Signaling Mode: isdn-pri
                                Connect: pbx                    Interface: peer-master
                                CentreVu Long Timers? n        Peer Protocol: Q-SIG
                                Interworking Message: PROgress Side: a
                                Interface Companding: alaw      CRC? n
                                Idle Code: 11111111            Channel Numbering: timeslot
                                                                DCP/Analog Bearer Capability: 3.1kHz

                                Slip Detection? n                Near-end CSU Type: other
    
```

- Configure a Signaling Group that will be assigned to the DS1 channels. The Signaling Group should be configured as follows:

```

display signaling-group 1
                                SIGNALING GROUP
                                Page 1 of 5

                                Group Number: 9                Group Type: isdn-pri
                                Associated Signaling? y          Max number of NCA TSC: 10
                                Primary D-Channel: 01b1716      Max number of CA TSC: 10
                                                                Trunk Group for NCA TSC: 1
                                Trunk Group for Channel Selection: 1 X-Mobility/Wireless Type: NONE
                                Supplementary Service Protocol: b
    
```

NCA-TSCs are used for QSIG.

CA-TSCs are used with DCS+

Trunk Group for NCA TSC should have a value to indicate trunk group only if using MWI Interrogation.

Important: The Max number of NCA TSC (Non-Carrier Associated Temporary Signaling Channels) should be 1 or higher. This allows a path (1 or more) for MWI to operate.

Create a trunk group, and assign the newly created DS1 channels to it. If the MAS will be configured to perform outcalls, ensure that the COR (Class of Restriction) assigned to this trunk group allows for outside trunk access. The trunk group must be configured as follows:

```

display trunk-group 1                                     Page 1 of 22
TRUNK GROUP

Group Number: 1                                         Group Type: isdn           CDR Reports: y
  Group Name: QSIG - E1                                  COR: 1                    TN: 1          TAC: 21
  Direction: two-way                                     Outgoing Display? y      Carrier Medium: PRI/BRI
  Dial Access? n                                         Busy Threshold: 99       Night Service:
Queue Length: 0
Service Type: tie                                         Auth Code? n             TestCall ITC: rest
Far End Test Line No:

TestCall BCC: 4
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
Calling Number - Delete:                               Insert:                   Numbering Format: unk-unk
  Bit Rate: 1200                                       Synchronization: async   Duplex: full
Disconnect Supervision - In? y Out? n
Answer Supervision Timeout: 0

```

Tip: If the numbering format is set to **unk-pvt** then the PBX looks to the Private-Numbering Table to build the number. The Network Level must not be left blank (in most cases this is set to **0**) or NO number will be sent.

This is where the PBX builds the called party number for the integration. If it is set to **unknown**, the PBX looks at the Public-Unknown table where an entry is required to build to the number. If there is no entry in this table to build the number, NO number is sent.

For Avaya CM 4.0 and later see related note below

```

display trunk-group 1                                     Page 2 of 22
TRUNK FEATURES
  ACA Assignment? n                                     Measured: none           Wideband Support? n
Internal Alert? n                                       Maintenance Tests? y
Data Restriction? n                                    NCA-TSC Trunk Member: 29 (Note1)
  Send Name: y                                         Send Calling Number: y
Used for DCS? n                                         Hop Dgt? n
Suppress # Outpulsing? n                               Numbering Format: unk-pvt (see Tip & NOTE2 below)
Outgoing Channel ID Encoding: exclusive                UUI IE Treatment: service-provider

Replace Restricted Numbers? n
Replace Unavailable Numbers? n
Send Called/Busy/Connected Number: y

Send UCID? n
Send Codeset 6/7 LAI IE? y                             Ds1 Echo Cancellation? n
Modify Reroute Number? y

Path Replacement with Retention? n
Path Replacement Method: always
Network (Japan) Needs Connect Before Disconnect? n

```

NOTE 1: NCA-TSC Trunk Member should have the highest timeslot from the upper E1 span in the trunk group. For example, if there is one E1 span in the MAS then it should be 29, and if there are 2 spans then its should be 59.

NOTE 2: The Numbering Format fields could be set according to the customer's environment. There are many variations plus other tables associated with these fields that must be considered also. Consult with a Software Specialists for proper programming.

There should be 29 total ports for a single E1 span per MAS.

There should be 59 total ports if two E1 spans per MAS.

Do not administer (add) the highest order timeslot/port as a member in the trunk group.

- For an MAS with a single E1 card **Do Not** add port 30. Only 29 ports should be in the trunk group.
- For an MAS equipped with two E1 cards **Do Not** add Port 60. Only 59 ports should be in the trunk group.

The reason for this is due to a limitation with the Dialogic drivers it requires that we leave out the highest order timeslot to enable MWI, even though MWI is sent on the *D-Channel*.

Therefore, only 29 ports are available on an MAS with one E1 card, or 59 ports on an MAS with two E1 cards.

```
display trunk-group 1
```

Page 6 of 22

TRUNK GROUP
Administered Members (min/max): 1/29
Total Administered Members: 29

GROUP MEMBER ASSIGNMENTS

	Port	Code	Sfx	Name	Night	Sig	Grp
1:	01B1701	TN464	F			1	
2:	01B1702	TN464	F			1	
3:	01B1703	TN464	F			1	
4:	01B1704	TN464	F			1	
5:	01B1705	TN464	F			1	
6:	01B1706	TN464	F			1	
7:	01B1707	TN464	F			1	
8:	01B1708	TN464	F			1	
9:	01B1709	TN464	F			1	
10:	01B1710	TN464	F			1	
11:	01B1711	TN464	F			1	
12:	01B1712	TN464	F			1	
13:	01B1713	TN464	F			1	
14:	01B1714	TN464	F			1	
15:	01B1715	TN464	F			1	

```
display trunk-group 1
```

Page 7 of 22

TRUNK GROUP
Administered Members (min/max): 1/29
Total Administered Members: 29

GROUP MEMBER ASSIGNMENTS

	Port	Code	Sfx	Name	Night	Sig	Grp
16:	01B1717	TN464	F			1	
17:	01B1718	TN464	F			1	
18:	01B1719	TN464	F			1	
19:	01B1720	TN464	F			1	
20:	01B1721	TN464	F			1	
21:	01B1722	TN464	F			1	
22:	01B1723	TN464	F			1	
23:	01B1724	TN464	F			1	
24:	01B1725	TN464	F			1	
25:	01B1726	TN464	F			1	
26:	01B1727	TN464	F			1	
27:	01B1728	TN464	F			1	
28:	01B1729	TN464	F			1	
29:	01B1730	TN464	F			1	

See side **NOTE**

Note: For full N+1 support in your MM environment you should set aside a number of ports from each MAS in your environment to be allocated for each of the different port types. When planning trunk group configuration for your multiple MAS environment care should be taken to ensure the same number of ports are allocated to the incoming trunk group for each MAS in your environment. For example, if MAS1 has 16-ports allocated to the incoming trunk group, MAS2 should also have 16-ports allocated to the incoming trunk group. When running in a multiple MAS environment with N+1 support, you should setup the trunks groups as follows:

Incoming Trunk group: should be configured containing all of the required incoming ports from each MAS in your environment, the ports should be entered into the trunk group so Trunk group port-1 is the first port on MAS1, Trunk group port-2 is the first port on MAS2, Trunk group port-3 is the second port on MAS1, etc..

Outgoing Trunk group: should be configured with all of the required outgoing ports from each MAS in your environment, used for MM client outcalls and TUI transfers.'

Refer to Consideration 8.10 for additional N+1 programming.

Change the ISDN Numbering - Private Network form to configure and ensure the PBX for the proper Network Level to be used. Below is a copy of the ISDN Numbering - Private Network form with the required field in **boldface**.

```
display isdn private-numbering
```

```
ISDN NUMBERING - PRIVATE FORMAT
```

```
Network Level: 0                PBX Identifier:  
Level 2 Code:                    Deleted Digits: 0  
Level 1 Code:
```

NOTES for Avaya CM 4.0 and later

Network Levels and Level codes are now found in *system-parameters features* under *Parameters for Creating QSIG Selection Numbers*

Note: If the numbering format is set to **unk-pvt** then the PBX looks to the Private-Numbering Table to build the number. Network Level must not be left blank (in most cases this is set to **0**) or NO number will be sent. This is where the PBX builds the proper number (i.e., user's station number) for the integration to open the proper mailbox.

