



Siemens Hicom 300 E CS Release 6.6 using T1 QSIG to Cisco Unified CallManager Express Release 4.0(3)

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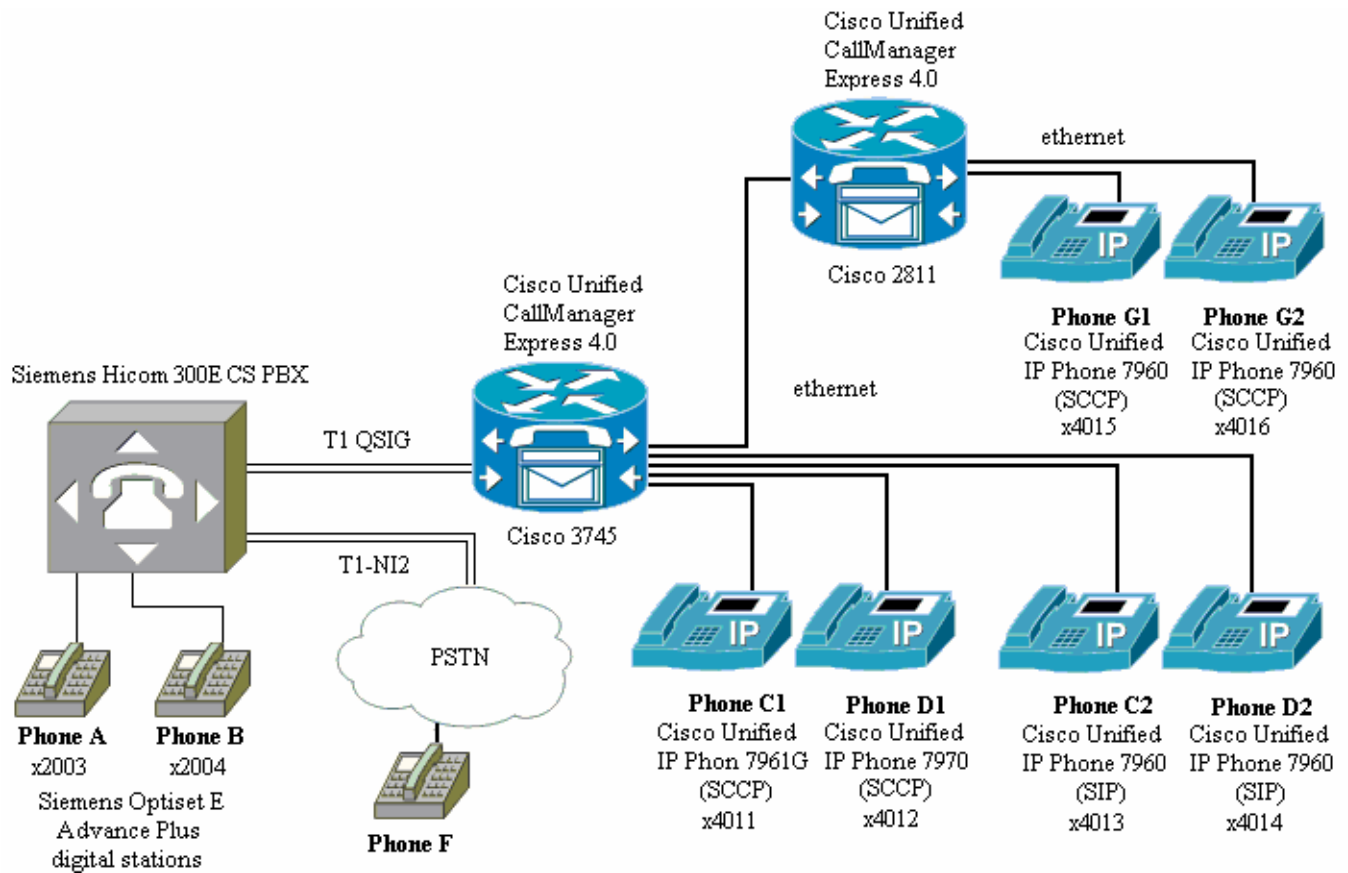
Introduction

- This is an Application Note for connectivity between a Siemens Hicom 300 E CS Release 6.6 PBX and Cisco Unified CallManager Express Release 4.0(3) using a Cisco 3745 voice gateway with QSIG protocol.
- The network topology diagram (Figure 1) shows the test setup for end-to-end interoperability with Cisco Unified CallManager Express Release 4.0(3) connected to the PBX via the 3745 T1 QSIG link. The 3745 IOS voice gateway was connected via H.323 to a Cisco 2811 IOS voice gateway. The two gateways were running Cisco Unified CallManager Express 4.0(3). Cisco Unified IP phones (models 7960, 7961G, and 7970) were connected to the 2 Cisco Unified CallManager Express gateways via SIP and SCCP, as per the figure. A NM-HDV and VWIC-1MFT-T1 were used for the T1 QSIG interface. Calls were made to test basic call, caller ID, conference, transfer, forward, and reroute features.
- This Application Note uses the 3745 voice gateway. However, the use of other Cisco voice gateways is also an option since Cisco Unified Call Manager Express QSIG implementation does not depend on the physical interface.
- The inclusion of Cisco SIP phones in this application note is for reference only. Cisco Unified Communications Manager Express 4.0(3) supports SIP end-points with limited number of features.



Network Topology

Figure 1. Test Network Topology.





Limitations

Basic Calls

- Overlap dialing is not supported in either direction.
- Connected Name is not supported on calls between PBX and Cisco Unified IP Phone running SIP.
- Alerting Name is not supported on calls between PBX and Cisco Unified IP Phone running SIP.
- Calling Line (Number) Identification Restriction (CLIR) is not supported by the Siemens PBX for outgoing calls. The PBX does honor CLIR on incoming calls, however.
- Calling Name Identification Restriction (CNIR) and Connected Number/Name Restriction are not supported.

Call Transfers

- The Siemens PBX will not perform a true blind transfer. It can perform a consultation transfer or early attended transfer.
- A blind local transfer originated from a call placed from a PBX station to a SCCP phone on the local Cisco Unified CallManager Express, and then transferred to a SIP phone on the same Cisco Unified CallManager Express (e.g., A calls C1, and C1 transfers to D2) does not complete. The call drops as soon as the number is dialed.
- A consultation or early-attended transfer originated from a call placed from a phone on the remote Cisco Unified CallManager Express to a SIP phone on the local Cisco Unified CallManager Express, and then transferred to a PBX phone (e.g., G1 calls C2, and C2 transfers to A) does not complete.
- For all consultation and early attended network/external transfers, and all consultation and early attended local transfers that involve a transfer from a SCCP phone to a SIP phone, the original calling name and number are not displayed on the final destination. The remaining local transfers and all blind transfers result in the original calling name and number information displaying properly.
- For many call transfers, the called (connected) name and number are not updated on the original phone after the transfer.

