

IP Telephony Advanced Troubleshooting

2003 Cisco Canada Technical Symposium

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Session Objectives

- Understand the functionality of call flows, from the perspective of Cisco CallManager when all works well
- understand the trace outputs and provide a troubleshooting methodology
- To be able to enable the debugs and provide the information requested by Cisco TAC in resolving your issues
- Understand Echo and steps to resolve Echo in IPT Networks

What You Should Know

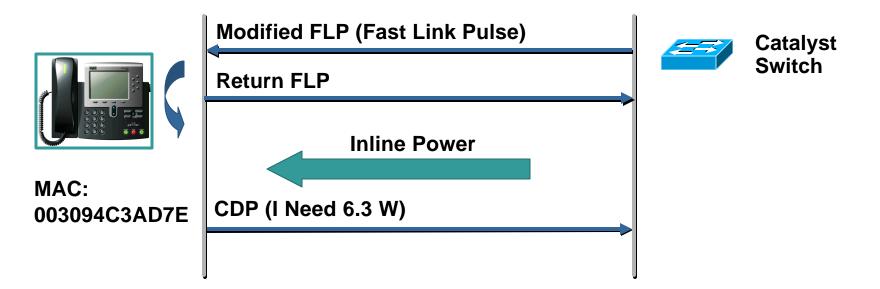
- Understanding of IP Telephony architecture, concepts and configuration
- Experience with CCM configuration and operation
- Cisco IOS[®] GW configuration and operation



- Phone Initialization
- Tools and Utilities to Configure, Monitor and Troubleshoot CCM
- IP Telephony Case Study and Troubleshooting Techniques
- Understand Echo and steps to resolve Echo in IPT Networks

IP Phone Initialization—Inline Power

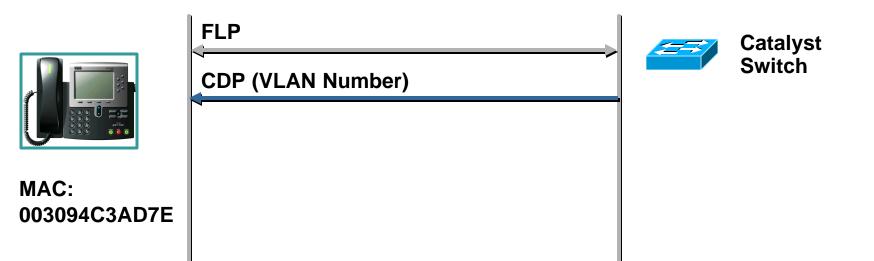
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Phone: Mute, Headset, Speaker Buttons Illuminated

IP Phone Initialization—AUX VLAN

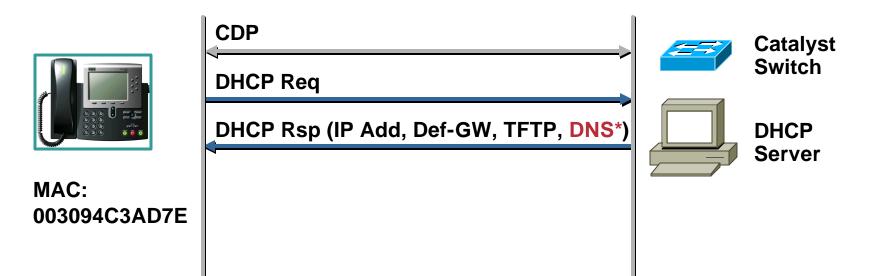
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Phone Displays: Configuring VLAN Check Settings: NetCfg->19 Operational VLAN ID

IP Phone Initialization—IP Configuration

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Phone Displays: Configuring IP *DNS is Optional Check settings: NetCfg-> 1 DHCP Server NetCfg-> 6 IP Address

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IP Phone Initialization—TFTP

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	CDP	Catalyst
	DHCP Req	Switch
	DHCP Rsp (IP Add, Def-GW, TFTP, DNS*)	DHCP Server
MAC: 003094C3AD7E	TFTP Read	Server
	TFTP Data (OS79XX.txt)	TFTP
	TFTP Read	Server
	TFTP Data (SEP003094C3AD7E.cnf.xml)	

Phone Displays: Configuring IP Error Verifying Config Info Check settings: NetCfg-> 8 TFTP Server

IP Phone Initialization TFTP Trace

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• TFTP traces provide more information on initialization process

TFTP|[opcode = 1] [Mode = octet] [thread count 0]| <CLID::StandAloneCluster> <NID::172.21.54.99><CT::3456:10.94.145.219:50953><IP::10.94.145.219> <DEV::SEP003094C25FCE.cnf.xml>

TFTP|file error[File Name or path not found 2]|<CLID::StandAloneCluster> <NID::172.21.54.99><CT::3456:10.94.145.219:50953><IP::10.94.145.219> <DEV::SEP003094C25FCE.cnf.xml>

• Note that the phone requests the configuration file

SEP0002FDAEFB9D.cnf.xml from the CallManager (version 3.1 and above)

Phone Loads in CallManager Releases prior to 3.1 request the File name: SEP0002FDAEFB9D.cnf

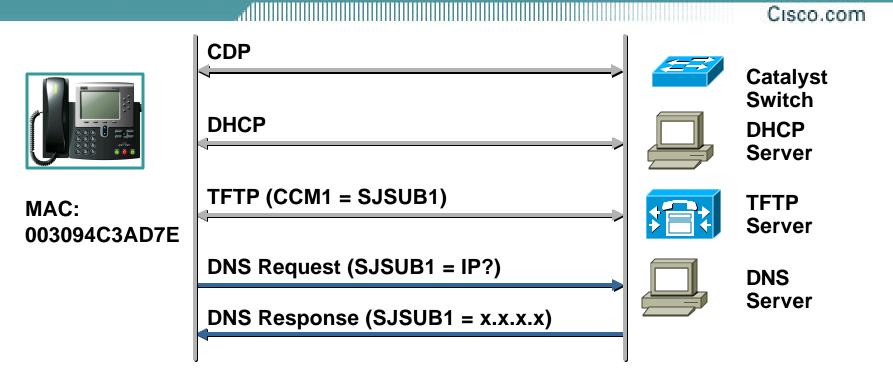
3.0.12 will understand both file formats

• Files served by TFTP

Are located in c:\Program Files\Cisco\TFTPPath

CCM 3.2: By default SEP.cnf.xml files cached in RAM and not written to disk

IP Phone Initialization—DNS



Phone displays: Configuring IP CallManager Name DNS Error Defaulting CM to TFTP Server Check settings: NetCfg-> 14 DNS Server 1

IP Phone Initialization—CCM Registration

MAC: 003094C3AD7E	CDP		Catalyst Switch	
	DHCP		DHCP Server	
	TFTP		TFTP Server	
	DNS		DNS	
	Skinny Register		Server	
	Skinny Registration Confirm		Call Manager	
Phone displays: Configuring CM List				
	Registration Rejected			
Opening 10.1.1.1				
	CM Down, Features Disable	ed		
Check set	tings: NetCfg-> 21 Call Manager 1			

Post Initialization—The Sound of Success

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Re-order or fast busy is NOT a phone problem

No route for dialed destination on CCM No matching route patterns on GW CODEC mismatch



- Phone Initialization
- Tools and Utilities to Configure, Monitor and Troubleshoot CCM
- IP Telephony Case Study and Troubleshooting Techniques
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Tools and Utilities to Configure, Monitor and Troubleshoot on CallManager

