



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

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# **Application Notes for Product Support Solutions' FlexxGate™ Media Gateway with Avaya Communication Manager using T1/E1 ISDN/QSIG and CAS Interfaces – Issue 1.0**

### **Abstract**

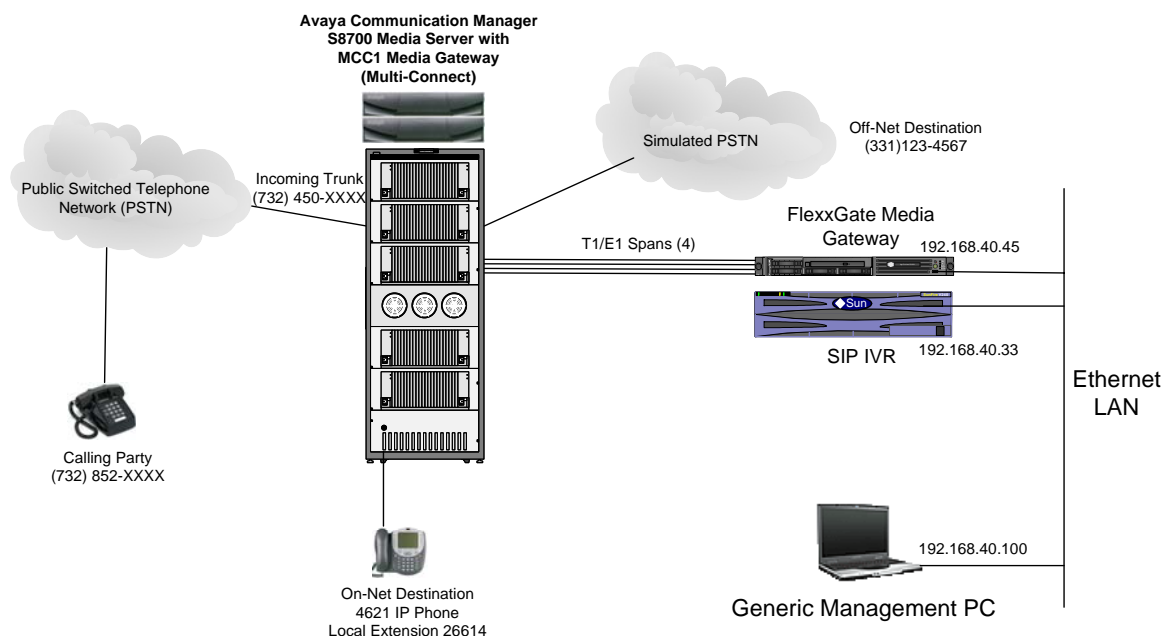
These Application Notes describe the configuration steps required for Product Support Solutions' FlexxGate™ media gateway connectivity solution to successfully interoperate with Avaya Communication Manager using T1/E1 ISDN/QSIG and CAS (Channel Associated Signaling) interfaces. Information in these Application Notes has been obtained through compliance testing and additional technical discussions. Testing was conducted via the Developer*Connection* Program at the Avaya Solution and Interoperability Test Lab.

# 1. Introduction

Product Support Solutions' FlexxGate™ media gateway is a TDM/VoIP connectivity solution. For this integration, the FlexxGate™ media gateway received inbound calls on T1 and E1 interfaces and converted the calls for delivery to an Interactive Voice Response (IVR) unit using SIP protocol. Subsequently, the SIP IVR transmitted SIP REFER messages and the FlexxGate™ used ISDN/QSIG Path Replacement, on spans configured for ISDN/QSIG, or flash-hook requests, on spans configured for CAS, to complete the transfers. Using these transfer methods results in the release of the bearer channels as soon as the transfers are complete.

In the scenarios that were tested, the FlexxGate™ media gateway interfaced with Avaya Communication Manager via a TN464 DS1 circuit pack. Four spans were used during the testing and were configured in the following manner: span 1 was configured for T1/CAS, span 2 was configured for E1/CAS, span 3 was configured for T1/ISDN/QSIG, and span 4 was configured for E1/ISDN/QSIG. The compliance testing focused on the ability of the FlexxGate™ to properly initiate transfers using either ISDN/QSIG Path Replacement or flash-hook signaling.

The FlexxGate™ media gateway utilizes a Digium TE412P telephony interface card that is physically inserted in the FlexxGate™ media gateway server. The Digium TE412P terminates up to 4 spans and, in the tested scenario, simultaneously terminated each of the four spans that were used. The other end of each span was terminated on a TN464 DS1 Interface circuit pack in an Avaya Communication Manager media gateway. In the tested scenarios, an Avaya MCC1 media gateway was used. Refer to **Figure 1**.



**Figure 1: Avaya DeveloperConnection Compliance Test Configuration**

## 2. Equipment and Software Validated

The following equipment and software were used in the tested configuration:

Equipment	Version Information
Avaya S8700 Media Servers	Avaya Communication Manager 4.0, load 730.5 w/update 13566
Avaya MCC1 Media Gateway	N/A
Avaya TN464F DS1 Interface	Vintage 6 (used for CAS spans)
Avaya TN464GP DS1 Interface	HW02 FW019 (used for ISDN/QSIG spans)
FlexxGate™ Media Gateway	V2.1.2
FlexxGate™ Server – HP DL360/G4 w/Digium TE412P	Red Hat Enterprise Linux ES Release 4 (Nahant Update 4)
Holly Voice Portal (HVP)	V4.2
Server for HVP Sun V240	Solaris 9

## 3. Configure Avaya Communication Manager<sup>1</sup>

### 3.1. Channel Associated Signaling (T1/CAS)

The procedures for configuring T1/CAS on Avaya Communication Manager include the following areas:

- Administer a DS1 Circuit Pack
- Administer Stations
- Administer a Hunt Group

#### 3.1.1. Administer a DS1 Circuit Pack (T1/CAS)

Note that the values described in this section are appropriate when adding a TN464 circuit pack with switches set to 24 channel and 120 ohms. Note that these switches must be set properly prior to inserting the circuit pack into the system. Refer to [2] for instructions on how to set the switches on TN464 circuit packs.

Administer a DS1 circuit pack to be used for connectivity to the FlexxGate™ media gateway. Use the “add ds1 1c12” command. The actual slot number may vary. In this case “1c12” is the slot number. Enter the following values for the specified fields, and retain the default values for all remaining fields. Enter a descriptive name, such as “FlexxGateSpan1”, in the **Name** field. Enter “ami-basic” in the **Line Coding** field, “d4” in the **Framing Mode** field, “robbed-bit” in the **Signaling Mode** field, and “mulaw” in the **Interface Companding** field. The appropriate entry for the **Line Compensation** field varies with the type of cable used. A value of 1 is

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<sup>1</sup> Note that while four spans were configured during integration testing, the number of spans and the method in which the spans will be configured on Avaya Communication Manager and FlexxGate™ will depend upon customer requirements.

appropriate for 22-gauge ABAM cable terminated on a DSX-1 cross-connect and a wire length between 0 and 133 feet<sup>2</sup>. Submit these changes.

```
add ds1 1c12

                                DS1 CIRCUIT PACK

      Location: 01C12              Name: FlexxGateSpan1
      Bit Rate: 1.544             Line Coding: ami-basic
Line Compensation: 1             Framing Mode: d4
      Signaling Mode: robbed-bit

Interface Companding: mulaw
      Idle Code: 11111111

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

### 3.1.2. Administer Stations

Add a station for each of the 24 channels using the “add station n” command, where “n” is an available extension number. In the following example, extension “22201” was used for channel 1. Enter a descriptive name in the **Name** field. Set the **Type** field to “DS1FD”. For the **Port** field, note that “01C1201” refers to the first channel of the DS1 circuit pack as administered in **Section 3.1.1**.

Repeat this procedure for the rest of the channels. The “duplicate station” command can be used to expedite the process. Submit these changes.

```
add station 22201                                Page 1 of 4

                                STATION

Extension: 22201      Lock Messages? n          BCC: 0
      Type: DS1FD      Security Code:           TN: 1
      Port: 01C1201    Coverage Path 1:         COR: 1
      Name: FlexxGate  Coverage Path 2:         COS: 1
                                Hunt-to Station: Tests? y