Call Forwards

- For "trombone" or "hairpin" calls from the Cisco Unified CallManager Express to PBX to Cisco Unified CallManager Express, the Siemens PBX will not allow a "trombone" or "hairpin" on call forward busy. It invokes a reroute.
- For "trombone" or "hairpin" calls from the PBX to the Cisco Unified CallManager Express to the PBX, call reroute does not occur. Even if Cisco Unified CallManager Express proposes reroute (SCCP phone only), the PBX ignores it.
- For many call forwards, the forwarding called name and number are not displayed on the final destination.
- For many call forwards, the called (connected) name and number are not updated on the original phone.
- Forwarded calls originated from a PBX extension to a remote Cisco Unified CallManager Express SCCP extension, and forwarded to a local Cisco Unified CallManager Express extension (e.g., A calls G1, and G1 forwards to C2), Cisco Unified CallManager Express performs a QSIG reroute, even though a QSIG reroute is not in order (i.e., there is no QSIG "hairpin" or "trombone").
- Forwarded calls originated from a PBX extension to a local Cisco Unified CallManager Express SCCP extension, and forwarded to another local Cisco Unified CallManager Express extension by call forward no reply (e.g., A calls C1, and C1 forwards to D1 or D2), Cisco Unified CallManager Express performs a reroute, and even though a reroute is not in order (i.e., there is no "hairpin" or "trombone").

MWI

- Cisco Unified Communications Manager Express 4.0(3) supports Cisco Unity integration with QSIG. However, in this instance, no testing was performed with Cisco Unified Communications Manager Express 4.0(3) as the message center PINX.
- There was no PBX voice mail system present at the time of testing. Therefore, no testing was performed with the PBX as the message center PINX.



System Components

Hardware Requirements

- Cisco 3745 IOS voice gateway
 - NM-HDV
 - VWIC-2MFT-T1
- Cisco 2811 IOS voice gateway
- (4) Cisco Unified IP phone 7960s
- (1) Cisco Unified IP phone 7961G
- (1) Cisco Unified IP phone 7970
- (1) Siemens 300 E CS PBX
 - (2) Siemens Optiset E Advance Plus digital station phones
 - (2) TMDN64P T1 trunk cards (for QSIG & PSTN links)

Software Requirements

- Cisco Unified CallManager Express Release 4.0(3)
- Cisco IOS Software, 3700 Software (C3745-IPVOICE-M), Version 12.4(11)T
- Cisco IOS Software, 2800 Software (C2800NM-IPVOICE-M), Version 12.4(11)T
- Siemens 300 E CS Release 6.6

G1, G2 – 7960 – SCCP

- Cisco7960 IP phone version 7.2(T0.23)
- Cisco 7960 IP phone app load P0030702T023
- Cisco 7960 IP phone boot load PC0303010200

C2, D2 – 7960 - SIP

- Cisco7960 DSP load ID PS03AT46
- Cisco 7960 IP phone app load POS3-07-5-00
- Cisco 7960 IP phone boot load PC030301

C1 – 7961G – SCCP

- Cisco7961G IP phone load file: TERM61.DEFAULT
- Cisco 7961G IP phone app load ID: Jar41.2-9-1-45.sbn
- Cisco 7961G IP phone boot load ID: 7961G_64-020704128Amd64meg.bin

D1 – 7970 – SCCP

- Cisco7970 IP phone load file: SCCP70.8-0-3S
- Cisco 7970 IP phone app load ID: jar70sccp.8-0-2.25.sbn
- Cisco 7970 IP phone boot load ID: 7970_64060118.bin



Features

Features Supported

- Basic Call, ENBLOC Dialing only
- CLIP-Calling Line (Number) Identification Presentation on Basic and Forwarded Calls
- CLIR-Calling Line (Number) Identification Restriction on Basic Calls (See Limitations section.)
- CNIP-Calling Name Identification Presentation on Basic and Forwarded Calls
- COLP-Connected Line (Number) Identification Presentation on Basic Calls (See Limitations section.)
- CONP-Connected Name Identification Presentation on Basic Calls (See Limitations section.)
- Alerting Name (See Limitations section.)
- Tandem PSTN call
- Consultation Transfer – Local (See Limitations Section)
- Consultation Transfer – Network/External (See Limitations Section)
- Early Attended Transfer – Local (See Limitations Section)
- Early Attended Transfer – Network/External (See Limitations Section)
- Blind Transfer – Local (See Limitations Section)
- Blind Transfer – Network/External (See Limitations Section)
- Call Forward Unconditional by Join – Local (See Limitations Section)
- Call Forward Unconditional by Join – Network/External (See Limitations Section)
- Call Forward Busy by Join – Local (See Limitations Section)
- Call Forward Busy by Join – Network/External (See Limitations Section)
- Call Forward No Reply by Join – Local (See Limitations Section)
- Call Forward No Reply by Join – Network/External (See Limitations Section)
- Call Forward Unconditional by Reroute – Network/External (See Limitations Section)
- Call Forward Busy by Reroute – Network/External (See Limitations Section)
- Call Forward No Reply by Reroute – Network/External (See Limitations Section)



Features Not Supported

- Overlap Dialing
- CNIR-Calling Name Identification Restriction
- COLR- Connected Line (Number) Identification Restriction
- CONR- Connected Name Identification Restriction
- Blind Transfers initiated from PBX
- CLIP-Calling Line (Number) Identification Presentation on Transferred Calls
- CNIP-Calling Name Identification Presentation on Transferred Calls
- COLP-Connected Line (Number) Identification Presentation on Transferred Calls
- CONP-Connected Name Identification Presentation on Transferred Calls
- COLP-Connected Line (Number) Identification Presentation on Forwarded Calls
- CONP-Connected Name Identification Presentation on Forwarded Calls
- Call Completion to Busy Subscriber (Call Back when Free)
- Call Completion on No Reply (Call Back Next Used)
- Path Replacement for Call Transfer by Join
- Path Replacement for Trombone Connection
- Path Replacement for Call Diversion by Forward Switch



Configuration

Configuring the Siemens 300 E CS Release 6.6

Below is the dialplan configuration:

DIS-DPLN:DGTS,;

