

EVA Console Manager (EVCOM) Users Guide

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1 Revision History

Revision history:

Author	Date	Description
Quark	2011/09/25	Initial version



2 Introduction

2.1 Overview

EVCOM, Enhance VoIP Architecture Console Manager, is an application designed to manipulate DSP functionalities through ADAM (Adaptive DSP Access Manager) API and provide to user as a demo program for ADAM programming reference.

Though EVCOM was meant to be a demo application for programmer, it is also a handy and powerful tool to test DSP functionalities without any signaling application hook-up.

EVCOM is provided in source code basis, so user can trace the source code and learn how to program each function.

This document provides usage guidance from user's point of view. The EVCOM commands can be categorized into four operation objects: DSP, Channel, Stream, and Interface. EVCOM provides the commands to change the configuration or invoke certain process of specific object. See Chapter 4 for each command detail description.

2.2 Glossary

ADAM	Adaptive DSP Access	Manager
------	---------------------	---------

Cadence A combination of signal on and off for certain time is called a cadence.

Caller ID (CID) A telephony signal standard to indicate subscriber (caller) identification,

usually telephone number, and other information, such as user name, calling date and time. ** There is Type-1 caller ID which is known as the onhook caller ID. The caller ID is transmitted during the telephone ringing; There is also Type-2 caller ID which is known as the "call-waiting" caller ID or the off-hook caller ID. The caller ID is transmitted during a call-waiting

request (only certain countries provide Type-2 caller ID service).

CPT Call progress tone. Telephony signals used to indicate the state of service.

i.e. Dial-tone indicates a line is ready for dialing out. Busy-tone indicates a

line is occupied and cannot reach its destination.

Channel A DSP process path connecting the PCM I/O from a physical audio

hardware to a network CODEC I/O.

CNG Comfort Noise Generation. By incorporating with VAD and silence

compression and generate artificial background noise to save bandwidth

and improve talking experience.

CODEC Coded/Encoded, usually implies a process of conversion between raw data

and compressed (coded) data.

DAA Data Access Arrangement. A hardware component emulate a POTS phone

to provide FXO function.

DSP Digital Signal Processor

DTMF Dual-Tone Multi-Frequency, a telephone standard to indicate (signaling)

digits.

DTMF Relay A RFC standard (RFC2833 obsolete by RFC4733) to transmit DTMF



information in RTP payload instead of in-band audio to provide reliable

DTMF transmission.

Echo Cancellation

A process to remove echo.

(Echo Canceller)

EVA Enhanced VoIP Architecture.

FXO Foreign Exchange Office, a telephony endpoint (Telephone) or device used

to signal Central Office (CO) its request or response of a phone call.

FXS Foreign Exchange Station, a telephony endpoint or device at Central Office

(CO) side to provide signal and power for FXO.

Interface An interface is an abstract of a physical audio hardware.

OP Code EVCOM operation code, a short conversion of EVCOM command.

P-time / P-rate Packetization time (rate) used to negotiate and indicate the length (ms) of

the audio in each packet payload.

SAS Subscriber Alert Signal. A signal to alert the user (telephone) a call is

waiting, may be followed with type-2 caller ID.

Silence A method to save bandwidth consumption by transmit silence indication

Compressionpacket (SID) instead of full RTP payload when user is not talking.SLICSubscriber Line Interface Circuit. A hardware component emulate CO

service to provide FXS function.

Stream Stream is a path or process to disassembly sequential coded data (i.e.

audo), transmit over network, and reassembly the coded data on the far-

end to restore the original information.

VAD Voice Activity Detection. A method to assess the audio level to determine if

a user is talking.



3 Start using EVCOM

First, insert DSP Kernel Modules.

**Note: Depending on the (chipset) platform being used, the DSP kernel modules may be different or have extra module dependency. Please check the "User Guide" or "Readme" document for each platform for further detail. (Here is one example)

```
# insmod extern_osal_kernel_gpl.ko
# insmod osal_kernel.ko
osal_module.c:35 OSAL module_init OK

# insmod ve_vtsp_hw.ko
# insmod ve_vtsp_rt.ko
#
```

After Kernel Modules are loaded, execute EVCOM program.

```
#./evcom
Initializing ADAM ...
ADAM Version: 0.1
Number of DSP: 1
Number of Interface: 3

EVCom >>
```

Type "dq" or "dspquery" to check current DSP initialization status and capability,

**Note: The DSP capability varies on different (chipset) platform.

```
EVCom >> dq
DSP features:
Active status: Inactive
DSP ID: (0)
Number of Channel: 3
Stream per Channel: 2

EVCom >>
```