```

display system-parameters features                                     Page 8 of 17
FEATURE-RELATED SYSTEM PARAMETERS

ISDN PARAMETERS
Send Non-ISDN Trunk Group Name as Connected Name? n
Display Connected Name/Number for ISDN DCS Calls? y
Send ISDN Trunk Group Name on Tandem Calls? n
Send Custom Messages Through QSIG? y

                QSIG/ETSI TSC Extension: 2998
MWI - Number of Digits Per Voice Mail Subscriber: 4 (see note below)
                Feature Plus Ext:
                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? y
Path Replacement with Measurements? y
                QSIG Path Replacement Extension: 2798
                Path Replace While in Queue/Vectoring? n

PARAMETERS FOR CREATING QSIG SELECTION NUMBERS
Network Level: 0
Level 2 Code:
Level 1 Code:
    
```

NOTE: This parameter must match the number of digits used for mailbox/extension length. For **multiple length extensions** leave this field blank (this requires Avaya CM 2.1 or later). However, please note **MM supports only one mailbox length**.

In **Avaya CM 4.0** the Private Numbering Form is now used to define the number format for **specific** trunk groups. In our example screen below, we have a 4-digit extension length that includes extensions from 2000 thru 5999. The 4-digit number will be part of the QSIG number information that will be passed to the MM for call integration.

If you set numbering format to "unk-pvt" on page 2 of the trunk group form (see earlier in this section), which is for the MM Trunk Group, this form must be completed so the CM knows how to build the *private format* number. For that reason, do not leave the form blank or the MM call integration will fail.

IMPORTANT

This screen supports Private Numbering Plans (PNP) allowing you to specify the digits to be put in the Calling Number information element (IE), the Connected Number IE, and the QSIG Party Number for extensions in the Private Numbering Plan.

Avaya CM supports private-network numbers up to 15 digits in length. If the total number — including the level 1 and 2 prefixes, the Private Prefix (formerly known as PBX identifier), and the extension — is more than 15 digits long, neither QSIG Party Numbers nor the information elements are created or sent.

```

change private-numbering 3                                         Page 1 of 2
NUMBERING - PRIVATE FORMAT

Ext  Ext      Trk      Private      Total
Len  Code     Grp(s)     Prefix       Len
4    2          99         4            4
4    3          99         4            4
4    4          99         4            4
4    5          99         4            4
Total Administered: 4
Maximum Entries: 540
    
```

Create a Route Pattern for the trunk group that was previously created for the DS1 channels. The Route Pattern must be configured as follows:

```
display route-pattern 1                               Page 1 of 3
                Pattern Number: 12

  Grp. FRL NPA Pfx Hop Toll No.  Inserted          DCS/ IXC
  No.   Mrk Lmt List Del  Digits          QSIG
                Dgts          Intw
1: 1    0                3                Y   user
2: 2    0                3                Y   user
3:                    (this is an example,   n   user
4:                    see note below)        n   user
5:                    n                       n   user
6:                    n                       n   user

  BCC VALUE  TSC CA-TSC  ITC BCIE Service/Feature BAND  No. Numbering LAR
  0 1 2 3 4 W   Request  Format  Subaddress
1: y y y y y n  Y  as needed rest          unk-unk  rehu
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
```

Note: If adding additional MAS servers, ensure their Trunk Groups are added to this Route Pattern and follow the same programming as the 1st Trunk Group.

Within the AAR Digit Analysis Table, create a dialed string that will map calls to the newly created Route Pattern. The dialed string created in the AAR Digit Analysis Table will be used later in the Hunt Group form that will define the MAS Hunt Group. Below is an example of an AAR dialed string in **boldface**.

```
display aar analysis 15                               Page 1 of 2
                Percent Full: 3

  Dialed      Total      Route      Call      Node      ANI
  String      Min       Max       Pattern   Type      Num      Reqd
4573000    7         7         1         aar              n
```

NOTE: Avaya IP phones may experience DTMF failure when dialing the Hunt Group. They can dial the UDP number or the AAR # above to bypass the Hunt Group and use DTMF capability.

Configure a Hunt Group to be used as the Call Coverage Point for the Call Coverage Path assigned to the MAS subscribers. This hunt group's extension number is going to be used as the MAS Access Number. Enter the dialed string created previously in the AAR Digit Analysis Table in the "Voice Mail Number" field on page 2 of the Hunt Group form. Also, in the "Routing Digit (e.g. AAR/ARS Access Code)" field of this form, enter your PBX's AAR Access Code as defined on page 1 of the Feature Access Codes form. This hunt group is configured with no members assigned to it, and should be configured as follows:

```

display hunt-group 1                               Page 1 of 60
                                         HUNT GROUP

      Group Number: 1                               ACD? n
      Group Name:   QSIG E1                          Queue? n (see note below)
      Group Extension: 3000                          Vector? n
      Group Type:   ucd-mia                          Coverage Path:
      TN: 1        Night Service Destination:
      COR: 1       MM Early Answer? n
      Security Code:
      ISDN Caller Display: grp-name
  
```

Note: Queue? should be set to "n" as recommended. Refer to Consideration 8.9 for further information.

```

display hunt-group 1                               Page 2 of 60
                                         HUNT GROUP

      Message Center: qsig-mwi
      Voice Mail Number: 4573000
      Routing Digits (e.g. AAR /ARS Access Code): 107

      Send Reroute Request: n

      LWC Reception: none
  
```

NOTE: The "Voice Mail Number" entered here must be administered in the "Voice Mail System Configuration->PBX->OutgoingCall" tab of the MAS. The number that is entered in the "Outgoing Call" tab is the Voice Mail Number.

- continued on next page -

- Create a Call Coverage Path that will be assigned to the subscribers' stations. This Call Coverage Path will have the MAS Hunt Group as the Call Coverage Point. Below is an example of a Call Coverage Path.

```

display coverage path 10
                                COVERAGE PATH
                                Coverage Path Number: 1
                                Next Path Number:
                                Hunt after Coverage? n
                                Linkage
                                Number of Rings: 4

COVERAGE CRITERIA
Station/Group Status   Inside Call   Outside Call
Active?                n             n
Busy?                  y             y
Don't Answer?         y             y
All?                   n             n
DND/SAC/Goto Cover?   y             y

COVERAGE POINTS
Terminate to Coverage Pts. with Bridged Appearances? n
Point1:h1              Point2:
Point4:                Point5:
Point3:                Point6:
    
```

- Configure the subscriber stations, assigning the newly created Call Coverage Path (i.e. 10) to them. Also, ensure the following setting for subscribers:

```

display station 3002
                                Page 1 of 4
                                STATION
Extension: 3002              Lock Messages? n          BCC: 0
Type: 8410D                 Security Code:             TN: 1
Port: 01A0102              Coverage Path 1: 10       COR: 1
Name: Smith John           Coverage Path 2:          COS: 1
                                Hunt-to Station:

STATION OPTIONS
Loss Group: 2               Personalized Ringing Pattern: 1
Data Module? n              Message Lamp Ext: 3002
Display Module? y
Speakerphone: 2-way         Mute Button Enable? y
Display Language: english
    
```

```

display station 3002                                     Page 2 of 4
                                                    STATION
FEATURE OPTIONS
    LWC Reception: none      Auto Select Any Idle Appearance? n
    LWC Activation? n        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 Call Alerting? n      Restrict Last Appearance? y
    Active Station Ringing: single
H.320 Conversion? n      ► Per Station CPN - Send Calling Number? y
MWI Served User Type: qsig-mwi      Display Client Redirection? n
                                      Select Last Used Appearance? n
                                      Coverage After Forwarding? s
    