- CCM administration
- Event log
- CCM serviceability

Performance monitor

Trace configuration

CCM traces

Collection

Q931 translator

EventLog

- Start-> Programs-> Administrative Tools -> Event Viewer -> Application Log
- EventLog is a Windows 2000 Server application that displays a log of Windows 2000 server and CCM
- Even if a service (including TFTP) can not read the database (where it gets trace configuration), it will add errors to the event log

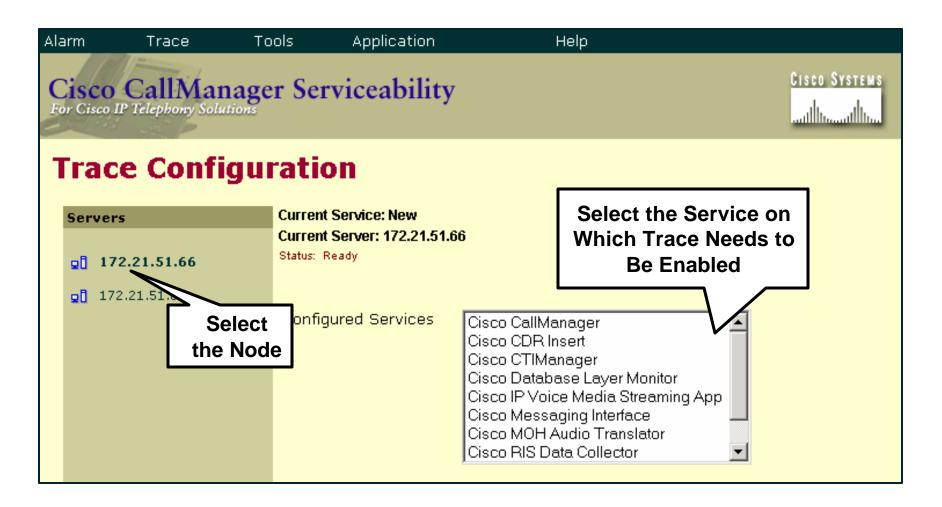
EventLog

	Action ⊻iew	🔁 📧 😭 🔀	ß			
	Tree	Application Log	8,815 event(s)			
	Event Viewer (Local)	Туре	Date	Time	Source	Cal
	Application Log	(Information	4/25/2002	6:54:48 AM	SceCli	Nor
	Security Log	(Information	4/24/2002	4:11:47 PM	MSDTC	СМ
nt Properties	? ×	(Information	4/24/2002	4:11:47 PM	MSDTC Client	СМ
		🚫 Error	4/24/2002	4:01:32 PM	Cisco CallManager	Nor
vent		😣 Error	4/24/2002	4:01:08 PM	Cisco CallManager	Nor
Date: 4/24/2002 Source: Cisco CallManage	·	Error	4/24/2002	3:57:06 PM	Cisco CallManager	No
Time: 15:57 Category: None		S Error	4/24/2002	3:57:01 PM	Cisco CallManager	No
Type: Error Event ID: 3	+	Information	4/24/2002	3:48:34 PM	MSDTC Client	CM
User: N/A		Information	4/24/2002	3:48:34 PM	MSDTC	CM
 Computer: CM31PUB		😣 Error	4/24/2002	3:25:43 PM	Cisco CallManager	No
		😣 Error	4/24/2002	3:22:33 PM	Cisco CallManager	No
Description:		😣 Error	4/24/2002	3:22:28 PM	Cisco Tftp	No
Error: DeviceTransientConnection - Transient connecti	on attempt. 🔺 📘	Information	4/24/2002	1:47:43 PM	SceCli	No
Connecting Port: 2000 Device name [Optional].:		Information	4/23/2002	9:23:37 PM	SceCli	No
Device IP address.: 10.17.168.108		Information	4/23/2002	5:09:32 AM	SceCli	No
Device type. [Optional]: 255 Reason Code [Optional].: 6		Information	4/22/2002	12:58:27 PM	SceCli	No
App ID: Cisco CallManager		Information	4/21/2002	8:21:21 PM	SceCli	No
		Information	4/21/2002	3:28:16 AM	SceCli	No
Data: 💿 <u>B</u> ytes 🔿 <u>W</u> ords		😣 Error	4/20/2002	1:46:53 PM	Cisco CTIManager	No
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Enabling Tracing in Cisco CallManager

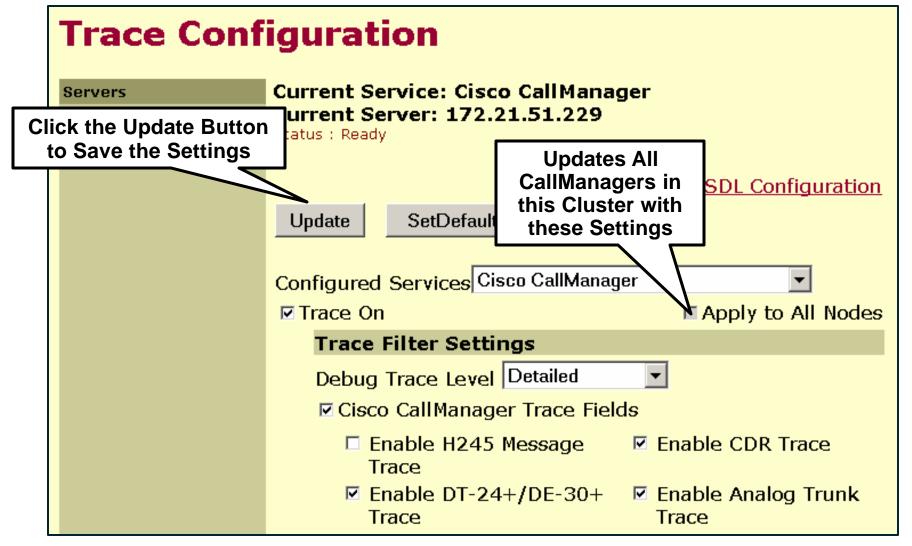
- For nearly all troubleshooting scenarios, you will need traces from the Cisco CallManager service; click the 'Set Default' button then change the trace level to either 'Arbitrary' or 'Detailed'
- Arbitrary and Detailed are nearly identical, except Detailed shows KeepAlives and some additional digit analysis data
- CCM trace (a.k.a.: System diagnostic interface) files provide the greatest level of detail

Trace Utility—Node, Service Selection



Enabling Tracing in Cisco CallManager

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Trace Utility—Debug Trace Level

Error—Used for all traces generated in

- abnormal path; minimum amount of CPU cycles
- Special—Non-repetitive messages; ex. all system and device initialization messages
- State transition—Call processing events
- Significant—Media layer events
- Arbitrary—Used for debugging excluding keepalives
- Detail—Detailed debug information