STATION OPTIONS

      Loss Group: 4
Off Premises Station? y
      R Balance Network? n

      Survivable COR: internal
Survivable Trunk Dest? y
```

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<sup>2</sup> See [1].

### 3.1.3. Administer a DDC Hunt Group

Add a hunt group using the “add hunt-group n” command, where “n” is an available hunt group number. Enter a descriptive name in the **Group Name** field. Set the **Group Type** field to “ddc”. A Direct Department Calling (DDC) type hunt group selects the first available extension (in the administered sequence) when answering a new call. Enter an available extension number in the **Group Extension** field. In the example below, the extension number assigned in the **Group Extension** field was 22200 and was used to allow calls to be answered by any one of the channels in the group. Note that multiple hunt-groups can be defined and channels assigned to each hunt group to allow for different IVR applications to be run on different groups of channels as required by the implementation. In addition, channels can be dialed individually using their individual extension numbers as assigned in **Section 3.1.2**.

add hunt-group 22		Page 1 of 60
HUNT GROUP		
Group Number: 22	ACD? n	
Group Name: FlexxGateSpan1	Queue? n	
Group Extension: 22200	Vector? n	
Group Type: ddc	Coverage Path:	
TN: 1	Night Service Destination:	
COR: 1	MM Early Answer? n	
Security Code:	Local Agent Preference? n	
ISDN/SIP Caller Display:		

Page Down to the **Group Member Assignments** page and enter the extensions of each of the channels that were added in **Section 3.1.2**. Submit these changes.

add hunt-group 22		Page 3 of 60
HUNT GROUP		
Group Number: 22	Group Extension: 22200	Group Type: ddc
Member Range Allowed: 1 - 1500	Administered Members (min/max): 1 /24	
Total Administered Members: 24		
GROUP MEMBER ASSIGNMENTS		
<b>Ext</b>	<b>Name(19 characters)</b>	<b>Ext</b>
1: 22201	FlexxGate	14: 22214
2: 22202	FlexxGate	15: 22215
3: 22203	FlexxGate	16: 22216
4: 22204	FlexxGate	17: 22217
5: 22205	FlexxGate	18: 22218
6: 22206	FlexxGate	19: 22219
7: 22207	FlexxGate	20: 22220
8: 22208	FlexxGate	21: 22221
9: 22209	FlexxGate	22: 22222
10: 22210	FlexxGate	23: 22223
11: 22211	FlexxGate	24: 22224
12: 22212	FlexxGate	25:
13: 22213	FlexxGate	26:
At End of Member List		

## 3.2. Channel Associated Signaling (E1/CAS)

The procedures for configuring E1/CAS on Avaya Communication Manager include the following areas:

- Administer a DS1 Circuit Pack
- Administer Stations
- Administer a Hunt Group

### 3.2.1. Administer a DS1 Circuit Pack (E1/CAS)

Note that the values described in this section are appropriate when adding a TN464 circuit pack with switches set to 32 channels and 120 ohms. Note that these switches must be set properly prior to inserting the circuit pack into the system. Refer to [2] for instructions on how to set the switches on TN464 circuit packs.

Administer a DS1 circuit pack to be used for connectivity to the FlexxGate™ media gateway. Use the “add ds1 1c13” command. The actual slot number may vary. In this case “1c13” is the slot number. Enter the following values for the specified fields, and retain the default values for all remaining fields. Enter a descriptive name, such as “FlexxGateSpan2”, in the **Name** field. Enter “ami-basic” in the **Line Coding** field, “CAS” in the **Signaling Mode** field, and “alaw” in the **Interface Companding** field. Submit these changes.

```
add ds1 1c13
```

DS1 CIRCUIT PACK	
Location: 01C13	Name: FlexxGateSpan2
Bit Rate: 2.048	Line Coding: ami-basic
Signaling Mode: CAS	
Interconnect: pbx	Country Protocol: 1
Interface Companding: alaw	CRC? n
Idle Code: 11111111	
Slip Detection? n	Near-end CSU Type: other

### 3.2.2. Administer Stations

Add a station for each of the 30 channels using the “add station n” command, where “n” is an available extension number. In the following example, extension “22226” was used for channel 1. Enter a descriptive name in the **Name** field. Set the **Type** field to “DS1FD”. For the **Port** field, note that “01C1301” refers to the first channel of the DS1 circuit pack as administered in **Section 3.2.1**.

Repeat this procedure for the rest of the channels. The “duplicate station” command can be used to expedite the process. Submit these changes.

add station 22226		Page 1 of 4
STATION		
Extension: 22226	Lock Messages? n	BCC: 0
<b>Type: DS1FD</b>	Security Code:	TN: 1
<b>Port: 01C1301</b>	Coverage Path 1:	COR: 1
<b>Name: FlexxGate</b>	Coverage Path 2:	COS: 1
	Hunt-to Station:	Tests? y
STATION OPTIONS		
Time of Day Lock Table:		
Loss Group: 4		
Off Premises Station? y		
R Balance Network? n		
Survivable COR: internal		
Survivable Trunk Dest? y		

### 3.2.3. Administer a DDC Hunt Group

Add a hunt group using the “add hunt-group n” command, where “n” is an available hunt group number. Enter a descriptive name in the **Group Name** field. Set the **Group Type** field to “ddc”. A Direct Department Calling (DDC) type hunt group selects the first available extension (in the administered sequence) when answering a new call. Enter an available extension number in the **Group Extension** field. In the example below, the extension number assigned in the **Group Extension** field was 22225 and was used to allow calls to be answered by any one of the channels in the group. Note that multiple hunt-groups can be defined and channels assigned to each hunt group to allow for different IVR applications to be run on different groups of channels as required by the implementation. In addition, channels can be dialed individually using the individual extension numbers as assigned in **Section 3.2.2**.

add hunt-group 23		Page 1 of 60
HUNT GROUP		
Group Number: 23	ACD? n	
<b>Group Name: FlexxGateSpan2</b>	Queue? n	
<b>Group Extension: 22225</b>	Vector? n	
<b>Group Type: ddc</b>	Coverage Path:	
TN: 1	Night Service Destination:	
COR: 1	MM Early Answer? n	
Security Code:	Local Agent Preference? n	
ISDN/SIP Caller Display:		

Page Down to the **Group Member Assignments** page and enter the extensions of each of the channels that were added in **Section 3.2.2**.

```

display hunt-group 23                                     Page 3 of 60
                                     HUNT GROUP
      Group Number: 23   Group Extension: 22225   Group Type: ddc
Member Range Allowed: 1 - 1500   Administered Members (min/max): 1 /30
                                     Total Administered Members: 30
GROUP MEMBER ASSIGNMENTS
      Ext      Name(19 characters)      Ext      Name(19 characters)
1: 22226      FlexxGate      14: 22239      FlexxGate
2: 22227      FlexxGate      15: 22240      FlexxGate
3: 22228      FlexxGate      16: 22241      FlexxGate
4: 22229      FlexxGate      17: 22242      FlexxGate
5: 22230      FlexxGate      18: 22243      FlexxGate
6: 22231      FlexxGate      19: 22244      FlexxGate
7: 22232      FlexxGate      20: 22245      FlexxGate
8: 22233      FlexxGate      21: 22246      FlexxGate
9: 22234      FlexxGate      22: 22247      FlexxGate
10: 22235      FlexxGate      23: 22248      FlexxGate
11: 22236      FlexxGate      24: 22249      FlexxGate
12: 22237      FlexxGate      25: 22250      FlexxGate
13: 22238      FlexxGate      26: 22251      FlexxGate

More Members Exist
  
```

Page Down to the next page and enter the remaining extensions of each of the channels that were added in **Section 3.2.2**. Submit these changes.

```

add hunt-group 23                                     Page 4 of 60
                                     HUNT GROUP
      Group Number: 23   Group Extension: 22225   Group Type: ddc
Member Range Allowed: 1 - 1500   Administered Members (min/max): 1 /30
                                     Total Administered Members: 30
GROUP MEMBER ASSIGNMENTS
      Ext      Name(19 characters)      Ext      Name(19 characters)
27: 22252      FlexxGate      40:
28: 22253      FlexxGate      41:
29: 22254      FlexxGate      42:
30: 22255      FlexxGate      43:
31:              44:
32:              45:
33:              46:
34:              47:
35:              48:
36:              49:
37:              50:
38:              51:
39:              52:

At End of Member List
  
```



### 3.3. T1/ISDN/QSIG

The procedures for configuring the T1/ISDN/QSIG trunk on Avaya Communication Manager include the following areas:

- Verify Avaya Communication Manager License
- Administer DS1 Circuit Pack
- Administer ISDN Trunk Group
- Administer ISDN Signaling Group
- Administer ISDN Trunk Group Members
- Administer System Features Form
- Administer Private Numbering Selection Numbers
- Administer Private Network Routing

#### 3.3.1. Verify Avaya Communication Manager License

Log into the System Access Terminal (SAT) to verify that the Avaya Communication Manager license has proper permissions for features illustrated in these Application Notes. Use the “display system-parameters customer-options” command to verify that the **ISDN-PRI** option is set to “y” on **Page 4**, as shown below. A system license file controls the settings on the customer-options form.

```
display system-parameters customer-options                               Page 4 of 11
                                OPTIONAL FEATURES

Emergency Access to Attendant? y                                     IP Stations? y
  Enable 'dadmin' Login? y
  Enhanced Conferencing? y                                           ISDN Feature Plus? y
    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? y                                           Malicious Call Trace? y
  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? y                                       Multifrequency Signaling? y
  Global Call Classification? n   Multimedia Call Handling (Basic)? y
  Hospitality (Basic)? y         Multimedia Call Handling (Enhanced)? y
Hospitality (G3V3 Enhancements)? y
  IP Trunks? y

IP Attendant Consoles? n
(NOTE: You must logoff & login to effect the permission changes.)
```

Page Down to Page 5 and verify that the **Private Networking** and **Uniform Dialing Plan** features are set to “y” as shown below.