```

Per Station CPN – Send Calling Number must be set to “y” for the integration to work properly.

Note: With QSIG integration the following fields are required for Message Waiting:

- LWC Reception: none**
- MWI Served User Type: qsig-mwi**

Single Line sets should have field “**Message Waiting Indicator**” set to “**led**” or “**neon**,” depending on the type of telephone set used. Also, the “**Number of Rings**” field should be set to a minimum of 4 rings, to allow Personal Assistance to work properly.

Save these PBX changes.

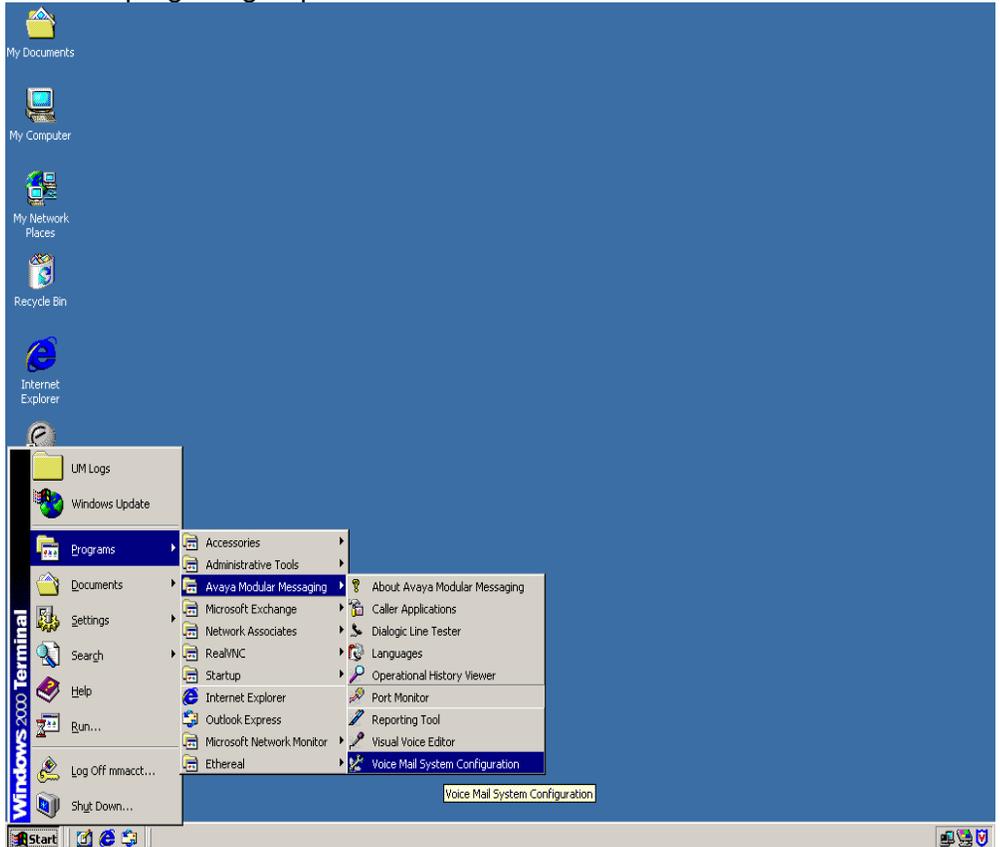
Please refer to the Consideration section at the end of this document for special PBX programming considerations.

Configuring the MAS

6.0 CONFIGURING THE MESSAGE APPLICATION SERVER

Configuring the MAS platform for proper PBX integration requires configuring several menus accessed within the Voice Mail System Configuration application, and a certified MM engineer.

- Access the **Voice Mail System Configuration** application from the MAS program group.



Special Note: Verify the CRC value in the QTE.PRM is set correctly. If left uncorrected the Dialogic E1 boards will not initialize properly. The file is found in:

C:\Program Files\Dialogic\data\qte.prm

Open the qte.prm file. About 23 lines into the file text you will see a setting under the line as show in bold below.

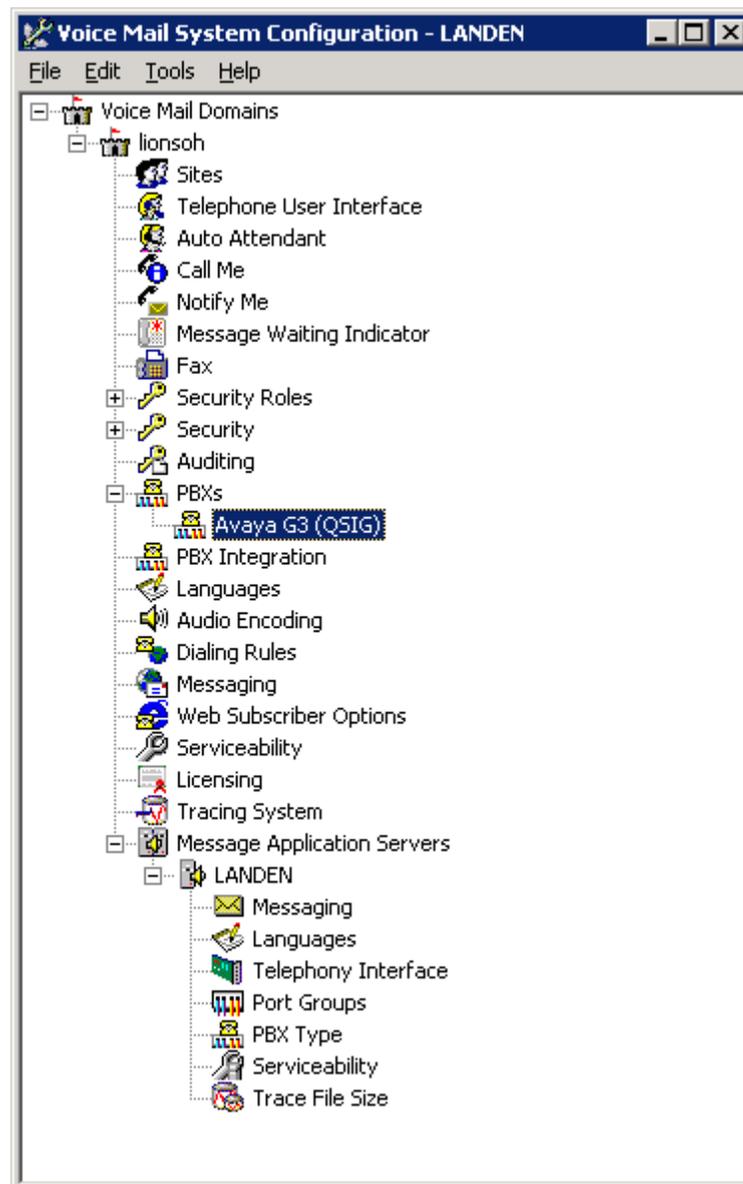
```