H500: AMO DPLN STARTED

DIGIT INTERPRETATION		VALID FOR DIAL PLAN 0			
		CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER		1 11111 1111222	RESULT	RSVD	ROUTE
		12345 67890 12345 6789012	(SKIP DIGIT)		
0	**	ANS		
1		.*.....**	CO		
2000 - 2023		.******	STN		
2024 - 2026		.******	STN	R	
2222		.******	ATNDIND		
3000 - 3023		.******	STN		
3024 - 3099		.******	STN	R	
36	*	GENANS		
37 - 41		.*.....**	CO		
43 - 48		.******	TIE		
49 - 59		.*.....**	CO		
71000		.******	STN		1
DIGIT INTERPRETATION		VALID FOR DIAL PLAN 0			
		CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER		1 11111 1111222	RESULT	RSVD	ROUTE
		12345 67890 12345 6789012	(SKIP DIGIT)		
71001 - 79999		.******	STN	R	
8000 - 8001		.******	STN		41
9		.******	TIE		
*0		*.....*	ACDWORK		
*2		*.....*	ACCTCODE		
*3		*.....*	PUDIR		



*4	*...* ...* *	CONFRNC			
52 *	MWCAN			
530 *	PMCANCEL			
532 *	PMCALLBK			
*563	*...* ...* *	BADLINE			
*564	*...* ...* *	ACDLOGON			

	DIGIT INTERPRETATION	VALID FOR DIAL PLAN 0			
--	----------------------	-----------------------	--	--	--

	CALL PROGRESS STATE	DIGIT ANALYSIS			
DIRECTORY NUMBER	1 11111 1111222	RESULT	RSVD	ROUTE	
	12345 67890 12345 6789012	(SKIP DIGIT)			

*565	*...* ...* *	ACDLOGOF			
*570	*...* ...* *	ACDPQ			
*571	*...* ...* *	ACDPS			
572 *	RING			
*580	*...* ...* *	ACDSQ			
*581	*...* ...* *	ACDSS			
*6	***** ...* *	ROLMPARK			
*7	*...* ...* *	CONSULT			
*80 - *89	.***** ...* *	PARK			
9* *	HOLD			
0	.*. *	BVSL			
**1	*...* ...* *	TOGGLE			

	DIGIT INTERPRETATION	VALID FOR DIAL PLAN 0			
--	----------------------	-----------------------	--	--	--

	CALL PROGRESS STATE	DIGIT ANALYSIS			
DIRECTORY NUMBER	1 11111 1111222	RESULT	RSVD	ROUTE	
	12345 67890 12345 6789012	(SKIP DIGIT)			

**3	...* ...* *	PU			
**41 - **48 *. *	CONFRMV			
**50	*...* ...* *	CAFGRAVL			
**51	*...* ...* *	CAFGRUNA			
**6	...* ...* *	INTERCOM			
**8	...* *	MWANS			
***4 *	CONFRMVL			
***5 *	MONSLNT			

**#65	*...* ...* *	CAFGROFF			
#01 *	RCHNL			



#02	RTERM		
#03	LTERM		

DIGIT INTERPRETATION	VALID FOR DIAL PLAN 0
----------------------	-----------------------

	CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER	1 1111 1111222	RESULT	RSVD	ROUTE
	12345 67890 12345 6789012	(SKIP DIGIT)		

#04	PRITEST		
*#274	*....*	WS		
*#50	*....*	CAFAVLB		
*#51	*....*	CAFUNAV		
*#53	*****	CIDBLK		
*#54	*****	CIDUBLK		
*#55	*....*	CAFFWD		
*#56	*....*	CAFFWDC		
#57	PIDON		
#58	PIDOFF		
#590	DCOSX		
#591	ACOSX		

DIGIT INTERPRETATION	VALID FOR DIAL PLAN 0
----------------------	-----------------------

	CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER	1 1111 1111222	RESULT	RSVD	ROUTE
	12345 67890 12345 6789012	(SKIP DIGIT)		

*#63	*****	CLEAR		
*#65	*....*	CAFLOGOF		
#735	RELOCATE		
#738	SET		
#97*	COXFER		
#0	*....*	ACDUNAV		
#1	*....*	ACBK		
#2	*....*	PRION		
#3	.***.*	SPDI		
#4	***.*	SNR		
#5*	ADND		
#61	.****	SPDC1		

DIGIT INTERPRETATION	VALID FOR DIAL PLAN 0
----------------------	-----------------------



	CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER	1 1111 1111222	RESULT	RSVD	ROUTE
	12345 67890 12345 6789012	(SKIP DIGIT)		

#62	.**** ..**.*	SPDC2		
#80	*....* ..**.*	BROADCST		
#81	*....* ..**.*	SPKRCALL		
#8378**.*	HWTEST		
#91**.*	CFWVABTH		
#92**.*	CFWVAEXT		
#93**.*	CFWVAINT		
#94**.*	CFWVB		
#95**.*	CFWVBNA		
#96**.*	CFWVNA		
#*056**.*	DATA56		
#*1	***** ..**.*	MWACT		

DIGIT INTERPRETATION VALID FOR DIAL PLAN 0

	CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER	1 1111 1111222	RESULT	RSVD	ROUTE
	12345 67890 12345 6789012	(SKIP DIGIT)		

#*2	*....* ..**.*	BUZZ		
#*329	.***.* *.**** **...	FAX	R	
#*4	*....* ..**.*	VCECALL		
#*75**.*	DIGIDAT		
#*76**.*	SWITCH		
#*77**.*	DTE		
#*78**.*	CODE		
#*79**.*	SPEED		
#*8	***** ***** ..**.*	MWCANORI		
#*90**.*	HUNTPROG		
#*92**.*	AHTVCE		
#*93**.*	DHTVCE		

DIGIT INTERPRETATION VALID FOR DIAL PLAN 0

	CALL PROGRESS STATE	DIGIT ANALYSIS		
DIRECTORY NUMBER	1 1111 1111222	RESULT	RSVD	ROUTE
	12345 67890 12345 6789012	(SKIP DIGIT)		



	#*94	*		AHTDTE			
	#*95	*		DHTDTE		
	#*96	*		AHTFAX		
	#*97	*		DHTFAX		
	#*99	**		HUNTCLR		
	##0		*....*	..**.*		ACDAVLB		
	##1	*		DCBK		
	##2		*....*	..**.		PRIOFF		
	##3	**		SPDIPROG		
	##4		*....*	..**.*		LNR		
	##5	*		DDND		
	##7		*....*		KNOVR		

	DIGIT INTERPRETATION						VALID FOR DIAL PLAN 0			

			CALL PROGRESS STATE				DIGIT ANALYSIS			
	DIRECTORY NUMBER		1 1111 1111222				RESULT			RSVD ROUTE
			12345 67890 12345 6789012				(SKIP DIGIT)			

	##8		*****	..**.		DTA		
	##91	**		CFWVAOFF		
	##*78	*	..**.*		RESET		
	###1		*....	..**.		TRACE		
	###20		**..*	**....		MILLWAT		
	###21		**..*	**....		LOOPBACK		
	###22		**..*	**....		SILENCE		
	###23		**..*	**....		COMBO		
	###4		*....		THRCONF		
	###6	*		MONTONE		

AMO-DPLN -10 DIALING PLANS, FEATURE ACCESS CODES

DISPLAY COMPLETED;



Below is the configuration of the trunk board. This is the configuration of the board supporting the QSIG trunk.