Type any invalid command will bring out the EVCOM command list,



EVCom >> ??			
DspInvoke	DspRevoke	DspQuery	DspConfigTone
ChanConfig	ChanPlayTone	ChanPlayCid	ChanPlayCid2
ChanPcmDump	StrmConfig	StrmStart	StrmstoP
StrmSendDtmfr	StrmPlayTone	InfcQuery	InfcConfigLine
InfcConfigHook InfcConfigRing		InfcRing	Quit
EVCom >>			

Before you can issue command to control the DSP, you must do "di" or "dspinvoke" first to initialize the DSP functionalities.

EVCom >> dct
DSP was not initialized, please run DspInvoke first!

EVCom >>

Type "di" or "dspinvoke" to initialize DSP core. Once initialization completed, the DSP is ready to work under your command in EVCOM.



```
EVCom >> di
Invoking DSP ...
DSP initialization completed! You can issue command to use DSP now.
EVCom >>
== Event Once ==: Line
Interface (2) detected Line-Active-Reversed.
== Event Once ==: Hook
Interface (2) on-hooked.
== Event Once ==: Hook
Interface (1) on-hooked.
== Event Once ==: Hook
Interface (0) on-hooked.
EVCom >> dq
DSP features:
Active status: Active
DSP ID: (0)
Number of Channel: 3
Stream per Channel: 2
EVCom >>
```

To shutdown the DSP, type "dr" or "dsprevoke" to terminate DSP process.

**Note: Once DSP is revoked, you cannot re-invoke DSP immediately. You must exit EVCOM, remove all DSP kernel modules, and start from again from inserting DSP modules.

```
EVCom >> dr
```

Once DSP is revoked, it cannot be invoked again until re-insert DSP modules, sure? (type 'yes' to revoke)

Revoking DSP ...

DSP shutdown completed! You can quit evcom safely now.

EVCom >> di

DSP has been revoked, please exit evcom, remove and re-insert DSP modules, and start again!

EVCom >>

To leave EVCOM, type "q" or "quit" to exit the program.



**Note: If DSP has not been revoked before quitting EVCOM, the exit process will revoke DSP automatically.

EVCom >> q
Quit evcom and shutdown DSP, sure? (type 'yes' to quit) yes
Exit evcom goodbye!
Closing ADAM : Execution success!
#



4 EVCOM Command

4.1 Dsplnvoke

Description:

Initialize and start the DSP process.

OP Code: di

Parameters & Attributes: None

Syntax:

<dspinvoke | di>

Usage example:

EVCom >> di

Invoking DSP ...

DSP initialization completed! You can issue command to use DSP now.

EVCom >>

4.2 DspRevoke

Description:

Terminate DSP process.

OP Code: dr

Parameters & Attributes: None

Syntax:

<dsprevoke | dr>

Usage example:

 ${\sf EVCom}>> {\sf dr}$

Once DSP is revoked, it cannot be invoked again until re-insert DSP modules, sure? (type 'yes' to revoke) yes

Revoking DSP ...

DSP shutdown completed! You can quit evcom safely now.

EVCom >>



4.3 DspQuery

Description:

Query DSP configuration.

OP Code: dq

Parameters & Attributes: None

Syntax:

<dspquery | dq>

Usage example:

EVCom >> dq

DSP features:

Active status: Active

DSP ID: (0)

Number of Channel: 3 Stream per Channel: 2

EVCom >>

4.4 DspConfigTone

Description:

Change DSP tone configuration.

OP Code: dct

Parameters	Name	Value	Description
Tone ID	tone	0~9	Note: tone[0] is reserved for silence and
10110 15	10110		cannot be configured.
Frequency [0 ~ 3]	f0 ~ f3	[frequency,power]	Frequency: 0~4000(Hz)
r requericy [o ~ 5]	10 ~ 13	[frequency,power]	Power: -40~0(db)
		[base_frequency,	Base frequency: 0 ~ 4000(Hz)
Modulation	mf	modulation_frequency,	Modulation frequency: 0 ~ 4000(Hz)
Frequency		modulation_power,	Modulation power: -40 ~ 0(db)
		modulation_depth]	Modulation depth: 0 ~ 256
	cad0 ~	[on_time,off_time,	On time: 0 ~ 65535(ms)
Cadence [0 ~ 2]	cad2	repeat]	Off time: 0 ~ 65535(ms)
			Repeat: 0 ~ 65535



<dspinvoke | di> tone=[tone_id] \$attr0=[value] \$attr1=[value] ...