;---
;--- CRC ENABLE SWITCH (Parameter type 0FH - E1 only)
;--- Turns the transmission of the CRC4 pattern on or off and searches
;--- for such a pattern in the received signal.
;--- Possible values for the data are as follows:
;---     00H = Turn CRC off.
;---     01H = Turn CRC on.
;--- The default data value is 00H (Turn CRC off).
000F 01

```

The example number shown above in blue is incorrect. Change this from **000F 01** to **000F 00**. Save then file and then reboot the MAS.

Expand all fields so all-applicable options are visible:

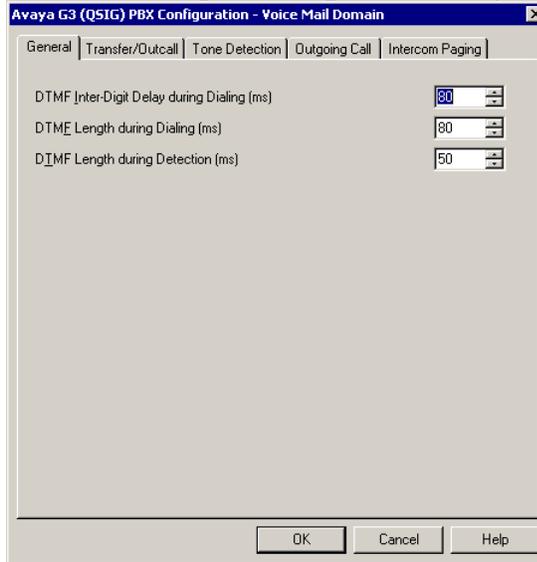


Note: Starting with MM 5.0 additional Fields such as *Sites and PBX Integration* will appear on the VMSC screen.

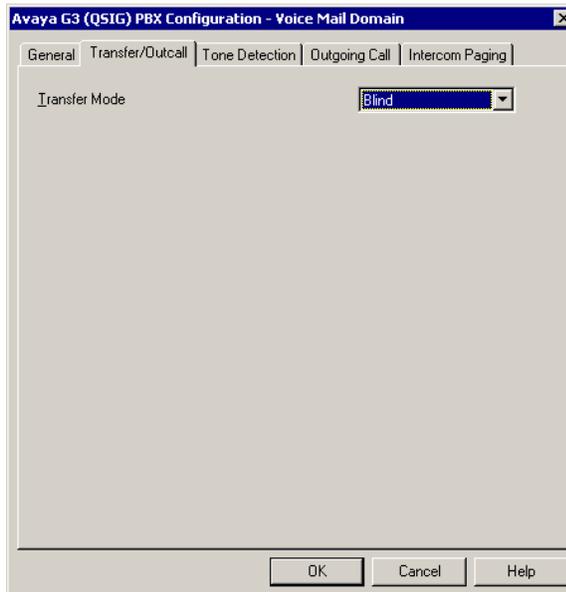
Ensure the new PBX is added as instructed by the Modular Messaging Installation guide. The new PBX should be:

Avaya G3 (QSIG)

1. Select the **Voice Mail Domain**
2. Expand **PBXs**
3. Select the newly created **Avaya G3 (Dialogic QSIG)**
4. Access the **General (QSIG) PBX Configuration** tab
5. **DTMF Inter-Digit Delay during Dialing (ms) = 80**
6. **DTMF Length during Dialing (ms) = 80**
7. **DTMF Length during Detection (ms) = 50**



- Next access the **Transfer/Outcall** tab
Transfer Mode = Blind



- Next access the **Tone Detection** tab
Maximum Silence before Hanging Up (ms) = 6000



- Next access the **Outgoing Call** tab

Avaya G3 (QSIG) PBX Configuration - Voice Mail Domain

General | Transfer/Outcall | Tone Detection | **Outgoing Call** | Intercom Paging

Layer1 Protocol: G.711 A-Law

BC Transfer Cap: Speech

Number Type: Unknown

Number Plan: Unknown

Origin Number: 4573000

OK Cancel Help

1. **Layer Protocol** = G.711 A-Law
2. **BC Transfer Cap** = Speech
3. **Number Type** = Unknown
4. **Number Plan** = Unknown
5. **Origin Number** = 4573000 (The number entered here should be the number entered in the “Voice Mail Number” field on page 2 of the Hunt Group form)

Note: Layer protocol fields should match the Interface Companding setting selected on Page1 of the DS1 Circuit Pack configuration screen on Page 7 of this configuration note.

- Next access the **Message Waiting Indicator (MWI)** tab

Support NOTE
for MM 1.1 & 2.0 only:

- Limit Requests should be checked.
- Maximum Requests per Minute should be set for 200.

1. **Enable Message Waiting Indicator (MWI)** = Enable by checking the box
2. **MAS MWI Server** = Enter the name of the MWI server created during the installation procedure.
3. **Scheduled MWI updates: Active or Inactive** = Configure as per customer requirements.*
4. **Limit requests** = Leave Unchecked
5. **Maximum requests per Minute** = <grayed out>
6. **Message Application Servers that Support MWI** = This box should contain a list of MAS servers capable of placing MWI requests.
7. Select **OK** to save changes

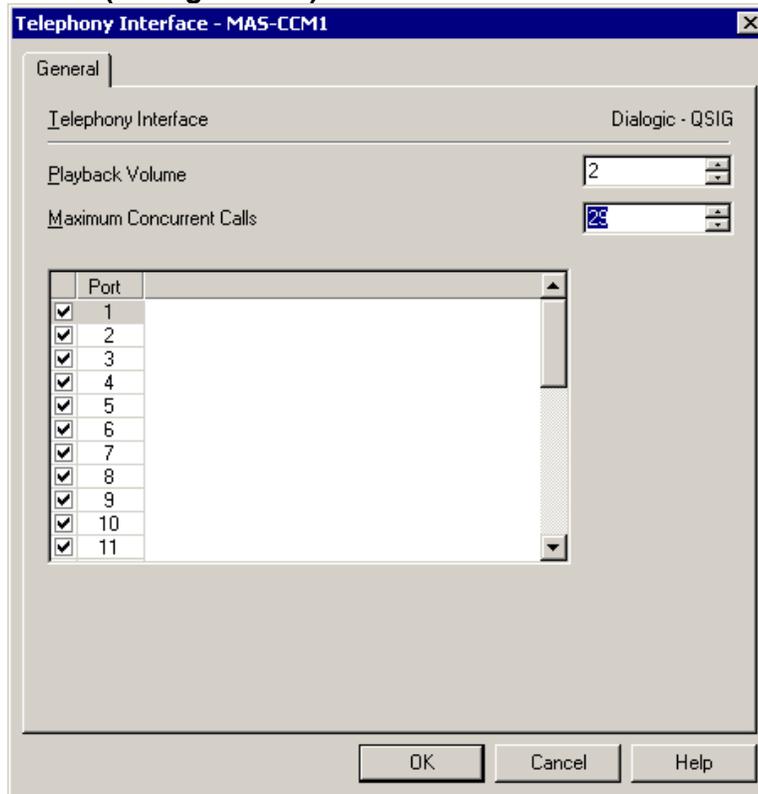
*Note: The Scheduled MWI updates parameter is only available on MM 3.x

Tip: To make the QSIG or set emulation telephony interface active, click the down arrow and click

Make Active.

If the QSIG or set emulation telephony interface is already active, this field does not appear.

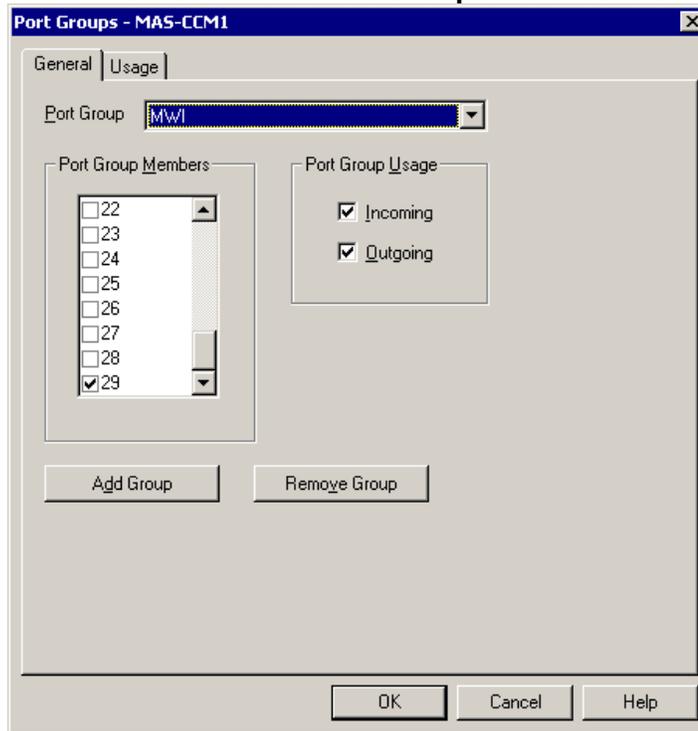
- Next access the **General** tab within the **Telephony Interface (Dialogic-QSIG)** tab



1. **Playback Volume = 2**
2. **Maximum Concurrent Calls =** Enter the number of ports connected to the PBX (i.e. 29)
3. **Port =** Ports are enabled by default
Note: The MAS service must be restarted to allow port disabling
4. Select **OK** to save changes

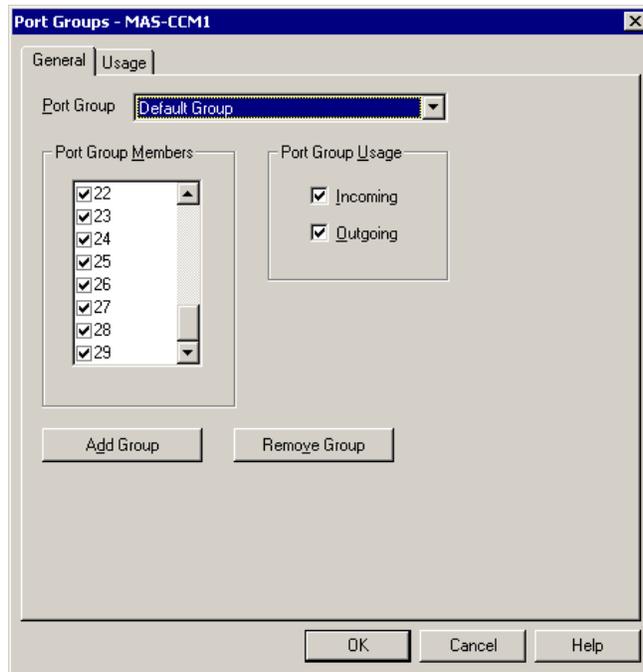
Note: For a QSIG link, the total number of ports specified must occupy the first available set of contiguous channels. For example, if there are two boards on a system, one with 10 ports enabled and one with 29 ports enabled, then the first 10 ports will be enabled on one boards, and all 29 on the second board. Use the port checkboxes to enable or disable the correct ports.

- Next access the **Port Groups** tab under the MAS name



NOTE: The MWI port within the MWI Port Group it's used by the MWI sub-system to control concurrent MWI requests. This does not affect incoming/outgoing traffic to the port in anyway.

1. Click **Add Group** button
2. Name Group **MWI**
3. Within the new **MWI** Port Group uncheck all **Ports** except the MWI port. This will be the upper most port of the trunk group. Port 29 on a single board or Port 59 if 2 boards (see side NOTE). This port can receive calls, so ensure it is programmed for both **Incoming** and **Outgoing** under the **Port Group Usage**
4. Return to the **Port Groups** General tab and ensure the **Default Group** under **Port Groups** is configured to meet the customer's need for **Incoming** and **Outgoing** under **Port Group Usage**.

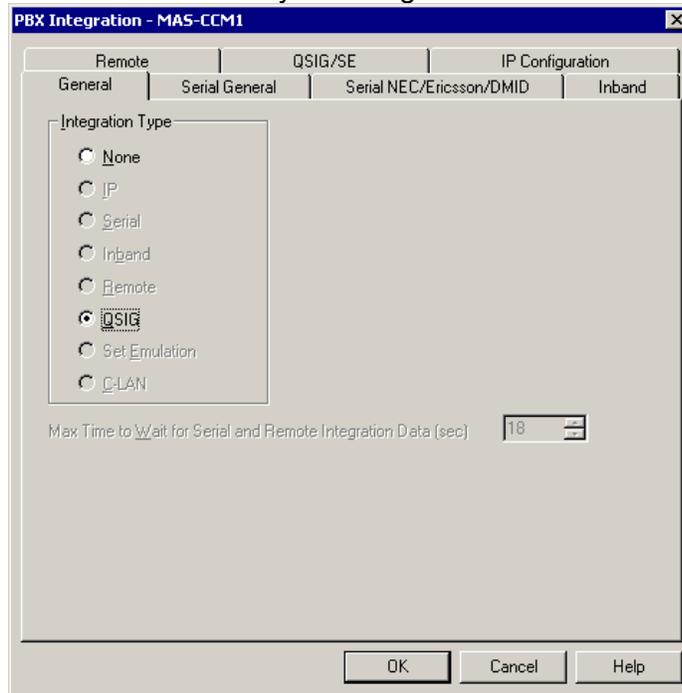


5. Access the **Port Group** General tab and check all **Ports** (1-29 for a single span or 1-59 if 2 spans) including the MWI (29 or 59) port.
6. Select **OK** to save changes

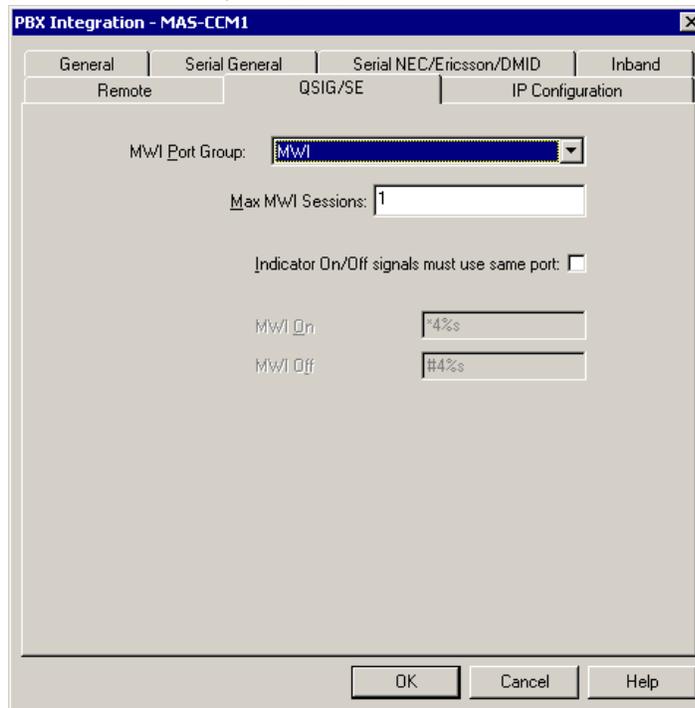
NOTE: This Port Group configuration applies to each MAS

- Next access the **General** tab within the **PBX Integration**

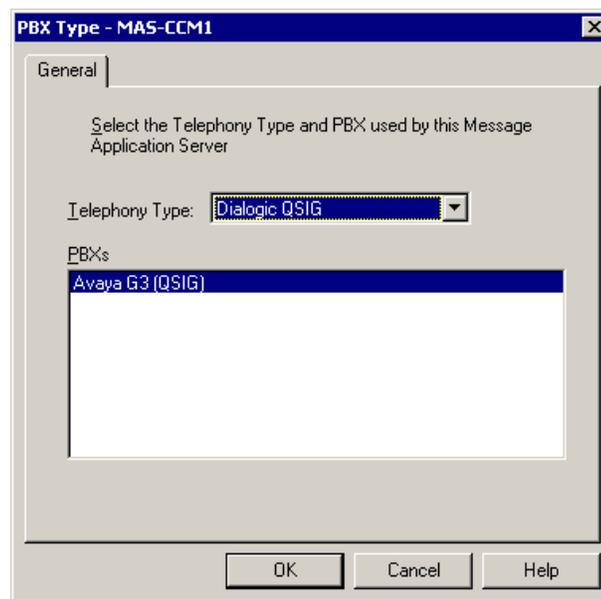
1. **QSIG** = Enable by checking the box



2. Access the **QSIG/DSE** tab



3. **Port Group Name** = MWI
 4. **Max MWI Sessions** = 1
 5. **Indicator On/Off signals must use same port** = Leave Blank
 6. **MWI On Field** = Leave as default (can't be changed)
 7. **MWI Off Field** = Leave as default (can't be changed)
 8. Select **OK** to save changes
- Next access the **General** tab within the **PBX Type** tab
 1. Telephony Type = **Dialogic QSIG**
 2. Under PBXs ensure **Avaya G3 (Dialogic QSIG)** is selected
 3. Select **OK** to save changes



After making these changes, return to “Configuring the voicemail system” within the Message Application Server Installation Guide. Ensure you are prompted to restart the Message Application Server services to apply these changes.

Important notes regarding
this integration

8.0 CONSIDERATIONS/ALTERNATIVES

8.1 QSIG Integration does not support forwarding/transfer from a Vector. Currently, if calls are routed from a vector to the QSIG link(s) connected to the MAS, the call will not pass the VDN as the called party ID. Applications requiring calls that are routed from vectors to mailboxes on the MAS can be configured so as to route calls to phantom extensions (X-ports) configured to call-cover all-calls to the MAS hunt group.

Note: Patch 7960 corrects this. Avaya CM 2.0.1 and later releases include this fix/patch.

8.2 Outcalls to pagers placed over analog trunks may fail. If the CO connecting the analog trunks to the PBX does not provide answer supervision, the MAS will not out-pulse DTMF digits to the pager terminal. This problem can be eliminated by installing a Call Classifier board in the PBX (if one is not already installed), enabling system parameter customer-option “Answer Supervision by Call Classifier”, and enabling “Answer Supervision” in the Trunk Group associated with the outgoing analog trunks accessed during the outcalls. Outcalls over digital trunks are not affected.