Trace Utility—SDI Trace Format, raw

Million Cisco.com

🔄 ccm00000111.txt - Notepad
Eile Edit Format Help
03/20/2003 20:17:54.553 Cisco CallManager StationInit: 51b7af4
OffHook. <clid::wwccm1-cluster><nid::cm-fl2><ct::1,100,95,1.1897913><ip::10.17.168.70><dev::sep0002f< td=""></dev::sep0002f<></ip::10.17.168.70></ct::1,100,95,1.1897913></nid::cm-fl2></clid::wwccm1-cluster>
DAEFB9D>
03/2012003 20:17:54.553 Cisco CallManager StationD: 51b7af4 DisplayText text=' 1000
'. <pre>'. </pre> CLID::WWCCM1-Cluster> <nid::10.1.1.1><ct::1,100,95,1.1897913><ip::10.17.168.70><dev::sep0002fdaef b9d=""></dev::sep0002fdaef></ip::10.17.168.70></ct::1,100,95,1.1897913></nid::10.1.1.1>
03/20/2008 20:17:54.553 Cisco
CallManager <pre>CLID::WWCCM1-Cluster><nid::10.1.1.1><ct::1,100,95,1.1897913><mn::directory< pre=""></mn::directory<></ct::1,100,95,1.1897913></nid::10.1.1.1></pre>
Number> <mv: 1000=""><dev::sep0002fdaefb9d></dev::sep0002fdaefb9d></mv:>
03/20/2003 20:17:54.553 Cisco CallManager CTI: RoutePatternToCtiGlobalCallID::findValue(): RP=1000:,
bRc=0, T=, #entries=0 <clid::wwccm1-cluster><nid::10.1.1.1></nid::10.1.1.1></clid::wwccm1-cluster>
03/20/2003 20:17:54.553 Cisco CallManager CTI: RoutePatternToCtiCallOrigin::findValue() : RP=1000:,
bRc=0, T=, #entries=0 <clid::wwccm1-cluster><nid::10.1.1.1></nid::10.1.1.1></clid::wwccm1-cluster>
03/20/2003 20:17 54.553 Cisco CallManager StationD: 51b7af4 SetLamp stimulus=9(Line)
lampMode=2 (I) < CI_TD, MMCCM1_Clusters <ntd, 10_17_168_70="" 10_1_1_1_s="" 1_100_05_1_1807013s="" <ct,="" <td,=""><</ntd,>
DEV::SEP0002 We Are Going to Detail this Entry from the "raw" Trace File
lineInstance Output; Please Note that in this View, Word Wrap Is Turned on
callReferend68.7
0> <dev::sep0002fdaefb9d></dev::sep0002fdaefb9d>
03/20/2003 20:17:54.553 Cisco CallManager StationD: 51b7af4 DisplayPromptStatus timeOutValue=0
70> <dev::sep0002fdaefb9d></dev::sep0002fdaefb9d>
03/20/2003 20:17:54.553 Cisco CallManager StationD: 51b7af4 SelectSoftKeys instance=1
reference=50331661 softKeySetIndex=4

Trace Utility— SDI Trace Format, detailed

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03/20/2003 20:17:54.553 Cisco CallManager |StationInit. 51b7af4 OffHook.| <CLID::WWCCM1-Cluster> <NID::10.1.1.1> <CT::1,100,95,1.1897931> <IP::10.17.168.70> <DEV::SEP0002FDAEFB9D>

Field Name	Example	
Date	" 03/20/2003 "	
Time	" 20:17:54.553 "	
System	"Cisco CallManager"	
Trace Text	"Trace/Event ID, Event Information"	
Cluster ID (CLID)	"WWCCM1-Cluster"	Device ID
Node ID (NID)	"10.1.1.1"	a.k.a. TCP
Correlation Tag (CT)	"1,100,95,1.1897913"	
Source Device	"10.17.168.70, SEP0002FDAEFB9D"	
Tag Mapping	"Destination Device Name, DN, Application defined Tags" Not used in this example	

How Do We Find Information in these Huge Text Files?

Cisco.com 💐 ccm00000191.txt - Notepad _ 🗆 × File Edit Format Help Cisco fallManager MediaManager - wait AuConnyrtReply - received 1 responses 04/22/2003 15:28:50 072 **18:00** Cases ConnectReply | < CLID: : CCM01-CLSTE |04/22/2003 15:28:50.072 Cisco CallManager|ConnectionManager - wait AuConnectReplv(36613487.3661348{ 04/22/2003 15:28:50.088 Cisco CallManager CTI; RoutePatternToCtiCommandData::findValue() : RP=70240 AuscaddMeSSrOTathe:DNOMe38 SelectSoftKeys instance=1 reference= 104/22/2003 04/22/2003 15:28:50.166 Cisco CallManager MediaManager - wait AuReConnectRequest CLID::CCM01-CLST 04/22/20<u>03</u> 15:28:50.<u>16</u>6 Cisco CallMan<u>age</u>r|Medi<u>a</u>Manager - wait_AuDisconnectRequ<u>e</u>st|<CLID::CCM01-CLS 04/22/2003 15:28:50.166 Cisco CallManager MediaManager - no proxy and St Cœurrences®® ēst. <CLID:::CCM01-CLSTR(wait AuDisconnectReply Qiero CallManager MediaManager - wait_AuDisconnectReply - received all disc 04/22/2003 **C** 28:50 YWM3co CallManager MediaCoordinator - wait_AuReConnectRequest | <CLID::CCM01-0 04/22/2003 15:28:50.166 Cisco CallManager|MediaCoordinator - wait_AuDisconnectReply - removing Med: 04/22/2003 15:28:50.166 Cisco CallManager MediaCoordinator - wait_AuConnectRequest() <CLID:;CCM01-(04/2**9**/2073OC48S'.OUTSSE2MC49=24FOUMO1at e neor (); the g Media 04/22/2003 Cisco CallManager MediaCoordinator AuConnectRequest - new MediaManag wait_ **CStigatin G**diaManager(1378984) started|<CLID::CCM01-CLSTR01-Cluste diaManager - wait_AuConnectRequest|<CLID::CCM01-CLSTR01 /22/2003 166 Cisco CallManager | MediaManager - wait_AuConnectRequest - normal connection 04/22/2003 15 04/22/2003 15:28:50.182 Cisco CallManager MediaManager wajt nnectReply | <CLID::CCM01-CLSTR01-(04/22/2055 (5:28: .stox11Mae:DeV4.GeP1t_luCmnec eived 1 responses 04/22/2003 12:28:50.182 Cisco CallManager MediaCoordinator wait AuConnectReply <CLID::CCM01-CLSTH</p> 04/22/2003 15:28:50.229 Cisco CallManager H245Interface(185602) OLC indication chan number = 1 <CL otag gets changed if the phone has to re-establish lust 04/22/2003 1 04/22/2003 1 Ction with the CallManager Manager StationD: 33bc279c SetSpeckerMo 8 TCP CONNECTION WITH TH 8:50.229 Cisco CallManager StationD: 04/22/2003 eakermode=1(On). |-04/22/2003 : Cisco CallManager <CLID::CCM01-CLSTR01-Cluster><NID::10.0.32.65><CT::2,100.</pre> 04/22/2003 Cisco CallManager CTI: RoutePatternToCtiGlobalCallID::findValue(): RP=94922 04/22/2003 1

How Do We Find Information in these Huge Text Files?

