



Siemens HiPath 4000 Release 1.0 using E1 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 HiPath 4000 Release 1 PBX and Cisco Unified CallManager Express Release 4.0(3) using a Cisco 3845 voice gateway with QSIG protocol.

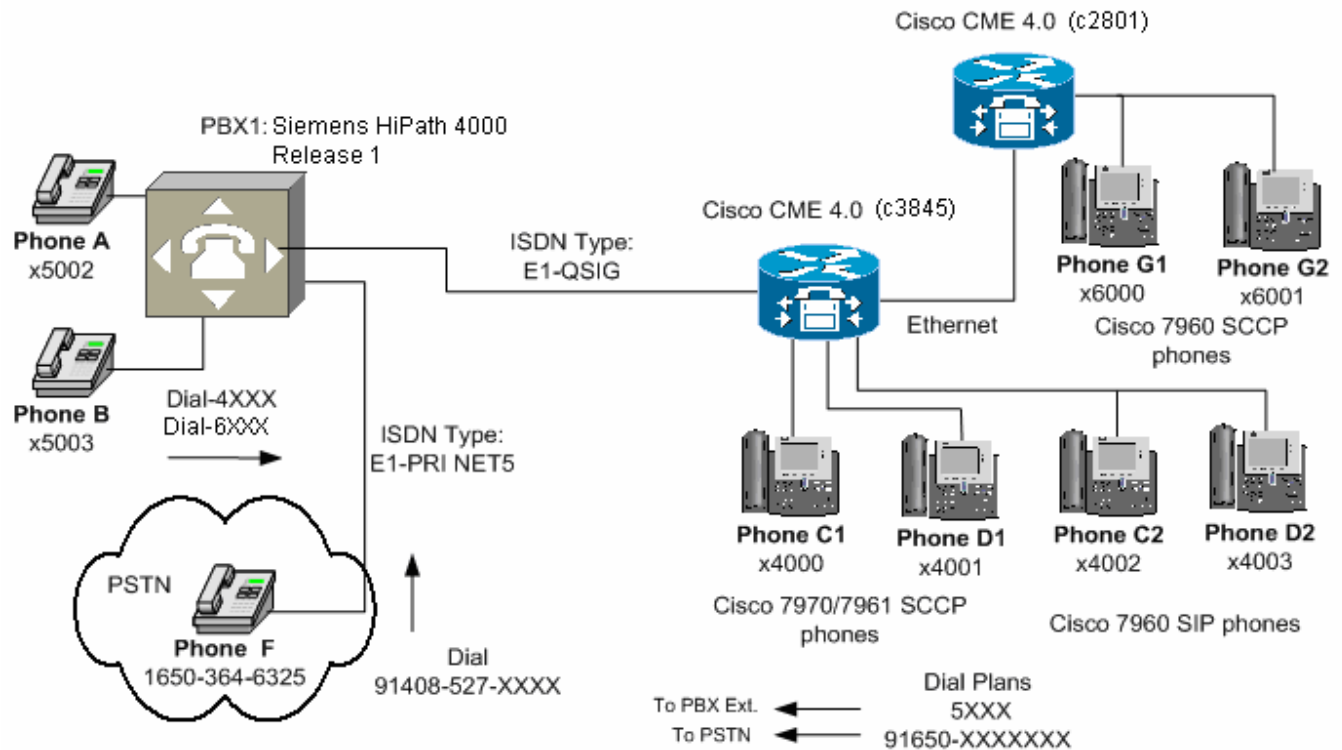
The network topology diagram (Figure 1) shows the test setup for end-to-end interoperability with Cisco Unified CallManager Express Release 4.0(3) connected to the PBX via the 3845 E1 QSIG link. The 3845 IOS voice gateway was connected via H.323 to a Cisco 2801 IOS voice gateway. The two gateways were 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 figure 1. A NM-HDV and VWIC-2MFT-E1 were used for the E1 QSIG interface. Calls were made to test basic call, caller ID, conference, transfer, Callforward, call back and reroute features.

This Application Note uses the 3845 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. CME 4.0(3) supports SIP end-points with limited number of features.

Network Topology

Figure 1. Network Topology or Test Setup – basic calls configuration.





Limitations

Basic Calls

Cisco Unified CallManager Express does not support overlap sending. It does support overlap receiving.

Calling Name Restriction is not supported for calls originating from from Cisco Unified CallManager Express end-points.

Alerting Name is not supported on calls between PBX and Cisco Unified IP Phone running SIP.

Called/Connected Name is not supported on calls between PBX and Cisco Unified IP Phone running SIP.

Call Transfers

On a call originating from SCCP phone calling a SIP phone and the SIP phone, after answering the call, attempts to transfer the call to a PBX extension the call transfer fails. (e.g. Phone C1 calls Phone C2, C2 attempts to transfer to Phone A. See figure 1).

Call Transfer attended; When a PBX phone calls a SCCP phone or a SIP phone and the call is transferred to another SIP phone the original Calling name and number are not updated. The same behavior is seen during an early attended transfer between a SCCP phone and a SIP phone, but the limitation is not seen during an early attended transfer between Phone a SIP phone to SIP phone. (e.g. Phone A calls Phone C1 or C2, xfer to Phone D2. See figure 1.)

Call Transfer attended, early attended or blind; When the PBX phone calls a SCCP phone or a SIP phone and the call is trasferred to a SSCP phone or SIP phone the connected number does not update on PBX phone, after the trasferred call is answered.

For all supervised and early-attended Network/External call transfers, the original calling name and number are not displayed on the final destination. The limitation is due to the CME not supporting the QSIG operation "CallTransferComplete" carried in the Q931 FACILITY message for name/number updating

Call Forwards

For call forward CFU CFB and CFNR, the forwarding called name and number are not displayed on the final destination when the final destination is a SIP phone. This is a CME to SIP phone limitation.