Usage Example:

EVCom >> dspconfigtone tone=1 type=reg f0=350,-20 f1=440,-16 cad0=1000,0,10

Tone (1) configuration: Tone type: Regular tone

(a) 2501 @ 20

Frequency(0) = 350Hz @ -20db

Frequency(1) = 440Hz @ -16db

Freqency(2) = 0Hz @ 0db

Frequency(3) = 0Hz @ 0db

Candence(0) = 1000(ms)/On, 0(ms)/Off, x 10 (times)

Candence(1) = 0(ms)/On, 0(ms)/Off, x 0 (times)

Candence(2) = 0(ms)/On, 0(ms)/Off, x 0 (times)

EVCom >> dct tone=2 type=mod mf=1300,25,-6,40 cad0=500,200,5

Tone (2) configuration:

Tone type: Modulated tone

Base frequency = 1300Hz

Modulation frequency = 25Hz

Modulation power = -6db

Modulation depth = 40

Candence(0) = 500(ms)/On, 200(ms)/Off, x 5 (times)

Candence(1) = 0(ms)/On, 0(ms)/Off, x 0 (times)

Candence(2) = 0(ms)/On, 0(ms)/Off, x 0 (times)

EVCom >>

4.5 ChanConfig

Description:

Change channel configuration.

OP Code: cc

Parameters	Name	Value	Description
Channel ID	ch	[0 ~ MAX_CHANNEL]	**Note: Depending on the exact channel numbers on the platform.
Detection Mask	det	dtmf[+ -], mdm[+ -], cpt[+ -], cid[+ -]	dtmf: Detect DTMF tone mdm: Detect modem (fax) tone. cpt: Detect call progress tone. (i.e. dial-tone, busy-tone, etc.) cid: Detect caller ID. (For FXO)



Parameters	Name	Value	Description
			**Use [+ -] to turn on/off the detector, i.e. dtmf+,cid-, which is to enable DTMF detector and disable caller ID detector. **Default: All On.
Echo Canceller	ес	[on off]	Enable or disable echo canceller. *Default: On
Tx Gain Amplify	tx	-20 ~ 20	Change Tx gain power between +/-20(db). *Default: 0
Rx Gain Amplify	rx	-20 ~ 20	Change Rx gain power between +/-20(db). *Default: 0

<chanconfig | cc> ch=[channel_id] \$attr0=[value] \$attr1=[value] ...

Usage Example:

EVCom >> chanconfig ch=1 ec=on det=mdm- tx=-3 rx=3

Channel (1) configuration:

Enabled Detectors:

Active : dtmf (DTMF_TONE)

Inactive : mdm (FAX/MODEM_TONE)
Active : cpt (CALL_PROGRESS_TONE)

Active : cid (Caller_ID)

EC : Enable Tx Gain: -3db Rx Gain: 3db

EVCom >>

4.6 ChanPlayTone

Description:

Generate a tone to the channel.

OP Code: cpt

Parameters	Name	Value	Description
Channel ID	ch	[0 ~ MAX_CHANNEL]	**Note: Depending on the exact channel
			numbers on the platform.
			Tone ID to be played. User can put several
Tone ID	tone	[0 ~ 9],	tone IDs and play in sequence, i.e
			tone=1,5,8,2,3
Repeat Time	rpt	0 ~ 65535	Times of repeat the tone sequence



<chanplaytone | cpt> ch=[channel_id] tone=[tone_id<,tone_id,...>] rpt=[repeat]

Usage Example:

EVCom >> cpt ch=0 tone=1,3,5,2,4 rpt=5

Tone repeat times: 5

Playing Tone (1).

Playing Tone (3).

Playing Tone (5).

Playing Tone (2).

Playing Tone (4).

Generate tone(s) on channel (0), please check the telephone!

EVCom >>

4.7 ChanPlayCid

Description:

Generate a caller ID to the channel.

**Note: This feature is for advance user who wants to manually transmit caller ID to the telephone. However, to successfully generate caller ID to the telephone, user might need to configure the SLIC to correct state (On-hook Transmission) so the SLIC can transmit caller ID correctly.

OP Code: cpc

Parameters & Attributes:

Parameters	Name	Value	Description
Channel ID	ch	[0 ~ MAX_CHANNEL]	**Note: Depending on the exact channel numbers on the platform.
Caller ID	cid	[number],	EVCOM only generate number

Syntax:

<chanplaycid | cpc> ch=[channel_id] cid=[number]

Usage Example:

EVCom >> cpc ch=0 cid=7654321

Generate caller ID [7654321] on channel (0), please check the telephone!