C:\WINNT\System32\cmd.exe	
E:\>findstr "SEPABCDEFB8931F" ccm×.txt	
ccm00000191.txt:04/22/2003 15:28:42.557 Cisco CallManager StationIni	
eypadButton kpButton=7. <clid::ccm01-clstd01-c< td=""><td>(CT</td></clid::ccm01-clstd01-c<>	(CT
,95,1.80254893> <ip::10.28.88.120><dev::sepabcdefb8931f></dev::sepabcdefb8931f></ip::10.28.88.120>	
ccm00000191.txt:04/22/2003 15:28:47.635 cises CallHanager StationInit:	
timulus stimulus=9(Line) stimulusInstance=1. <clid::ccm01-clstr01-clus< td=""><td></td></clid::ccm01-clstr01-clus<>	
10.0.32.65> <ct::2,100,95,1.80254923><ip::10.28.88.120><deu::sepabcdefb< td=""><td></td></deu::sepabcdefb<></ip::10.28.88.120></ct::2,100,95,1.80254923>	
ccm00000191.txt:04/22/2003 15:28:47.635 Cisco CallManager StationInit:	
ffHook. <clid::ccm01-clstr01-cluster><nid::10.0.32.65><ct::2,100,95,1.< td=""><td>80254924><</td></ct::2,100,95,1.<></nid::10.0.32.65></clid::ccm01-clstr01-cluster>	80254924><
IP::10.28.88 Now We Know what the Device ID Is for that Phone	
composition in a company of the set of the s	-CLSTR01-C
	> <mu::7024< td=""></mu::7024<>
073709 X DEU: Only; Scenarios of what to Look for Will Be More	
Obvieue ee Me Deview the Ceee Studies	timLine: 0
0254923> <ip::10.28.88.120><deu::sepabcdefb8931f></deu::sepabcdefb8931f></ip::10.28.88.120>	100,95,1.8
ccm00000191.txt:04/22/2003 15:28:47.635 Cisco CallManager StationD:	830b438 S
etSpeakerMode speakermode=1(On). <clid::ccm01-clstr01-cluster><nid::10< td=""><td></td></nid::10<></clid::ccm01-clstr01-cluster>	
CT::2,100,95,1.80254923> <ip::10.28.88.120><deu::sepabcdefb8931f></deu::sepabcdefb8931f></ip::10.28.88.120>	.0.02.00/
ccm00000191.txt:04/22/2003 15:28:47.635 Cisco CallManager StationD:	830b438 D
isplayText text=' 7024073709 '. <clid::ccm01-clstr< td=""><td></td></clid::ccm01-clstr<>	
> <nid::10.0.32.65><ct::2,100,95,1.80254923><ip::10.28.88.120><deu::sep< td=""><td></td></deu::sep<></ip::10.28.88.120></ct::2,100,95,1.80254923></nid::10.0.32.65>	
1F>	
ccm00000191.txt:04/22/2003 15:28:47.635 Cisco CallManager StationD:	830b438 D
isplayText text=' 7024073709 '. <clid::ccm01-clstr< td=""><td></td></clid::ccm01-clstr<>	



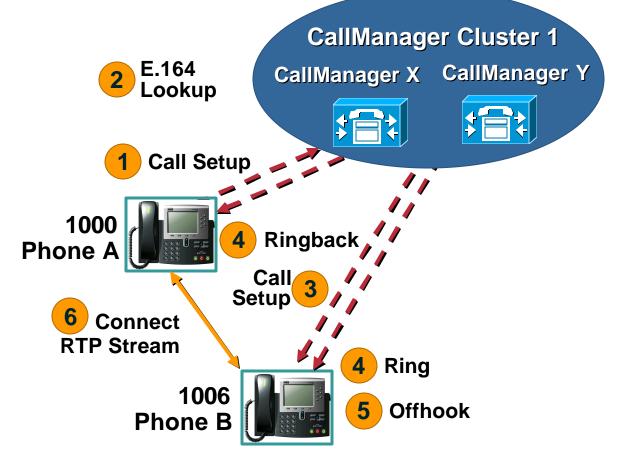
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Case Study: IP Phone to IP Phone Intra-Cluster (Successful Call)

- IP phone to IP phone call flow within a cluster
- IP phone to IP phone call flow SCCP messages
- IP phone registration and call flow messages through the CCM traces

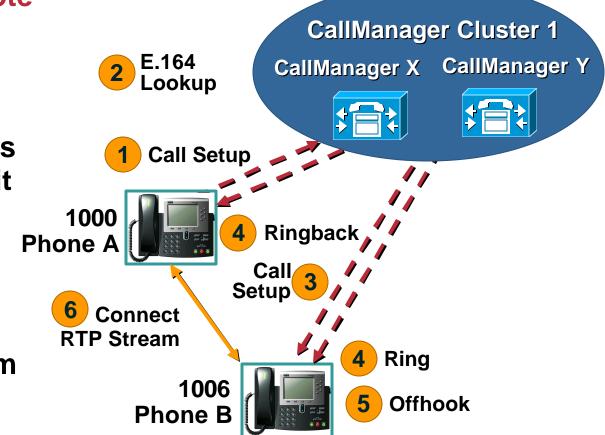
IP-Phone to IP-Phone Call Flow

- Phone A is registered to CallManager X
- Phone B is registered to CallManager X
- Call is placed from Phone A (1000) to Phone B (1006)



IP-Phone to IP-Phone Call Flow

- Very important note
- All traces were generated on CallManager X, since both phones are registered to it
- If phone B were registered to CallManager Y, we would have to collect traces from both servers



Simple Intra-Cluster Call Flow (SCCP)

	Cisco.com
IP Phone A Station Off Hook	CM IP Phone B
Station Display Text	
Station Play Tone (Dial Tone) Station Set Lamp (Steady)	
Station Digit Dialled	
Station Stop Tone (Dial Tone)	
Station Digit Dialled	
Station Digit Dialled	
	Station Call Information Station Set Lamp (Blink)
	Station Set Ringer (On)
Station Call Info Station Start Tone (alerting)	Station Off Hook
Ctation Ston Tone	Station Set Lamp (Steady) Station Set Ringer (Off)
Station Step Form	Station Open Receive Channel
Station Start Media Transmission	Station Open Receive Channel Ack
Station Open Receive Channel Ack	Station Start Media Transmission
	rsation
Station On Hook	Station Close Receive Channel
Station Set Lamp (Off)	Ctation Stop Media Transmission
Station Close Receive Channel	Station Set Lamp (Off)
Station Stop Media Transmission	Station On Hook

Intra Cluster Call Flow Trace— Phone Goes Off Hook

Cisco.com

Phone 1000 goes to OffHook

03/20/2002 23:14:49.517 Cisco CallManager

StationInit: 520443c OffHook.

<CLID::WWCCM1-

Cluster><NID::172.21.51.216><CT::1,100,95,1.1243><IP::10.17.168.70><DEV::SEP0002FD AEFB9D>

Header that is common to all the messages

03/20/2002 23:14:49.517 Cisco CallManager

Trailer is common to all the messages to and from this phone

<CLID::WWCCM1-Cluster><NID::172.21.51.216><CT::1,100,95,1.1243><IP::10.17.168.70><DEV::SEP0002FD AEFB9D>

- Header and trailer messages will be omitted from now on to simplify the display
- The **520443c** is unique ID for this phone
- StationInit indicates that CallManager received a TCP message from a skinny station

Intra Cluster Call Flow Trace— SCCP Call States

- 1—Off Hook 2—On Hook 3—Ring Out 4—Ring In 5—Connected 6—Busy 7—Congestion
 - 8—Hold
 - 9—Call Waiting
 - 10—Call Transfer
 - 11—Call Park
 - 12—Proceed
 - 13—Call Remote Multiline
 - 14—Invalid Number

Intra Cluster Call Flow Trace— Dialing Starts

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- User starts entering the digits—First digit '1' is dialed [StationInit: 520443c KeypadButton kpButton=1.]
- CallManager stops sending the dial tone to the phone 1000 and collects the other digits entered through the keypad (We dialed 1006)

StationD: 520443c StopTone.

