



Adaptive DSP Access Manager (ADAM) Programmer Reference Manual

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Table of Contents

Table of Contents	2
1 Revision History	5
2 Introduction.....	7
2.1 EVA Framework	7
2.2 Object Model.....	8
2.3 ADAM API Command and Event.....	9
2.4 Glossary.....	12
3 ADAM Management.....	14
3.1 adamInit	14
3.2 adamExit.....	14
3.3 adamQuery	15
3.4 adamPollEvent	15
4 DSP Control	17
4.1 adamDspInvoke.....	17
4.2 adamDspRevoke	17
4.3 adamDspQuery	18
4.4 adamDspConfigTone.....	19
4.5 adamDspConfigCPT.....	20
4.6 adamDspReplaceData	22
4.7 adamPcmLoopback	23
5 Channel Control	24
5.1 adamChanQuery	24
5.2 adamChanConfig.....	25
5.3 adamChanPlayTone.....	27
5.4 adamChanStopTone	28
5.5 adamChanPlayDtmf	28
5.6 adamChanPlayDtmfCid	29
5.7 adamChanPlayCid.....	30
5.8 adamChanPlayType2Cid.....	31
5.9 adamChanDumpPcm	32
5.10 adamRtpLoopback	33
5.11 adamDspLoopback.....	33
6 Stream Control	35
6.1 adamStrmQuery	35
6.2 adamStrmConfig.....	36
6.3 adamStrmStart.....	37
6.4 adamStrmStop.....	38
6.5 adamStrmStopAll.....	39
6.6 adamStrmSendDtmfr	40
6.7 adamStrmPlayTone	40
6.8 adamStrmStopTone	41

6.9	adamStrmQueryMediaInfo	42
6.10	adamStrmResetMediaInfo	43
6.11	adamStrmPlayDtmf	44
7	Interface Control	45
7.1	adamInfcQuery	45
7.2	adamInfcConfigLine	46
7.3	adamInfcConfigHook	47
7.4	adamInfcConfigRing	47
7.5	adamInfcRing	48
7.6	adamInfcStopRing	49
7.7	adamInfcConfigHookTs	50
7.8	adamInfcLineTest	51
7.9	adamInfcCULineTest	51
7.10	adamInfcSlicTypeQuery	52
7.11	adamInfcRingParams	52
7.12	adamInfcDcFeedParams	53
7.13	adamInfcConfigHwProf	53
8	ADAM EVENT Processing	55
9	Appendix: Constant and Enumeration	58
9.1	Constant	58
9.2	activeState_e	58
9.3	blockMode_e	59
9.4	chanId_e	59
9.5	cidFormat_e	59
9.6	codec_e	59
9.7	dspld_e	60
9.8	dtmf_e	60
9.9	ecTail_e	60
9.10	enableControl_e	61
9.11	evaBool_e	61
9.12	eventCode_e	61
9.13	eventEdge_e	61
9.14	exCode_e	62
9.15	hookState_e	62
9.16	infclId_e	63
9.17	infcType_e	63
9.18	ipVer_e	63
9.19	jbMode_e	63
9.20	lineState_e	63
9.21	polDir_e	64
9.22	pTime_e	64
9.23	strmDir_e	64
9.24	strmId_e	64
9.25	t38State_e	64
9.26	toneCode_e	65
9.27	toneDir_e	65

9.28	toneType_e.....	66
9.29	MWIValue_e	66
9.30	hwProf_e.....	66
10	Appendix: Data Structure	68
10.1	cadence_t	68
10.2	chanConfig_t.....	68
10.3	cid_t	69
10.4	cpt_t.....	70
10.5	dspFeature_t.....	72
10.6	eventContext_u	73
10.7	event_t.....	75
10.8	hookThreshold_t.....	76
10.9	infcConfig_t.....	77
10.10	jbConfig_t.....	78
10.11	mediaInfo_t.....	79
10.12	netAddr_t	79
10.13	ringProfile_t.....	80
10.14	session_t.....	80
10.15	strmAttr_t	81
10.16	strmConfig_t	82
10.17	t38Ctrl_t	83
10.18	tone_t.....	83
10.19	toneSeq_t	84
10.20	infcLineTest_t	85
10.21	SlicParams_t.....	85
10.22	InfcRingParams _t.....	85
10.23	InfcDcFeedParams _t.....	86
10.24	infcUnicomLineTest _t.....	86
10.25	replaceData_t	88
11	Appendix: Default Call Progress Tone Profile	89

1 Revision History

Revision history:

Revision	Author	Date	ADAM_VERSION	Description
1.0.0	Quark	2011/09/28	0.1.3	Initial version
1.0.1	Serena	2011/11/11	0.1.3	Add new features, 1. T.38 event. 2. RTCP port in netAddr_t. 3. CPT Tone detection configuration API. 4. Hook state threshold configuration API. 5. CID format configuration. 6. Remote DTMF event.
1.0.2	PTChen	2011/11/14	0.1.3	Add toneCode_e element, 1. TONE_FAX_TIMEOUT 2. TONE_FAX_CM
1.0.3	PTChen	2011/12/30	0.1.3	Add session_t element, 1. rtpTosVal 2. rtcpTosVal Add strmAttr_t element, dtmfRemove
1.0.4	PTChen	2012/01/06	0.1.3	Add CID format – CID_FORMAT_ETSI_RPAS.
1.0.5	LyricTian	2012/02/15	0.1.3	Add cid_t element, 1. blockInfo
1.0.6	PTChen	2012/10/03	0.1.3	MTK VoIP version
1.0.7	Pork.chen	2013/04/07	0.1.3	Add adamInfcConfigHwProf
1.0.8	Lin.chen	2014/05/28	0.1.4	Add adamChanPlayTone argument
1.0.9	Weifeng.xu	2014/06/13	0.1.4	Add adamChanPlayDtmfCid
1.0.10	Peter	2014/11/17	0.1.7	Add adamStrmStopAll
1.0.11	Peter	2014/11/27	0.1.8	Add adamPcmLoopback Add adamRtpLoopback Add adamDspLoopback
1.0.12	Peter	2015/04/21	0.1.9	Add lec_nlp enable switch Add lec_aes enable switch