EVCom >>

4.8 ChanPlayCid2

Description:

Generate a type 2 caller ID with SAS signal to the channel.



OP Code: cpc2

Parameters & Attributes:

Parameters	Name	Value	Description
Channel ID	ch	[0 ~ MAX_CHANNEL]	**Note: Depending on the exact channel
Charmerib			numbers on the platform.
Caller ID	cid	[number],	EVCOM only generate number

Syntax:

<chanplaycid2 | cpc2> ch=[channel_id] cid=[number]

Usage Example:

EVCom >> cpc2 ch=1 cid=0987654321

Generate caller ID [0987654321] on channel (1), please check the telephone!

EVCom >>

4.9 ChanPcmDump

Description:

Enable PCM dump for debugging.

** Note: User can use this function to dump channel PCM data to designated network address (PC) and capture PCM dump log for debugging with sniffer software, such as Wireshark.

OP Code: cpd

Parameters & Attributes:

Parameters	Name	Value	Description
Channel ID	ch	[0 ~ MAX_CHANNEL]	**Note: Depending on the exact channel
			numbers on the platform.
			IP address of destination PC to receive PCM
IP address	lp	[IP_Address],	dump log.
			Set ip =0.0.0.0 will disable the dump.

Syntax:

<chanpcmdump | cpd> ch=[channel_id] ip=[ip]

Usage Example:

EVCom >> chanpcmdump ch=0 ip=192.168.1.2 Dump channel (0) PCM to (192.168.1.2).

EVCom >>



4.10 StrmConfig

Description:

Change a stream configuration.

**Note: User can change a stream configuration anytime no matter if the stream is started or not and the configuration change will take effect immediately.

OP Code: sc

Parameters & Attributes:

Parameters	Name	Value	Description
Stream ID	st	[channel_id:stream_id]	**Note:
		[0~MAX_CH:0~MAX_ST]	Channel ID and Stream ID depending on
			the exact channel and stream numbers on
			the platform.
Source Session	src	[IP_Address:port]	Source IP address and port used to
Addresss			transmit/receive RTP packet.
			** Note: RTCP use [port + 1]
Destination	dst	[IP_Address:port]	Destination IP address and port used to
Session Address			transmit/receive RTP packet.
			** Note: RTCP use [port + 1]
CODEC	codec	<g711a g711u="" g722="" td="" ="" <=""><td>Payload type to be used for stream</td></g711a>	Payload type to be used for stream
		g726 g729 t38>	transmission.
PTIME	ptime	<10 20 30 40 50	Packetizaiton time, up to 60ms.
		60>	
DTMF Relay	dtmfr	<on off="" =""></on>	Enable or disable DTMF relay.
Silence	scomp	<on off="" =""></on>	Enable or disable silence compression.
Compression			
Stream Direction	dir	<sr in="" ro="" so="" =""></sr>	sr: Send-Receive
			so: Send-Only
			ro: Receive-Only
			in: Inactive
Jitter Buffer	jb	[, jb_init_size,	a=active, f=fixed,
		jb_max_size]	JB size is between 0~800 ms

Syntax:

<strmconfig | sc> st=[channel_id:stream_id] \$attr0=[value] \$attr1=[value] ...



Usage Example:

Channel 0 -> Stream 0 Configuration:

Stream state: Inactive

Source address: 192.168.1.1:5566

Destination address: 192.168.1.2:5566

Codec: G.711a Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >>

4.11 StrmStart

Description:

Start a streaming process.

OP Code: ss

Parameters & Attributes:

Parameters	Name	Value	Description
			**Note:
Otros and ID	ot.	[channel_id:stream_id]	Channel ID and Stream ID depending on
Stream ID	st	[0~MAX_CH:0~MAX_ST]	the exact channel and stream numbers on
			the platform.

Syntax:

<strmstart | ss> st=[channel_id:stream_id]

Usage Example:

EVCom >> ss st=0:0

Enable streaming process on channel (0) -> stream (0).

EVCom >>

4.12 StrmstoP

Description:

Stop a streaming process.



OP Code: sp

Parameters & Attributes:

Parameters	Name	Value	Description
			**Note:
Ctra are ID	ot.	[channel_id:stream_id]	Channel ID and Stream ID depending on
Stream ID	st	[0~MAX_CH:0~MAX_ST]	the exact channel and stream numbers on
			the platform.