For call forwards, the connected name and number are not updated on the originating phone when the originating phone is a SIP phone. On SCCP, phones only connected name is updated. This is a CME to PBX interworking limitation

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 (e.g., A calls C1, and C1 forwards to D1 or D2), Cisco Unified CallManager Express performs a reroute, even if a reroute is disabled, and even though a reroute is not in order (i.e., there is no "hairpin" or "trombone").

For calls hairpinned at a SIP extension (PBX phone calls Cisco Unified CallManager Express SIP phone that forwards back to another PBX phone), the call completes, but Cisco Unified CallManager Express does not perform a reroute, even if reroute is enabled.

Calls initiated from the PBX side using overlap-dialing and destined to a Cisco Unified CallManager Express SIP or SCCP phone that is Callforwarded back to another PBX phone will not perform a qsig reroute, even if reroute is enabled and the call is eligible for a reroute.

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 3845 IOS voice gateway

NM-HDV

VWIC-2MFT-E1

Cisco 2801 IOS voice gateway

(4) Cisco Unified IP phone 7960s

(2) Cisco Unified IP phone 7961G

(1) Siemens HiPath 40000

DIU-N2

Software Requirements

Cisco Unified CallManager Express Release 4.0(3)

Cisco IOS Software, 3800 Software (C3845-IPVOICE-M), Version 12.4(4)XC4

Cisco IOS Software, 2801 Software (C2801-IPVOICE-M), Version 12.4(4)XC4

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 P0S3-07-5-00

Cisco 7960 IP phone boot load PC030301

C1 – 7961G – SCCP

Cisco7961G IP phone load file: TERM41.7-0-3-0S.loads

Cisco 7961G IP phone app load ID: Jar41.2-9-1-45.sbn

Cisco 7961G IP phone boot load ID: 7961G_64-020704128Amd64meg.bin



Features

Features Supported

Basic Call, ENBLOC and Overlap sending (From PBX to Cisco Unified CallManager Express only. See Limitations section for details.)

CLIP-Calling Line (Number) Identification Presentation and CNIP-Calling Name Identification Presentation on Basic Calls

CLIR-Calling Line (Number) Identification Restriction and CNIR-Calling Name Identification Restriction on Basic Calls (From PBX to Cisco Unified CallManager Express only. See Limitations section for details.)

COLP-Connected Line (Number) Identification Presentation and CONP-Connected Name Identification Presentation on Basic Calls (SCCP phones only. See Limitations section for details.)

Alerting Name on Basic Calls (SCCP phones only. See Limitations section for details.)

Tandem PSTN call

Consultation Transfer – Local and Network/External (See Limitations section for details.)

Early Attended Transfer – Local and Network/External (See Limitations section for details.)

Blind Transfer – Local and Network/External (From Cisco Unified CallManager Express only. See Limitations section for details)

Call Forward by forward unconditional, busy and No Reply by Join (forward switch) – Local and Network/External (See Limitations section for details.)

Call Forward unconditional, busy and no reply by Reroute – Network/External (See Limitations section for details.)



Features Not Supported

Basic Call, Overlap sending originating from Cisco Unified CallManager Express IP phone to PBX

CNIR-Calling Name Identification Restriction from Cisco Unified CallManager Express to PBX

COLR- Connected Line (Number) Identification Restriction

CONR- Connected Name Identification Restriction

CONP-Connected Name Identification Presentation (for calls between PBX and Cisco Unified IP Phones running SIP)

Alerting Name (for calls between PBX and Cisco Unified IP Phones running SIP)

H323/QSIG tandem transfers via SIP phone

Name and Number updates on transferred calls

CLIP-Calling Line (Number) Identification Presentation on Forwarded Calls to a PBX station.

Connected Name and Number updates on forwarded calls

Call Forward by Reroute for QSIG "trombone" from a Cisco Unified CallManager Express SIP extension

Call Forward by Reroute during a call originated by overlap sending/receiving

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 (accomplished by consultation transfer)

Path Replacement for Call Diversion by Forward Switch



Configuration

Configuring the Siemens HiPath 4000 PBX

1. Add the new access code to Dialing Plans using WABE + LDPLN.
2. Add the new trunk board using BCSU.
3. Configure Class of Trunk using COT.
4. Configure Class of Parameter for device handler using COP.
5. Configure Class of Service using COSSU.
6. Add the new trunk group access code using BUEND.
7. Configure trunk using TDCSU.
8. Configure Reference Clock using REFTA.
9. Configure trunk Least Cost Routing using LDAT + RICHT.
10. Configure LCR Out-dial Rules using LODR.
11. Enable In-Band DTMF signaling for the Digital Stations using SBCSU.
12. Configure Digital Station for MWI application.
13. Configure Message Center's Service Access Number for MWI application.

DPLN

```
<dis-wabe:gen;  
DIS-WABE:GEN;  
H500: AMO WABE STARTED
```

DIGIT INTERPRETATION		VALID FOR ALL DIAL PLANS			
CODE		CALL PROGRESS STATE			
		1	11111	11112	22
		0	12345	67890	12345 67890 12
					DIGIT ANALYSIS RESULT
					RESERVED/CONVERT DNI/ADD-INFO
					*=OWN NODE
0		. *****	. *****	** ..	CO
001	- 009	*	NETRTE
111		. *****	*****	** ..	TIE
12		. *****	*****	** ..	TIE
13	- 14	. *****	*****	** ..	TIE
21	*	KNOVRKY
22	*	DNDKY
222		. *****	*****	** ..	TIE
23	*	FWDKY
24	*	MBKY
25	*	MSGRKY
26	*	DAKY
27	*	DSSKY
28	*	VCRKY
29	*	VCKY
30	*	CONFKY

DIGIT INTERPRETATION		VALID FOR ALL DIAL PLANS			
CODE		CALL PROGRESS STATE			
		1	11111	11112	22
		0	12345	67890	12345 67890 12
					DIGIT ANALYSIS RESULT
					RESERVED/CONVERT DNI/ADD-INFO
					*=OWN NODE