```
display system-parameters customer-options                                Page 5 of 11
                                OPTIONAL FEATURES

                                Multinational Locations? n
Multiple Level Precedence & Preemption? n                                Station and Trunk MSP? y
                                Multiple Locations? n                                Station as Virtual Extension? n

                                Personal Station Access (PSA)? y
                                Posted Messages? n                                System Management Data Transfer? n
                                PNC Duplication? n                                Tenant Partitioning? n
                                Port Network Support? y                                Terminal Trans. Init. (TTI)? y
                                Processor and System MSP? y                                Time of Day Routing? y
                                Private Networking? y                                Uniform Dialing Plan? y
                                Processor Ethernet? y                                Usage Allocation Enhancements? y
                                Remote Office? n                                TN2501 VAL Maximum Capacity? y
                                Restrict Call Forward Off Net? y                                Wideband Switching? y
                                Secondary Data Module? y                                Wireless? n

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

Page Down to Page 8 and verify that the **Basic Call Setup**, **Basic Supplementary Services**, and **Supplementary Services with Rerouting** features are set to “y” as shown below.

```
display system-parameters customer-options                                Page 8 of 11
                                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? n
                                Value-Added (VALU)? n

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

### 3.3.2. Administer DS1 Circuit Pack (T1/QSIG)

Note that the values described in this section are appropriate when adding a TN464 circuit pack with switches set to 24 channels and 120 ohms. Refer to [2] for instructions on how to set the switches on TN464 circuit packs.

Administer a DS1 circuit pack to be used for connectivity to the FlexxGate™ media gateway. Use the “add ds1 1b17” command. The actual slot number may vary. In this case “1b17” is the slot number. Enter the following values for the specified fields, and retain the default values for all remaining fields. Enter a descriptive name, such as “FlexxGateSpan3” in the **Name** field, “b8zs” in the **Line Coding** field, “esf” in the **Framing Mode** field, “isdn-pri” in the **Signaling Mode** field, “pbx” in the **Connect** field, “peer-master”<sup>3</sup> in the **Interface** field, “Q-SIG” in the **Peer Protocol** field, and “a” in the **Side** field. Submit these changes.

add ds1 1b17		Page 1 of 2
DS1 CIRCUIT PACK		
Location: 01B17	Name: FlexxGateSpan3	
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: mulaw	CRC? n	
Idle Code: 11111111		
DCP/Analog Bearer Capability: 3.1kHz		
T303 Timer(sec): 4		
Slip Detection? n		
Near-end CSU Type: other		
T303 Timer(sec): 4		
Slip Detection? n		
Near-end CSU Type: other		
Block Progress Indicator? n		

### 3.3.3. Administer ISDN Trunk Group

Administer an ISDN trunk group. Use the “add trunk-group n” command, where “n” is an available trunk group number. In the test configuration, trunk group 200 was used. Enter the following values for the specified fields, and retain the default values for all remaining fields. Enter “isdn” in the **Group Type** field, a descriptive name, such as “FlexxGateSpan3” in the

<sup>3</sup> A setting of “peer-master” was used in the tested configuration. When the **Interface** field is set to “peer-master”, the FlexxGate media gateway must be set to slave. This setting on the FlexxGate will be provisioned by Product Support Solutions, Inc. Avaya Communication Manager can also work in the reverse configuration.

**Group Name** field, an available trunk access code in the **TAC** field, “two-way” in the **Direction** field, and “tie” in the **Service Type** field.

add trunk-group 200		Page 1 of 21	
TRUNK GROUP			
Group Number: 200	<b>Group Type: isdn</b>	CDR Reports: y	
<b>Group Name: FlexxGateSpan3</b>	COR: 1	TN: 1	<b>TAC: 1200</b>
<b>Direction: two-way</b>	Outgoing Display? n	Carrier Medium: PRI/BRI	
Dial Access? n	Busy Threshold: 255	Night Service:	
Queue Length: 0			
<b>Service Type: tie</b>	Auth Code? n	TestCall ITC: rest	
	Far End Test Line No:		
TestCall BCC: 4			

Page Down to Page 2 and enter the following values for the specified fields, and retain the default values for all remaining fields. Enter “b” in the **Supplementary Service Protocol** field and “ascend”<sup>4</sup> in the **Trunk Hunt** field.

add trunk-group 200		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		
<b>Supplementary Service Protocol: b</b>	Digit Handling (in/out): enbloc/enbloc		
<b>Trunk Hunt: ascend</b>			
		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			
Administer Timers? n			

<sup>4</sup> In the tested configuration, the **Trunk Hunt** field was set to “ascend”. This setting controls how Avaya Communication Manager performs a trunk hunt when searching for available channels within a facility in an ISDN trunk group. A value of “ascend” enables a linear trunk hunt search from the lowest to highest numbered channels. The FlexxGate media gateway was set to choose channels from highest to lowest to help avoid the possibility of glare conditions. Avaya Communication Manager can also work in the reverse configuration.

Page Down to Page 3 and enter the following values for the specified fields, and retain the default values for all remaining fields. Enter “y” in the **Send Name** field, “y” in the **Send Calling Number** field, “unk-pvt” in the **Format** field and “y” in the **Send Connected Number** field.

add trunk-group 200		Page 3 of 21
TRUNK FEATURES		
ACA Assignment? n	Measured: none	Wideband Support? n
	Internal Alert? n	Maintenance Tests? y
	Data Restriction? n	NCA-TSC Trunk Member:
	<b>Send Name: y</b>	<b>Send Calling Number: y</b>
Used for DCS? n	Hop Dgt? n	Send EMU Visitor CPN? n
Suppress # Outpulsing? n	<b>Format: unk-pvt</b>	
Outgoing Channel ID Encoding: preferred	UII IE Treatment: service-provider	
	Replace Restricted Numbers? n	
	Replace Unavailable Numbers? n	
	<b>Send Connected Number: y</b>	
	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		
Show ANSWERED BY on Display? y		
	Network (Japan) Needs Connect Before Disconnect? n	

Page Down to Page 4 and ensure the **Path Replacement** field is set to “y”. Submit these changes.

add trunk-group 200		Page 4 of 21
QSIG TRUNK GROUP OPTIONS		
TSC Method for Auto Callback: drop-if-possible		
Diversion by Reroute? y		
<b>Path Replacement? y</b>		
Path Replacement with Retention? n		
Path Replacement Method: better-route		
SBS? n		
Display Forwarding Party Name? y		
Character Set for QSIG Name: eurofont		

### 3.3.4. Administer ISDN Signaling Group

Administer an ISDN signaling group for the DS1 circuit pack. Use the “add signaling-group n” command, where “n” is an available signaling group number. For the **Primary D-Channel** field, enter the slot number for the DS1 circuit pack as administered in **Section 3.3.2** and port “24”. Note that TSCs were not used during this integration testing and, therefore, the TSC related fields are not relevant for this integration.

Maintain the default values for the remaining fields, and submit these changes.

```
add signaling-group 200
```

#### SIGNALING GROUP

```

Group Number: 200          Group Type: isdn-pri
Associated Signaling? y    Max number of NCA TSC: 0
Primary D-Channel: 01B1724 Max number of CA TSC: 0
Trunk Group for NCA TSC:
Trunk Group for Channel Selection:
TSC Supplementary Service Protocol: a

```

### 3.3.5. Administer ISDN Trunk Group Members

Use the “change trunk-group n” command, where “n” is the trunk group number that was administered in **Section 3.3.3**. Navigate to the **GROUP MEMBER ASSIGNMENTS** page of the **TRUNK GROUP** screen, and enter the ports of the DS1 circuit pack into the corresponding **Port** fields. The corresponding **Code** and **Sfx** fields will be populated automatically.

Repeat this procedure for the desired number of trunk group members. The number of members assigned should match the total number of active channels between Avaya Communication Manager and the FlexxGate™ media gateway on this span. This includes the channels that will be used by FlexxGate™ to transfer calls away from the FlexxGate™ back to the Avaya system.