Syntax:

<strmstop | sp> st=[hannel_id:stream_id]

Usage Example:

EVCom >> sp st=0:0

Disable streaming process on channel (0) -> stream (0).

EVCom >>

4.13 StrmSendDtmfr

Description:

Send DTMF relay (RFC2833/4733) packet to the stream.

**Note: The stream has to be started first before executing this command.

OP Code: ssd

Parameters & Attributes:

Parameters	Name	Value	Description
			**Note:
Stream ID	st	[channel_id:stream_id]	Channel ID and Stream ID depending on
Siream iD	Si	[0~MAX_CH:0~MAX_ST]	the exact channel and stream numbers on
			the platform.
DTMF Digit	dtmf	[O O * # A P C D]	DTMF digit to be send in DTMF relay
DTIVIE DIGIL	aum	[0 ~ 9, *, #, A, B, C, D]	packet.
Duration	dur	0 ~ 65536	DTMF duration (ms). Minimum length must
Duration	uui	0 ~ 00000	be greater or equal to 10ms.

Syntax:

<strmsenddtmfr | ssd> st=[channel_id:stream_id] dtmf=<[0~9] | * | # | [a~d]> dur=[duration(ms), dur >= 10ms]

Usage Example:



EVCom >> ss st=0:0

Enable streaming process on channel (0) -> stream (0).

EVCom >> ssd st=0:0 dtmf=5 dur=5000

Generate DTMF_5 relay packet to channel (0) -> stream (0), dur: 5000(ms) please check with sniffer!

EVCom >>

4.14 StrmPlayTone

Description:

Generate tone to the stream.

OP Code: spt

Parameters & Attributes:

Parameters	Name	Value	Description
			**Note:
Stream ID	st	[channel_id:stream_id]	Channel ID and Stream ID depending on
SileaniiD	Si	[0~MAX_CH:0~MAX_ST]	the exact channel and stream numbers on
			the platform.
			Tone ID to be played. User can put several
Tone ID	tone	[0 ~ 9],	tone IDs and play in sequence, i.e
			tone=1,5,8,2,3
Repeat Time	rpt	0 ~ 65535	Times of repeat the tone sequence

Syntax:

<strmplaytone | spt> st=[channel_id:stream_id] tone=[tone_id<,tone_id,...>] rpt=[repeat]

Usage Example:

EVCom >> spt st=0:0 tone=1 rpt=5

Playing Tone (1).

Generate tone(s) on channel (0) -> stream (0), please check the peer!

EVCom >>

4.15 InfcQuery

Description:

Query interface configuration.

OP Code: iq



Parameters	Name	Value	Description	
Interface ID	;f	[0 ~	**Note: Depending on the exact interface	
interface ID	lt lt	MAX_INTERFACE]	numbers on the platform.	

<infcquery | iq> if=[interface_id]

Usage Example:

EVCom >> iq if=0

Interface (0) configuration:

Interface type: FXS

Line State: Line-Active-Forward

Hook State: ON-HOOK Ring configuration:

Cadence0: 500(ms)/on, 1000(ms)/off Cadence1: 0(ms)/on, 0(ms)/off Cadence2: 0(ms)/on, 0(ms)/off

Caller ID: number=[]

Caller ID Generation at (0)th break

EVCom >> iq if=1

Interface (1) configuration:

Interface type: FXS

Line State: Line-Active-Forward

Hook State: ON-HOOK Ring configuration:

Cadence0: 500(ms)/on, 1000(ms)/off Cadence1: 0(ms)/on, 0(ms)/off Cadence2: 0(ms)/on, 0(ms)/off

Caller ID: number=[]

Caller ID Generation at (0)th break

EVCom >> iq if=2

Interface (2) configuration:

Interface type: FXO

Line State: Line-Active-Reversed

Hook State: ON-HOOK

EVCom >>

4.16 InfcConfigLine

Description:

Change interface line state, only works for FXS interface.



OP Code: icl

Parameters & Attributes:

Parameters	Name	Value	Description	
Interface ID	if	[0 ~	**Note: Depending on the exact interface	
interiace ib	"	MAX_INTERFACE]	numbers on the platform.	
			down: Power down the line	
			fwd: Set line power active forward.	
	line		rev: Set line power active reverse.	
Line State		<down fwd="" rev="" ring<="" td="" =""><td>ring: Set line ringing. Note: Based on</td></down>	ring: Set line ringing. Note: Based on	
Line State		sleep>	interface ring configuration.	
			sleep: Set line in low power mode. Note:	
			Depending on the platform if the feature	
			supported.	