StationD: 520443c SelectSoftKeys instance=1 reference=16777230 softKeySetIndex=6 validKeyMask=-1.

|Digit analysis: match(fqcn="1000", cn="1000", pss="PA:Employee:CER", dd="1")|

|Digit analysis: potentialMatches=PotentialMatchesExist|

StationInit: 520443c KeypadButton kpButton=0.

|Digit analysis: match(fqcn="1000", cn="1000", pss="PA:Employee:CER", dd="10")|

|Digit analysis: potentialMatches=PotentialMatchesExist|

StationInit: 520443c KeypadButton kpButton=0.

|Digit analysis: match(fqcn="1000", cn="1000", pss="PA:Employee:CER", dd="100")|

|Digit analysis: potentialMatches=PotentialMatchesExist|

StationInit: 520443c KeypadButton kpButton=6.

|Digit analysis: match(fqcn="1000", cn="1000", pss="PA:Employee:CER", dd="1006")|

Intra Cluster Call Flow Trace— Digit Analysis

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• A match was found: Here are the CallManager digit analysis results

|Digit analysis: analysis results|

|PretransformCallingPartyNumber=1000

|CallingPartyNumber=1000

|DialingPartition=Employee

|DialingPattern=1006

|DialingRoutePatternRegularExpression=(1006)

|DialingWhere=

|PatternType=Enterprise

|PotentialMatches=NoPotentialMatchesExist *

|DialingSdlProcessId=(1,34,20)

|PretransformDigitString=1006

* NoPotentialmatchesExist Means that the Dialed String Is Not Partially Matching a Pattern within this Call's pss

Simple Intra-Cluster Call Flow (SCCP)

IP Phone CCM IP Phone Station Off Hook Station Display Text Station Display Text Station Display Tone (Dial Tone) Station Digit Dialled Station Set Lamp (Steady) Station Set Lamp (Steady) Station Digit Dialled Station Set Lamp (Blink) Station Digit Dialled Station Set Limp (Blink) Station Set Limp (Blink) Station Stati Digit Dialled Station Set Limp (Blink) Station Set Limp (Blink) Station Stati Tone (Alerting) Station Set Limp (Blink) Station Set Limp (Blink) Station Stat Tone (Alerting) Station Set Limp (Steady) Station Stat Tone (Alerting) Station Open Receive Channel Station Stat Media Transmission Station Open Receive Channel Station Open Receive Channel Ack Station Start Media Transmission Station Open Receive Channel Ack Station Stop Media Transmission Station On Hook Station Stop Media Transmission Station Stop Media Transmission Station Stop Media Transmission				
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Station Off Hook Station Display Text Station Play Tone (Dial Tone) Station Set Lamp (Steady) Station Stop Tone (Dial Tone) Station Digit Dialled Station Digit Dialled Station Digit Dialled Station Call Information Station Set Lamp (Blink) Station Set Lamp (Off) Station Start Tone (Alerting) Station Stat Tone (Alerting) Station Open Receive Channel Station Open Receive Channel Station Open Receive Channel Ack Station On Hook Station Close Receive Channel Station Set Lamp (Off) Station Stop Media Transmission Station Set Lamp (Off) Station Stop Media Transmission Station Close Receive Channel Station Stop Media Transmission	IP Pho	ne (CCM	IP Phone
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- Phone Initialization
- Tools and Utilities to Configure, Monitor and Troubleshoot CCM
- IP Telephony Case Study and Troubleshooting Techniques
- Understand Echo and steps to resolve Echo in IPT Networks

Troubleshooting Echo

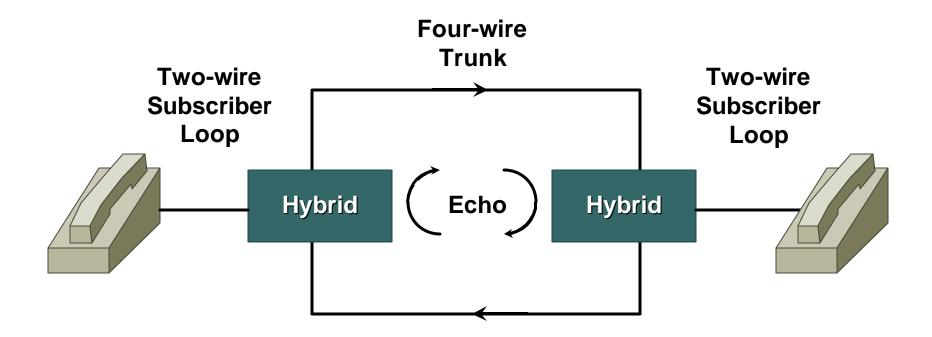
- Echo Sources and Definition
- Eliminating Echo
- Tools and Methodology for Troubleshooting Echo

Echo—What Is Echo? What Is Echo?

- Objectionable reflected voice energy in speaker's receive path
- Function of signal strength and delay
- Why is echo prevalent in packet telephony?

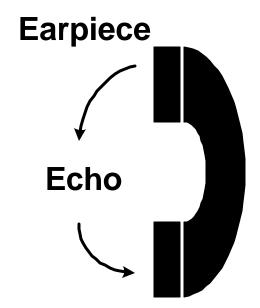
Sources of Echo: Hybrid Echo

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Sources of Echo: Acoustic Echo

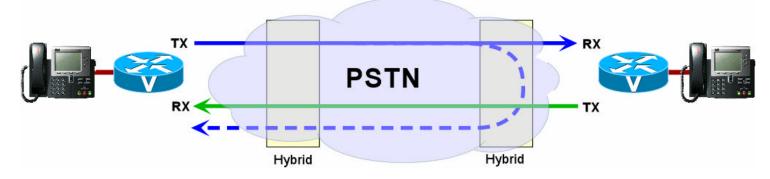
 Poor acoustic isolation between the earpiece microphone and the microphone in handsets and hands-free devices introduce acoustic echo



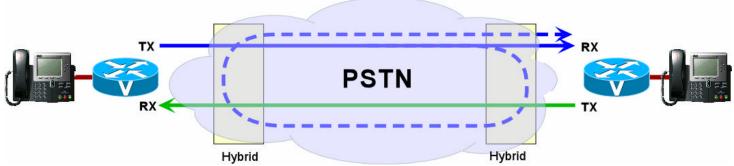
Types of Echo

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Talker Echo—The effect is the talker hears his/her own voice delayed by the total echo path delay time



Listener Echo—The effect is the person listening hears the talker and an echo of the talker



What Makes Echo a Problem?