DIS-BCSU:TMD,1,2,25;

H500: AMO BCSU STARTED

```
-----
| DETAILS OF TMD BOARD AT ADDRESS (LTG.LTU.SLOT) = 1. 2. 25 |
|
|   CABTYP =    1           TIMTYP = SYST           SIGTYP = MOS |
|   FRAME  =   ESF         TABS  =    NO          FCTID  =    2 |
|   BI8SUB =   YES        BIVDET =    NO           |
|-----|
|   RDRATIO =    6         RDTH  =   2500         RDQUAL = 15000 |
|   YLSEND  =  5000        YLTH  =    400         YLQUAL =   100 |
|   LOS     =   150        AOS   =   4000           |
|   SESDISTH =   10       SESREQTH =   10           |
|   OESDISTH =   30       OESDISIN = 24-00-00        |
|   OESREQTH =  4000      OESREQIN = 04-00-00        |
|-----|
|   NETUSR  = NETWK        ACKTIM = 1000           DLVTIM =  5000 |
|   OCTMAX  =   260        RETMAX =    3           WINDOW =    7 |
|   CRIDC   =              TTSC   =              NSFIV  =         |
|   NSFTSC  =              PFDGT  =                |
|-----|
|           IGN = 0         IID  = 0                |
|-----
```

AMO-BCSU -10 BOARD CONFIGURATION, SWITCHING UNIT



Below is the configuration of the Class of Trunk (COT) table as configured on the PBX:

DIS-COT:;

H500: AMO COT STARTED

D	A	D	D	D	M	S	V	E	E	A	R
I	N	S	S	I	D	A	L	S	S	N	F
T	S	A	A	S	R	T	S	P	P	I	L
	R		S				A	A	D	D	A
							T	N	N	N	S
							I	I	I	H	

COT										S	S	
-----	--	--	--	--	--	--	--	--	--	---	---	--

0												
1		X										
2		X		X								
3	X											
4		X		X								
5		X		X								
6		X										
14	X			X								

AMO-COT -10 CLASS OF TRUNK FOR CALL PROCESSING



Below is a the configuration of the Class of Parameter (COP) table as configured on the PBX:

DIS-COP;;

H500: AMO COP STARTED

	S			E	A				
	T			E	S	N			
	A	S	V	S	P	I	D	DD	S
	D	Z	L	P	D	D	T	TT	U
	I	A	A	S	S	A	N	N	O
COP	A	N	C	A	A	N	I	I	N
FF	V								
IDX	L	S	K	T	T	I	S	S	E
L	12	1234							

0									
1	X								
2	X	X				X	X	X	
4	X					X		X	
5	X	X	X			X	X	X	
6	X	X			X			X	
50	X					X	X	X	

AMO-COP -10 CLASS OF PARAMETER



Below is the configuration of the trunk group associated with the QSIG trunk:

TGRP = 41;

DIS-TGACC:41;

H500: AMO TGACC STARTED

```
+-----+
| TGRP NUMBER      :    41   TGRP NAME   : QSIG           /N   MAXIMUM NO:   23 |
| SUBGROUP NUMBER:     5   DEVICE TYPE: CORNET B           DIR TYPE  : BOTH |
| ACD THRESHOLD   :     *   TRACENO      :                0   USAGE TYPE: TERR |
| ALLOCATED TO AT LEAST ONE ROUTE                                     GDTR RULE :    0 |
| SELECTION        :  LOW   CFBLOCK      : DISABLED                               |
| THE FOLLOWING PORTS (LTG-LTU-SLOT-CIRCUIT) ARE ALLOCATED:              |
+-----+
| 1- 2- 25- 1| 1- 2- 25- 2| 1- 2- 25- 3| 1- 2- 25- 4| 1- 2- 25- 5| 1- 2- 25- 6|
+-----+
| 1- 2- 25- 7| 1- 2- 25- 8| 1- 2- 25- 9| 1- 2- 25-10| 1- 2- 25-11| 1- 2- 25-12|
+-----+
| 1- 2- 25-13| 1- 2- 25-14| 1- 2- 25-15| 1- 2- 25-16| 1- 2- 25-17| 1- 2- 25-18|
+-----+
| 1- 2- 25-19| 1- 2- 25-20| 1- 2- 25-21| 1- 2- 25-22| 1- 2- 25-23| - - - |
+-----+
```

AMO-TGACC-10 TRUNK GROUP ACCESS CODE



Below is the configuration of the LCR Route that the QSIG trunk is assigned to:

DIS-LROUT:41;

H500: AMO LROUT STARTED

LCR ROUTE DEFINITION TABLE

ROUTENUM = 41	SCHED A = X	AORT =	INFORMATION	
ROUTEELE = 1	B =	AUTH = 1	TRANS CAP = S3V	
BEARER = ONE	C =	ONHKQ = Y	TRKSIG = COR	
BANDWTH = 1	D =	OFFHKQ = Y	SCCID =	
TRUNKGRP = 41	E =	ODRNUM = 1	SVCVCE = NON	
MASTGRP = 3	F =	APLTYP = VFD	SVCN-V = NON	
ROUTSERV = N	G =		FACNUM =	
	H =			

ROUTENUM = 41	SCHED A = X	AORT =	INFORMATION	
ROUTEELE = 2	B =	AUTH = 1	TRANS CAP = S3V	
BEARER = ONE	C =	ONHKQ = Y	TRKSIG = COR	
BANDWTH = 1	D =	OFFHKQ = Y	SCCID =	
TRUNKGRP = 41	E =	ODRNUM = 1	SVCVCE = NON	
MASTGRP = 3	F =	APLTYP = VD	SVCN-V = NON	
ROUTSERV = N	G =		FACNUM =	
	H =			