Syntax:

<infcconfigline | icl> if=[interface_id] line=<down | fwd | rev | ring | sleep>

Usage Example:

EVCom >> icl if=0 line=rev

Line State: Line-Active-Reversed

EVCom >>

== Event Once ==: Line

Interface (0) detected Line-Active-Reversed.

EVCom >>

4.17 InfcConfigHook

Description:

Change interface hook state, only works for FXO interface.

OP Code: ich

Parameters	Name	Value	Description
Interface ID	;ŧ	[0 ~	**Note: Depending on the exact interface
interface ib	"	MAX_INTERFACE]	numbers on the platform.
			on: Set On-hook
Hook State	hook	<on flash="" off="" =""></on>	off: Set Off-hook
			flash: Set Hook-flash



<infcconfighook | ich> if=[interface_id] hook=<on | off | flash>

Usage Example:

EVCom >> ich hook=off if=2 Hook State: OFF-HOOK

EVCom >>

== Event Once ==: Hook Interface (2) off-hooked.

== Event Once ==: Hook Interface (0) off-hooked.

EVCom >>

4.18 InfcConfigRing

Description:

Change interface ring profile configuration, only works for FXS interface.

OP Code: icr

Parameters & Attributes:

Parameters	Name	Value	Description
Interface ID	if	[0 ~	**Note: Depending on the exact interface
		MAX_INTERFACE]	numbers on the platform.
Ring Cadence	cad0 ~	[on_time,off_time]	Signal on/off time (ms).
Tring Gadenee	cad2		Olgital Olivon time (ma).
Ring Duration	dur	0 ~ 65535	(ms)
Caller ID	cid	[number]	Caller ID number.
Caller ID At (N-th	cidb	0 ~ 255	Set N-th ring-break to generate caller ID.
Ring-break)	Club	0 ~ 200	Set N-th hing-break to generate caller ib.

Syntax:

<infcconfigring | icr> if=[interface_id] \$attr0=[value] \$attr1=[value] ...

Usage Example:



EVCom >> icr if=0 dur=500 cad0=500,1000 cad1=0,0 cad2=0,0 cid=1234 cidb=0

Ring configuration:

Cadence0: 500(ms)/On, 1000(ms)/Off Cadence1: 0(ms)/On, 0(ms)/Off Cadence2: 0(ms)/On, 0(ms)/Off

Ring duration= 500(ms)
Caller ID: number=[1234]

Caller ID Generation at (0)th ring break.

EVCom >>

4.19 InfcRing

Description:

Control interface ringing, only work for FXS type interface.

OP Code: ir

Parameters & Attributes:

Parameters	Name	Value	Description
Interface ID	if	[0 ~ MAX_INTERFACE]	**Note: Depending on the exact interface numbers on the platform.
Ring Duration	dur	0 ~ 65535	(ms) Note: optional, if not provided, it will use the ring configuration. Set dur=0 to disable ring.

Syntax:

<infcring | ir> if=[interface_id] dur=[duration(ms)], Note: duration is optional.

Usage Example:

EVCom >> icr if=0 dur=10000

Ring configuration:

Cadence0: 500(ms)/On, 1000(ms)/Off Cadence1: 0(ms)/On, 0(ms)/Off Cadence2: 0(ms)/On, 0(ms)/Off Ring duration= 10000(ms)

Caller ID: number=[]

Caller ID Generation at (0)th ring break.

EVCom >>



4.20 Quit

Description:

Shutdown DSP process and leave this application.

OP Code: q

Parameters & Attributes: None

Syntax: <quit | q>

Usage example:

EVCom >> q

Quit evcom and shutdown DSP, sure? (type 'yes' to quit)



5 Event

5.1 Event Message

EVCOM reports DSP and Interface event in the below format,

[T:<timestamp>] [Event Header]:[Event Type] [Object] [Event Description]

Examples:

[T:000000710] == Event Once ==: Hook Interface (0) on-hooked.

[T:0000017350] ++ Event Begin ++: Tone Channel (1) tone[DTMF_1] detected.

[T:0000018380] -- Event End --: Tone Channel (1) tone[DTMF_1] detected.

Description:

Field	Data	Description
T:Timestamp	0 ~ 4294967296	DSP (or CPU) ticks used to
		note or calculate the time of a
		event.
Event Header	== Event Once ==	Indicate whether it is a
	++ Event Begin ++	beginning or an end of an
	Event End	event, or an instant event.
Event Type	CID	Indicate the event subject.
	Tone	
	Line	
	Hook	
Object	Channel	Indicate the subject object
	Stream [Channel -> Stream]	where the event is reported for.
	Interface	
Event Description	Caller ID	Caller ID / Tone: See channel
	Tone	event.
	Line State	Line/Hook State: See interface
	Hook State	event

5.2 Channel Event

5.2.1 CID

Caller ID event, EVCOM reports a CID event when a Type-I/II caller ID is detected on a channel.