```
change trunk-group 200
```

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#### TRUNK GROUP

Administered Members (min/max): 1/23

#### GROUP MEMBER ASSIGNMENTS

Total Administered Members: 23

	Port	Code	Sfx	Name	Night	Sig	Grp
1:	01B1701	TN464	G				200
2:	01B1702	TN464	G				200
3:	01B1703	TN464	G				200
4:	01B1704	TN464	G				200
5:	01B1705	TN464	G				200
6:	01B1706	TN464	G				200
7:	01B1707	TN464	G				200
8:	01B1708	TN464	G				200
9:	01B1709	TN464	G				200
10:	01B1710	TN464	G				200
11:	01B1711	TN464	G				200
12:	01B1712	TN464	G				200
13:	01B1713	TN464	G				200
14:	01B1714	TN464	G				200
15:	01B1715	TN464	G				200

Page Down to the next page and enter the remaining ports. Note that port 24 is reserved for the signaling channel and is not added as a group member. Submit these changes.

change trunk-group 200					Page 6 of 21
TRUNK GROUP					
Administered Members (min/max):					1/23
GROUP MEMBER ASSIGNMENTS					Total Administered Members: 23
	Port	Code	Sfx	Name	Night
16:	01B1716	TN464	G		Sig Grp
17:	01B1717	TN464	G		200
18:	01B1718	TN464	G		200
19:	01B1719	TN464	G		200
20:	01B1720	TN464	G		200
21:	01B1721	TN464	G		200
22:	01B1722	TN464	G		200
23:	01B1723	TN464	G		200
24:					
25:					
26:					
27:					
28:					
29:					
30:					

### 3.3.6. Administer System Features Form

Use the “change system-parameters features” command and enter “all” in the **Trunk-to-Trunk Transfer** field.

display system-parameters features					Page 1 of 17
FEATURE-RELATED SYSTEM PARAMETERS					
Self Station Display Enabled? n					
<b>Trunk-to-Trunk Transfer: all</b>					
Automatic Callback - No Answer Timeout Interval (rings): 9					
Call Park Timeout Interval (minutes): 1					
Off-Premises Tone Detect Timeout Interval (seconds): 10					
AAR/ARS Dial Tone Required? y					
Music/Tone on Hold: music Type: port 02A0715					
Music (or Silence) on Transferred Trunk Calls? all					
DID/Tie/ISDN/SIP Intercept Treatment: attd					
Internal Auto-Answer of AttD-Extended/Transferred Calls: transferred					
Automatic Circuit Assurance (ACA) Enabled? n					
Abbreviated Dial Programming by Assigned Lists? n					
Auto Abbreviated/Delayed Transition Interval (rings): 2					
Protocol for Caller ID Analog Terminals: Bellcore					
Display Calling Number for Room to Room Caller ID Calls? n					

Page Down to page 8 and enter an available extension number in the **QSIG PATH REPLACEMENT EXTENSION** field. In this case, extension “22770” was used. Enter “0” in the **PARAMETERS FOR CREATING QSIG SELECTION NUMBERS NETWORK LEVEL** field. Submit these changes.

display system-parameters features		Page 8 of 17
FEATURE-RELATED SYSTEM PARAMETERS		
ISDN PARAMETERS		
Send Non-ISDN Trunk Group Name as Connected Name? n	<b>PARAMETERS FOR CREATING QSIG SELECTION NUMBERS</b>	
Display Connected Name/Number for ISDN DCS Calls? n	<b>Network Level: 0</b>	
Send ISDN Trunk Group Name on Tandem Calls? n	Level 2 Code:	
	Level 1 Code:	
QSIG/ETSI TSC Extension: 22699		
MWI - Number of Digits Per Voice Mail Subscriber: 5		
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? n		
Path Replacement with Measurements? y		
<b>QSIG Path Replacement Extension: 22700</b>		
Path Replace While in Queue/Vectoring? n		

### 3.3.7. Administer Private Numbering

Use the “change private-numbering 0” command to access the **NUMBERING – PRIVATE FORMAT** table. In the tested configuration, FlexxGate™ transferred calls to on-pbx extensions that were 5 digits in length and started with digit 2. Therefore, a new entry was created as follows: “5” was inserted in the **EXT LEN** column, “2” in the **EXT CODE** column and “5” in the **TOTAL LEN** column. This explicitly specifies to Avaya Communication Manager that the extensions in the range of 20000 through 29999 are local endpoints that pertain to the local system. This configuration is necessary for Avaya Communication Manager to initiate QSIG Path Replacement Proposals. Note that the extension number assigned in the **QSIG Path Replacement Extension** field in **Section 3.3.6** must also be included in this range. The entry in the **QSIG Path Replacement Extension** field in **Section 3.3.6** is used so that QSIG Path Replacement can still function when the transfer destination is not a local endpoint. Submit these changes.

change private-numbering 0					Page 1 of 2
NUMBERING – PRIVATE FORMAT					
Ext Len	Ext Code	Trk Grp(s)	Private Prefix	Total Len	
0	attd			0	Total Administered: 2
5	2			5	Maximum Entries: 540



### 3.3.8. Administer Private Routing to the T1/ISDN/QSIG Span

#### 3.3.8.1 Administer the AAR Feature Access Code<sup>5</sup>

Use the “change feature-access-code” command to access the **FEATURE ACCESS CODE** form. Enter “8” in the **Auto Alternate Routing (AAR) Access Code** field. Submit this change.

change feature-access-codes	Page 1 of 8
FEATURE ACCESS CODE (FAC)	
Abbreviated Dialing List1 Access Code:	*70
Abbreviated Dialing List2 Access Code:	
Abbreviated Dialing List3 Access Code:	
Abbreviated Dial - Prgm Group List Access Code:	
Announcement Access Code:	*45
Answer Back Access Code:	*31
Auto Alternate Routing (AAR) Access Code: 8	
Auto Route Selection (ARS) - Access Code 1:	9
Access Code 2:	
Automatic Callback Activation:	Deactivation:
Call Forwarding Activation Busy/DA:	All: #444
Deactivation:	#445
Call Forwarding Enhanced Status:	Act:
Deactivation:	
Call Park Access Code:	*10
Call Pickup Access Code:	*22
CAS Remote Hold/Answer Hold-Unhold Access Code:	*04
CDR Account Code Access Code:	*55
Change COR Access Code:	
Change Coverage Access Code:	*83
Contact Closure Open Code:	
Close Code:	

#### 3.3.8.2 Administer the Uniform Dialing Plan

Use the “change uniform-dialplan 2” command to access the **UNIFORM DIAL PLAN TABLE**. In the tested configuration, an entry was created to convert extension 22291 into a private network number of 2262291 in the following manner: enter “22291” in the **Matching Pattern** column, “5” in the **Len** column, “1” in the **Del** column, “226” in the **Insert Digits** column, “aar” in the **Net** column, and “n” in the **Conv** column. Submit these changes.

change uniform-dialplan 2					Page 1 of 2	
UNIFORM DIAL PLAN TABLE						
					Percent Full: 0	
Matching			Insert			Node
Pattern	Len	Del	Digits	Net	Conv	Num
22291	5	1	226	aar	n	

<sup>5</sup> Note that in the following sections, AAR Feature Access Code “8”, extension “22291”, AAR dialed string “2262291”, route pattern “201”, trunk group “200”, vector directory number “2201”, and vector “201”, were used in the tested configuration. However, these values will differ from site to site.

### 3.3.8.3 Administer the AAR Digit Analysis Table

Use the “change aar analysis 2262291” command to access the **AAR DIGIT ANALYSIS TABLE**. In the tested configuration, an entry was created to route private network calls to 2262291 using route pattern 201 in the following manner: enter “2262291” in the **Dialed String** column, “7” in the **Total Min** column, “7” in the **Total Max** column, “201” in the **Route Pattern** column, “aar” in the **Call Type**. Other values were left at the defaults. Submit these changes.

change aar analysis 2262291						Page	1 of	2
						AAR DIGIT ANALYSIS TABLE		
						Percent Full:		
						7		
<b>Dialed String</b>	<b>Total Min</b>	<b>Total Max</b>	<b>Route Pattern</b>	<b>Call Type</b>	<b>Node Num</b>	<b>ANI</b>	<b>Reqd</b>	
2262291	7	7	201	aar		n		

### 3.3.8.4 Administer the Route Pattern

Use the “change route-pattern 201” command to access route pattern 201. In the tested configuration, an entry was created to route calls to trunk 200 in the following manner: enter “200” in the **Grp No** column, “0” in the **FRL** column, “7” in the **No. Del Dgts** column, and “2201” in the **Inserted Digits**. The inserted digits “2201” represent the Dialed Number Identification Service (DNIS) digits to be passed to the FlexxGate™ during the initial call setup. Other values were left at the defaults. Submit these changes.

change route-pattern 201												Page	1 of	3
												Pattern Number: 201 Pattern Name: FlexGate2201		
												SCCAN? n Secure SIP? n		
Grp	FRL	NPA	Pfx	Hop	Toll	No.	Inserted					DCS/	IXC	
No			Mrk	Lmt	List	Del	Digits					QSIG		
							Dgts					Intw		
1:	200	0				7	2201					n	user	
2:												n	user	
3:												n	user	
4:												n	user	
5:												n	user	
6:												n	user	
BCC	VALUE	TSC	CA-TSC				ITC	BCIE	Service/Feature	PARM	No.	Numbering	LAR	
0	1	2	M	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		

### 3.3.8.5 Administer a Vector Directory Number

Use the “add vdn n” command, where “n” is an available extension number. In the tested configuration, a vector directory number was used as the initial pilot number to route calls to the FlexxGate™. Enter a description name such as “FlexxGateVDN2201” in the **Name** field. Enter

“201” in the **Vector Number** field. Other values were left at the defaults. Submit these changes.

add vdn 2201	Page 1 of 3
VECTOR DIRECTORY NUMBER	
Extension: 2201	
Name*: FlexxGateVDN2201	
Vector Number: 201	
Meet-me Conferencing? n	
Allow VDN Override? n	
COR: 1	
TN*: 1	
Measured: none	
VDN of Origin Annc. Extension*:	
1st Skill*:	
2nd Skill*:	
3rd Skill*:	
* Follows VDN Override Rules	

### 3.3.8.6 Administer a Vector

Use the “change vector 201” command. In the tested configuration, a vector was used to route calls to the FlexxGate™. Enter a description name such as “FlexxGateVec201” in the **Name** field. For vector Step 1, enter “wait-time” and populate the rest of the values as shown below. For vector Step 2, enter “route-to” “number” and enter “82262291” as shown below. The number “82262291” corresponds to the AAR Feature Access Code configured in **Section 3.3.8.1** and the AAR Routing Table Entry configured in **Section 3.3.8.3**. For vector Step 3, enter “busy”.<sup>6</sup> Submit these changes.

change vector 201	Page 1 of 6
CALL VECTOR	
Number: 201	Name: FlexxGateVec201
Multimedia? n	Meet-me Conf? n Lock? n
Basic? y	EAS? y G3V4 Enhanced? y ANI/II-Digits? y ASAI Routing? y
Prompting? y	LAI? y G3V4 Adv Route? y CINFO? y BSR? y Holidays? y
Variables? y	3.0 Enhanced? y
01 wait-time	2 secs hearing ringback
02 route-to	number 82262291 with cov n if unconditionally
03 busy	
04	
05	

## 3.4. E1/ISDN/QSIG

Since the procedures for configuring the E1/ISDN/QSIG trunk are very similar to the procedures for configuring the T1/ISDN/QSIG trunk in **Section 3.3**, the reader will be referred to the

<sup>6</sup> Additional call vectoring steps can be added to take advantage of conditional routing per customer requirements.

appropriate sub-sections under T1/ISDN/QSIG configuration. Only those sections that differ will be highlighted in this section. The procedures include the following areas:

- Verify Avaya Communication Manager License
- Administer DS1 Circuit Pack
- Administer ISDN Trunk Group
- Administer ISDN Signaling Group
- Administer ISDN Trunk Group Members
- Administer System Features Form
- Administer Private Numbering Selection Numbers
- Administer Private Network Routing

### 3.4.1. Verify Avaya Communication Manager License

See Section 3.3.1.

### 3.4.2. Administer DS1 Circuit Pack (E1/ISDN/QSIG)

Note that the values described in this section are appropriate when adding a TN464 circuit pack with switches set to 32 channels and 120 ohms. Refer to [2] for instructions on how to set the switches on TN464 circuit packs.

Administer a DS1 circuit pack to be used for connectivity to the FlexxGate™ media gateway. Use the “add ds1 1b16” command. The actual slot number may vary. In this case “1b16” is the slot number. Enter the following values for the specified fields, and retain the default values for all remaining fields. Enter a descriptive name, such as “FlexxGateSpan4”, in the **Name** field. Enter “hdb3” in the **Line Coding** field, “isdn-pri” in the **Signaling Mode** field, “pbx” in the **Connect** field, “peer-master” in the **Interface** field, “Q-SIG” in the **Peer Protocol** field, “alaw” in the **Interface Companding** field, and “a” in the **Side** field. Submit these changes.