END OF LCR ROUTE DEFINITION TABLE DISPLAY

AMO-LROUT-10 ROUTE DEFINITION DETERMINATION PACKAGE



Below is the configuration of the LCR Outdial Rule assigned to the ISDN PRI trunks in the PBX:

DIS-LODR;

H500: AMO LODR STARTED

<< DISPLAY LCR OUTDIAL RULE >>

ODR NO	COMMAND	BRANCH VALUE
1	ECHOALL	
	END	

----- END OF DISPLAY -----

AMO-LODR -10 AMO LCR ODR FOR SWITCHING UNIT

DISPLAY COMPLETED;

Below is the configuration of one of the B-channels of the PSTN trunk:

DISP-TCSU:1-2-103-1,,,,;

H500: AMO TCSU STARTED

+-----+					
	PEN: 1- 2-103- 1	INS: Y	BOARD: TMDN64P	DEV: PRIB	TGRP: 40
+-----+					
	TRKID : 0040	TCCID :			
	CCT : T1PRI /0040				
	ACDATA : 0	DITIDX : 0	LOCANA :		
	ATNTYP : ISDN	DPLN : 0	REMANA :		
	COPNO : 0	ITR : 0	SIDANI : N		
	COSNO : 75	LCRCOSD : 5	SRTIDX : 3		
	COTNO : 0	LCRCOSV : 5	TRTBL : DIDCR		
	DEDSVC : NONE	FACILITY : *			
+-----+					

AMO-TCSU -120 TRUNK CONFIGURATION, SWITCHING UNIT

DISPLAY COMPLETED;



Below is the configuration of the D-channel assigned to the PSTN trunk

DISP-TCSU:1-2-103-24,;

H500: AMO TCSU STARTED

```
+-----+
| PEN: 1- 2-103-24  INS: Y   BOARD: TMDN64P      DEV: PRID      |
+-----+
| TCCID  :                                                  |
| CCT    : T1PRI                                           |
|
| ACDATA : 0          DEDSCC  :          INTERFID : 0          |
| COPNO  : 5          DITIDX  :          ITR       : 0          |
| COTNO  : 6          DPLN    : 0          PROTOCOL : NI2        |
|
| TMR301 : 300 SEC.   TMR308  : 4 SEC.   TMR313  : 4 SEC.      |
| TMR303 : 4 SEC.     TMR309  : 90 SEC.  TMR316  : 30 SEC.      |
| TMR305 : 30 SEC.    TMR310  : 30 SEC.  TMR322  : 4 SEC.      |
| TDELAY : 3000 MSEC. BEARER: ONE                                     |
| NCT    : N          TNCT    :                                                  |
+-----+
```

AMO-TCSU -185 TRUNK CONFIGURATION, SWITCHING UNIT

DISPLAY COMPLETED;

Below is the configuration of the LCR Route assigned to the PSTN trunk:

DISP-LROUT:40;

H500: AMO LROUT STARTED

LCR ROUTE DEFINITION TABLE

```
+-----+
| ROUTENUM = 40          SCHED A = X  AORT   =          INFORMATION      |
| ROUTEELE = 1          B =          AUTH   = 1      TRANS CAP = SU3V      |
| BEARER    = ONE        C =          ONHKQ  = Y      TRKSIG = PRI          |
| BANDWTH   = 1          D =          OFFHKQ = Y      SCCID  =              |
| TRUNKGRP  = 40          E =          ODRNUM = 1      SVCVCE = NON          |
| MASTGRP   = 1          F =          APLTYP = VFD     SVCN-V = NON          |
| ROUTSERV  = N          G =                                  FACNUM =        |
|
|                      H =                                  |
+-----+
```

END OF LCR ROUTE DEFINITION TABLE DISPLAY

AMO-LROUT-185 ROUTE DEFINITION DETERMINATION PACKAGE

DISPLAY COMPLETED;



Below is the configuration of the T1 trunk board assigned to the PSTN trunk:

DISP-BCSU:TMD,1,2,103;

H500: AMO BCSU STARTED

```
-----
| DETAILS OF TMD BOARD AT ADDRESS (LTG.LTU.SLOT) = 1. 2.103 |
|
| CABTYP = 1          TIMTYP = SYST          SIGTYP = MOS      |
| FRAME = ESF         TABS = NO             FCTID = 2         |
| BI8SUB = YES        BIVDET = NO           |
|-----|
| RDRATIO = 6         RDTH = 2500          RDQUAL = 15000      |
| YLSEND = 5000       YLTH = 400          YLQUAL = 100        |
| LOS = 150           AOS = 4000          |
| SESDISTH = 10       SESREQTH = 10        |
| OESDISTH = 30        OESDISIN = 24-00-00 |
| OESREQTH = 4         OESREQIN = 04-00-00 |
|-----|
| NETUSR = USER      ACKTIM = 1000        DLVTIM = 30000      |
| OCTMAX = 260        RETMAX = 3           WINDOW = 1         |
| CRIDC =             TTSC =              NSFIV =            |
| NSFTSC =            PFDGT =              |
|-----|
| IGN = 0             IID = 1              |
|-----|
```

AMO-BCSU -120 BOARD CONFIGURATION, SWITCHING UNIT

DISPLAY COMPLETED;



Below is the configuration of the trunk group assigned to the PSTN trunk:

DISP-TGACC:40;

H500: AMO TGACC STARTED

```
+-----+
| TGRP NUMBER      :    40   TGRP NAME  : T1PRI           /N   MAXIMUM NO:   23 |
| SUBGROUP NUMBER:    4   DEVICE TYPE: PRI B           DIR TYPE  : BOTH |
| ACD THRESHOLD   :    *   TRACENO    :                0   USAGE TYPE: TERR |
| ALLOCATED TO AT LEAST ONE ROUTE                               GDTR RULE :    0 |
| SELECTION       :  LOW   CFBLOCK    :  ENABLED                               |
| THE FOLLOWING PORTS (LTG-LTU-SLOT-CIRCUIT) ARE ALLOCATED:           |
+-----+
| 1- 2-103- 1| 1- 2-103- 2| 1- 2-103- 3| 1- 2-103- 4| 1- 2-103- 5| 1- 2-103- 6|
+-----+
| 1- 2-103- 7| 1- 2-103- 8| 1- 2-103- 9| 1- 2-103-10| 1- 2-103-11| 1- 2-103-12|
+-----+
| 1- 2-103-13| 1- 2-103-14| 1- 2-103-15| 1- 2-103-16| 1- 2-103-17| 1- 2-103-18|
+-----+
| 1- 2-103-19| 1- 2-103-20| 1- 2-103-21| 1- 2-103-22| 1- 2-103-23|  -  -  -  |
+-----+
```