8.3 Transfers to ringing use additional ports. When performing unsupervised transfer, and the transferred-to extension forwards back to the MAS, additional ports are tied up on the MAS server, as “Path Replacement” does not occur. Two additional ports are used for each number dialed through the automated attendant, and all these ports are in use until the transferred call is answered or the caller disconnects from the message server. Customers should consider implementing supervised transfers, or installing additional ports. Note that with supervised transfers, callers are not provided with music on hold, but are instead prompted to wait during the silence. The called party will hear a “Connecting” prompt as he/she answers the call.

8.4 When multiple PBX’s are arranged in a QSIG network, care must be taken to configure the QSIG tie trunks properly. In order to provide full feature functionality to all subscribers, the trunk group(s) assigned to the QSIG tie trunks connecting all PBX’s in the network must match the configuration of the trunk group form (page 1 and 2 of the form) assigned to the MAS QSIG trunks. An example of the trunk group administration form is illustrated on the PBX programming section of this document.

8.5 Message Waiting Interrogation is a feature that allows the PBX to request message waiting indicator status of subscribers’ stations from

the MAS. The following conditions will result in the PBX requesting message waiting indicators status:

- **Switch reload.** 15 minutes after a switch reload, a message waiting interrogation request will be sent to the MAS.
- **Interruption of service on a QSIG link.** 15 minutes after service is re-established on the QSIG link, a message waiting interrogation request is sent to the MAS.
- **Scheduled daily maintenance routine.** Once daily, during the scheduled maintenance period (usually around 1:00 A.M. on most switches), a message waiting interrogation request is sent to the MAS.

8.6 MAS can support multiple Avaya PBX's in a “centralized” voice mail arrangement only if all PBX's are connected using QSIG. If Avaya G3/MultiVantage PBX's uses DCS for networking, QSIG integration cannot provide full functionality to remote PBX subscribers. Please refer to Section 9.0 to support a centralized environment using DCS+.

When running MWI on your Voice Server you should setup a separate trunk group containing 2channels from one of your QSIG trunks, these channels should not be apart of any other trunk group. You should then create a new MWI port group containing these channels and make the relevant changes in MAS VMCS.

8.7 Support for Transfer to QSIG Voice Mail feature. In order to provide support for the Transfer to QSIG Voice Mail feature on your QSIG integrated MAS solution, you need to ensure the following criteria are met:

Your switch will need to be running Definity software R8 or later.

- 1. You will need to have the following features enabled on your switch:**
 - ISDN-PRI or ISDN-BRI
 - QSIG Supplementary Services
 - Transfer to QSIG Voice Mail
- 2. A Feature Access code for Transfer to QSIG Voice Mail, will need to be configured.**
- 3. Your MAS Voice Server will need to be using the QSIG-MWI hunt group integration as described in this document.**

Example on how the feature works:

Extension 7001 has voice mail but calls received are forwarded to extension 7002 rather than directly to voice mail.

Callers to 7001 are answered at 7002 but the caller wishes to leave a voice mail for the person at 7001.

The person at 7002 transfers the caller directly to the voice mail for 7001 by pressing the transfer button, then the code for transfer to voice mail (*8,*50, etc.), then presses transfer again.

The caller immediately hears the greeting of 7001 (no additional ringing) and can subsequently leave a message.

Note: For full details on how to configure and implement the Transfer to QSIG Voice Mail feature on your Avaya switch, please consult your Avaya Sales or Support representative.

8.8 Possible reasons for QSIG Path Replacement failure:

Note: Path Replacement is a function of the PBX. There are no Path Replacement settings on MM.

- When the QSIG Message Application Server Auto Attendant, Caller Application or Find Me features are transferring a call, you will see a second channel in Port Monitor appear busy until the transfer has been completed. After the transfer has been completed you will see that both channels are now idle in Port Monitor. This shows that the QSIG Path Replacement feature has completed successfully.
- However there are some switch configurations that will cause Path Replacement to fail, therefore your MM Server will stay bridge onto the transferred call keeping two channels busy in Port Monitor. The following list of features can cause Path Replacement to fail:
 - **Call Vector.** A call that has experienced SS-CT (Supplementary Service Call Transfer) that terminated to a vector and received answer treatment can have its path replaced. This is allowed only after a true user answers the call so that vector processing is completed. A vector step of recorded announcement or wait listening to music could cause a CONNECT message to be sent back and make the far end think ANF-PR (Path Replacement Additional Network Feature) could take place. If the PR Propose is received while still in vector processing, the vectoring PBX will deny the ANF-PR attempt but will initiated its own attempt at ANF-PR when a user does answer the call.
 - **Malicious Call Trace (MCT).** While Malicious Call Trace (MCT) is active and if there is a trunk involved in the call, MCT

feature does not allow the trunk resources to be released from the switch-side to facilitate the tracing activity. Also, a MCT controller on switch A may request a controller on switch B to continue tracing a call that was tandem through switch B by providing the trunk member id they wish to have traced. So, ANF-PR shall not be done while MCT is active on the call.

- **Restriction Features.** *Class of Restriction (COR).* The restrictions placed on routing calls are in affect for ANF-PR. Any call that cannot be originated or terminated because of COR on a regularly dialed basis will not be originated or terminated when that call is made ON BEHALF OF that terminal by ANF-PR.
 - *Voice Terminal Restrictions.* Voice Terminal Restrictions for the reroute of an ANF-PR call will be enforced.
 - *Inward.* If a terminal is Inward restricted then that terminal would not be able to accept an incoming call. This would include a new path SETUP message. Therefore a Requesting PBX should not bother proposing ANFPR, since it is destined to fail.
 - *Manual Terminating Line.* If a terminal is Manual Terminating restricted then that terminal would not be able to accept an incoming call except from an attendant. Thus a new path SETUP would be denied. Therefore a Requesting PBX should not bother proposing ANF-PR, since it is destined to fail.