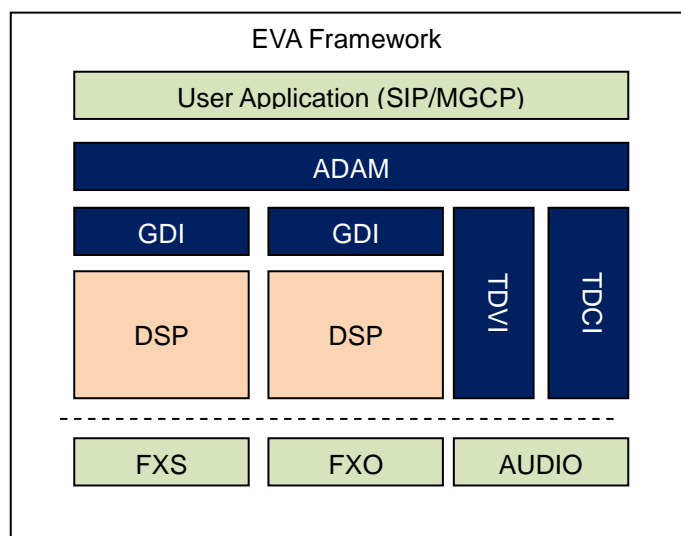
				Add lec bypass enable switch Add lec bypass_infax enable switch
1.0.13	linChen	2015/04/30	0.1.10	Add adamInfcRingParams
1.0.14	yafeiRen	2015/05/12	0.1.11	Add adamInfcLineTest
1.0.15	April	2015/06/23	0.1.12	Add adamInfcDcFeedParams
1.0.16	Yafei.Re n	2015/08/28	0.1.13	Add adamInfcCULineTest
1.0.17	Shelven .lu	2016/01/14	0.1.14	Add adamDspReplaceData
1.0.18	Eric.guo	2016/02/18	0.1.15	Modify adamChanPlayTone, add tone play duration.
1.0.19	Guanhu. Zou	2016/02/19	0.1.16	Add adamStrmPlayDtmf
1.0.20	April.wu	2016/05/10	0.1.17	Update ADAM_Reference_Manual.

2 Introduction

ADAM, Adaptive DSP Access Manager, is one of the modules in EVA framework who provides a consistent single access point to manipulate variant DSP(s). It does not just provide consistent API to application such that application requires no change when DSP changed, but also gives the developer the freedom to add his/her own logic to extend the VoIP related functions, such as, call logs, stream recording, etc. Also, ADAM is designed to be able to handle multiple DSP simultaneously, therefore, it is highly flexible to design scalable VoIP product from low channels CPE to high density gateway products.

2.1 EVA Framework

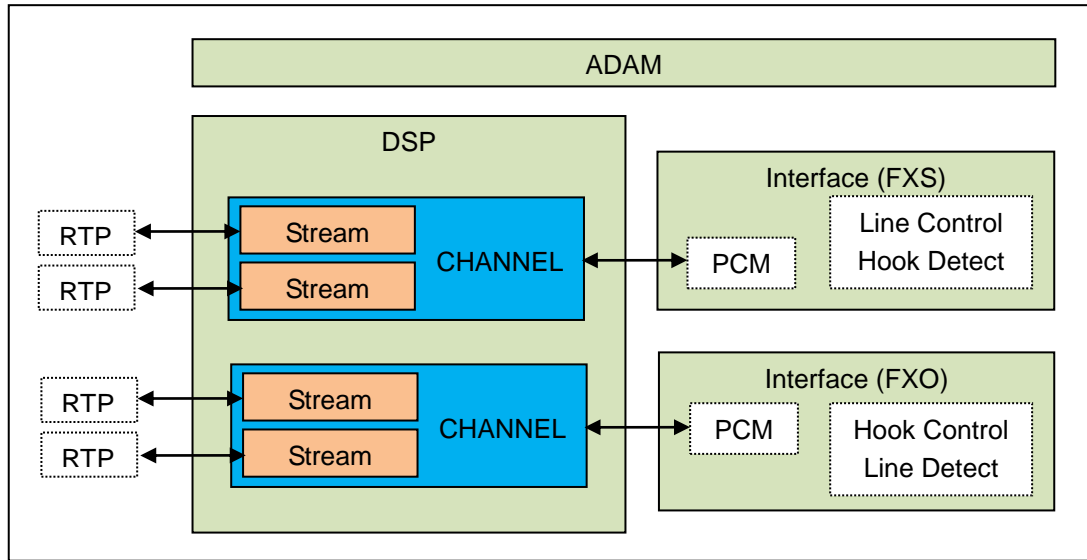
EVA, Enhanced VoIP Architecture, is a framework to redefine VoIP components in clear layers and object model. It provides the portability, scalability, flexibility, and transparency in developing VoIP product. The ADAM can provide a consistent DSP access interface to one or many DSPs and the application requires no change when the underlying DSP changes. ADAM is also open to developers to allow developer to add his/her own logic in ADAM to twist, convert, or replace the DSP logics. For example, some DSP handles telephony hardware interface control, but developer may obsolete the DSP control and implement his/her own interface control in ADAM internal function(s). However, to the application who calls ADAM APIs, it does not know the function has been replaced by different implementation.



The GDI, Generic DSP Interface, is the middle layer framework for consistent DSP functionalities abstract. ADAM control DSP(s) through GDI API. GDI framework is open to user who wants to add different DSP modules. However, GDI implementation might be closed due to license issue based on the DSP module provider. Fortunately, users do not need to see the GDI internal implementation which is mainly calling the DSP API, but just need to know the DSP functions are delivered as it should as defined in GDI API.

EVA is an evolving open framework to developer. New API will be added to provide better functionalities. We wish you enjoy using EVA framework and welcome your contribution to EVA. For more information about EVA, please refer to EVA_Framework_Introduction.

2.2 Object Model



EVA uses object model to simplify the DSP control process. ADAM controls and management DSP operation based on this object model concept.

Interface:

Interface is a telephony hardware user interface abstract which could be FXS, FXO, DECT module, or other telephony handset types. Interface is mainly responsible to reflect the user operation, such as off hook the phone or generate specific event to notify application for certain operation. On the other hand, ADAM controls interface to deliver specific notification to the user, such as ringing.