```
add ds1 1b16

                                DS1 CIRCUIT PACK

      Location: 01B16                      Name: FlexxGateSpan4
      Bit Rate: 2.048                    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
```

### 3.4.3. Administer ISDN Trunk Group

Administer an ISDN trunk group. Use the “add trunk-group n” command, where “n” is an available trunk group number. In the test configuration, trunk group 210 was used. Enter the following values for the specified fields, and retain the default values for all remaining fields. Enter “isdn” in the **Group Type** field, a descriptive name, such as “FlexxGateSpan4”, in the **Group Name** field, an available trunk access code in the **TAC** field, “two-way” in the **Direction** field, and “tie” in the **Service Type** field.

add trunk-group 210		Page 1 of 21	
TRUNK GROUP			
Group Number: 210	<b>Group Type: isdn</b>	CDR Reports: y	
<b>Group Name: FlexxGateSpan4</b>	COR: 1	TN: 1	<b>TAC: 1210</b>
<b>Direction: two-way</b>	Outgoing Display? n	Carrier Medium: PRI/BRI	
Dial Access? n	Busy Threshold: 255	Night Service:	
Queue Length: 0			
<b>Service Type: tie</b>	Auth Code? n	TestCall ITC: rest	
	Far End Test Line No:		
TestCall BCC: 4			

Page Down to Page 2 and enter the following values for the specified fields, and retain the default values for all remaining fields. Enter “b” in the **Supplementary Service Protocol** field and “ascend”<sup>7</sup> in the **Trunk Hunt** field.

add trunk-group 210		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		
<b>Supplementary Service Protocol: b</b>	Digit Handling (in/out): enbloc/enbloc		
<b>Trunk Hunt: ascend</b>			
		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			
Administer Timers? n			

<sup>7</sup> In the tested configuration, the **Trunk Hunt** field was set to “ascend”. This setting controls how Avaya Communication Manager performs a trunk hunt when searching for available channels within a facility in an ISDN trunk group. A value of “ascend” enables a linear trunk hunt search from the lowest to highest numbered channels. The FlexxGate media gateway was set to choose channels from highest to lowest to help avoid the possibility of glare conditions. Avaya Communication Manager can also work in the reverse configuration.

Page Down to Page 3 and enter the following values for the specified fields, and retain the default values for all remaining fields. Enter “y” in the **Send Name** field, “y” in the **Send Calling Number** field, “unk-pvt” in the **Format** field, and “y” in the **Send Connected Number** field.

add trunk-group 210		Page 3 of 21
TRUNK FEATURES		
ACA Assignment? n	Measured: none	Wideband Support? n
	Internal Alert? n	Maintenance Tests? y
	Data Restriction? n	NCA-TSC Trunk Member:
	<b>Send Name: y</b>	<b>Send Calling Number: y</b>
Used for DCS? n	Hop Dgt? n	Send EMU Visitor CPN? n
Suppress # Outpulsing? n	<b>Format: unk-pvt</b>	
Outgoing Channel ID Encoding: preferred	UII IE Treatment: service-provider	
	Replace Restricted Numbers? n	
	Replace Unavailable Numbers? n	
	<b>Send Connected Number: y</b>	
	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		
Show ANSWERED BY on Display? y		
	Network (Japan) Needs Connect Before Disconnect? n	

Page Down to Page 4 and ensure the **Path Replacement** field is set to “y”. Submit these changes.

add trunk-group 210		Page 4 of 21
QSIG TRUNK GROUP OPTIONS		
TSC Method for Auto Callback: drop-if-possible		
Diversion by Reroute? y		
<b>Path Replacement? y</b>		
Path Replacement with Retention? n		
Path Replacement Method: better-route		
SBS? n		
Display Forwarding Party Name? y		
Character Set for QSIG Name: eurofont		

### 3.4.4. Administer ISDN Signaling Group

Administer an ISDN signaling group for the DS1 circuit pack. Use the “add signaling-group n” command, where “n” is an available signaling group number. For the **Primary D-Channel** field, enter the slot number for the DS1 circuit pack as administered in **Section 3.4.2** and port “16”. Note that TSCs were not used during this integration testing and, therefore, the TSC related fields are not relevant for this integration.

Maintain the default values for the remaining fields, and submit these changes.

```
add signaling-group 210
```

#### SIGNALING GROUP

```

Group Number: 210          Group Type: isdn-pri
Associated Signaling? y    Max number of NCA TSC: 0
Primary D-Channel: 01B1616 Max number of CA TSC: 0
Trunk Group for NCA TSC:
Trunk Group for Channel Selection:
TSC Supplementary Service Protocol: a

```

### 3.4.5. Administer ISDN Trunk Group Members

Use the “change trunk-group n” command, where “n” is the trunk group number that was administered in **Section 3.4.3**. Navigate to the **GROUP MEMBER ASSIGNMENTS** page of the **TRUNK GROUP** screen, and enter the ports of the DS1 circuit pack into the corresponding **Port** fields. The corresponding **Code** and **Sfx** fields will be populated automatically.

Repeat this procedure for the desired number of trunk group members. The number of members assigned should match the total number of active channels between Avaya Communication Manager and the FlexxGate™ media gateway on this span. This includes the channels that will be used by FlexxGate™ to transfer calls away from the FlexxGate™ back to the Avaya system.

```
change trunk-group 210
```

Page 5 of 21

#### TRUNK GROUP

Administered Members (min/max): 1/30

Total Administered Members: 30

#### GROUP MEMBER ASSIGNMENTS

	Port	Code	Sfx	Name	Night	Sig Grp
1:	01B1601	TN464	G			210
2:	01B1602	TN464	G			210
3:	01B1603	TN464	G			210
4:	01B1604	TN464	G			210
5:	01B1605	TN464	G			210
6:	01B1606	TN464	G			210
7:	01B1607	TN464	G			210
8:	01B1608	TN464	G			210
9:	01B1609	TN464	G			210
10:	01B1610	TN464	G			210
11:	01B1611	TN464	G			210
12:	01B1612	TN464	G			210
13:	01B1613	TN464	G			210
14:	01B1614	TN464	G			210
15:	01B1615	TN464	G			210

Page Down to the next page and enter the remaining ports. Note that port 16 is reserved for the signaling channel and is not added as a group member. Submit these changes.

change trunk-group 210					Page 6 of 21
TRUNK GROUP					
Administered Members (min/max):					1/30
GROUP MEMBER ASSIGNMENTS					Total Administered Members: 30
Port	Code	Sfx	Name	Night	Sig Grp
16: 01B1617	TN464	G			210
17: 01B1618	TN464	G			210
18: 01B1619	TN464	G			210
19: 01B1620	TN464	G			210
20: 01B1621	TN464	G			210
21: 01B1622	TN464	G			210
22: 01B1623	TN464	G			210
23: 01B1624	TN464	G			210
24: 01B1625	TN464	G			210
25: 01B1626	TN464	G			210
26: 01B1627	TN464	G			210
27: 01B1628	TN464	G			210
28: 01B1629	TN464	G			210
29: 01B1630	TN464	G			210
30: 01B1631	TN464	G			210

### 3.4.6. Administer System Features Form

See Section 3.3.6.

### 3.4.7. Administer Private Numbering

See Section 3.3.7.

### 3.4.8. Administer Private Routing to the E1/ISDN/QSIG Span

#### 3.4.8.1 Administer the AAR Feature Access Code<sup>8</sup>

See Section 3.3.8.1.

#### 3.4.8.2 Administer the Uniform Dialing Plan

See Section 3.3.8.2.

#### 3.4.8.3 Administer the AAR Digit Analysis Table

See Section 3.3.8.3.

#### 3.4.8.4 Administer the Route Pattern

Use the “change route-pattern 201” command to access route pattern 201. In the tested configuration, an entry was created to route calls to trunk 210 in the following manner: enter “210” in the **Grp No** column, “0” in the **FRL** column, “7” in the **No. Del Dgts** column, and “2201” in the **Inserted Digits**. The inserted digits “2201” represent the Dialed Number

<sup>8</sup> Note that in the following sections, AAR Feature Access Code “8”, extension “22291”, AAR dialed string 2262291”, route pattern “201”, trunk group “210”, vector directory number “2201”, and vector “201”, were used in the tested configuration. However, these values will differ from site to site.



Identification Service (DNIS) digits to be passed to the FlexxGate™ during the initial call setup. Other values were left at the defaults. Submit these changes.

change route-pattern 201										Page	1 of 3
Pattern Number: 201 Pattern Name: FlexGate2201											
SCCAN? n Secure SIP? n											
Grp	FRL	NPA	Pfx	Hop	Toll	No.	Inserted	DCS/	IXC		
No			Mrk	Lmt	List	Del	Digits	QSIG			
							Dgts	Intw			
1:	210	0				7	2201	n	user		
2:								n	user		
3:								n	user		
4:								n	user		
5:								n	user		
6:								n	user		
BCC VALUE		TSC	CA-TSC		ITC		BCIE	Service/Feature	PARM	No. Numbering	LAR
0	1	2	M	4	W					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	

### 3.4.8.5 Administer a Vector Directory Number

See Section 3.3.8.5.

### 3.4.8.6 Administer a Vector

See Section 3.3.8.6.

## 4. Configure FlexxGate™ T1/E1 Board and Protocol

The procedures for configuring the FlexxGate™ media gateway include the following areas:

- Span Configuration – Physical Layer
- Signaling and Channel Configuration
- Dial Plan Configuration

### 4.1. Span Configuration – Physical Layer

The /etc/zaptel.conf file controls the physical layer configuration of the individual spans connected to the Digium TE412P telephony interface card. Four spans were used during the testing and were configured in the following manner: span 1 was configured for T1/CAS, span 2 was configured for E1/CAS, span 3 was configured for T1/ISDN/QSIG, and span 4 was configured for E1/ISDN/QSIG.

Log into the FlexxGate™ media gateway using root level permissions and edit the /etc/zaptel.conf file. The programming of this file is beyond the scope of these Application Notes. Product Support Solutions, Inc. will provision this file as necessary to meet customer

requirements. The values that were used during integration testing are listed below and are described in [4], available from Product Support Solutions, Inc.