AMO-TGACC-120 TRUNK GROUP ACCESS CODE

DISPLAY COMPLETED;



Below is the configuration of one of your test stations:

```
DISP-SCSU:2004,ALL;
H500:  AMO SCSU  STARTED
STNO      2004      NAME      DONALD DUCK                      ACT DEV
COS1        60      COSX        1    DIAL          -    DLIDX          -    DEVFUNC  OPTI
COS2        60      SPDC1       1    DPLN          0    TA             N    PEN       1-2-37-4
LCRCOSV1    12      SPDC2       2    HTLNIDX       -    TADLIDX        -    PUBSCR    2004
LCRCOSV2    12      SPDI       30   ITR           0    TAINS          -    ACTCDE    000000000000
LCRCOSD1    12      HANDSFR   Y    SPECL         -    ACCLASS        -    NTYPE     NAT
LCRCOSD2    12      INS        Y    PUGRP         -    QPRIOR         -    RPTYPE
DSSALERT    N      DTS        N    STD           1    FAXSERV        N/A   HDSTYPE   NONE
NWBALNO     -      CDIDX      -    WINKOFF       -    SEIZE          -    DTE DL   VER
CFWDV       N      CFWDD      N    DND           N    CALLWAIT       N    VCE DL   VER    0
VCP         OFF    MSGWLMP    -    PHONMAIL      N    COMGRP         0    DNIDSP          Y
MAINO       -      CUI        Y    KEYM          0    TSI            1    LOCODE        -
DCFVBUSY    N      API        N    EVMS          N    EVMSIDX        0    OPTITYPE     OADVPL
TATYPE      -                      TATYPE2      -                      FLASH        -
PATTERN     -      TFAGROUP   -    ATMADDR       -                      SPKALERT     Y
RELTEST     -
FIXED CFW1   -                      FIXED CFW2   -                      VAR CFW      -
STATION-HUNT N
UCD-HUNT     N
PILOT-HUNT   N
NIGHTVARIANT N
-----
AMO-SCSU -185      SUBSCRIBER CONFIGURATION IN THE SWU
DISPLAY COMPLETED;
```



Configuring the Local Cisco Unified CallManager Express (Cisco 3745)

LOCAL-3745#sho ver

Cisco IOS Software, 3700 Software (C3745-IPVOICE-M), Version 12.4(4)XC4, RELEAS

Synched to technology version 12.4(5.13)T

Technical Support: <http://www.cisco.com/techsupport>

Copyright (c) 1986-2006 by Cisco Systems, Inc.

Compiled Mon 24-Jul-06 19:48 by ealyon

ROM: System Bootstrap, Version 12.2(8r)T2, RELEASE SOFTWARE (fc1)

ROM: Cisco IOS Software, 3700 Software (C3745-IPVOICE-M), Version 12.4(4)XC4, R)

LOCAL-3745 uptime is 18 hours, 58 minutes

System returned to ROM by bus error at PC 0x6101C330, address 0xD0D0D0D

System image file is "flash:c3745-ipvoice-mz.124-4.XC4.bin"

Cisco 3745 (R7000) processor (revision 2.0) with 241664K/20480K bytes of memory.

Processor board ID JMX0813L0Z3

R7000 CPU at 350MHz, Implementation 39, Rev 3.3, 256KB L2, 2048KB L3 Cache

2 FastEthernet interfaces

48 Serial interfaces

2 Channelized T1/PRI ports

2 Voice FXS interfaces

2 Voice DID interfaces

DRAM configuration is 64 bits wide with parity enabled.

151K bytes of NVRAM.

62720K bytes of ATA System CompactFlash (Read/Write)

Configuration register is 0x0



LOCAL-3745#wr t

Building configuration...

Current configuration : 5012 bytes

!

version 12.4

service timestamps debug datetime msec

service timestamps log datetime msec

no service password-encryption

!

hostname LOCAL-3745

!

boot-start-marker

boot system flash:c3745-ipvoice-mz.124-4.XC4.bin

boot-end-marker

!

logging buffered 99999999 debugging

no logging console

enable password cisco

!

no aaa new-model

!

resource policy

!

no network-clock-participate slot 1

no network-clock-participate slot 3

voice-card 1

no dspfarm



```
!  
voice-card 3  
  
dspfarm  
  
!  
ip cef  
  
!  
!  
no ip dhcp use vrf connected  
  
!  
ip dhcp pool ephone3  
    host 172.20.15.203 255.255.255.0  
    client-identifier 0100.170e.c858.d4  
    default-router 172.20.15.1  
    option 150 ip 172.20.15.196  
  
!  
ip dhcp pool ephone4  
    host 172.20.15.204 255.255.255.0  
    client-identifier 0100.15f9.c856.1a  
    default-router 172.20.15.1  
    option 150 ip 172.20.15.196  
  
!  
ip dhcp pool ephone1  
    host 172.20.15.201 255.255.255.0  
    client-identifier 0100.15fa.0cb1.dc  
    default-router 172.20.15.1  
    option 150 ip 172.20.15.196  
  
!  
ip dhcp pool ephone2  
    host 172.20.15.202 255.255.255.0
```



```
client-identifier 0100.15fa.0cb5.d9
default-router 172.20.15.1
option 150 ip 172.20.15.196
!
ip dhcp pool ephone7
host 172.20.15.207 255.255.255.0
client-identifier 0100.15c6.96dd.6b
default-router 172.20.15.1
option 150 ip 172.20.15.196
!
!
no ip domain lookup
ip dhcp-server query lease retries 5
ip dhcp-server 172.20.15.196
isdn switch-type primary-qsig
!
!
voice call carrier capacity active
!
voice service pots
<supplementary-service qsig call-forward>1
!
voice service voip
qsig decode
allow-connections h323 to h323
allow-connections h323 to sip
allow-connections sip to h323
allow-connections sip to sip
```

¹ Omitted to force QSIG call forward by join (no reroute).



```
< no supplementary-service h450.2> 2
<no supplementary-service h450.3 > 2

supplementary-service h450.12

h323

sip

  registrar server expires max 600 min 60

!

!

!

voice register global

mode cme

source-address 172.20.15.196 port 5060

max-dn 100

load 7960-7940 POS3-07-5-00

tftp-path flash:

create profile sync 0841241944151508

!

voice register dn 1

  number 4011

< call-forward b2bua busy 2118> 3

<call-forward b2bua noan 2118 timeout 7> 4

  name Local IP1

  huntstop

!

voice register dn 2

  number 4012

  name Local IP2
```

² Inserted to force IP call forward by join (no reroute).