Though PCM is exchange through the interface hardware, the Interface abstract does not get involve with any PCM control or manipulation. (** See TDCI 'Telephony Device Control Interface' and TDVI 'Telephony Device Voice Interface' in EVA_Framework_Introduction for further detail.)

Channel:

Channel is one independent DSP process which handles a PCM raw input and output of an Interface. A DSP may consist of one or many channels and each channel's DSP process should be able to configure independently without interference.

The PCM raw input may go through various DSP process, such as echo cancellation, tone detection, voice activity detection, noise floor estimation, gain adjustment, etc. Relative event may be generated if the PCM raw input meets the condition. And the processed PCM raw data might be mixed with a decoded stream PCM raw data if conference is required. Then, Channel pass this PCM output raw data to a Stream process and encode it to coded format and transmit it to the destination peer.

On the other direction, Channel takes a decoded PCM raw data from a Stream and put these PCM raw data through certain DSP process, such as, comfort noise generation, packet loss compensation, tone detection, noise floor estimation, etc. Then, it might mix with another Stream decoded and DSP processed PCM raw data if conference is required. And it goes through some further DSP process, such as gain adjustment, echo reference. Then finally, pass the PCM raw data output to the interface.

User need to control Channel object to decide which DSP process needs to be enable and which to be turned off.

Stream:

Stream is to represent an encoding/decoding process and its pre-/post- process. Stream is usually attached to a Channel's DSP process. A Channel may consist of one or many Stream depending on the DSP, for CPE product, it is usually two for supporting 3-way conference call.

Each Stream configuration can be configured independently without interference. For example, in a 3-way conference call, user can configure the stream direction of a Stream to put one Stream on hold.

Stream configuration can be configured at anytime regardless the Stream active state and the configuration changes take effect immediately.

DSP:

DSP is the abstract of a DSP main body which could be a physical DSP hardware or DSP software. There are certain configurations are DSP-wide, such as tone generation template and tone detection template. Moreover, DSP is the main host of Channel and Stream. To use their functions, DSP must be initialized first and shutdown properly to release the resource when the DSP service is no longer required.

2.3 ADAM API Command and Event

User may control DSP(s) and Interface(s) through ADAM API Commands and get to know DSP and Interface status by reported events. By handling the event properly with application logic and control DSP with ADAM API when application required DSP control; user can develop VoIP application easily without worrying the internal process of DSP.

According to the Object Model, ADAM APIs are categorized into five categories, ADAM Generic, DSP, Channel, Stream, and Interface. Each API perform specific task to its subject object. Some of them may be used equivalently, such as user can call `adamStrmStopTone` to stop a tone generation, or call `adamStrmPlayTone` with silence tone to get the same result. Another example is `adamInfcRing`, alternatively user can call `adamInfcConfigLine` and set the line state to RING to get the same effect. In another word, users do not need to use all API but can select the API he/she preferred and use the same API in different occasions. Here is a list of ADAM APIs:

Command List:

Command	Description
ADAM	
<code>adamInit</code>	Initialize ADAM and get DSP(s) handles.
<code>adamExit</code>	Quit ADAM, release any allocated resources.
<code>adamQuery</code>	Get ADAM capability, such as number of DSP hooked and interface numbers, etc.
<code>adamPollEvent</code>	The single event access point to retrieve event from DSP(s) and Interface(s).
DSP	

Command	Description
adamDspInvoke	Initialize and start the DSP process.
adamDspRevoke	Shutdown and terminate the DSP process.
adamDspQuery	Get DSP capability information, such as CODEC and detectors support, etc.
adamDspConfigTone	Configure the tone template to a DSP.
adamDspConfigCPT	Configure the CPT template to a DSP
adamDspReplaceData	For dsp replace pcm data to a special file.
adamPcmLoopback	For pcm loopback test.
Channel	
adamChanQuery	Get a channel configuration.
adamChanConfig	Change a channel configuration, such as its detectors' active state and Tx/Rx Gain.
adamChanPlayTone	Generate tone(s) on a channel (to interface).
adamChanStopTone	Stop a tone generation.
adamChanPlayDtmf	Generate DTMF on a channel (to interface).
adamChanPlayDtmfCid	Play (generate) Caller ID to specific channel by DSP ID and Channel ID
adamChanPlayCid	Manually generate caller ID signal on a channel.
adamChanPlayType2Cid	Generate Type-II caller ID on a channel.
adamChanDumpPcm	Enable raw Tx/Rx PCM data dump to a network peer.
adamRtpLoopback	Enable or Disable rtp loopback on a specified channel, used to check rtp processing.
adamDspLoopback	Enable or Disable DSP loopback on a specified channel, used to check dsp processing.
Stream	
adamStrmQuery	Get the configuration of a stream and its active state.
adamStrmConfig	Change the configuration of a stream.
adamStrmStart	Start the streaming process of a channel.
adamStrmStop	Stop the streaming process of a channel.
adamStrmStopAll	Stop all streaming process of all channel.
adamStrmSendDtmfr	Manually send DTMF relay packet to the network peer.
adamStrmPlayTone	Generate tone(s) to the network peer.
adamStrmStopTone	Stop the tone generation to the network peer.
adamStrmQueryMediaInfo	Get accumulated Media(RTP) information.
adamStrmResetMediaInfo	Clean accumulated Media(RTP) information.
adamStrmPlayDtmf	Generation DTMF on far-end direction.
Interface	
adamInfcQuery	Get interface configuration and line/hook state.
adamInfcConfigLine	Change the line state of an FXS interface, limited to certain states only.
adamInfcConfigHook	Change the hook state of an FXO interface.

Command	Description
adamInfcConfigRing	Change the ring configuration of an FXS interface.
adamInfcRing	Set FXS interface to RING state.
adamInfcStopRing	Stop FXS ringing.
adamInfcConfigHookTs	Configure hook state threshold
adamInfcLineTest	Get Line Test information.
adamInfcSlicTypeQuery	Get slic information: slic fxnum fxonum and slic type.
adamInfcRingParams	Set ringParams.
adamInfcDcFeedParams	Set dcFeedParams.
adamInfcCULineTest	China Unicom inside and outside line test.
adamInfcConfigHwProf	Change country code.