3000 - 3010	. ***** **... .. *	STN	DESTNO 30 DNNO 0- 0-222
3011 - 3020	. ***** **... .. *	STN	DESTNO 31 DNNO 0- 0- 31
3021 - 3030	. ***** **... .. *	STN	DESTNO 32 DNNO 0- 0- 32
3031 - 3040	. ***** **... .. *	STN	DESTNO 33 DNNO 0- 0- 33
3041 - 3050	. ***** **... .. *	STN	DESTNO 35 DNNO 0- 0- 35
31* ..	NAMEKY	
32* ..	PARKKY	

DIGIT INTERPRETATION VALID FOR ALL DIAL PLANS

CODE	CALL PROGRESS STATE 1 11111 11112 22	DIGIT ANALYSIS RESULT	RESERVED/CONVERT DNI/ADD-INFO *=OWN NODE
0 12345 67890 12345 67890 12			
33* ..	CCKY	
34* ..	HTKY	
35* ..	STKY	
36* ..	REMKY	
36 - 37	. ***** **... .. *	CO	
38* ..	TIMEKY	
39	. ***** **... .. *	TIE	
4000 - 4050	. ***** **... .. *	STN	DESTNO 111 DNNO 0- 0-111
4051 - 4566	. ***** **... .. *	STN	DESTNO 222 DNNO 0- 0-222
4567	. ***** **... .. *	STN	DESTNO 34 DNNO 0- 0-200

DIGIT INTERPRETATION VALID FOR ALL DIAL PLANS

CODE	CALL PROGRESS STATE 1 11111 11112 22	DIGIT ANALYSIS RESULT	RESERVED/CONVERT DNI/ADD-INFO *=OWN NODE
0 12345 67890 12345 67890 12			
4568 - 4999	. ***** **... .. *	STN	DESTNO 222 DNNO 0- 0-222
5000 - 5040	. ***** **... .. *	STN	DESTNO 0 DNNO 0- 0-555*
5500 - 5501	. ***** **... .. *	STN	DESTNO 56 DNNO 0- 0-560
555	. ***** **... .. *	OWNNODE	
560	. ***** **... .. *	TIE	
59	. ***** **... .. *	TIE	
6000 - 6009	. ***** **... .. *	STN	R DESTNO 0 DNNO 0- 0-555*

DIGIT INTERPRETATION VALID FOR ALL DIAL PLANS

CODE	CALL PROGRESS STATE 1 11111 11112 22	DIGIT ANALYSIS RESULT	RESERVED/CONVERT DNI/ADD-INFO
------	---	--------------------------	----------------------------------



				0 12345 67890 12345 67890 12	RESULT	*=OWN NODE
7000	-	7002	.	***** **...	STN	DESTNO 56 DNNO 0- 0-560
8000	-	8050	.	***** **...	STN	DESTNO 222 DNNO 0- 0-222
8060			.	***** **...	TIE	
8070			.	***** **...	TIE	
83			.	***** **...	SPDC1	
84			.	***** **...	SPDC2	
88			.	***** **...	SCONSI	R
89			.	***** **...	SCONSCO	R
9			.	***** **...	TIE	
*13			.	***** **...	AHTVCE	
*15			.	***** **...	SPLIT	
*16			.	***** **...	AREM	
*17			.	***** **...	TRACE	
DIGIT INTERPRETATION VALID FOR ALL DIAL PLANS						
				CALL PROGRESS STATE	DIGIT ANALYSIS	RESERVED/CONVERT
CODE				1 11111 11112 22	RESULT	DNI/ADD-INFO
				0 12345 67890 12345 67890 12		*=OWN NODE
*18			.	***** **...	ACOSX	
*19			.	***** **...	KNOVR	
*20			.	***** **...	ADND	
*25			.	***** **...	FWDTERM	
*29			.	***** **...	AFFWDVCE	

AMO-WABE -111 DIALLING PLANS, FEATURE ACCESS CODES
DISPLAY COMPLETED;

Enbloc Sending Dial Plan, DPLN

<DISPLAY-LDPLN:TYPE=LDP,LDPNO=4;
DISPLAY-LDPLN:TYPE=LDP,LDPNO=4;
H500: AMO LDPLN STARTED

LDPNO :	4	LDP : 111-X	
		SPC : 22	
		FDSFIELD : 0	SDSFIELD : 0 PINDP : N
DPLN	LROUTE	LAUTH	
0	1	1	
1	1	1	
2	1	1	
3	1	1	
4	1	1	
5	1	1	
6	1	1	
7	1	1	
8	1	1	
9	1	1	
10	1	1	
11	1	1	
12	1	1	
13	1	1	
14	1	1	



```
| 15 | 1 | 1 |
+-----+
```

```
AMO-LDPLN-111      ADMINISTRATION LCR DIALPLAN
DISPLAY COMPLETED;
```




Access code for Overlap Sending Dial Plan, DPLN

```
<DISPLAY-LDPLN:TYPE=LDP,LDPNO=12;  
DISPLAY-LDPLN:TYPE=LDP,LDPNO=12;  
H500: AMO LDPLN STARTED
```

LDPNO : 12			LDP : 59-X
			SPC : 22
			FDSFIELD : 0 SDSFIELD : 0 PINDP : N
DPLN	LROUTE	LAUTH	
0	27	1	
1	27	1	
2	27	1	
3	27	1	
4	27	1	
5	27	1	
6	27	1	
7	27	1	
8	27	1	
9	27	1	
10	27	1	
11	27	1	
12	27	1	
13	27	1	
14	27	1	
15	27	1	

```
AMO-LDPLN-111 ADMINISTRATION LCR DIALPLAN  
DISPLAY COMPLETED;
```

PSTN Dial Plan, DPLN

```
<DISPLAY-LDPLN:TYPE=LDP,LDPNO=5;  
DISPLAY-LDPLN:TYPE=LDP,LDPNO=5;  
H500: AMO LDPLN STARTED
```