```
# Span 1 = T1 - D4/AMI
# Channels 1-24 set to e&m
span=1,1,0,d4,ami
fxsks=1-24

# Span 2 = E1 - CAS/AMI
# Channels 25-55 set to e&m
span=2,2,0,cas,ami
fxsks=25-55

# Span 3 = T1 - ISDN (ESF/B8ZS)
# B Channels 56-78 / D Channel 79
span=3,3,0,esf,b8zs
bchan=56-78
dchan=79

# Span 4 = E1 - ISDN
# B Channels 80-94 & 96-110 / D Channel 95
span=4,4,0,ccs,hdb3,crc4
bchan=80-94,96-110
dchan=95
```

## 4.2. Signaling and Channel Configuration

The /etc/asterisk/zapata.conf file controls the signaling layer configuration for all calls arriving at the FlexxGate™ media gateway. Four spans were used during the testing and were configured in the following manner: span 1 was configured for T1/CAS, span 2 was configured for E1/CAS, span 3 was configured for T1/ISDN/QSIG, and span 4 was configured for E1/ISDN/QSIG.

Log into the FlexxGate™ media gateway using root level permissions and edit the /etc/zapata.conf file. For spans 1 and 2, the **switchtype** variables were set to “5ess” and the “signalling” variables were set to “fxs\_ks”. For spans 3 and 4, the **switchtype** variables were set to “qsig” and the “signalling” variables were set to “pri\_cpe”. The programming of this file is beyond the scope of these Application Notes. Product Support Solutions, Inc. will provision this file as necessary to meet customer requirements. The values that were used during integration testing are listed below and are described in [4], available from Product Support Solutions, Inc.

```
; *****
; **** Span 1 ****
; *****
language=en
switchtype=5ess
signalling=fxs_ks
usecallerid=yes
rxgain=0.0
txgain=0.0
echocancel=no
echocancelwhenbridged=no
echotraining=no
transfer=yes
transfertobusy=yes
threewaycalling=yes
cancallforward=yes
immediate=no
busydetect=no
callprogress=no
```

```

relaxdtmf=no
;facilityenable=yes
;overlapdial=yes
group=0
channel=1-24
; *****
; **** End Span 1 ****
; *****

; *****
; **** Span 2 ****
; *****
language=en
switchtype=5ess
signalling=fxs_ks
usecallerid=yes
rxgain=0.0
txgain=0.0
echocancel=no
echocancelwhenbridged=no
echotraining=no
transfer=yes
transfertobusy=yes
threewaycalling=yes
cancallforward=yes
immediate=no
busydetect=no
callprogress=no
relaxdtmf=no
;facilityenable=yes
;overlapdial=yes
group=1
channel=25-55
; *****
; **** End Span 2 ****
; *****

; *****
; **** Span 3 ****
; *****
language=en

; ISDN - no Path Replacement!
;switchtype=national

; ISDN - Q.Sig with Path Replacement!
switchtype=qsig

signalling=pri_cpe
usecallerid=yes
rxgain=0.0
txgain=0.0
echocancel=no
echocancelwhenbridged=no
echotraining=no
transfer=yes
transfertobusy=yes
threewaycalling=yes
cancallforward=yes
immediate=no
busydetect=no
callprogress=no
relaxdtmf=no
;facilityenable=yes
;overlapdial=yes
group=2
channel=56-78
; *****
; **** End Span 3 ****
; *****

```

```
; *****
; **** Span 4 ****
; *****
language=en

; E1-ISDN - no Path Replacement!
;switchtype=euroisdn

; ISDN - Q.Sig with Path Replacement!
switchtype=qsig

signalling=pri_cpe
usecallerid=yes
rxgain=0.0
txgain=0.0
echocancel=no
echocancelwhenbridged=no
echotraining=no
transfer=yes
transfertobusy=yes
threewaycalling=yes
cancallforward=yes
immediate=no
busydetect=no
callprogress=no
relaxdtmf=no
;facilityenable=yes
;overlapdial=yes
group=3
channel=80-94,96-110
; *****
; **** End Span 4 ****
; *****
```

### 4.3. Dial Plan Configuration

The `/etc/asterisk/extensions.conf` file controls the dial plan configuration of the individual spans connected to the Digium TE412P telephony interface card. This file provides the flexibility to map incoming channels to the appropriate IVR script and can manipulate or forward any DNIS digits that are received from Avaya Communication Manager, as programmed in **Section 3.3.8.4** and **Section 3.4.8.4**, for processing by the appropriate IVR script.

Log into the FlexxGate™ media gateway using root level permissions and edit the `/etc/asterisk/extensions.conf` file. The programming of this file is beyond the scope of these Application Notes. Product Support Solutions, Inc. will provision this file as necessary to meet customer requirements.

```
; *****
; *                               INBOUND FROM AVAYA                               *
; *****

; example ...
; send ANY inbound call to 2202@hollylab
;exten => s,1,Dial(SIP/2203@hollylab)
; ... end of example

; for standard T1 and E1 inbound (without DNIS)
; check inbound channel and send call to appropriate app for that channel - for a specific span
exten => s,1,NoOp({CHANNEL}) ; show current channel designation
exten => s,2,Set(CHNO=${CUT(CHANNEL/,2)}) ; Get JUST channel number from Channel designation
exten => s,3,Set(CHNO=${CUT(CHNO,-,1)}) ;
exten => s,4,NoOp({CHNO}) ; show JUST channel number
```

```

; Span 1
exten => s,5,GotoIf($[${CHNO} < 9]?:7)      ;if channel is less than 9(1-8) send DNIS 2201 to
holly
exten => s,6,Dial(SIP/2201@hollylab)
exten => s,7,GotoIf($[${CHNO} < 17]?:9)      ;if channel is less than 17(9-16) send 2202 to holly
exten => s,8,Dial(SIP/2201@hollylab)
exten => s,9,GotoIf($[${CHNO} < 25]?:11)     ;if channel is less than 25(17-24) send 2203 to
holly
exten => s,10,Dial(SIP/2201@hollylab)
; Span 2
exten => s,11,GotoIf($[${CHNO} < 36]?:13)    ;if channel is less than 36(S2,1-11) send 2201 to
holly
exten => s,12,Dial(SIP/2201@hollylab)
exten => s,13,GotoIf($[${CHNO} < 46]?:15)    ;if channel is less than 46(S2,12-21) send 2202 to
holly
exten => s,14,Dial(SIP/2201@hollylab)
exten => s,15,GotoIf($[${CHNO} < 56]?:17)    ;if channel is less than 56(S2,22-30) send 2202 to
holly
exten => s,16,Dial(SIP/2201@hollylab)
exten => s,17,NoOp(Fell through - Channel ${CHNO} did not match dialplan)

; For ISDN lines with Inbound DNIS - change 1001-1003 to any inbound DNIS as received from the
Avaya
; Get appropriate DNIS and send onto associated Holly app
; App 1
exten => 2201,1,Dial(SIP/2201@hollylab)
; App 2
exten => 2202,1,Dial(SIP/2202@hollylab)
; App 3
exten => 2203,1,Dial(SIP/2203@hollylab)

; *****
; *                                TRANSFERS OUT FROM HOLLY                                *
; *****
;
; All the dialplans below are to be used with the Holly 2201 Transfer App
; The number dialed when asked by Holly "Please enter your transfer number" is the exten =>
number at
; the start of each dialplan below.

; T1 and E1 hookflash CAS transfers (Spans 1&2) - use 8101 to transfer from the Holly back to the
Avaya (at extension 1234)
; 1234 can be changed to any valid - any length - extension the Avaya is set to receive for xfer
testing
exten => 8101,1,flash()
exten => 8101,2,SendDTMF(1234)
exten => 8101,3,Wait(2)
exten => 8101,4,Hangup()

; T1 bridged/trombone CAS transfers (Spans 1&2) - use 8101 to transfer from the Holly back to the
Avaya (at extension 1234)
; 1234 can be changed to any valid - any length - extension the Avaya is set to receive for xfer
testing
exten => 8301,1,Dial(Zap/g0/ww1234)

; E1 bridged/trombone CAS transfers (Spans 1&2) - use 8101 to transfer from the Holly back to the
Avaya (at extension 1234)
; 1234 can be changed to any valid - any length - extension the Avaya is set to receive for xfer
testing
exten => 8302,1,Dial(Zap/g1/ww1234)

; T1 ISDN xfer (Span 3) - Use 8202 to transfer from the Holly back to the Avaya (at extension
1234)
; for trombone (dual-line) or Path Replacment xfers based on setting of switchtype in
/etc/zaptel.conf

```

```

; 1234 can be changed to any valid - any length - extension the Avaya is set to receive for xfer
testing
exten => 8202,1,Dial(Zap/G2/26614)

;exten => _2XXXX,1,Dial(Zap/G2/${EXTEN})
;exten => _91331XXXXXXX,1,Dial(Zap/G2/${EXTEN})

; E1 ISDN xfer (Span 4) - Use 8202 to transfer from the Holly back to the Avaya (at extension
1234)
; for trombone (dual-line) or Path Replacment xfers based on setting of switchtype in
/etc/zaptel.conf
; 1234 can be changed to any valid - any length - extension the Avaya is set to receive for xfer
testing
exten => 8203,1,Dial(Zap/g3/26614)