Event List:

EVENT	Description
EVENT_CODE_INVALID	Return value in NON-BLOCKING mode when no valid event available.
EVENT_CODE_TONE	Notify application of a tone detection event. Including DTMF, Modem (FAX), Call Progress Tone (CPT).
EVENT_CODE_CID	Notify application of receiving CID signal and CID context.
EVENT_CODE_LINE	Notify application of line state change.
EVENT_CODE_HOOK	Notify application of hook state change.
EVENT_CODE_T38	Notify application of a T38 event.
EVENT_CODE_JB_UPDATE	Update jitter buffer statistic information.
EVENT_CODE_NON_RTP_RECVD	Notify application of receiving unidentified packet in RTP port and packet context, could be used for sending/receiving STUN packet for RTP port open; Not used now.
EVENT_CODE_RTCP_SEND	Notify application of RTCP sending event and RTCP context; Not used now.
EVENT_CODE_RTCP_RECVD	Notify application of RTCP receiving event and RTCP context; Not used now.
EVENT_CODE_STREAM_UPDATE	Update stream statistic information; Not used now.
EVENT_CODE_TIMER	Generate notification to application based on DSP ticks and user configured interval; Not used now.
EVENT_CODE_PERFORMANCE	To update DSP benchmark information, such as DSP uptime and average MHz consumption; Not used now.
EVENT_CODE_ERROR	To report DSP error; Not used now.

EVENT	Description
EVENT_CODE_DL_DTMF	DL DTMF TONE
EVENT_CODE_JB_CHANGE	Jitter change.
EVENT_CODE_TONE_END	Tone end.
EVENT_CODE_CALLERID_SEND	CallerID send.
EVENT_CODE_FIRST_RTP_DELIVERD	First rtp packet income.
EVENT_CODE_PCM	For fxs replace data.

2.4 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 on-hook 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 (Echo Canceller)	A process to remove echo.
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 and may be followed with type-2 caller ID.
Silence Compression	A method to save bandwidth consumption by transmit silence indication packet (SID) instead of full RTP payload when user is not talking.
SLIC	Subscriber Line Interface Circuit. A hardware component emulates CO service to provide FXS function.
Stream	Stream is a path or process to disassembly sequential coded data (i.e. audio), 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 ADAM Management

3.1 adamInit

Prototype:

exCode_e adamInit(void);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
None	

Description:

Initialize ADAM to get all DSP handles.

Example Code:

```
printf("Initializing ADAM ... \n");

if (EXEC_SUCCESS != adamInit()) {
    printf("ADAM initialization failed!!\n");
}
```

3.2 adamExit

Prototype:

exCode_e adamExit(void);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
None	

Description:

Terminate ADAM and release any allocated resources.

Example Code:

```
printf("Shutdown ADAM ... \n");

if (EXEC_SUCCESS != adamExit()) {
    printf("ADAM shutdown failed!!\n");
}
```

3.3 adamQuery

Prototype:

exCode_e adamQuery(adamConfig_t *pAdamConf);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
pAdamConf	Pointer to an adamConfig_t instance to receive the ADAM configuration.

Description:

Query ADAM information, such as version, number of DSP support, etc.

Example Code:

```
if (EXEC_SUCCESS == adamQuery(&adConf)) {
    printf("ADAM Version: %s\n", adConf.version);
    printf("Number of DSP: %d\n", adConf.dspNum);
    printf("Number of Interface: %d\n", adConf.infcNum);
}
```

3.4 adamPollEvent

Prototype:

```
exCode_e adamPollEvent(blockMode_e mode, event_t *pEvent);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Mode	Event polling mode: NON-BLOCKING – the function will return immediately regardless if a valid event presented. Return EXEC_SUCCESS when there is a valid event and EXEC_FAIL when no valid event received. BLOCKING – the function will not return until a valid event is received. In BLOCKING mode, the return valid is always EXEC_SUCCESS.
pEvent	Pointer to an event_t instance to receive the event data.

Description:

Polling event from DSP.

Example Code:

(See *Chapter 8 - ADAM EVENT Processing*)

4 DSP Control

4.1 adamDspInvoke

Prototype:

```
exCode_e adamDspInvoke(dspld_e dsp);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.

Description:

Invoke DSP will initialize the DSP and start DSP process.

Example Code:

```
printf("Invoking DSP ...\n\n");

if (EXEC_SUCCESS != adamDspInvoke(DSP_MTK)) {
    printf("Error: DSP initialization failed!\n");
}
```

4.2 adamDspRevoke

Prototype:

```
exCode_e adamDspRevoke(dspld_e dsp);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.

Description:

Revoke DSP will shutdown the DSP process.

****NOTE:** Depending on the DSP capability, once DSP is revoked, it might not be able to re-invoke again unless system reboot or other external process executed.

Example Code:

```
printf("Revoking DSP ...\n\n");
if (EXEC_SUCCESS != adamDspRevoke(DSP_MTK)) {
    printf("Error: DSP shutdown failed!\n");
}
```

4.3 adamDspQuery

Prototype:

exCode_e adamDspQuery(dspId_e dsp, activeState_e *dspActive, dspFeature_t *pFeature);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
dspActive	DSP active state
pFeature	Pointer to a dspFeature_t instance to receive the DSP supported feature information.

Description:

Query the DSP capability information.

Example Code:

```

activeState_e active;
dspFeature_t feature;

if (EXEC_SUCCESS != adamDspQuery(DSP_MTK, &active, &feature)) {
    printf("Error: Cannot get DSP status, application will quit now!\n\n");
    exit(-1);
}

printf("DSP features: \n");
printf("Active status: %s\n", (active));
printf("DSP ID: (%d)\n", feature.dspId);
printf("Number of Channel: %d\n", feature.numOfChan);
printf("Stream per Channel: %d\n", feature.strmsPerChan);

```

4.4 adamDspConfigTone

Prototype:

exCode_e adamDspConfigTone(dspId_e dsp, uint16 toneId, tone_t *pTone);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Dsp	DSP ID.
toneId	Tone ID, ** 0~29, but tone ID [0] is reserved for silence which cannot be configured.
pTone	Pointer to a tone_t instance which contains the tone configuration information.

Description:

Configure a tone (generation) template in the DSP.

Example Code:

```
tone_t mTone;
int toneld;

toneld = 1;
mTone.toneType = TONE_REGULAR;

mTone.regular.toneFreq[0] = 350; /* Hz */
mTone.regular.tonePwr[0] = -18 * 2; /* -18db */
mTone.regular.toneFreq[1] = 440; /* Hz */
mTone.regular.tonePwr[1] = -18 * 2; /* -18db */

mTone.makeTime[0] = 500;
mTone.breakTime[0] = 1000;
mTone.repeat[0] = 5;

if (EXEC_SUCCESS != adamDspConfigTone(DSP_MTK, toneld, &mTone)) {
    printf("Error: adamDspConfigTone failed!\n");
    return;
}
```

4.5 adamDspConfigCPT

Prototype:

exCode_e adamDspConfigCPT(dspld_e dsp, cpt_t *pCpt);

Return Values:

EXEC_SUCCESS

EXEC_FAIL

INVALID_PARAM

Arguments:

Name	Description
Dsp	DSP ID.
pCpt	Pointer to a cpt_t instance which contains the CPT configuration information.

Description:

Configure a call progress tone (detection) template in the DSP.

Example Code:

Ringback tone detection configuration

```

cpt_t mCpt;

memset(&mCpt, 0, sizeof(cpt_t));

mCpt.type = TONE_RINGBACK;

mCpt.dual.toneFreq[0] = 440; /* Hz */
mCpt.dual.toneDev[0] = 22; /* 5% frequency deviation */
mCpt.dual.toneFreq[1] = 480; /* Hz */
mCpt.dual.toneDev[1] = 24; /* 5% frequency deviation */

mCpt.dual.minMake[0] = 950;
mCpt.dual.maxMake[0] = 1050;
mCpt.dual.minBreak[0] = 2550;
mCpt.dual.minBreak[0] = 3050;
mCpt.dual.power = -39;

if (EXEC_SUCCESS != adamDspConfigCPT(DSP_MTK, &mCpt)) {
    printf("Error: adamDspConfigCPT failed!\n");
    return;
}

```

SIT tone detection configuration

```

cpt_t mCpt;

memset(&mCpt, 0, sizeof(cpt_t));

mCpt.type = TONE_SIT;

mCpt.sit.toneFreq[0] = 914; /* Hz */
mCpt.sit.toneDev[0] = mCpt.sit.toneFreq[0] / 20; /* 5% frequency deviation*/
mCpt.sit.toneFreq[1] = 985; /* Hz */
mCpt.sit.toneDev[1] = mCpt.sit.toneFreq[1] / 20; /* 5% frequency deviation*/
mCpt.sit.toneFreq[2] = 1370; /* Hz */
mCpt.sit.toneDev[2] = mCpt.sit.toneFreq[2] / 20; /* 5% frequency deviation*/
mCpt.sit.toneFreq[3] = 1428; /* Hz */
mCpt.sit.toneDev[3] = mCpt.sit.toneFreq[3] / 20; /* 5% frequency deviation*/
mCpt.sit.toneFreq[4] = 1776; /* Hz */
mCpt.sit.toneDev[4] = mCpt.sit.toneFreq[4] / 20; /* 5% frequency deviation*/

mCpt.sit.minShortDur = 276 - 25;
mCpt.sit.maxShortDur = 276 + 25;
mCpt.sit.minLongDur = 380 - 25;
mCpt.sit.maxLongDur = 380 + 25;
mCpt.sit.power = -39;

if (EXEC_SUCCESS != adamDspConfigCPT(DSP_MTK, &mCpt)) {
    printf("Error: adamDspConfigCPT failed!\n");
    return;
}
    
```

4.6 adamDspReplaceData

Prototype:

exCode_e adamDspReplaceData(dspId_e dsp, chanId_e ch, replaceData_t *pReplaceConf);

Return Values:

INVALID_PARAM
FUNC_UNSUPPORTED

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pReplaceConf	Pointer to a replaceData_t instance which contains the filePath, stage and direction information.

Description:

Play a specific file (raw audio data) to local channel or remote peer.

Example Code:

```
chanId_e      ch;
replaceData_t pReplaceConf;
memset(&pReplaceConf, 0, sizeof(replaceData_t));

ch = 0;
pReplaceConf.enable= 1; /*enable = on*/
pReplaceConf.dir= 0;    /*direction = rx*/
pReplaceConf.stage= 0;
strcpy(pReplaceConf.filePath, "/tmp/a.pcm");

adamDSPReplaceData(DSP_ID, ch, &pReplaceConf);
```

4.7 adamPcmLoopback

Prototype:

exCode_e adamPcmLoopback(dspId_e dsp, enableControl_e en);

Return Values:

INVALID_PARAM
FUNC_UN SUPPORT

Arguments:

Name	Description
dsp	DSP ID.
en	Function control switch.

Description:

Enable downlink voice loopback in SLIC back (go through PCM processing) to uplink, for PCM debugging.

Example Code:

```
enableControl_e en = 0; /*0:disable; 1:enable*/

adamPcmLoopback(DSP_MTK, en);
```

5 Channel Control

5.1 adamChanQuery

Prototype:

```
exCode_e adamChanQuery(dspId_e dsp, channelId_e ch, chanConfig_t *pChanConf);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pChanConf	Pointer to a chanConfig_t instance to receive the channel configuration information.