LDPNO : 5			LDP : 9-1650-XXXXXXX
			SPC : 22
			FDSFIELD : 0 SDSFIELD : 0 PINDP : N
DPLN	LROUTE	LAUTH	
0	165	1	
1	165	1	
2	165	1	
3	165	1	
4	165	1	
5	165	1	
6	165	1	
7	165	1	
8	165	1	
9	165	1	
10	165	1	
11	165	1	
12	165	1	
13	165	1	
14	165	1	
15	165	1	



AMO-LDPLN-111 ADMINISTRATION LCR DIALPLAN
DISPLAY COMPLETED;

BCSU

PRI Board

<DISPLAY-BCSU:TYPE=TBL,LTG=1,LTU=3,SLOT=25;
DISPLAY-BCSU:TYPE=TBL,LTG=1,LTU=3,SLOT=25;
H500: AMO BCSU STARTED

```
ADDRESS : LTG 1 LTU 3 SOURCE GROUP 1
```

	ASSIGNED	MODULE	FCT	HWY	INSERTED			MODULE
PEN	MODULE	TYPE	ID	BDL	MODULE	STATE	HW-INFO	STATUS
25	Q2196-X	DIU-N2	1	A	Q2196-X	1	-06 -	READY

AMO-BCSU -111 BOARD CONFIGURATION, SWITCHING UNIT
DISPLAY COMPLETED;
<

PSTN board

<DISPLAY-BCSU:TYPE=TBL,LTG=1,LTU=2,SLOT=79;
DISPLAY-BCSU:TYPE=TBL,LTG=1,LTU=2,SLOT=79;
H500: AMO BCSU STARTED

```
ADDRESS : LTG 1 LTU 2 SOURCE GROUP 1
```

	ASSIGNED	MODULE	FCT	HWY	INSERTED			MODULE
PEN	MODULE	TYPE	ID	BDL	MODULE	STATE	HW-INFO	STATUS
79	Q2196-X	DIU-N2	1	A	Q2196-X	1	-06 -	READY

AMO-BCSU -111 BOARD CONFIGURATION, SWITCHING UNIT
DISPLAY COMPLETED;

Class of Trunk, COT

<dis-cot:22;
DIS-COT:22;
H500: AMO COT STARTED

```
COT: 22 INFO:
DEVICE: INDEP            SOURCE: DB
PARAMETER:
  PRIORITY FOR AC WILL BE DETERMINED FROM MESSAGE            PRI
  RECALL IF USER HANGS UP IN CONSULTATION CALL            RCL
  TRUNK CALL TRANSFER            XFER
  TRUNK SIGNALING ANSWER            ANS
  CHANGEOVER FROM HOLD TO RING TONE            CHRT
  KNOCKING OVERRIDE POSSIBLE            KNOR
  CALL EXTEND FOR BUSY, RING OR CALL STATE            CEBC
  NETWORKWIDE AUTOMATIC CALLBACK ON BUSY            CBBN
  NETWORKWIDE AUTOMATIC CALLBACK ON FREE            CBFN
  DON'T RELEASE CALL TO BUSY HUNT GROUP            BSHT
  CONNECTION TO ROUTE OPTIMIZATION NODE            ROPT
  TSC-SIGNALING FOR NETWORKWIDE FEATURES (MANDATORY)            TSCS
  INCOMING CDR BY ZONE OR FROM LINE            ICZL
  AOC PER CALL (AUTOMATICAL OR ON REQUEST), MAND. CORNET-NQ            AOCC
  LINE WITH IMPLICIT NUMBERS            LINO
  NO TONE            NTON
```




```
AMO-COT -111          CLASS OF TRUNK FOR CALL PROCESSING
DISPLAY COMPLETED;
```

**For Call Forwarding by Reroute need to add FNAN and FWDN to COT22
For Path Replacement need to add ROPT to COT 22 on the Hipath 4000.**

```
<CHANGE-COT:COTNO=22,COTTYPE=COTADD,PAR=FNAN&FWDN;
CHANGE-COT:COTNO=22,COTTYPE=COTADD,PAR=FNAN&FWDN&ROPT;
H500: AMO COT   STARTED
H07: CHANGED COT STILL LINKED WITH FOLLOWING TRUNKS:
```

COT	TRUNK	BCGR	DEVICE
22	1- 3- 55- 0	1	CDGCONN
	1- 3- 55- 1	1	CDGCONN
	1- 3- 25- 0	1	S2CONN
	1- 3- 25- 1	1	S2CONN

```
H06: COT 22 CHANGED
```

```
AMO-COT -111          CLASS OF TRUNK FOR CALL PROCESSING
CHANGE COMPLETED;
```

```
<DISPLAY-COT:COTNO=22;
DISPLAY-COT:COTNO=22;
H500: AMO COT   STARTED
```

COT: 22 INFO:	
DEVICE: INDEP	SOURCE: DB
PARAMETER:	
PRIORITY FOR AC WILL BE DETERMINED FROM MESSAGE	PRI
RECALL IF USER HANGS UP IN CONSULTATION CALL	RCL
TRUNK CALL TRANSFER	XFER
TRUNK SIGNALING ANSWER	ANS
CHANGEOVER FROM HOLD TO RING TONE	CHRT
KNOCKING OVERRIDE POSSIBLE	KNOR
CALL EXTEND FOR BUSY, RING OR CALL STATE	CEBC
NETWORKWIDE AUTOMATIC CALLBACK ON BUSY	CBBN
NETWORKWIDE AUTOMATIC CALLBACK ON FREE	CBFN
NETWORKWIDE CALL FORWARDING PERMITTED	FWDN
NETWORKWIDE FORWARDING NO-ANSWER	FNAN
DON'T RELEASE CALL TO BUSY HUNT GROUP	BSHT
CONNECTION TO ROUTE OPTIMIZATION NODE	ROPT
TSC-SIGNALING FOR NETWORKWIDE FEATURES (MANDATORY)	TSCS
INCOMING CDR BY ZONE OR FROM LINE	ICZL
AOC PER CALL (AUTOMATICAL OR ON REQUEST), MAND. CORNET-NQ	AOCC
LINE WITH IMPLICIT NUMBERS	LINO
NO TONE	NTON

```
AMO-COT -111          CLASS OF TRUNK FOR CALL PROCESSING
DISPLAY COMPLETED;
```