; For 5 digit extensions starting with 2, refer back on same span:
exten => _2XXXX!,1,NoOp(${CHANNEL}) ; dial the number given from the SIP-REFER message
exten => _2XXXX!,n,Set(CHNO=${CUT(CHANNEL,,2)})
exten => _2XXXX!,n,Set(CHNO=${CUT(CHNO,-,-1)})
exten => _2XXXX!,n,NoOp(${CHNO})
exten => _2XXXX!,n,GotoIf(${CHNO} < 25)?:10 ;if Span 1 (1-24), do hook-flash
exten => _2XXXX!,n,flash()
exten => _2XXXX!,n,SendDTMF(${EXTEN})
exten => _2XXXX!,n,Wait(0)
exten => _2XXXX!,n,Hangup()
exten => _2XXXX!,n,GotoIf(${CHNO} < 56)?:15 ;if Span 2 (25-55), do hook-flash
exten => _2XXXX!,n,flash()
exten => _2XXXX!,n,SendDTMF(${EXTEN})
exten => _2XXXX!,n,Wait(0)
exten => _2XXXX!,n,Hangup()
exten => _2XXXX!,15,GotoIf(${CHNO} < 79)?:17 ;if Span 3 (56-78), dial out on Span 3 (G2)
exten => _2XXXX!,n,Dial(Zap/G2/${EXTEN})
exten => _2XXXX!,17,Dial(Zap/G3/${EXTEN}) ;else must be Span 4, dial out on Span 4
(G3)

; Refer external Long Distance calls for area code 331 back on same span:
exten => _91331XXXXXXX,1,NoOp(${CHANNEL}) ; dial the number given from the SIP-REFER message
exten => _91331XXXXXXX,n,Set(CHNO=${CUT(CHANNEL,,2)})
exten => _91331XXXXXXX,n,Set(CHNO=${CUT(CHNO,-,-1)})
exten => _91331XXXXXXX,n,NoOp(${CHNO})
exten => _91331XXXXXXX,n,GotoIf(${CHNO} < 25)?:10 ;if Span 1 (1-24), do hook-flash
exten => _91331XXXXXXX,n,flash()
exten => _91331XXXXXXX,n,SendDTMF(${EXTEN})
exten => _91331XXXXXXX,n,Wait(0)
exten => _91331XXXXXXX,n,Hangup()
exten => _91331XXXXXXX,n,GotoIf(${CHNO} < 56)?:15 ;if Span 2 (25-55), do hook-flash
exten => _91331XXXXXXX,n,flash()
exten => _91331XXXXXXX,n,SendDTMF(${EXTEN})
exten => _91331XXXXXXX,n,Wait(0)
exten => _91331XXXXXXX,n,Hangup()
exten => _91331XXXXXXX,15,GotoIf(${CHNO} < 79)?:17 ;if Span 3 (56-78), dial out on Span 3
(G2)
exten => _91331XXXXXXX,n,Dial(Zap/G2/${EXTEN})
exten => _91331XXXXXXX,17,Dial(Zap/G3/${EXTEN}) ;else must be Span 4, dial out on Span 4
(G3)

```

## 5. Interoperability Compliance Testing

The interoperability compliance testing focused on the following areas:

- The ability of FlexxGate™ to receive inbound calls from Avaya Communication Manager via T1/E1 ISDN/QSIG and CAS interfaces.
- The ability of FlexxGate™ to properly transfer calls using QSIG Path Replacement or flash-hook signaling.

## 5.1. General Test Approach

The feature test cases were conducted by dialing Public Switched Telephone Network (PSTN) calls that arrived on trunk facilities to Avaya Communication Manager. For T1/CAS and E1/CAS spans, calls were dialed to hunt group extensions and for T1/ISDN/QSIG and E1/ISDN/QSIG spans, calls were dialed to vector directory numbers that routed calls over the appropriate span to the FlexxGate™ media gateway. The FlexxGate™ converted these calls for processing by the IVR using SIP based protocols. The caller was prompted to enter either a five digit extension or a long distance telephone number. In either case, calls were transferred to the intended destination and the bearer channels between Avaya Communication Manager and the FlexxGate™ media gateway were properly dropped.

Subjective verification of the talk path was performed from the caller to the intended transfer destination. Various call detail record reports available using the FlexxGate™ media gateway browser interface were verified.

## 5.2. Test Results

All of the executed test cases passed.

## 6. Verification Steps

This section provides various checks that can be performed to verify basic connectivity between Avaya Communication Manager and the FlexxGate™ media gateway.

### 6.1. Verify Avaya Communication Manager T1/CAS and E1/CAS

Verify the status of the T1/CAS and E1/CAS channels by using the “status station n” command, where “n” is the extension number administered in **Section 3.1.2**, for T1/CAS channels, and **Section 3.2.2**, for E1/CAS channels. While the channels are idle, verify the **Service State** for each of the stations is “in-service/on-hook” as shown below.

status station 22201		Page 1 of 3	
GENERAL STATUS			
Administered Type: DS1FD		Service State: in-service/on-hook	
Connected Type: N/A			
Extension: 22201			
Port: 01C1201		Parameter Download: not-applicable	
Call Parked? no		SAC Activated? no	
Ring Cut Off Act? no			
Active Coverage Option: 1			
EC500 Status: N/A		Off-PBX Service State: N/A	
Message Waiting:			
Connected Ports:			
Limit Incoming Calls? no			
User Cntrl Restr: none		HOSPITALITY STATUS	
Group Cntrl Restr: none		Awaken at:	
		User DND: not activated	
		Group DND: not activated	
		Room Status: non-guest room	

## 6.2. Verify Avaya Communication Manager T1/ISDN/QSIG and E1/ISDN/QSIG

Verify the status of the trunk group by using the “status trunk n” command, where “n” is the trunk group number administered in **Section 3.3.3** for T1/ISDN/QSIG or in **Section 3.4.3** for E1/ISDN/QSIG. While the trunks are connected and idle, verify the **Service State** for each connected trunk is “in-service/idle” as shown below.

status trunk 200				Page 1
TRUNK GROUP STATUS				
Member	Port	Service State	Mtce Connected Ports Busy	
0200/001	01B1701	in-service/idle	no	
0200/002	01B1702	in-service/idle	no	
0200/003	01B1703	in-service/idle	no	
0200/004	01B1704	in-service/idle	no	
0200/005	01B1705	in-service/idle	no	
0200/006	01B1706	in-service/idle	no	
0200/007	01B1707	in-service/idle	no	
0200/008	01B1708	in-service/idle	no	
0200/009	01B1709	in-service/idle	no	
0200/010	01B1710	in-service/idle	no	
0200/011	01B1711	in-service/idle	no	
0200/012	01B1712	in-service/idle	no	
0200/013	01B1713	in-service/idle	no	
0200/014	01B1714	in-service/idle	no	
press CANCEL to quit -- press NEXT PAGE to continue				



Page Down to Page 2 and verify the remaining trunk group members are “in-service/idle” as shown below.

```
status trunk 200
```

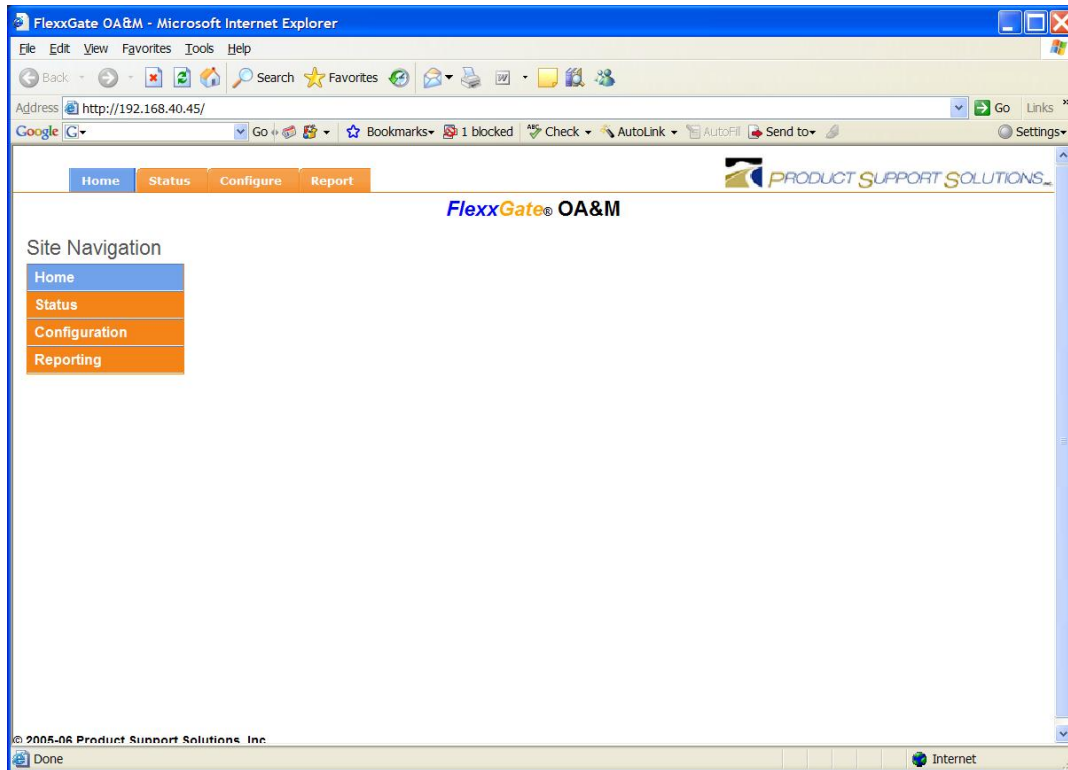
TRUNK GROUP STATUS

Member	Port	Service State	Mtce Connected Ports Busy
0200/015	01B1715	<b>in-service/idle</b>	no
0200/016	01B1716	<b>in-service/idle</b>	no
0200/017	01B1717	<b>in-service/idle</b>	no
0200/018	01B1718	<b>in-service/idle</b>	no
0200/019	01B1719	<b>in-service/idle</b>	no
0200/020	01B1720	<b>in-service/idle</b>	no
0200/021	01B1721	<b>in-service/idle</b>	no
0200/022	01B1722	<b>in-service/idle</b>	no
0200/023	01B1723	<b>in-service/idle</b>	no