³ Inserted for call forward busy from SIP extension.

⁴ Inserted for call forward no reply from SIP extension.



```
huntstop

!

voice register dn 3

call-forward b2bua busy 3015

!

voice register pool 1

id mac 0015.FA0C.B1DC

type 7960

number 1 dn 1

max registrations 42

dtmf-relay rtp-nte

description Cisco7960

codec g711ulaw

!

voice register pool 2

id mac 0015.FA0C.B5D9

type 7960

number 1 dn 2

max registrations 42

dtmf-relay rtp-nte

description Cisco7960

codec g711ulaw

!

!

!

controller T1 3/0

framing esf

linecode b8zs

pri-group timeslots 1-24
```



```
!  
controller T1 3/1  
  
framing esf  
  
linecode b8zs  
  
pri-group timeslots 1-24  
  
!  
!  
!  
!  
  
interface FastEthernet0/0  
  
ip address 172.20.15.196 255.255.255.0  
  
duplex auto  
  
speed auto  
  
!  
  
interface FastEthernet0/1  
  
no ip address  
  
shutdown  
  
duplex auto  
  
speed auto  
  
!  
  
interface Serial3/0:23  
  
no ip address  
  
encapsulation hdlc  
  
isdn switch-type primary-qsig  
  
isdn overlap-receiving  
  
isdn incoming-voice voice  
  
no cdp enable  
  
!  
  
interface Serial3/1:23
```



```
no ip address

encapsulation hdlc

isdn switch-type primary-qsig

isdn overlap-receiving

isdn protocol-emulate network

isdn incoming-voice voice

isdn T310 120000

no cdp enable

!

ip route 0.0.0.0 0.0.0.0 172.20.15.1

!

ip http server

ip http authentication local

ip http path flash:

!

!

!

tftp-server flash:P003-07-5-00.bin

tftp-server flash:P003-07-5-00.sbn

tftp-server flash:P0S3-07-5-00.bin

tftp-server flash:P0S3-07-5-00.sb2

tftp-server flash:P0S3-07-5-00.loads

< tftp-server flash: any load file that is not on the phone and is needed >

< tftp-server slot0: any load file that is not on the phone and is needed>

!

control-plane

!

!

!
```



```
voice-port 1/0/0
timing digit 75
timing inter-digit 65
!
voice-port 1/0/1
!
voice-port 1/1/0
!
voice-port 1/1/1
!
voice-port 3/0:23
!
voice-port 3/1:23
!
!
dial-peer voice 3023 pots
destination-pattern 2...
incoming called-number ....
<clid restrict> 5
< supplementary-service qsig call-forward > 6
direct-inward-dial
port 3/0:23
forward-digits all
!
dial-peer voice 1 voip
preference 1
destination-pattern 4...
```

⁵ Inserted for CLID restrict cases only.

⁶ Omitted to force QSIG call forward by join (no reroute).



```
session target ipv4:172.20.15.159
```

```
dtmf-relay h245-alphanumeric
```

```
codec g711ulaw
```

```
no vad
```

```
!
```

```
dial-peer voice 5050 pots
```

```
destination-pattern 5050
```

```
direct-inward-dial
```

```
port 3/0:23
```

```
forward-digits all
```

```
!
```

```
dial-peer voice 5 pots
```

```
destination-pattern 5...
```

```
direct-inward-dial
```

```
port 3/0:23
```

```
forward-digits all
```

```
!
```

```
dial-peer voice 3700 pots
```

```
destination-pattern 37..
```

```
direct-inward-dial
```

```
port 3/0:23
```

```
forward-digits all
```

```
!
```

```
!
```

```
sip-ua
```

```
!
```

```
!
```

```
telephony-service
```

```
load 7960-7940 P003-07-5-00
```



```
max-ephones 25
max-dn 50
ip source-address 172.20.15.196 port 2000
max-conferences 8 gain -6
call-forward pattern .T
transfer-system full-consult
transfer-pattern .... <blind>7
create cnf-files version-stamp 7960 Sep 11 2006 16:53:04
!
!
ephone-dn 3 dual-line
number 4013
name Local IP3
< call-forward busy 2118>8
<call-forward noan 2118 timeout 7>9
huntstop channel
!
!
ephone-dn 4 dual-line
number 4014
name Local IP4
huntstop channel
!
!
ephone 3
mac-address 0017.0EC8.58D4
type 7961
```

⁷ Inserted to enable blind transfers, as opposed to early attended transfers.

⁸ Inserted for call forward busy from SCCP extension.

⁹ Inserted for call forward no reply from SCCP extension.



```
keep-conference
button 1:3
!
!
!
ephone 4
mac-address 0015.F9C8.561A
type 7970
keep-conference
button 1:4
!
!
!
line con 0
exec-timeout 0 0
line aux 0
line vty 0 4
exec-timeout 0 0
password cisco
login
transport input telnet
!
!
end
```

LOCAL-3745#



Configuring the Remote Cisco Unified CallManager Express (Cisco 2811)

REMOTE-2811#sho ver

Cisco IOS Software, 2800 Software (C2800NM-IPVOICE-M), Version 12.4(4)XC4, RELE)

Synched to technology version 12.4(5.13)T

Technical Support: <http://www.cisco.com/techsupport>

Copyright (c) 1986-2006 by Cisco Systems, Inc.

Compiled Mon 24-Jul-06 18:33 by ealyon

ROM: System Bootstrap, Version 12.4(1r) [hqluong 1r], RELEASE SOFTWARE (fc1)

ROM: Cisco IOS Software, 2800 Software (C2800NM-IPVOICE-M), Version 12.4(4)XC4,)

REMOTE-2811 uptime is 10 weeks, 22 hours, 30 minutes

System returned to ROM by power-on

System restarted at 16:23:28 UTC Thu Sep 7 2006

System image file is "flash:c2800nm-ipvoice-mz.124-4.XC4.bin"

Cisco 2811 (revision 53.51) with 251904K/10240K bytes of memory.

Processor board ID FHK0946F0MZ

2 FastEthernet interfaces

2 Voice FXS interfaces

DRAM configuration is 64 bits wide with parity enabled.

239K bytes of non-volatile configuration memory.

62592K bytes of ATA CompactFlash (Read/Write)

Configuration register is 0x2

REMOTE-2811#



REMOTE-2811#

REMOTE-2811#wr t

Building configuration...

Current configuration : 3553 bytes

!