Class of Parameter for Device Handler, COP

```
<DISPLAY-COP:COPNO=21;
DISPLAY-COP:COPNO=21;
H500: AMO COP   STARTED
```

COP: 21 INFO:	
DEVICE: INDEP	SOURCE: DB
PARAMETER:	
LINE WITH END-OF-DIAL	EOD
SPECIAL MODE	SFRM
CODE CALLING RELEASE AFTER EVERY TASK	CCR
REGISTRATION OF LAYER 3 ADVISORIES	L3AR

TA

TA

```

AMO-COP    -111          CLASS OF PARAMETER FOR DEVICE HANDLER
DISPLAY COMPLETED;

```

Class of Service, COSSU

```
<DISPLAY-COSSU:TYPE=COS,COS=32;  
DISPLAY-COSSU:TYPE=COS,COS=32;  
H500:  AMO COSSU STARTED
```

COS	VOICE	FAX	DTE
32	>		
	TA TNOTCR	NOCO NOTIE	TA TNOTCR BASIC MSN CDRINT MULTRA

AMO-COSSU-111 CLASSES OF SERVICE
DISPLAY COMPLETED;

```
<DISPLAY-COSSU:TYPE=LCOSV,LCOSV=1;
DISPLAY-COSSU:TYPE=LCOSV,LCOSV=1;
H500: AMO COSSU STARTED
```

LCOS	LAUTH												COPIN
V	1		2		3		4		5		6		NUM
	1	2	3	4	5	6	7	8	9	0	1	2	3
	>SERVICE INFORMATION												
1	X	0
	>LCR ATTENDANT FOR VOICE												

AMO-COSSU-111 CLASSES OF SERVICE
DISPLAY COMPLETED;



Trunk Group Access Code, BUEND

<DISPLAY-BUEND:TGRP=26,FORMAT=L;

DISPLAY-BUEND:TGRP=26,FORMAT=L;

H500: AMO BUEND STARTED

----- FORMAT = L -----					
TGRP NUMBER :	26	TGRP NAME :	PRI ECMA 3	MAXIMUM NO. :	30
		CHARCON :	NEUTRAL		
SUBGROUP NO.:	9	DEVICE TYPE :	S2CONN	TRACENO :	0
RESERVED :	N	SEARCH MODE :	CIRCULAR	ACD THRESHOLD :	*
NUMBER OF ASSOCIATED ROUTES	:	6		PRIORITY :	2
TDDRFLAG :	ON	TDDRTHRESHOLD:	3	SOURCEGROUPIDX :	1
GDTRRULE :	0	ACDPMGRP :	0		
THE FOLLOWING TRUNKS (LTG-LTU-SLOT-CCT) HAVE BEEN ALLOCATED:					
1- 3- 25-0	1	1- 3- 25-0	2	1- 3- 25-0	3
1- 3- 25-0	4	1- 3- 25-0	5	1- 3- 25-0	6
1- 3- 25-0	7	1- 3- 25-0	8	1- 3- 25-0	9
1- 3- 25-0	10	1- 3- 25-0	11	1- 3- 25-0	12
1- 3- 25-0	13	1- 3- 25-0	14	1- 3- 25-0	15
1- 3- 25-0	16	1- 3- 25-0	17	1- 3- 25-0	18
1- 3- 25-0	19	1- 3- 25-0	20	1- 3- 25-0	21
1- 3- 25-0	22	1- 3- 25-0	23	1- 3- 25-0	24
1- 3- 25-0	25	1- 3- 25-0	26	1- 3- 25-0	27
1- 3- 25-0	28	1- 3- 25-0	29	1- 3- 25-0	30

AMO-BUEND-111 TRUNK GROUP
DISPLAY COMPLETED;

<DISPLAY-BUEND:TGRP=27,FORMAT=L;

DISPLAY-BUEND:TGRP=27,FORMAT=L;

H500: AMO BUEND STARTED

----- FORMAT = L -----					
TGRP NUMBER :	27	TGRP NAME :	PRI ECMA 4	MAXIMUM NO. :	30
		CHARCON :	NEUTRAL		
SUBGROUP NO.:	10	DEVICE TYPE :	S2CONN	TRACENO :	0
RESERVED :	N	SEARCH MODE :	CIRCULAR	ACD THRESHOLD :	*
NUMBER OF ASSOCIATED ROUTES	:	4		PRIORITY :	2
TDDRFLAG :	ON	TDDRTHRESHOLD:	3	SOURCEGROUPIDX :	1
GDTRRULE :	0	ACDPMGRP :	0		
THE FOLLOWING TRUNKS (LTG-LTU-SLOT-CCT) HAVE BEEN ALLOCATED:					
1- 3- 25-1	1	1- 3- 25-1	2	1- 3- 25-1	3
1- 3- 25-1	4	1- 3- 25-1	5	1- 3- 25-1	6
1- 3- 25-1	7	1- 3- 25-1	8	1- 3- 25-1	9
1- 3- 25-1	10	1- 3- 25-1	11	1- 3- 25-1	12
1- 3- 25-1	13	1- 3- 25-1	14	1- 3- 25-1	15
1- 3- 25-1	16	1- 3- 25-1	17	1- 3- 25-1	18
1- 3- 25-1	19	1- 3- 25-1	20	1- 3- 25-1	21
1- 3- 25-1	22	1- 3- 25-1	23	1- 3- 25-1	24
1- 3- 25-1	25	1- 3- 25-1	26	1- 3- 25-1	27
1- 3- 25-1	28	1- 3- 25-1	29	1- 3- 25-1	30