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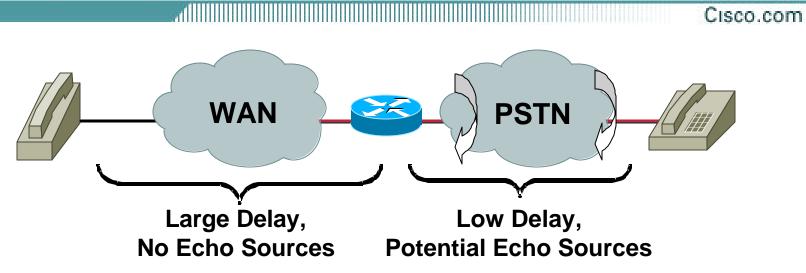
For Echo to Be a Problem, All of the Following Conditions Must Exist:

- An analog leakage path between analog Tx and Rx paths
- Sufficient delay in echo return for echo to be perceived as annoying

(delays < 25ms are masked as sidetone)

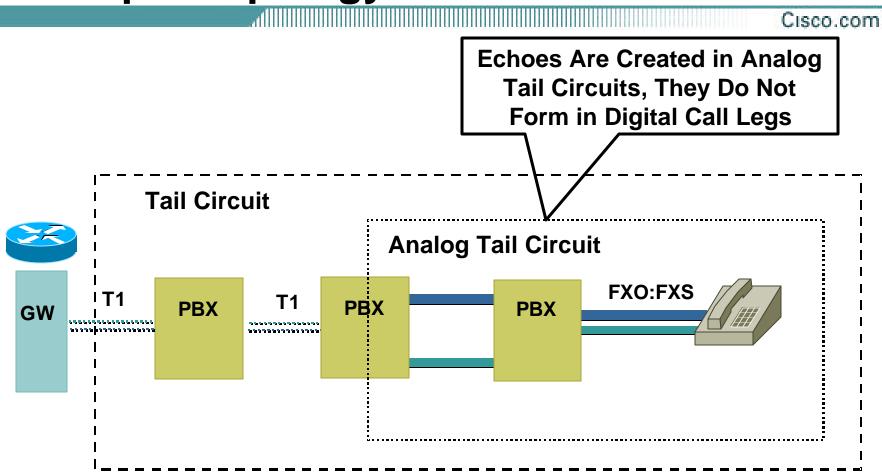
 Sufficient echo amplitude to be perceived as annoying

The Impact of Packet Voice on Echo Perception



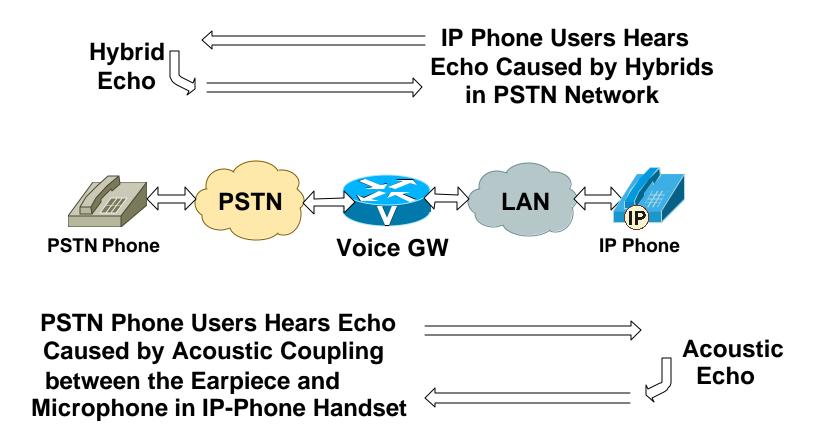
- The packet segment of the voice connection introduces large delay (typically > 30 ms in each direction)
- The introduction of delay causes echoes (from analog tail circuits) that are normally indistinguishable from side tone to become perceptible
- Because the delay introduced by packet voice is unavoidable, the voice gateways must prevent the echo

Example Topology of a Tail Circuit

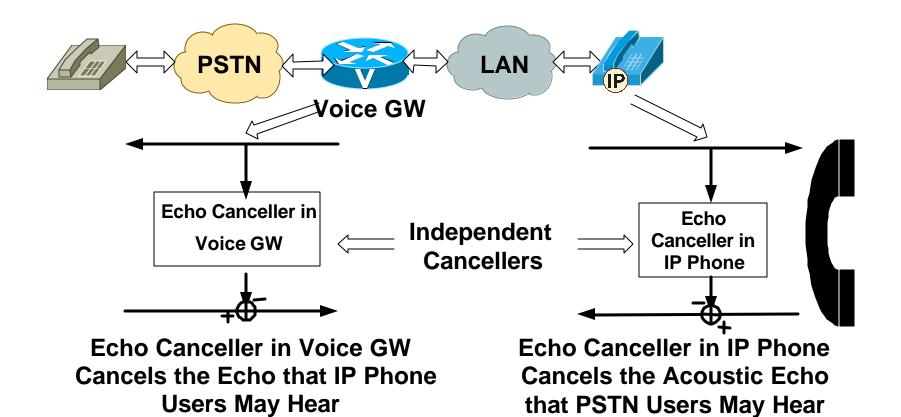


Echo in IP-PBX Environment

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Echo Cancellation in an IP-PBX Environment

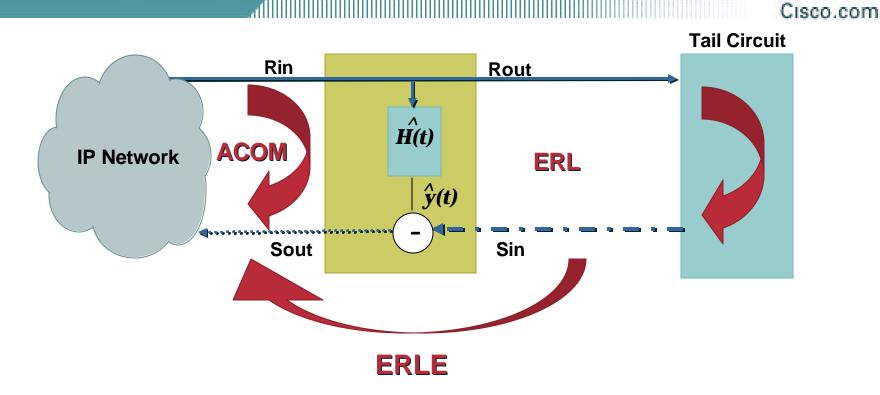


Troubleshooting Echo

- Echo Sources and Definition
- Eliminating Echo
- Tools and Methodology for Troubleshooting Echo

Eliminating Echo—Terminology

.411111111



ERL = Echo Return Loss = Rout—Sin ERLE = Echo Return Loss Enhanced= Rout—Sout ACOM—Acombined = Rin—Sout

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Eliminating Echo—Terminology

• Echo Cancel Coverage is the amount of time the Echo Canceller will 'Remember' a signal that has been output; This parameter must be set to a value greater than the time it takes the echo to return back to the gateway

Characteristics of Conventional Echo Cancellers

- The convergence rate depends on the properties of the reference (Rin) signal
- Silence and telephony signaling tones (DTMF, ringback, etc...) are poor reference signals and will not train the echo canceller significantly
- The convergence rate is inversely proportional to the duration of coverage provisioned
- Violation of the minimum assumed ERL prevents convergence due to falsely detecting doubletalk (when both parties talk at the same time)
- Adaptation during doubletalk will cause divergence

Characteristics of the dspware Echo Canceller before 12.2(13)T

- Echo coverage is "provisionable" via CLI in 8 ms steps up to 32 ms; The greater the coverage provisioned, the slower the convergence rate
- Limitations of DSP MIPs and memory prohibit the expansion of coverage
- The minimum ERL is assumed to exceed 6 dB
- The EC does not perform well with echo environments that a) exceed 32ms in duration, b) are time varying (conferencing changes, acoustic echo), c) have an ERL < 6dB