 - *Origination.* If a terminal is Origination restricted then that terminal would not be able to make an outgoing call. Thus originating a new path SETUP should be denied. Therefore a Cooperating PBX should reject the path replacement proposal/request.
 - *Outward.* If a station is outward restricted at the cooperating end of an ANF-PR call then no ANF-PR SETUP should be attempted and ANF-PR should fail. Any call that cannot be originated outward (because of COR) on a regularly dialed basis will not be allow to originate outward when that call is made ON BEHALF OF that terminal by ANF-PR.
 - *Termination.* If a terminal is Termination restricted then that terminal would not be able to accept an incoming call. Thus an incoming new path SETUP would be denied. Therefore a Requesting PBX should not bother proposing ANF-PR, since it is destined to fail.
 - *TAC (Trunk Access Code).* When an outgoing call is made using a TAC, or a call was extended by an attendant using DTGS (Direct Trunk Group Selection), the user has

intentionally chosen a particular Trunk Group for the outgoing call. ANF-PR will not replace the path in this case.

8.9 The Communication Manager does not support call queuing on QSIG trunks. Hence, calls cannot be queued to MAS ports. The user audible behavior is that during peak traffic, when all MAS ports are busy, a caller will hear a fast busy. They should hang up and try at a later time.

8.10 When configuring an N+1 MM environment the following changes should be made to the MAS Service in the services applet, all steps should be completed:

1. Double click the Monitor icon on the desktop.
2. Select Services
3. Locate the Message Application Service
4. Right click the Message Application Service and select properties from the menu.
5. Select the Recovery tab.
6. For First Failure select Run File.
7. Select the Browse button and locate the QSIGRecover file:
\\Avaya_Support\Tools\QSIGRecover\QSIGRecover.exe
8. Enter the following in the Command Line Parameters box:
If you have one E1 board in your MAS:
/recover /boards 1
If you have two E1 boards in your MAS:
/recover /boards 2
If you have three E1 boards in your MAS:
/recover /boards 3
9. Check the Append fail count to end of command line checkbox.
10. Repeat steps 6 to 8 for Second Failure.
11. For Subsequent Failures select Reboot The Computer.
12. Select OK to save changes.
13. Select Run from the Start Menu
14. Enter cmd and select OK
15. At the command prompt navigate to the
Avaya_Support\Registry_Keys folder.
16. Enter the following: stopdriversonshutdown.reg and press the return key. Note: This registry file will apply a change to the registry key to stop the Dialogic drivers when the MAS service is stopped.
17. Close the command prompt.

8.11 Trunk-to-Trunk is not required to support Find Me if the minimum releases indicated below are met. Previously, when a public network call arrived at an Avaya™ Communication Manager system and was

routed via coverage to a QSIG trunk connected to a Modular Messaging system equipped with the Find Me feature, the Find Me feature would place the call to the user and connect the calling and called parties. When the Communication Manager received the Transfer Complete messages from the Messaging system, and Path Replacement was enabled on the Communication Manager, it would proceed with the Path Replacement. While performing this task, the Trunk-to-Trunk Transfer parameter would be checked, and if set to “none”, the call between the calling party and the found user would be torn down. A change was put into the following releases and load numbers to correct this:

- 1.3 Load 537.0
- 2.0 Load 226.0
- 2.1 Load 411.0

When the Communication Manager System is running on one of these loads or a later one, the Path Replaced call will not be torn down.

8.12 Call Call transfers may not display the Call ID to ringing phones.

The Call ID is not provided until the subscriber answers the phone.

- This issue was resolved in MM 3.0 where the Calling Party Number (number only, not the name) is displayed.
- Starting with MM 5.1, both name and calling party number will be displayed on ringing phones. Please note, for name to be displayed the PBX must send the MAS the Calling Name using the QSIG Name Identification Supplementary Services Protocol (ECMA-164).

8.13 The Communication Manager supports up to 28 lines of input within the DCS to QSIG TSC Gateway. This limitation affects how many entries can be configured for remote locations in a centralized voice mail environment.

8.14 If the Avaya CM is networked to a non-Avaya PBX using H.323 trunking, please consult PSN001841u.

8.15 If the Avaya CM is networked to non-Avaya PBXs that use MM for voice messaging the Avaya CM integrated to MM should be at the following minimum releases:

- Avaya CM 3.1.4 SP 2 or later
- Avaya CM 3.1.5 SP0 or later
- Avaya CM4.0.3 base or later
- Avaya CM 4.0.4 base or later
- Avaya CM 5.0 SP3 or later
- Avaya CM 5.1.0 base or later such as 5.1.1 and 5.1.2

8.16 Re-Route Request set to “y” (page 2 of the Hunt Group Form) eliminates potential call tromboning where 3-legs may remain active and Path Replacement in effect appears to have failed. Should remote

sites calling the MM experience call delay or calls not completing this may be due to a re-route request being sent back to the remote and that site not being administered with a dial plan to recognize the MM pilot number. Changing Send Re-Route Request to “n” is a common solution but recreates the tromboned call. To avoid issues it is best to have a proper dial plan so the remote PBX knows how to reroute the call when requested.

8.17 MM Caller Applications with Vectors — when using MM Caller Applications with Vectors, we have found that in some cases Path Replacement may not occur. We have found that adding a “wait 10 seconds hearing music” step to the beginning of a vector” provides the time needed to hold the call in vector processing and allow path replacement to occur before hitting the messaging step. This answer supervision is to so we can path replace the call, before the vector sends the call anywhere else. No other option on the wait step will work.

- continued on next page -

9.0. ADDENDUM ON CONFIGURING MULTIPLE PBXS (DCS+ NETWORKING/INTERNETWORKING)

The following information is not intended for new installations. This Addendum assumes the customer has an existing Network already in place. Please refer to the appropriate PBX installation guide for brand new networking installations. Obtain a configuration printout of the existing network to use as a reference.

Ensure the integration is working properly within the PBX where the MAS will reside (Hub Node) before continuing with the networking configuration. Configure the remote switch (i.e. Node 2), Definity G3, Prologix, etc., by following the screens below.

The Hub Node will not require changes to the Trunk Group; however configure the Signaling Group, which will be assigned to the DS1 channels. The following is an example of the changes highlighted in **boldface**:

NOTICE:

The screens in this Config Note are only for illustration purposes.

It is recommended that a qualified technician review the customer's CM QSig programming for accuracy.