AMO-BUEND-111 TRUNK GROUP
DISPLAY COMPLETED;



Trunk Configuration, TDCSU

For Master-side Configuration

```
<dis-tdcsu:1-3-25-1;
```

```
DIS-TDCSU:1-3-25-1;
```

```
H500: AMO TDCSU STARTED
```

DIGITAL TRUNK (FORMAT=L)					
DEV	= S2CONN	PEN	= 1-03-025-1	TGRP	= 27
PROTVAR	= PSS1V2	INS	= Y	SRCHMODE	= CIR
COTNO	= 22	COPNO	= 21	DPLN	= 0
ITR	= 1	COS	= 32	LCOSV	= 1
LCOSD	= 1	CCT	= PRI ECMA 4	DESTNO	= 111
SEGMENT	= 1	DEDSVC	=	DEDSVC	= NONE
FACILITY	=	DITIDX	=	SRTIDX	=
TRTBL	= GDTR	SIDANI	= N	ATNTYP	= TIE
CBMATR	= NONE	NWMUXTIM	= 10	TCHARG	= N
SUPPRESS	= 0	DGTPR	=	CHIMAP	= N
ISDNIP	=	ISDNNP	=		
PNPL2P	=	PNPL1P	=	PNPAC	=
TRACOUNT	= 31	SATCOUNT	= MANY	NNO	= 111
ALARMNO	= 0	FIDX	= 1	CARRIER	= 1
ZONE	= EMPTY	COTX	= 22	FWDX	= 5
DOMTYPE	=	DOMAINNO	=	TPROFNO	=
INIGHT	=			CCHDL	=
UUSCCX	= 16	UUSCCY	= 8	FNIDX	= 1
CLASSMRK	= EC & G711	& G729OPT		SRCGRP	=
TCCID	=				
BCNEG	= N	BCGR	= 1	LWPAR	= 1
LWPP	= 0	LWLT	= 0	LWPS	= 0
LWR1	= 0	LWR2	= 0		
SVCDOM	=				
BCHAN	= 1 && 30				

AMOUNT OF B-CHANNELS IN THIS DISPLAY-OUTPUT: 30

```
AMO-TDCSU-111 DIGITAL TRUNKS
```

```
DISPLAY COMPLETED;
```

```
<DISPLAY-LWPAR:FORMAT=L,BLNO=1,TYPE=DIUS2;
```

```
DISPLAY-LWPAR:FORMAT=L,BLNO=1,TYPE=DIUS2;
```

```
H500: AMO LWPAR STARTED
```

LOADWARE PARAMETERS		CIRCUIT TYPE: DIUS2		SOURCE:DB		BLOCK: 1	
LNTYPE	= COPPER	VERSION	= S2	QUAL	= ON		
MASTER	= Y	DCHAN1	= 16	DCHAN2	= 0		
PATTERN	= D5H	QUAL1	= 10 SEC.	QUAL2	= 10 MIN.		
SMD	= Y	PERMACT	= Y	FCBAB	= DFH		
CDG	= N	FIXEDTEI	= 0	CNTRNR	= 255		
TEIVERIF	= N	CRC4REP	= N				
DEV	= INDEP						
INFO	=						

```
AMO-LWPAR-111 LOADWARE PARAMETERS FOR NETWORKING MODULES
```

```
DISPLAY COMPLETED;
```




For Slave-side Configuration

```
<dis-tdcsu:1-3-25-1;
DIS-TDCSU:1-3-25-1;
H500: AMO TDCSU STARTED
```

DIGITAL TRUNK (FORMAT=L)					
DEV	=	S2CONN	PEN	=	1-03-025-1
TGRP	=	27			
PROTVAR	=	PSS1V2	INS	=	Y
COTNO	=	22	COPNO	=	21
ITR	=	1	COS	=	32
LCOSD	=	1	CCT	=	PRI ECMA 4
SEGMENT	=	1	DEDSVC	=	NONE
FACILITY	=		DITIDX	=	
TRTBL	=	GDTR	SIDANI	=	N
CBMATTR	=	NONE	NWMUXTIM	=	10
SUPPRESS	=	0	DGTPR	=	
ISDNIP	=		ISDNNP	=	
PNPL2P	=		PNPL1P	=	
TRACOUNT	=	31	SATCOUNT	=	MANY
ALARMNO	=	0	FIDX	=	1
ZONE	=	EMPTY	COTX	=	22
DOMTYPE	=		DOMAINNO	=	
INIGHT	=		CCHDL	=	
UUSCCX	=	16	UUSCCY	=	8
CLASSMRK	=	EC & G711		=	G729OPT
TCCID	=		SRCHMODE	=	CIR
			DPLN	=	0
			LCOSV	=	1
			DESTNO	=	111
			DEDSVC	=	NONE
			SRTIDX	=	
			ATNTYP	=	TIE
			TCHARG	=	N
			CHIMAP	=	N
			PNPAC	=	
			NNO	=	111
			CARRIER	=	1
			FWDX	=	5
			TPROFNO	=	
			FNIDX	=	1
			SRGGRP	=	
BCNEG	=	N	BCGR	=	1
LWPP	=	0	LWLT	=	0
LWR1	=	0	LWR2	=	0
SVCDOM	=		LWPAR	=	0
BCHAN	=	1 && 30	LWPS	=	0

```
AMOUNT OF B-CHANNELS IN THIS DISPLAY-OUTPUT: 30
AMO-TDCSU-111 DIGITAL TRUNKS
DISPLAY COMPLETED;
```

```
<DISPLAY-LWPAR:FORMAT=L,BLNO=0,TYPE=DIUS2;
DISPLAY-LWPAR:FORMAT=L,BLNO=0,TYPE=DIUS2;
H500: AMO LWPAR STARTED
```