Characteristics of the dspware Echo Suppressor

- Can eliminate echo due to convergence times of echo canceller
- Echo Suppressors allow one way communication voice-port cli: echo suppressor < # seconds >
- Must be disabled for modem communication
- Automatically disables during double talk
- Available in 12.2(8)T5 for NM-HDV and 12.2(11)T2 for most other voice interfaces (i.e. NM-2V)

Echo Cancellation Enhancements

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- Extends coverage to 64ms
- Allows required ERL drop from 6dB to 0 (via CLI)
- G.168 Complaint
- Much faster convergence time
- Echo suppressor not available (and is not needed) with ECAN Enhancements

cli: voice-card <slot>

codec complexity {high | medium} ecan-extended

Echo Cancellation Enhancements -Platform Support

- 12.2(13)T adds support on NM-HDV, 7200/7500, ICS7750 MRP, High Complexity on 1700, High Complexity on MC3810
- 12.2(15)ZJ and second release of 12.3T add additional platform support for NM-HDA and AIM-VOICE
- 12.2(8)YH, 12.2(13)ZH, and the first release of 12.3T add medium complexity support for 1700
- 12.3(1) adds support for medium complexity on MC3810.
- Second release of 12.3T adds support for High Complexity Analog and Medium Complexity digital on the Cat4k AGM.

Troubleshooting Echo

- Echo Sources and Definition
- Eliminating Echo
- Tools and Methodology for Troubleshooting Echo

Narrowing Your Focus

- First, find out what the exact symptoms of the echo problem are
- Every echo case is different and must be treated on a case-by-case basis
- Isolate the problem and try to find a scenario where the echo is reproducible
- There may be several different echo problems present—isolate each one and troubleshoot one at a time

Questions to Ask

When does the echo occur?

- Throughout call?
- Only beginning of call?
- Does a single number (set of numbers) result in echo often?
- Which side hears the echo?
- Does the party hearing the echo hear themselves or the far end echoed?

What to Collect?

- Need to gather data while a call is experiencing echo
- Have users collect call-ING and call-ED numbers
- Need to produce 1004Hz tone and adjust dB levels (# or * dtmf buttons give a rough estimate if test gear is not available)
- PCM trace on PSTN tail-circuit producing the echo is very useful
- Sniffer trace
- 'show voice call x/y:z.ts' or 'show call active voice' → several iterations
- Check ERL—adjust input gain if necessary

Eliminating Echo

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So, How Do I Get Rid of Echo?

You need to make sure you give the echo canceller enough information to distinguish between echo and normal conversation; The only parameters you have control over are:

- Input Level (Input Gain)
- Output Level (Output Attenuation)
- Echo Canceller Coverage
- Echo Suppressor (12.2(8)T5 and later)
- Enhanced Echo Canceller (12.2(13)T and later)
- Impedance

Measuring Echo Using a 7960 IP Phone

- To enable the tone generator, first unlock the phone by entering **# while the phone is idle; Wait at least 10 seconds before proceeding to the next step
- Type **3 on the 7960/40 keypad while the phone is not on a call; This will enable the 'Tone' softkey for as long as this phone is registered to CallManager
- Place a call to the source of echo
- Once the call is established, hit the 'i' button twice; This will bring up the statistics for the call
- If the **3 worked, you should have a 'Tone' softkey available; Press it and the phone will begin to generate a 1004 Hz tone at -15 dB
- The only way to stop the tone is to end the call
- Once the tone is being generated, keep the call up and then follow the procedures that follow to measure the dB levels

Measuring Echo in Cisco IOS

• 1004Hz tone coming from IP Call Leg:

Show voice call x/y:z.ts

```
vnt-3725-34a#sho voice call 1/0:23.23
1/0:23 23
vtsp level 0 state = S_CONNECT
callid 0x8003 B23 state S_TSP_CONNECT clld 99915506 cllg 41008
- snip -
***DSP LEVELS***
TDM Bus Levels(dBm0): Rx -15.5 from PBX/Phone, Tx -15.5 to PBX/Phone
TDM ACOM Levels(dBm0): +0.0, TDM ERL Level(dBm0): +0.0
TDM Bgd Levels(dBm0): -81.9, with activity being voice
```

Measuring Echo in Cisco IOS

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Use the Cisco IOS command 'show call active voice' to look at the input and output levels for your call; The following is a call to a loopback number that just echos back whatever we send with no attenuation

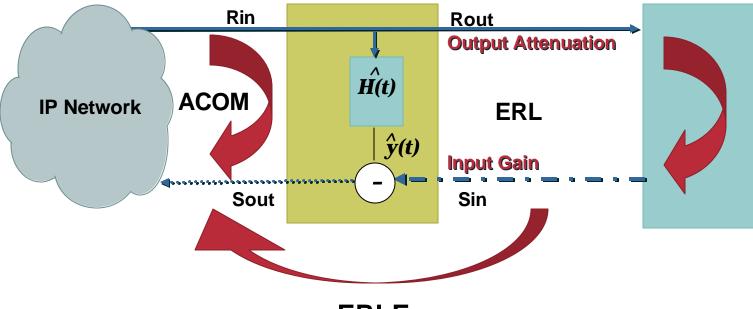
```
vdtl-3640-27a#sh call active voice
- snip -
OutSignalLevel=-15
InSignalLevel=-15
ERLLevel=25
- snip -
```

- The test tone is being output as -15 and is being looped back to us with 0dB loss, so it is coming back at -15 dB
- The ERL value here doesn't mean anything at this point since the ecan does not consider the input signal to be echo; The actual ERL is NOT really 25dB
- The OutSignalLevel shows the value of the level AFTER the output attenuation has been applied to the signal and InSignalLevel shows the level AFTER the input gain has been applied

Eliminating Echo

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ERLE

• Perform dB level adjustment procedure; Remember:

Output Attenuation of a signal is performed AFTER the echo canceller has 'seen' the original output signal

Input Gain of a signal is performed BEFORE the echo canceller has 'seen' the echo

Measuring Echo in Cisco IOS

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If we configure 1 dB of attenuation in each direction as follows:

```
voice-port 1/1:23
input gain -1
output attenuation 1
```

The resulting levels are as follows:

```
vdtl-3640-27a#sh call active voice
- snip -
OutSignalLevel=-16 -15 dB minus 1 from output attenuation = -16
InSignalLevel=-17 -16 dB minus 1 from input gain = -17
ERLLevel=11
- snip -
```

- Notice the OutSignalLevel is -16 because we attenuated the -15 dB signal by 1 dB; The InSignalLevel is -17 dB due to the input gain of -1
- At this point our real ERL is 2dB, however the Echo Canceller still does not acknowledge the input signal as echo

Measuring Echo in Cisco IOS

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If we configure 2 dB of attenuation in each direction as follows:

```
voice-port 1/1:23
input gain -2
output attenuation 2
```

The resulting levels are as follows:

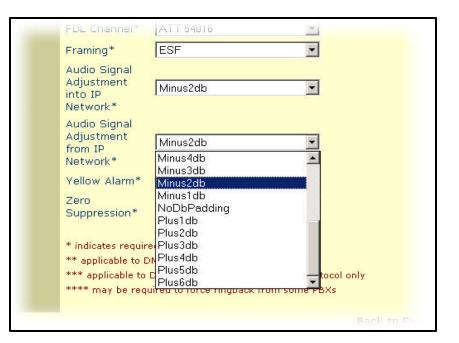
```
vdtl-3640-27a#sh call active voice
- snip -
OutSignalLevel=-17  -15 dB minus 2 from output attenuation = -17
InSignalLevel=-19  -17 dB minus 2 from input gain = -19
ERLLevel=4
-snip -
```