[T:0002617420] == Event Once ==: CID

Channel (2) detected caller ID:

Number: 123

5.2.2 Tone

Tone event, EVCOM reports a tone event when a DTMF or Modem/FAX tone is detected on a channel. Call progress tone (CPT) detection only works on FXO interface.

A tone event will be reported twice at the beginning and the end of detection with DSP (or CPU) ticks (ms). User can calculate how long the tone last by the ticks provided.

[T:0003135790] ++ Event Begin ++: Tone

Channel (1) tone $[DTMF_4]$ detected.

[T:0003136360] -- Event End --: Tone Channel (1) tone[DTMF_4] detected.

[T:0003268200] ++ Event Begin ++: Tone

Channel (0) tone[CED] detected.

[T:0003269580] -- Event End --: Tone Channel (0) tone[CED] detected.

5.3 Stream Event

Not used in EVCOM.

5.4 Interface Event

5.4.1 Line

Line State:

Line State	Description
LINE_DOWN	Indicate a line power feed is out.
LINE_ACTIVE_FWD	Indicate a line in active sate with polarity forward power feed.
LINE_ACTIVE_REV	Indicate a line in active sate with polarity reverse power feed.
LINE_BUSY	Indicate a line is currently used by user (off-hook).
LINE_RING	Indicate a line is ringing.
LINE_RING_PAUSE	Indicate a line is in the break of ringing.
LINE_SLEEP	Indicate a line is in low power mode. (Depends on the hardware
	support)
LINE_ERROR	Indicate a line is problematic and cannot get its state.



Example:

EVCom >> icl line=down if=1 Line State: Line-Power-Down

EVCom >> icl line=fwd if=1 Line State: Line-Active-Forward

EVCom >>

[T:0003800400] == Event Once ==: Line Interface (1) detected Line-Active-Forward.

EVCom >> icl line=rev if=1

Line State: Line-Active-Reversed

EVCom >>

[T:0003814310] == Event Once ==: Line Interface (1) detected Line-Active-Reversed.

5.4.2 Hook

Hook State:

Hook State	Description
HOOK_SEIZE	Indicate the user has lifted the handset and reached off-hook
	threshold.
HOOK_RELEASE	Indicate the user has hung up the handset and reached the on-
	hook threshold.
HOOK_FLASH	Indicate the user has press hook flash key and meet the hook-
	flash threshold.
HOOK_PULSE_[1 ~ 20]	Indicate a hook pulse has met its make/break threashold.
HOOK_ERROR	Indicate an unknown hook state.

Example:

[T:0003886310] == Event Once ==: Hook

Interface (1) off-hooked.

[T:0005666040] == Event Once ==: Hook

Interface (1) hook-flashed.

[T:0005669090] == Event Once ==: Hook

Interface (1) on-hooked.



5.5 Jitter Buffer Event

EVCOM reports a Jitter Buffer event periodically when stream is connected. There are packet drop, packet loss compensated, and JB size information showed for user reference. Example:

JB Channel (1) Stream (0):

Packet Drop: 0

Packet Loss Compensated: 0

JB Size: 80.

JB Channel (1) Stream (0):

Packet Drop: 0

Packet Loss Compensated: 46

JB Size: 80.

JB Channel (1) Stream (0):

Packet Drop: 0

Packet Loss Compensated: 145

JB Size: 80.

5.6 T38 Event

EVCOM reports a T38 event when a T38 event is detected in FAX process.

T38 State:

T38 State	Description
T38_DISCONN	Indicate the T38 stream is disconnected.

Example:

T38 channel (1) stream (0) status(T38 Disconnected).



6 Test Cases

6.1 Create Two-way call

In this example, we setup a two-way stream between interface_0 (FXS) and interface_1 (FXS) to create a two-way call.

EVCom >> sc st=0:0 src=127.0.0.1:5000 dst=127.0.0.1:6000 dir=sr

Channel 0 -> Stream 0 Configuration:

Stream state: Active

Source address: 127.0.0.1:5000 Destination address: 127.0.0.1:6000

Codec: G.711a Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> sc st=1:0 src=127.0.0.1:6000 dst=127.0.0.1:5000 dir=sr

Channel 1 -> Stream 0 Configuration:

Stream state: Inactive

Source address: 127.0.0.1:6000 Destination address: 127.0.0.1:5000

Codec: G.711u Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> ss st=0:0

Enable streaming process on channel (0) -> stream (0).