LOADWARE PARAMETERS		CIRCUIT TYPE: DIUS2		SOURCE:DB		BLOCK: 0	
LNTYPE	= COPPER	VERSION	= S2		QUAL	= ON	
MASTER	= N	DCHAN1	= 16		DCHAN2	= 0	
PATTERN	= D5H	QUAL1	= 10	SEC.	QUAL2	= 10	MIN.
SMD	= N	PERMACT	= Y		FCBAB	= DFH	
CDG	= N	FIXEDTEI	= 0		CNTRNR	= 255	
TEIVERIF	= N	CRC4REP	= N				
DEV	= INDEP						
INFO	=						

```
AMO-LWPAR-111 LOADWARE PARAMETERS FOR NETWORKING MODULES
DISPLAY COMPLETED;
```




PSTN Trunk Configuration, TDCSU

```
<dis-tdcsu:1-2-79-1;
DIS-TDCSU:1-2-79-1;
H500: AMO TDCSU STARTED
```

DIGITAL TRUNK (FORMAT=L)					
DEV	= S2CONN	PEN	= 1-02-079-1	TGRP	= 23
PROTVAR	= ETSI	INS	= Y	SRCHMODE	= CIR
COTNO	= 21	COPNO	= 21	DPLN	= 0
ITR	= 1	COS	= 10	LCOSV	= 32
LCOSD	= 32	CCT	= PRI ECMA 2	DESTNO	= 100
SEGMENT	= 8	DEDSKC	=	DEDSVC	= NONE
FACILITY	=	DITIDX	=	SRTIDX	=
TRTBL	= GDTR	SIDANI	= N	ATNTYP	= TIE
CBMATTR	= NONE	NWMUXTIM	= 10	TCHARG	= N
SUPPRESS	= 0	DGTPR	=	CHIMAP	= N
ISDNIP	=	ISDNNP	=		
PNPL2P	=	PNPL1P	=	PNPAC	=
TRACOUNT	= 31	SATCOUNT	= MANY	NNO	= 1
ALARMNO	= 0	FIDX	= 1	CARRIER	= 1
ZONE	= EMPTY	COTX	= 21	FWDX	= 10
DOMTYPE	=	DOMAINNO	=	TPROFNO	=
INIGHT	=			CCHDL	=
UUSCCX	= 16	UUSCCY	= 8	FNIDX	= 1
CLASSMRK	= EC & G711 & G729OPT			SRCGRP	=
TCCID	=				
BCNEG	= N	BCGR	= 1	LWPAR	= 1
LWPP	= 0	LWLT	= 0	LWPS	= 0
LWR1	= 0	LWR2	= 0		
SVCDOM	=				
BCHAN	= 1 && 30				

AMOUNT OF B-CHANNELS IN THIS DISPLAY-OUTPUT: 30

```
AMO-TDCSU-111      DIGITAL TRUNKS
DISPLAY COMPLETED;
```




Configuring the Local Cisco Unified CallManager Express (Cisco 3845)

c3845CME#sh run

Building configuration...

Current configuration : 3838 bytes

!

version 12.4

service timestamps debug datetime msec

service timestamps log datetime msec

no service password-encryption

!

hostname c3845CME

!

boot-start-marker

boot system flash:c3845-ipvoice-mz.124-4.XC4.bin

boot-end-marker

!

logging buffered 10000000 debugging

no logging console

enable password cisco

!

no aaa new-model

!

resource policy

!

network-clock-participate wic 0

network-clock-select 1 E1 0/0/1

ip cef

!

!

no ip dhcp use vrf connected

ip dhcp excluded-address 200.1.1.1

!

ip dhcp pool phone

network 200.1.1.0 255.255.255.0



```
option 150 ip 200.1.1.1
default-router 200.1.1.1
!
!
no ip domain lookup
!
isdn switch-type primary-qsig
voice-card 0
no dspfarm
!
!
!
!
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
  < no supplementary-service h450.2>2
  <no supplementary-service h450.3 >2
  h323
  sip
  registrar server expires max 600 min 60
!
!
!
!
!
voice register global
```

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

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



```
mode cme
source-address 200.1.1.1 port 5060
max-dn 100
max-pool 192
load 7960-7940 POS3-07-5-00
tftp-path flash:
create profile sync 000524734349230A
!
voice register dn 1
number 4000
name Zidane
huntstop
!
voice register dn 2
number 4001
name Platini
huntstop
!
voice register pool 1
id mac 000F.9054.2FC2
type 7960
number 1 dn 1
max registrations 240
dtmf-relay rtp-nte
description Zidane
!
voice register pool 2
id mac 0012.4362.BF71
type 7960
number 1 dn 2
max registrations 240
dtmf-relay rtp-nte
description Platini
!
!
```




```
!  
!  
!  
controller E1 0/0/0  
!  
controller E1 0/0/1  
clock source line primary  
pri-group timeslots 1-31  
!  
!  
!  
!  
interface GigabitEthernet0/0  
ip address 172.20.8.26 255.255.255.0  
duplex auto  
speed auto  
media-type rj45  
negotiation auto  
!  
interface GigabitEthernet0/1  
ip address 200.1.1.1 255.255.255.0  
duplex auto  
speed auto  
media-type rj45  
negotiation auto  
!  
interface Serial0/0/1:15  
no ip address  
encapsulation hdlc  
isdn switch-type primary-qsig  
isdn overlap-receiving  
isdn incoming-voice voice  
isdn contiguous-bchan  
no cdp enable  
!
```




```
ip default-gateway 172.20.8.1
ip route 0.0.0.0 0.0.0.0 172.20.8.1
ip route 201.2.2.0 255.255.255.0 172.20.8.27
!
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
!
control-plane
!
!
!
voice-port 0/0/1:15
!
!
!
!
!

dial-peer voice 6000 voip
destination-pattern 4..[89]
session target ipv4:201.2.2.1
no vad
!

dial-peer voice 5000 pots
destination-pattern 500[23]
```




```
< supplementary-service qsig call-forward > 3
direct-inward-dial
port 0/0/1:15
forward-digits all
!
dial-peer voice 95558000 pots
destination-pattern 95553...
no digit-strip
port 0/0/1:15
forward-digits 4
!
!
!
telephony-service
load 7960-7940 P0030702T023
load 7961 TERM41.7-0-3-0S
max-ephones 96
max-dn 192
ip source-address 200.1.1.1 port 2000
system message ABC Corp
max-conferences 8 gain -6
call-forward pattern .T
moh music-on-hold.au
dn-webedit
time-webedit
transfer-system full-consult
transfer-pattern ....
secondary-dialtone 9
create cnf-files version-stamp 7960 Oct 10 2006 15:14:21
!
!
ephone-dn 3 dual-line
number 4002
label 4002
```