- Notice the OutSignalLevel is -17 because we attenuated the -15 dB signal by 2 dB; The InSignalLevel is -19 dB due to the input gain of -2
- Our expected ERL of 4dB is now correct
- The G.113 specification recommends a minimum ERL of 15 dB

- Same principles apply when troubleshooting echo on a 6608, however the commands are different
- From the CatOS prompt, type 'show port voice active <mod>/<port>' to see the active calls on that port
- The 6608 does not show you input and output levels, however it does show the ERL and ACOM
- ERL and ACOM values are displayed as 10x the dB value, so a value of 61 is really 6.1 dB
- Echo cancel coverage is not configurable on the 6608; It is roughly equivalent to 32 ms in Cisco IOS
- The 6608 always has the echo suppressor enabled to reduce convergence echo

Adjusting Levels on Cisco CallManager

- On Skinny/MGCP gateways, the input and output dB levels are adjusted from the CallManager administration web page
- The Output level is adjusted using the 'Audio Signal Adjustment from IP Network' menu; (Same as output attenuation, but positive values increase volume and negative values decrease volume)
- The Input level is adjusted using the 'Audio Signal Adjustment into IP Network' menu; (Same as input gain)
- You must reset the gateway for the changes to take effect



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Input Gain of -1 and Output Attenuation of 1

```
vdtl-Cat6k-PBX1> (enable) sh port voice active 4/8
Port 4/8:
  Channel #23:
                                              : 172.18.104.74
    Remote IP address
                                              : 24876
    Remote UDP Port
    ACOM Level Current
                                              : 45
    Call State
                                              : voice
    Codec Type
                                              : G711 ULAW PCM
    Coder Type Rate
                                              : 20
    ERL Level
                                              : 61
    Voice Activity Detection
                                              : disabled
    Echo Cancellation
                                              : enabled
    Fax Transmit Duration (ms)
                                              : 0
    Hi Water Playout Delay
                                              : 65
    Low Water Playout Delay
                                              : 64
    Receive Bytes
                                              : 373920
    Receive Delay
                                              : 64
    Receive Packets
                                              : 2337
    Transmit Bytes
                                              : 374240
    Transmit Packets
                                              : 2339
    Tx Duration (ms)
                                              : 42090
    Voice Tx Duration (ms)
                                              : 42090
```

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Input Gain of -3 and Output Attenuation of 3

```
vdtl-Cat6k-PBX1> (enable) sh port voice active 4/8
Port 4/8:
  Channel #23:
                                              : 172.18.104.74
    Remote IP address
    Remote UDP Port
                                              : 28146
    ACOM Level Current
                                              : 370
    Call State
                                              : voice
    Codec Type
                                              : G711 ULAW PCM
    Coder Type Rate
                                              : 20
    ERL Level
                                              : 60
    Voice Activity Detection
                                              : disabled
    Echo Cancellation
                                              : enabled
    Fax Transmit Duration (ms)
                                              : 0
    Hi Water Playout Delay
                                              : 65
    Low Water Playout Delay
                                              : 64
    Receive Bytes
                                              : 315520
                                              : 64
    Receive Delay
    Receive Packets
                                              : 1972
    Transmit Bytes
                                              : 315680
    Transmit Packets
                                              : 1973
    Tx Duration (ms)
                                              : 37990
                                              : 37990
    Voice Tx Duration (ms)
```

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Input Gain of -4 and Output Attenuation of 4

```
vdtl-Cat6k-PBX1> (enable) sh port voice active 4/8
Port 4/8:
  Channel #23:
                                              : 172.18.104.74
    Remote IP address
                                              : 24526
    Remote UDP Port
    ACOM Level Current
                                              : 349
    Call State
                                              : voice
    Codec Type
                                              : G711 ULAW PCM
    Coder Type Rate
                                              : 20
    ERL Level
                                              : 79
    Voice Activity Detection
                                              : disabled
    Echo Cancellation
                                              : enabled
    Fax Transmit Duration (ms)
                                              : 0
    Hi Water Playout Delay
                                              : 65
    Low Water Playout Delay
                                              : 64
    Receive Bytes
                                              : 503840
    Receive Delay
                                              : 64
    Receive Packets
                                              : 3149
    Transmit Bytes
                                              : 504160
    Transmit Packets
                                              : 3151
    Tx Duration (ms)
                                              : 62440
    Voice Tx Duration (ms)
                                              : 62440
```

Echo Canceller Coverage

- The echo canceller coverage time can be changed on IOS platforms using the echo-cancel coverage voice-port command
 - Router(config-voiceport)#echo-cancel coverage ? 16 16 milliseconds echo canceller coverage 24 24 milliseconds echo canceller coverage 32 32 milliseconds echo canceller coverage 64 64 milliseconds echo canceller coverage 8 8 milliseconds echo canceller coverage
- Echo canceller coverage is hard-coded to 32 ms on the 6608 gateway

Impedance Adjustment

- Echo can also be caused by an impedance mismatch if both sides are not configured identically
- Default of 600r Ohm is consistent with most lines on the PSTN and PBX's

```
vnt-1760-32(config-voiceport)#impedance ?
```

600c	600 Ohms complex
600r	600 Ohms real
900c	900 Ohms complex
complex1	complex 1
complex2	complex 2
vnt-1760-32a	a#sho voice port 2/0 inc Impedance
Impedance :	is set to 600r Ohm

Rules of Thumb

- Echo on one end is typically generated at other end
- Introduced by 2 to 4 wire conversion in hybrid and impedance mismatch or via acoustic feedback
- Bits don't leak—Echo is not introduced on digital links
- Must have sufficient ERL for ECANs to engage
- Longer coverage yields longer convergence time—more so on legacy echo cancellers. Configure the coverage so that it is long enough to cover the worst-case for your environment, but no higher
- Use **3 on 7960/40 to use the built-in 1004 Hz tone generator
- # or * DTMF tones approximate 1004Hz @ 0dB tones (on Analog phones if test gear isn't available)

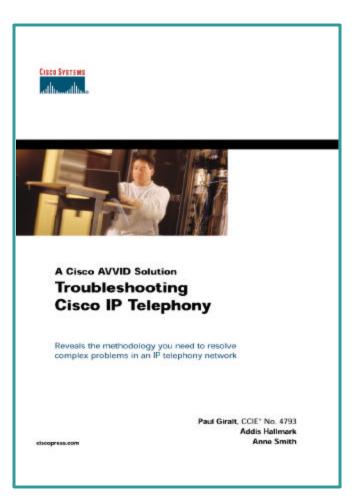
Summary

- Remember what is supposed to happen when there is no problem; simulate, re-create, stare and compare, etc...
- Traces are your friend!!!
- Assume that there is a perfectly logical explanation for the behavior you are investigating: 99% of the time, you'll be right
- For the remaining 1%, TAC will want to see the traces; be prepared to share the traces if need be
- Echo is the most common voice quality problem faced in the packet telephony networks. Proper analysis and methodical approach helps to resolve echo problems
- For further troubleshooting and information read the technical tips on Cisco TAC web site at <u>http://www.cisco.com/tac</u>

More Information

For more details on reading CCM traces, Troubleshooting Call Routing, and Echo, see here:

Troubleshooting Cisco IP Telephony ISBN 1-58705-075-7



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