```

change signaling-group 7                               Page 1 of 5
                                SIGNALING GROUP

Group Number: 7          Group Type: isdn-pri
Associated Signaling? y          Max number of NCA TSC: 16
Primary D-Channel: 01A0824      Max number of CA TSC: 31
                                Trunk Group for NCA TSC: 4

Trunk Group for Channel Selection: 4
Supplementary Service Protocol: a
    
```

```

change signaling-group 7                               Page 2 of 5
                                ADMINISTERED NCA TSC ASSIGNMENT

Service/Feature:          As-needed Inactivity Time-out (min):
TSC   Local
Index Ext.  Enabled Established   Dest. Digits   Appl.         Mach.
1:    2299   y      permanent   2100          dcs           1
2:    3081   y      permanent   3083          qsig-mwi     2
3:
4:
5:
etc..
    
```

Note: The **Group Type** will depend on the customer's environment. The trunks can either be T1, E1, IP, etc.. For example, they could be:
 Group Type: **isdn-pri** or Group Type: **h.323**
 Additionally, you should consult with a Software Specialist to ensure the Numbering Format of the trunk group is configured appropriately to route Call ID to the Modular Messaging from remote switches.

On the Hub Node program the ISDN QSIG TO DCS TSC GATEWAY. This defines the stations that are DCS to another Node (in our example DCS Node 2). This allows MWI to be directed via DCS to Node 2 from the MAS.

```
change isdn qsig-dcs-tsc-gateway
```

QSIG TO DCS TSC GATEWAY						
Subscriber Number	Sig Grp	TSC Index	Subscriber Number	Sig Grp	TSC Index	
63xxx	7	2				
6409	7	2				
6411	7	2				
6412	7	2				
6415	7	2				
6430	7	2				

Important: Use caution when completing this task. The data in this field will display exactly as it is entered. This is critical because when the switch makes a selection it will use the first match.

On the Hub Node program the ISDN DCS TO DCS TSC GATEWAY. This defines the stations using DCS to another Node (in our example DCS Node 2). This allows incoming calls from DCS Node 2 to be directed to the MAS.

```
change isdn dcs-qsig-tsc-gateway
```

DCS TO QSIG TSC GATEWAY									
Mach ID	Sig Grp	TSC Index	Voice Mail Number	AAR/ ARS Access Code	Mach ID	Sig Grp	TSC Index	Voice Mail Number	AAR/ ARS Access Code
2	7	2	4575678	107					

❑ The Remote switch (DCS Node 2) does not require changes on the DCS Trunk Group incoming to the Hub Node. However the Signaling Group for this Trunk Group requires changes. The following is an example of the changes highlighted in **boldface**:

```
change signaling-group 7                               Page 1 of 5
                SIGNALING GROUP

Group Number: 7      Group Type: isdn-pri
  Associated Signaling? y      Max number of NCA TSC: 16
  Primary D-Channel: 01A0824      Max number of CA TSC: 31
                                     Trunk Group for NCA TSC: 7
Trunk Group for Channel Selection: 7
  Supplementary Service Protocol: a
```

```
display signaling-group 7                               Page 2 of 5
                ADMINISTERED NCA TSC ASSIGNMENT

Service/Feature:      As-needed Inactivity Time-out (min):
TSC  Local
Index Ext.  Enabled Established  Dest. Digits  Appl.      Mach.
1:   2299   y      permanent  2100         dcs        1
2:   6451   y      permanent  3081         audix      1
3:                   n
4:                   n
5:                   n
etc..
```

❑ Create a Hunt Group for messaging from the remote Node.