³ Omitted to force QSIG call forward by join (no reroute).



```
description Pele
name Pele
call-forward busy 4009
call-forward noan 4009 timeout 10
huntstop channel
!
!
ephone-dn 4 dual-line
number 4003
label 4003
description Beckenbauer
name Beckenbauer
huntstop channel
!
!
ephone 3
mac-address 0017.0EEE.2F5E
type 7961
keep-conference
button 1:3
!
!
!
ephone 4
mac-address 0015.2B8F.351B
type 7961
keep-conference
button 1:4
!
!
!
line con 0
password cisco
login
stopbits 1
```




```
line aux 0
stopbits 1
line vty 0 4
exec-timeout 0 0
password cisco
login
!
scheduler allocate 20000 1000
!
end
```

```
c3845CME#
```




Configuring the Cisco Unified CallManager Express 2 (Cisco 2811)

c2801CME#sh run

Building configuration...

Current configuration : 2802 bytes

!

! Last configuration change at 17:42:21 UTC Mon Oct 30 2006

! NVRAM config last updated at 17:42:24 UTC Mon Oct 30 2006

!

version 12.4

service timestamps debug datetime msec

service timestamps log datetime msec

no service password-encryption

!

hostname c2801CME

!

boot-start-marker

boot system flash:c2801-ipvoice-mz.124-4.XC4.bin

boot-end-marker

!

logging buffered 100000000 debugging

no logging console

enable password cisco

!

no aaa new-model

!

resource policy

!

ip cef

!

!

no ip dhcp use vrf connected

ip dhcp excluded-address 201.2.2.1

!

ip dhcp pool phone



```
network 201.2.2.0 255.255.255.0
option 150 ip 201.2.2.1
default-router 201.2.2.1
!
!
no ip domain lookup
!
!
voice-card 0
!
!
!
voice service pots
!
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
no supplementary-service h450.2
no supplementary-service h450.3
h323
sip
registrar server expires max 600 min 60
!
!
!
!
!
!
!
!
!
```




```
!  
!  
!  
!  
!  
!  
!  
!  
!  
  
interface FastEthernet0/0  
ip address 172.20.8.27 255.255.255.0  
duplex auto  
speed auto  
!  
interface FastEthernet0/1  
ip address 201.2.2.1 255.255.255.0  
duplex auto  
speed auto  
!  
ip default-gateway 172.20.8.1  
ip route 0.0.0.0 0.0.0.0 172.20.8.1  
ip route 200.1.1.0 255.255.255.0 172.20.8.26  
!  
ip http server  
ip http authentication local  
ip http path flash:  
!  
!  
!  
  
tftp-server flash:P0030702T023.bin  
tftp-server flash:P0030702T023.loads  
tftp-server flash:P0030702T023.sb2  
tftp-server flash:P0030702T023.sbn
```




```
!  
control-plane  
!  
!  
!  
!  
!  
!  
!  
dial-peer voice 4000 voip  
destination-pattern 4..[0123]  
session target ipv4:200.1.1.1  
no vad  
!  
dial-peer voice 5000 voip  
destination-pattern 5...  
session target ipv4:200.1.1.1  
no vad  
!  
dial-peer voice 9 voip  
destination-pattern 9.....  
session target ipv4:200.1.1.1  
no vad  
!  
!  
!  
telephony-service  
load 7960-7940 P0030702T023  
load 7941 TERM41.7-0-3-0S  
max-ephones 30  
max-dn 150  
ip source-address 201.2.2.1 port 2000  
system message CBA Corp  
max-conferences 8 gain -6  
call-forward pattern .T
```




```
moh music-on-hold.au
dn-webedit
time-webedit
transfer-system full-consult
transfer-pattern ....
secondary-dialtone 9
create cnf-files version-stamp 7960 Oct 12 2006 11:41:08
!
!
ephone-dn 1 dual-line
number 4008
label 4008
description Ronaldinho
name Ronaldinho
huntstop channel
!
!
ephone-dn 4 dual-line
number 4009
label 4009
description Tevez
name Tevez
call-forward noan 5002 timeout 10
huntstop channel
!
!
ephone 1
mac-address 000F.9069.DB2C
type 7960
keep-conference
button 1:1
!
!
!
ephone 4
```




```
mac-address 0030.94C3.31AD
```

```
type 7960
```

```
keep-conference
```

```
button 1:4
```

```
!
```

```
!
```

```
!
```

```
line con 0
```

```
password cisco
```

```
login
```

```
line aux 0
```

```
line vty 0 4
```

```
exec-timeout 0 0
```

```
password cisco
```

```
login
```

```
!
```

```
scheduler allocate 20000 1000
```

```
end
```

```
c2801CME#
```




Acronyms

Acronym	Definitions
CFB	Call Forward when Busy
CFNR	Call Forward when No Reply
CFU	Call Forward Unconditional
IOS	Internetworking Operating System
PBX	Private Branch Exchange
PRI	Primary Rate ISDN
SIP	Session Initiation Protocol



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