! Last configuration change at 19:10:31 UTC Thu Nov 16 2006

! NVRAM config last updated at 19:10:33 UTC Thu Nov 16 2006

!

version 12.4

service timestamps debug datetime msec

service timestamps log datetime msec

no service password-encryption

!

hostname REMOTE-2811

!

boot-start-marker

boot system flash:c2800nm-ipvoice-mz.124-4.XC4.bin

boot-end-marker

!

enable password cisco

!

no aaa new-model

!

resource policy

!

!

!

ip cef



```
no ip dhcp use vrf connected
```

```
!
```

```
ip dhcp pool ephone5
```

```
host 172.20.15.205 255.255.255.0
```

```
client-identifier 0100.15fa.0cb7.46
```

```
default-router 172.20.15.1
```

```
option 150 ip 172.20.15.159
```

```
!
```

```
ip dhcp pool ephone6
```

```
host 172.20.15.206 255.255.255.0
```

```
client-identifier 0100.15fa.63bf.84
```

```
default-router 172.20.15.1
```

```
option 150 ip 172.20.15.159
```

```
!
```

```
!
```

```
no ip domain lookup
```

```
ip dhcp-server query lease retries 5
```

```
ip dhcp-server 172.20.15.159
```

```
!
```

```
!
```

```
voice-card 0
```

```
no dspfarm
```

```
!
```

```
!
```

```
!
```

```
voice service voip
```

```
qsig decode
```

```
allow-connections h323 to h323
```

```
allow-connections h323 to sip
```



allow-connections sip to h323

allow-connections sip to sip

supplementary-service h450.12

< **no supplementary-service h450.2** inserted here to force call by join>¹⁰

<**no supplementary-service h450.3** inserted here to force call by join>¹⁰

h323

sip

!

!

interface FastEthernet0/0

ip address 172.20.15.159 255.255.255.0

duplex auto

speed auto

!

interface FastEthernet0/1

no ip address

shutdown

duplex auto

speed auto

!

ip route 0.0.0.0 0.0.0.0 172.20.15.1

!

ip http server

!

!

!

tftp-server flash:P0030702T023.bin

¹⁰ Inserted to force IP call forward by join (no reroute).



```
tftp-server flash:P0030702T023.loads
```

```
tftp-server flash:P0030702T023.sb2
```

```
tftp-server flash:P0030702T023.sbn
```

```
< tftp-server flash: any load file that is not on the phone and is needed >
```

```
< tftp-server slot0: any load file that is not on the phone and is needed>
```

```
!
```

```
control-plane
```

```
!
```

```
!
```

```
!
```

```
voice-port 0/1/0
```

```
!
```

```
voice-port 0/1/1
```

```
!
```

```
!
```

```
!
```

```
!
```

```
!
```

```
dial-peer voice 1 voip
```

```
destination-pattern 2...
```

```
session target ipv4:172.20.15.196
```

```
dtmf-relay h245-alphanumeric
```

```
codec g711ulaw
```

```
!
```

```
dial-peer voice 3011 voip
```

```
destination-pattern 3011
```

```
session target ipv4:172.20.15.196
```

```
dtmf-relay h245-alphanumeric
```




```
codec g711ulaw
```

```
!
```

```
dial-peer voice 3014 voip
```

```
destination-pattern 3014
```

```
session target ipv4:172.20.15.196
```

```
dtmf-relay h245-alphanumeric
```

```
codec g711ulaw
```

```
!
```

```
dial-peer voice 3012 voip
```

```
destination-pattern 3012
```

```
session target ipv4:172.20.15.196
```

```
dtmf-relay h245-alphanumeric
```

```
codec g711ulaw
```

```
!
```

```
dial-peer voice 3013 voip
```

```
destination-pattern 3013
```

```
session target ipv4:172.20.15.196
```

```
dtmf-relay h245-alphanumeric
```

```
codec g711ulaw
```

```
!
```

```
dial-peer voice 4300 voip
```

```
destination-pattern 43..
```

```
session target ipv4:172.20.15.196
```

```
dtmf-relay h245-alphanumeric
```

```
codec g711ulaw
```

```
!
```

```
dial-peer voice 5214 voip
```

```
destination-pattern 5...
```

```
session target ipv4:172.20.15.196
```



```
dtmf-relay h245-alphanumeric
codec g711ulaw
!
dial-peer voice 2 voip
destination-pattern 4...
session target ipv4:172.20.15.196
dtmf-relay h245-alphanumeric
codec g711ulaw
!
dial-peer voice 5 voip
destination-pattern 5...
session target ipv4:172.20.15.196
!
dial-peer voice 3700 voip
destination-pattern 37..
session target ipv4:172.20.15.196
dtmf-relay h245-alphanumeric
codec g711ulaw
!
!
sip-ua
!
!
telephony-service
load 7960-7940 P0030702T023
max-ephones 25
max-dn 50
ip source-address 172.20.15.159 port 2000
max-conferences 8 gain -6
```



```
call-forward pattern .T
transfer-system full-consult
transfer-pattern .... <blind>11
create cnf-files version-stamp Jan 01 2002 00:00:00
!
!
ephone-dn 5 dual-line
number 4015
name Remote IP5
<call-forward busy 3603>12
< call-forward noan 3603 timeout 7>13

!
!
ephone-dn 6 dual-line
number 4016
name Remote IP6

!
!
ephone 5
mac-address 0015.FA0C.B746
type 7960
keep-conference
button 1:5

!
!
!
```

¹¹ Inserted to enable blind transfers, as opposed to early attended transfers.

¹² Inserted for call forward busy from SCCP extension.

¹³ Inserted for call forward no reply from SCCP extension.



```
ephone 6
mac-address 0015.FA63.BF84
type 7960
keep-conference
button 1:6
!
!
!
line con 0
line aux 0
line vty 0 4
password cisco
login
!
scheduler allocate 20000 1000
!
end

REMOTE-2811#
```



Acronyms

Acronym	Definitions
BRI	Basic Rate ISDN
CAMA	Centralized Automatic Message Accounting
CAS	Channel Associated Signaling
CFB	Call Forward when Busy
CFNR	Call Forward when No Reply
CFU	Call Forward Unconditional
CO	Central Office
FGD	Feature Group "D"
FXO	Foreign Exchange – Office
FXS	Foreign Exchange – Station
IOS	Internetworking Operating System
MCID	Malicious Caller ID
MGCP	Media Gateway Control Protocol
MoH	Music on Hold
MWI	Message Waiting Indication
PBX	Private Branch Exchange
PRI	Primary Rate ISDN
PSAP	Public Service Access Point
SIP	Session Initiation Protocol
ToH	Tone on Hold



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