```
display hunt-group 2                               Page 1 of 10
                HUNT GROUP

Group Number: 2      ACD? n
Group Name: S3400 VOICEMAIL REMOTE      Queue? n
Group Extension: 6300      Vector? n
Group Type: ucd-mia      Coverage Path:
  TN: 1      Night Service Destination:
COR: 8      MM Early Answer? n
Security Code:
ISDN Caller Display:
```

```
display hunt-group 2                               Page 2 of 10
                HUNT GROUP

Message Center: rem-vm
Voice Mail Extension: 3000

Send Reroute Request: y

Calling Party Number to INTUITY AUDIX? y
LWC Reception: none
```

Note: Ensure the voice mail extension is the pilot number of the voicemail system and not the lead number of the qsig-mwi hunt group at the host site.

```

display hunt-group 2                                     Page 3 of 10
                HUNT GROUP
      Group Number: 2      Group Extension: 6300      Group Type: ucd-mia
      Member Range Allowed: 1 - 200      Administered Members (min/max): 0 /0
                                          Total Administered Members: 0
GROUP MEMBER ASSIGNMENTS
      Ext      Name (24 characters)                Ext      Name (24 characters)
      1 :                                           14 :
      2 :                                           15 :
      3 :                                           16 :
      4 :                                           17 :
      5 :                                           18 :
      6 :                                           19 :
      7 :                                           20 :
      8 :                                           21 :
      9 :                                           22 :
     10 :                                           23 :
     11 :                                           24 :
     12 :                                           25 :
     13 :                                           26 :

      At End of Member List
    
```

The key in the remote Node is the method the remote Voice Mail Extension of the Hunt Group routing is configured. Change the Uniform Dial Plan and add RNX 457 and direct this to the AAR

```

display uniform-dialplan 0                               Page 1 of 2
                UNIFORM DIAL PLAN TABLE
                                          Percent Full: 0
Matching      Insert      Node      Matching      Insert      Node
Pattern  Len Del  Digits  Net Conv Num  Pattern  Len Del  Digits  Net Conv Num
  2         4  0   221   aar  n
 30         4  0   221   aar  n
 3000     4  0   457 aar n
 31         4  0   221   aar  n
  5         4  0   221   aar  n
 63         4  0           ext  n
 64         4  0           ext  n
                                     n
                                     n
                                     n
                                     n
    
```

In the remote Node AAR Analysis table route 4573000 to route pattern 2.

```
display aar analysis 2
```

Page 1 of 2

AAR DIGIT ANALYSIS TABLE

Percent Full: 9

Dialed String	Total Min	Total Max	Route Pattern	Call Type	Node Num	ANI Reqd
4573000	7	7	2	aar	1	n

In the remote Node Route Pattern 2 insert the MAS AAR Access code of 107.

```
display route-pattern 2
```

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Pattern Number: 2

Grp. No.	FRL	NPA	Pfx	Hop	Toll	No. Del	Inserted Dgts	DCS/ QSIG	IXC
1:	7	0					107	y	user
2:								n	user
3:								n	user
4:								n	user
5:								n	user
6:								n	user

	BCC	VALU	TSC	CA-TSC	ITC	BCIE	Service/Feature	BAND	No. Dgts	Numbering	LAR
	0	1	2	3	4	W	Request		Format		
1:	y	y	y	y	y	n	y as-needed	rest	unk-unk	Subaddress	none
2:	y	y	y	y	y	n		rest			none
3:	y	y	y	y	y	n		rest			none
4:	y	y	y	y	y	n		rest			none
5:	y	y	y	y	y	n		rest			none
6:	y	y	y	y	y	n		rest			none

- Create a Call Coverage Path that will be assigned to the subscribers' stations. This Call Coverage Path will have the Remote Voice Mail Hunt Group as the Call Coverage Point. Below is an example of a Call Coverage Path.

```

display coverage path 3
                                COVERAGE PATH
                                Coverage Path Number: 3
                                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: 4
                                All?                    n             n
                                DND/SAC/Goto Cover?     Y             Y
                                COVERAGE POINTS
                                Terminate to Coverage Pts. with Bridged Appearances? n
                                Point1: h2              Point2:
                                Point4:                 Point5:
                                                    Point3:
                                                    Point6:

```

- Configure the remote subscriber stations, assigning the newly created Call Coverage Path to them. They should also have LWC and the AUDIX Name set to AUDIX.

-continued on next page -

CHANGE HISTORY		
Revision	Issue Date	Reason for Change
Version F	11/13/03	Updated Route Pattern to support multiple MAS. Changed LAR from "none" to "rehu"
Version G	12/17/03	Updates to meet Release 1.1
Version H	01/06/04	Additional MM1.1 updates.
Version I	04/01/04	Added DCS+ or QSIG as the only supported networks for multiple PBX support. Also, changed the Port Group requirements to indicate reserve ports for Outgoing ports.
Version J	05/14/04	Added Consideration 8.11, 8.12 & 8.13
Version K	06/25/04	Removed support of the IP600 (S8100) PBX
Version L	04/22/05	Included NOTE on page 3, updated 8.12
Version M	09/02/05	Updated remote subscribers' configuration.
Version N	10/18/05	Updated to support ACM 3.0
Version O	01/26/06	Corrected parameters on various screens in Section 5.0.
Version P	04/11/06	Added: <ul style="list-style-type: none"> • MM 3.0 to support release section 2.0 • New MWI screen shot with Scheduled MWI updates parameter noted for MM3.0
Version Q	07/20/06	Added note (in red) to Consideration 8.1; also removed confusing portion of side note regarding Trunk Group members on page 9
Version M	02/01/07	Changed in Section 5.0: <ul style="list-style-type: none"> • Added note for multiple length extensions on system-parameter features screen. • Added sidebar adjacent signaling group form to explain NCA-TSC, CA-TSC, and Trunk Group for NCA-TSC Section 6.0 - Changed information clarifying what origin number entry should be on outgoing tab for PBX Type
Version N	10/18/05	Updated to support ACM 3.0
Version O	01/26/06	Corrected parameters on various screens in Section 5.0.
Version P	04/11/06	Added: <ul style="list-style-type: none"> • MM 3.0 to support release section 2.0 • New MWI screen shot with Scheduled MWI updates parameter noted for MM3.0
Version R	02/01/07	Changed in Section 5.0: <ul style="list-style-type: none"> • Added note for multiple length extensions on system-parameter features screen. • Added sidebar adjacent signaling group form to explain NCA-

		TSC, CA-TSC, and Trunk Group for NCA-TSC Section 6.0 - Changed information clarifying what origin number entry should be on outgoing tab for PBX Type
Version S	4/12/07	Added sidebar about CCRON needing to be turned on in system-parameters customer-options screen in addition to the one already placed adjacent the <i>display system-parameters coverage-forwarding</i> screen.
Version T	5/21/07	Added new screens for Avaya CM 4.0 and related private-numbering format; updated sidebars; changed Dial Access parameter in trunk group screen to N.
Version U	6/1/07	Updated Consideration 8.12
Version V	10/29/07	Updated info for station form
Version W	3/08	Updated Consideration 8.12; Updated Title to S87x0 / 8x00
Version X	05/05/08	Updated to support MM 4.0
Version Y	05/16/08	Removed Dialogic Driver release from Section 2.0
Version Z	06/11/08	Updated sidebar note in Section 5.0 detailing administration of trunk group and to leave out highest order timeslot/port.
Version AA	07/28/08	Added Consideration 8.14
Version AB	11/14/08	Added note about setting “Per Station CPN – Send Calling Number?” on Station Form to Yes in Section 5.0
Version AC	2/02/09	Updated to support MM 5.0; Limit Requests for MWI changed in Section 6.0; Added note about reviewing/changing qte.prm file at beginning of Section 6.0; Added Consideration 8.15
Version AD	02/12/09	Changed screen shot for VMSC and added sidebar in Section 6.0.
Version AE	03/18/09	Added Consideration 8.16 regarding Re-Route Request
Version AF	04/08/09	Added Consideration 8.17 regarding potential issues that may arise when using MM Caller Applications and Vectors
Version AG	07/09	Updated to support MM 5.1; changed wording in note on sidebar for VMSC in Section 6.0; Added note regarding collocated PBX/MM on page 1
Version AH	07/20/09	Updated Consideration 8.12 to reflect calling party number/name display on ringing phone and PBX requirements needed
Version AI	01/19/10	Added note indicator to titles of Section 2.0 and 3.1; added corresponding note in sidebar; removed word “supported” in same sections for MAS releases and PBX Software releases.

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