```
Command successfully completed
```

### 6.3. Verify FlexxGate™ Spans

Open a browser interface and enter the URL for the FlexxGate™ media gateway as shown below to access the FlexxGate™ OA&M screen. In the tested configuration, the URL for the FlexxGate™ media gateway was <http://192.168.40.45/>.



Click on **Status** on the toolbar at the top of the browser window.

The **Status Monitor** screen will be displayed. Verify that the **Status** column shows “OK” for each of the spans as shown below.

The screenshot shows the FlexxGate Status Monitor web application running in Microsoft Internet Explorer. The browser address bar shows <http://192.168.40.45/status.php>. The page has a navigation bar with links: Home, Status, Configure, and Report. The main title is "FlexxGate® Status Monitor" for "FlexxGate1". There is a "Refresh" button set to 30 seconds and an "Update" button.

**TDM**

Board	Span	Status	Info
0	1	OK	No Info! Not a PRI!
0	2	OK	No Info! Not a PRI!
0	3	OK	PRI is Provisioned, Up, Active
0	4	OK	PRI is Provisioned, Up, Active

Last Update: Fri Apr 6 13:50:18 EDT 2007 local machine time.

**VoIP**

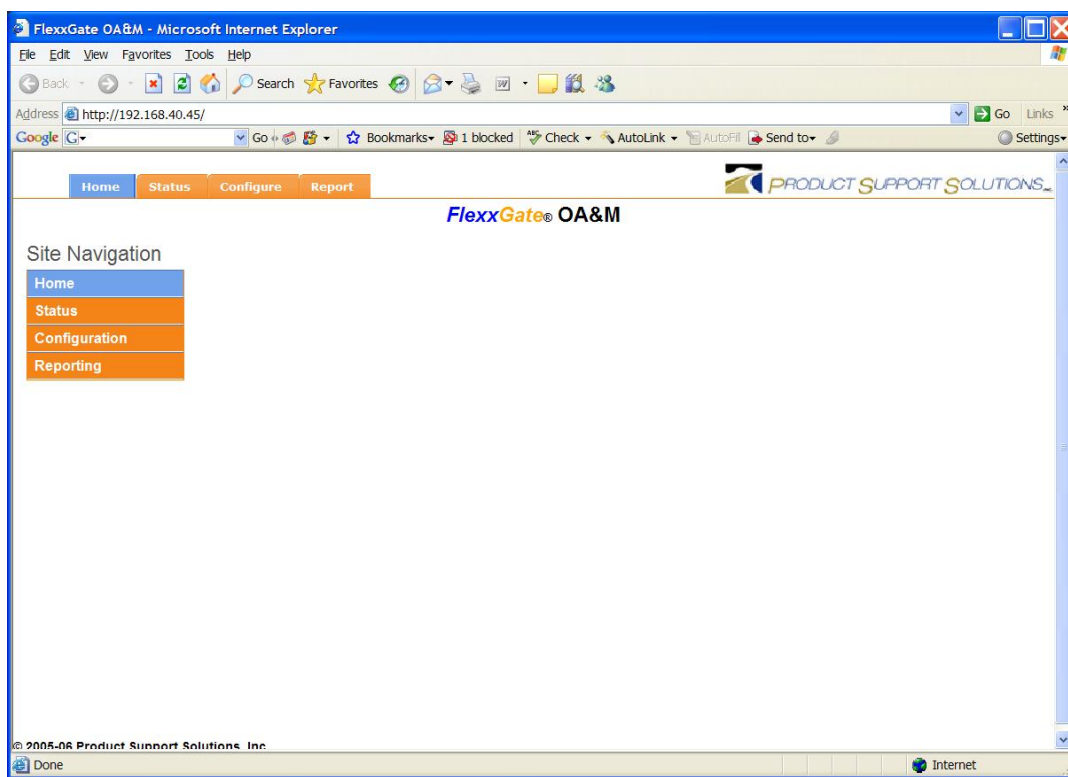
Name	Host	Port	Status (response)	Info
hollylab	192.168.40.33	5060	OK (1 ms)	

Last Update: Fri Apr 6 13:50:18 EDT 2007 local machine time.

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## 6.4. Verify FlexxGate™ Call Detail Records

Make a test call to the DDC hunt group extension added in **Section 3.1.3** for T1/CAS or **Section 3.2.3** for E1/CAS. Or, make a test call to the vector directory number added in **Section 3.3.8.5** for T1/ISDN/QSIG or **Section 3.4.8.5** for E1/ISDN/QSIG. Verify that a prompt is played back to the caller. The prompt that will be heard will depend upon customer requirements. In the test configuration, the IVR prompted the caller to enter transfer destination digits. Enter the transfer destination and verify that the call is transferred to the intended destination. Allow a few minutes for call detail records to be populated into the call detail records database on the FlexxGate™. Open a browser interface and enter the URL for the FlexxGate™ media gateway as shown below to access the FlexxGate™ OA&M screen. In the tested configuration, the URL for the FlexxGate™ media gateway was <http://192.168.40.45>.



Click on **Report** on the toolbar at the top of the browser window.

The **FlexxGate™ Reporting** screen will be displayed. Verify the default selection of “Selection of the month”. Click on the **Search** button.

FlexxGate® Reporting - FlexxGate1 - Microsoft Internet Explorer

File Edit View Favorites Tools Help

Address http://192.168.40.45/report.php

Google Go Links

Home Status Configure **Report**

Link 4

**FlexxGate® Reporting**  
**FlexxGate1**

Reports Menu

- CDR REPORT**
- CALLS COMPARE
- MONTHLY TRAFFIC
- DAILY LOAD

Selection of the month

From : April-2007 To : April-2007

Selection of the day

From : 01 To : 01

April-2007 April-2007

DESTINATION

Exact Begins with Contains Ends with

SOURCE

Exact Begins with Contains Ends with

CLI

Exact Begins with Contains Ends with

USERFIELD

Exact Begins with Contains Ends with

ACCOUNTCODE

Exact Begins with Contains Ends with

CHANNEL

DURATION

Search Result: Minutes Seconds

Done Internet

Under the **Call Logs** section, verify that call detail records are found and listed. A call detail record should exist for each new call that has arrived at the FlexxGate™ system. As shown below, the search results will contain the channel number, the caller identification, if available, and the call disposition.

The screenshot shows the FlexxGate Reporting interface in a Microsoft Internet Explorer browser window. The address bar shows the URL: [http://192.168.40.45/report.php?s=0&t=&order=caldate&sens=DESC&current\\_page=0](http://192.168.40.45/report.php?s=0&t=&order=caldate&sens=DESC&current_page=0). The interface has a navigation bar with links: Home, Status, Configure, and Report. The main heading is "FlexxGate Reporting FlexxGate1".

On the left is a "Reports Menu" with links: CDR REPORT, CALLS COMPARE, MONTHLY TRAFFIC, and DAILY LOAD. The "CDR REPORT" link is selected.

The main area contains search filters:

- Selection of the month:** From: April-2007, To: April-2007
- Selection of the day:** From: 01 April-2007, To: 01 April-2007
- DESTINATION:** [Text Field] Exact Begins with Contains Ends with
- SOURCE:** [Text Field] Exact Begins with Contains Ends with
- CLI:** [Text Field] Exact Begins with Contains Ends with
- USERFIELD:** [Text Field] Exact Begins with Contains Ends with
- ACCOUNTCODE:** [Text Field] Exact Begins with Contains Ends with
- CHANNEL:** [Text Field]
- DURATION:** [Text Field] > > = = < < [Text Field] > > = = < <

Below the filters is a "Search" button and radio buttons for "Result: Minutes" (selected) and "Seconds".

Below the search area, it says "Number of calls : 48".

The "Call Logs" section shows a table with the following data:

Calldate	Channel	Source	Clid	Lastapp	Lastdata	Dst	APP Disposition	Duration	Userfield	Accountcode
1. 2007-04-06 12:55:42	Zap/56-...	73285230497328523049	Dial	Zap/G2/26614	26614	ANSWERED	00:11			
2. 2007-04-06 12:55:42	Zap/1-1...		Dial	SIP/2201@hollylab	s	ANSWERED	00:42			

## 7. Support

Technical support on FlexxGate™ can be obtained through the following:

- Email the PSS support center via [support@psshhelp.com](mailto:support@psshhelp.com)
- Web at <http://www.psshhelp.com>
- Call the PSS support center:
  - In the US (24x7) at 1.888.455.2285
  - In the UK (24x7) at 0.808.234.6787

## 8. Conclusion

These Application Notes describe the configuration steps required for Product Support Solutions' FlexxGate media gateway connectivity solution to successfully interoperate with Avaya Communication Manager 4.0 using T1/E1 ISDN/QSIG and CAS interfaces.

## 9. Additional References

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

[1] *Administrator Guide for Avaya Communication Manager*, Document 03-300509, Issue 3, February 2007, available at <http://support.avaya.com>

[2] *Job Aid: Option Switch Settings*, Document 555-245-774, Issue 3, January 2005, available at <http://support.avaya.com>

[3] *Avaya Call Center Release 4.0 Call Vectoring and Expert Agent Selection (EAS) Guide*, Document 07-600780, Release 4.0, February 2007, available at <http://support.avaya.com>

[4] *Product Support Solutions, Inc., FlexxGate™ Media Gateway Operations Guide, Revision 0.04*, available from Product Support Solutions, Inc.

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