Description:

Query the configuration of a channel

Example Code:

```
chanConfig_t config;
channelId_e      ch = 0;

if (EXEC_SUCCESS != adamChanQuery(DSP_MTK, ch, &config)) {
    printf("Execution failed! Cannot retrieve channel configuration.\n\n");
    return;
}

printf("Channel (%d) configuration:\n", ch);
printf("Enabled Detectors:\n");
printf("Detect (DTMF_TONE)=%d\n", (config.detectMask & DETECT_TONE_DTMF));
printf("Detect (FAX/MODEM_TONE)=%d\n", (config.detectMask & DETECT_TONE_MODEM));
printf("Detect (CALL_PROGRESS_TONE)=%d\n", (config.detectMask & DETECT_TONE_CPT));
printf("Detect (Caller_ID)=%d\n", (config.detectMask & DETECT_CID));
printf("EC Control =%d\n", (config.ecEnable));
printf("Tx Gain: %ddb\n", config.ampTx/2);
printf("Rx Gain: %ddb\n", config.ampRx/2);
```


5.2 adamChanConfig

Prototype:

```
exCode_e adamChanConfig(dspId_e dsp, channelId_e ch, chanConfig_t *pChanConf);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pChanConf	Pointer to a chanConfig_t instance which contains the configuration information.

Description:

Change the configuration of a channel.

Example Code:

```

chanConfig_t config;
chanId_e ch = 0;

config.ampTx = (-3) * 2; /* -3db */
config.ampRx = 3 * 2; /* +3db */
if (EC_ON) {
    config.ecEnable = CONTROL_ENABLE; /* Enable Echo Cancellation */
}
else {
    config.ecEnable = CONTROL_DISABLE; /* Disable Echo Cancellation */
}

if (DTMF_DETECT_ON) {
    config.detectMask |= DETECT_TONE_DTMF; /* Enable DTMF Detection */
}
else {
    config.detectMask &= ~(DETECT_TONE_DTMF); /* Disable DTMF Detection */
}

if (MODEM_DETECT_ON) {
    config.detectMask |= DETECT_TONE_MODEM; /* Enable FAX/Modem Detection */
}
else {
    config.detectMask &= ~(DETECT_TONE_MODEM); /* Disable FAX/Modem Detection */
}

if (CPT_DETECT_ON) {
    config.detectMask |= DETECT_TONE_CPT; /* Enable Call Progress Tone Detection */
}
else {
    config.detectMask &= ~(DETECT_TONE_CPT); /* Disable Call Progress Tone Detection */
}

if (CID_DETECT_ON) {
    config.detectMask |= DETECT_CID; /* Enable Caller ID Detection, only for FXO interface */
}
else {
    config.detectMask &= ~(DETECT_CID); /* Disable Caller ID Detection, only for FXO interface */
}

if (EXEC_SUCCESS != adamChanConfig(DSP_MTK, ch, &config)) {
    printf("Error: adamChanConfig failed!\n");
    return;
}

```

5.3 adamChanPlayTone

Prototype:

```
exCode_e adamChanPlayTone(dspId_e dsp, chanId_e ch, toneSeq_t *pToneSeq, uint32 repeat,
uint32 dur);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pToneSeq	Pointer to a toneSeq_t instance which contains a sequence of tones and number of tones.
Repeat	Times to repeat the tone sequence generation.
dur	Tone play duration.

Description:

Generate tone(s) on a channel.

Example Code:

```
chanId_e ch=0;
toneSeq_t toneList;
const int sz = 3;
int mToneId[sz] = {3, 2, 1};
int rpt = 2;
uint32 dur = 2000;

toneList.numOfTone = sz;
toneList.toneIdSeq = (uint8*)mTone;

if (EXEC_SUCCESS != adamChanPlayTone(DSP_MTK, ch, &toneList,rpt,dur)) {
    printf("Error: adamChanPlayTone failed! \n");
    return;
}
```

5.4 adamChanStopTone

Prototype:

```
exCode_e adamChanStopTone(dspld_e dsp, chanId_e ch);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID

Description:

Stop tone generation on a channel.

Example Code:

```
if (EXEC_SUCCESS != adamChanStopTone(DSP_MTK, 0)) {
    printf("Error: adamChanStopTone failed! \n");
    return;
}

/* Another alternative is to play tone[0] (Silence) to stop a tone play */
toneSeq_t toneList = {
    . numOfTone = 1,
    . toneIdSeq = {0}
};

if (EXEC_SUCCESS != adamChanPlayTone(DSP_MTK, CH0, &toneList, 1)) {
    printf("Error: adamChanPlayTone failed! \n");
    return;
}
```

5.5 adamChanPlayDtmf

Prototype:

```
exCode_e adamChanPlayDtmf(dspld_e dsp, chanId_e ch, char digit, uint32 dur);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
digit	DTMF number
dur	DTMF tone duration (ms)

Description:

Generation DTMF on a channel.

Example Code:

```
chanId_e ch=0;
char digit=7;
uint32 dur=500;

if (EXEC_SUCCESS != adamChanPlayDtmf(DSP_MTK, CH0, digit, dur)) {
    printf("Error: adamChanPlayDtmf failed! \n");
    return;
}
```

5.6 adamChanPlayDtmfCid

Prototype:

exCode_e adamChanPlayDtmfCid (dspId_e dsp, chanId_e ch, cid_t *pCid);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID

pCid	Pointer of a cid_t instance which contains caller ID information.
------	---

Description:

Play (generate) Caller ID to specific channel by DSP ID and Channel ID

Example Code:

```
chanId_e ch=0;
cid_t cid ;
cid. Number = "1234";
cid.makeTime = 40;
cid.breakTime = 40;
cid.dtmfPower = 0;
cid.dtmfT3= 350;

if (EXEC_SUCCESS != adamChanPlayDtmfCid(DSP_ID, ch, &cid)){
    printf("Error: adamChanPlayDtmfCid failed! \n");
    return;
}
```

5.7 adamChanPlayCid

exCode_e adamChanPlayCid(dspId_e dsp, chanId_e ch, cid_t *pCid);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pCid	Pointer of a cid_t instance which contains caller ID information.

Description:

Manually generate a caller ID on the channel (to interface) without any leading signal or interface control (ring).

**** NOTE:** This API is provided for advance user who would like to do manually CID transmission and control the SLIC operation on his/her own. To correctly transmit CID manually, the SLIC must also be configure properly prior to call this API.