EVCom >> ss st=1:0

Enable streaming process on channel $(1) \rightarrow stream (0)$.

EVCom >>

6.2 Create Three-way call

Start two streams on the same channel and it will be conference automatically. User can use stream direction to control a stream on-hold.



EVCom >> sc st=0:0 src=192.168.1.56:5000 dst=192.168.1.200:5000 dir=sr

Channel 0 -> Stream 0 Configuration:

Stream state: Active

Source address: 192.168.1.56:5000

Destination address: 192.168.1.200:5000

Codec: G.711a Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> sc st=**0:1** src=**192.168**.1.**56:6**000 dst=**192.168.1.201:5**000 dir=sr

Channel 1 -> Stream 0 Configuration:

Stream state: Inactive

Source address: **192.168**.1**.56**:**6**000 Destination address: **192.168**.1**.201**:**5**000

Codec: G.711u Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> ss st=0:0

Enable streaming process on channel (0) -> stream (0).

EVCom >> ss st=**0:1**

Enable streaming process on channel (1) -> stream (0).

EVCom >>

6.3 Echo cancellation Test

Create a two-way call same as 6.1. Enable and disable the echo cancellation on one channel and check the difference at the far-end.



(Do the same steps in 6.1 to create a two-way call)

.....

EVCom >> cc ch=0 ec=off Channel (0) configuration:

Enabled Detectors:

Active : dtmf (DTMF_TONE)

Active : mdm (FAX/MODEM_TONE)
Active : cpt (CALL_PROGRESS_TONE)

Active : cid (Caller_ID)

EC : Disable Tx Gain: 0db Rx Gain: 0db

EVCom >> cc ch=**0** ec=**on** Channel (**0**) configuration:

Enabled Detectors:

Active : dtmf (DTMF_TONE)

Active : mdm (FAX/MODEM_TONE)
Active : cpt (CALL_PROGRESS_TONE)

Active : cid (Caller_ID)

EC : Disable
Tx Gain: 0db
Rx Gain: 0db

6.4 FAX Pass-Through Test

Repeat the same step in 6.1 to create a two-way call first, then change the codec to g711a or g711u. Manually disable echo canceller. Connect two FAX machine to the interfaces and manually start the FAX transmission and receiving.



(Do the same steps in 6.1 to create a two-way call)

EVCom >> sc st=0:0 codec=g711u Channel 0 -> Stream 0 Configuration:

Stream state: Active

Source address: 127.0.0.1:5000 Destination address: 127.0.0.1:6000

Codec: G.711u Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> sc st=1:0 codec=g711u Channel 1 -> Stream 0 Configuration:

Stream state: Active

Source address: 127.0.0.1:6000 Destination address: 127.0.0.1:5000

Codec: G.711u Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> cc ch=0 ec=off Channel (0) configuration:

Enabled Detectors:

Active : dtmf (DTMF_TONE)

Active : mdm (FAX/MODEM_TONE)
Active : cpt (CALL_PROGRESS_TONE)

Active : cid (Caller_ID)

EC : Disable
Tx Gain: 0db
Rx Gain: 0db

EVCom >> cc ch=1 ec=off Channel (1) configuration:

Enabled Detectors:

Active : dtmf (DTMF_TONE)

Active : mdm (FAX/MODEM_TONE)
Active : cpt (CALL_PROGRESS_TONE)

Active : cid (Caller_ID)

EC : Disable Tx Gain: 0db Rx Gain: 0db



6.5 FAX Relay (T.38) Test

Repeat the same step in 6.1 to create a two-way call first, then change the codec to t38. Connect two FAX machine to the interfaces and manually start the FAX transmission and receiving.

(Do the same steps in 6.1 to create a two-way call)

.....

EVCom >> sc st=0:0 codec=t38

Channel 0 -> Stream 0 Configuration:

Stream state: Active

Source address: 127.0.0.1:5000 Destination address: 127.0.0.1:6000

Codec: T.38 Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >> sc st=1:0 codec=t38

Channel 1 -> Stream 0 Configuration:

Stream state: Active

Source address: 127.0.0.1:6000 Destination address: 127.0.0.1:5000

Codec: T.38 Ptime: 20

Silence compression: Active

DTMF Relay: Active

Stream direction: Send & Receive

EVCom >>