Example Code:

```
chanId_e ch = 0;
cid_t cid = {
    .format = CID_FORMAT_BELLCORE_FSK,
    .number = "88888888",
    .name = "",
    .dateTime = ""
};

if (EXEC_SUCCESS != adamChanPlayCid(DSP_MTK, ch, &cid)) {
    printf("Error: adamChanPlayCid failed! \n");
    return;
}
```

5.8 adamChanPlayType2Cid

exCode_e adamChanPlayType2Cid(dspId_e dsp, chanId_e ch, cid_t *pCid);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pCid	Pointer of a cid_t instance which contains caller ID information.

Description:

Generate a Type-II (off-hook) caller ID on the channel (to interface).

Example Code:

```
chanId_e ch = 0;
cid_t cid = {
    .format = CID_FORMAT_BELLCORE_FSK,
    .number = "88888888",
    .name = "",
    .dateTime = ""
};

if (EXEC_SUCCESS != adamChanPlayType2Cid(DSP_MTK, ch, &cid)) {
    printf("Error: adamChanPlayType2Cid failed! \n");
    return;
}
```

5.9 adamChanDumpPcm

Prototype:

exCode_e adamChanDumpPcm(dspld_e dsp, chanId_e ch, netAddr_t *pDstAddr);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
pDstAddr	Pointer to a netAddr_t instance which contains the information of destination endpoint to be received the dump data. ** Configure IP address as: 0.0.0.0 to disable the dump.

Description:

Enable channel PCM dump process for debugging. When PCM dump enabled, the PCM Tx/Rx raw data of a channel will be sent to the designated network address and port in RTP format. User may capture these packets with sniffer tool, i.e. Wireshark, extract the payload and restore the audio.

** NOTE: To disable a dump process, configure the destination IP address to 0x0 (0.0.0.0).

Example Code:


```
chanId_e ch;
netAddr_t mAddr = {
    .ver = IPV4,
    .addrV4 = inet_addr("192.168.1.200");
};

if (EXEC_SUCCESS != adamChanDumpPcm(DSP_MTK, ch, &mAddr)) {
    printf("Error: adamChanDumpPcm failed! \n");
}
```

5.10 adamRtpLoopback

Prototype:

exCode_e adamRtpLoopback(dspld_e dsp, chanId_e ch, enableControl_e en);

Return Values:

EXEC_SUCCESS

EXEC_FAIL

INVALID_PARAM

Arguments:

Name	Description
dsp	DSP ID
ch	Channel
en	Enable or disable loopback

Description:

Enable downlink rtp data loopback in RTP module (no DSP processing) back to uplink, for RTP debugging.

Example Code:

```
enableControl_e en = 1; /*0:disable; 1:enable*/
chanId_e ch = 0;

adamRtpLoopback (DSP_MTK, ch,en);
```

5.11 adamDspLoopback

Prototype:

exCode_e adamDspLoopback(dspld_e dsp, chanId_e ch, enableControl_e en);

Return Values:

EXEC_SUCCESS

EXEC_FAIL

INVALID_PARAM

Arguments:

Name	Description
dsp	DSP ID
ch	Channel
en	Enable or disable loopback

Description:

Enable downlink raw data loopback before PCM(no PCM processing) back to uplink, for DSP debugging.

Example Code:

```
enableControl_e en = 1; /*0:disable; 1:enable*/
chanId_e      ch = 0;

adamDspLoopback (DSP_MTK, ch,en);
```

6 Stream Control

6.1 adamStrmQuery

Prototype:

```
exCode_e adamStrmQuery(dspId_e dsp, channelId_e ch, strmId_e strm, activeState_e *pStrmActive,
strmConfig_t *pStrmConf);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
strm	Stream ID
pStrmActive	Pointer to an activeState_e instance to receive stream active state.
pStrmConf	Pointer to a strmConfig_t instance to receive the stream configuration.

Description:

Query the configuration of a stream.

Example Code:

```
chanId_e ch = 0;
strmId_e st = 0;
strmConfig_t config;
activeState_e active;

if (EXEC_SUCCESS != adamStrmQuery(DSP_MTK, ch, st, &active, &config)) {
    printf("Execution failed! Cannot retrieve stream configuration.\n\n");
}

printf("Channel %d -> Stream %d Configuration:\n", ch, st);
printf("Stream state: %d\n", active);
printf("Source address: 0x%x port: %d RTPport: %d\n", config.session.srcAddr.addrV4,
config.session.srcAddr.port, config.session.srcAddr.rtcpPort);
printf("Destination address: 0x%x port: %d RTPport: %d\n", config.session.dstAddr.addrV4,
config.session.dstAddr.port, config.session.srcAddr.rtcpPort);
printf("RTP tos: %x RTCP tos: %x\n", config.session.rtpTosVal, config.session.rtcpTosVal);
printf("Codec: %d\n", config.strmAttr.payloadSelect);
printf("ulPtime: %d dlPtime: %d\n", config.strmAttr.ulPtime, config.strmAttr.dlPtime);
printf("Silence compression: %d\n", config.strmAttr.silenceComp);
printf("DTMF Relay: %d DTMF Remove: %d\n", config.strmAttr.dtmfRelay,
config.strmAttr.dtmfRelay, config.strmAttr.dtmfRemove);
printf("Stream direction: %d\n", config.strmAttr.direction);
printf("Jitter Buffer Configuration: %d, Initial Size: %ld Max Size: %ld\n", config.jbConf.mode,
config.jbConf.szJbInit, config.jbConf.szJbMax );
```

6.2 adamStrmConfig

Prototype:

exCode_e adamStrmConfig(dspId_e dsp, chanId_e ch, strmId_e strm, strmConfig_t *pStrmConf);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
strm	Stream ID
pStrmConf	Pointer to a strmConfig_t instance which

	contains the stream configuration information.
--	--

Description:

Change the configuration of a stream.

Example Code:

```
chanId_e ch = 0;
strmId_e st = 0;
strmConfig_t config;

config.session.srcAddr.addrV4 = inet_addr("192.168.1.1");
config.session.srcAddr.port = 5000;
config.session.srcAddr.rtcpPort = 5001;
config.session.dstAddr.addrV4 = inet_addr("192.168.1.100");
config.session.dstAddr.port = 5000;
config.session.dstAddr.rtcpPort = 5001;
config.session.rtpTosVal = 0x4;
config.session.rtcpTosVal = 0x4;
config.strmAttr.payloadSelect = CODEC_G711A;
config.strmAttr.ulPtime = PTIME_20MS;
config.strmAttr.dlPtime = PTIME_20MS;
config.strmAttr.silenceComp = CONTROL_ENABLE;
config.strmAttr.dtmfRelay = CONTROL_ENABLE;
config.strmAttr.dtmfRemove = CONTROL_ENABLE;
config.strmAttr.direction = STRM_SENDRECV;
config.jbConf.mode = JB_FIXED;
config.jbConf.szJbInit = 50;
config.jbConf.szJbMax = 100;

if (EXEC_SUCCESS != adamStrmQuery(DSP_MTK, ch, st, &config)) {
    printf("Execution failed! Cannot set stream configuration.\n\n");
}
```

6.3 adamStrmStart

Prototype:

exCode_e adamStrmStart(dspId_e dsp, chanId_e ch, strmId_e strm);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY

UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
strm	Stream ID

Description:

Start streaming process of a channel.

Example Code:

```
if (EXEC_SUCCESS != adamStrmStart(DSP_MTK, CH0, STRM0)) {
    printf("Execution failed! Cannot start stream .\n\n");
}
```

6.4 adamStrmStop

Prototype:

exCode_e adamStrmStop(dspId_e dsp, chanId_e ch, strmId_e strm);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
strm	Stream ID

Description:

Stop streaming process of a channel.

Example Code:

```
if (EXEC_SUCCESS != adamStrmStop(DSP_MTK, CH0, STRM0)) {
    printf("Execution failed! Cannot stop stream .\n\n");
}
```

6.5 adamStrmStopAll

Prototype:

```
exCode_e adamStrmStopAll(dspId_e dsp);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.

Description:

Stop all streaming process of all channel, this api is used in signal_handler, if a signal that may cause userspace application exit, all streams should be stopped before process exiting.

Example Code:

```
void signal_handler(int signum)
{
    psignal (signum, "Get signal");
    if (EXEC_SUCCESS != adamStrmStopAll (DSP_ID)) {
        printf("Error: stream stop failed!\n");
    }

    sleep(1);
    exit(1);
}

int main ()
{
    ... ..

    signal(SIGHUP, signal_handler);
    signal(SIGABRT, signal_handler);
    signal(SIGINT, signal_handler);
    signal(SIGTERM, signal_handler);
    signal(SIGQUIT, signal_handler);
    signal(SIGSEGV, signal_handler);
    signal(SIGFPE, signal_handler);

    ... ..
}
```

6.6 adamStrmSendDtmfr

Prototype:

```
exCode_e adamStrmSendDtmfr(dspId_e dsp, chanId_e ch, strmId_e strm, dtmf_e dtmf, uint32 dur);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
strm	Stream ID
dtmf	DTMF digit: 0~9, *, #, A, B, C, D, Dial tone~CAS tone
dur	Duration (ms).

Description:

Manually generate DTMF or Tone relay packet (RFC2833/4733) to remote peer.

Example Code:

```
if (EXEC_SUCCESS != adamStrmSendDtmfr(DSP_MTK, CH0, STRM0, DTMF_1, 1000)) {
    printf("Execution failed! Cannot send stream dtmfr .\n\n");
}
```

6.7 adamStrmPlayTone

Prototype:

```
exCode_e adamStrmPlayTone(dspId_e dsp, chanId_e ch, strmId_e strm, toneSeq_t *pToneSeq,
uint32 repeat);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
strm	Stream ID
pToneSeq	Pointer to a toneSeq_t instance which contains a sequence of tones and number of tones.
Repeat	Times to repeat the tone sequence generation.

Description:

Generate tone(s) to a stream.

Example Code:

```
chanId_e ch=0;
toneSeq_t toneList;
const int sz = 3;
int mToneId[sz] = {3, 2, 1};
int rpt = 2;

toneList.numOfTone = sz;
toneList.toneIdSeq = (uint8*)mTone;

if (EXEC_SUCCESS != adamStrmPlayTone(DSP_MTK, ch, st, &toneList, rpt)) {
    printf("Execution failed! Cannot send stream dtmfr .\n\n");
}
```

6.8 adamStrmStopTone

Prototype:

exCode_e adamStrmStopTone(dspId_e dsp, chanId_e ch, strmId_e strm);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID.

strm	Stream ID.
------	------------

Description:

Stop tone generation to a stream.

Example Code:

```
if (EXEC_SUCCESS != adamStrmStopTone(DSP_MTK, CH0, STRM0)) {
    printf("Error: adamStrmStopTone failed! \n");
    return;
}

/* Another alternative is to play tone[0] (Silence) to stop a tone play */
toneSeq_t toneList = {
    . numOfTone = 1,
    . toneIdSeq = {0}
};

if (EXEC_SUCCESS != adamStrmPlayTone(DSP_MTK, CH0, STRM0, &toneList, 1)) {
    printf("Error: adamStrmPlayTone failed! \n");
    return;
}
```

6.9 adamStrmQueryMediaInfo

Prototype:

```
exCode_e adamStrmQueryMediaInfo(dspId_e dsp, chanId_e ch, strmId_e strm, mediaInfo_t
*pMediaInfo);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID.
strm	Stream ID.
pMediaInfo	Pointer to a mediaInfo_t instance to receive the media information.

Description:

Get current media information.

Example Code:

```
mediaInfo_t mediaInfo;
if (EXEC_SUCCESS != adamStrmQueryMediaInfo(DSP_MTK, CH0, STRM0, & mediaInfo)) {
    printf("Error: adamStrmQueryMediaInfo failed! \n");
    return;
}

Printf("Query Media Info on channel (%d) -> stream (%d).\n", ch, st);
Printf("RtpError (%lld)\n", mediaInfo.rtpError);
Printf("Packet Loss (%lld)\n", mediaInfo.packetLoss);
Printf("Packet Recv (%lld)\n", mediaInfo.packetRecv);
Printf("Packet Loss Rate(%lld)\n", mediaInfo.packetLossRate);
Printf("maxJitter (%lld)\n", mediaInfo.maxJitter);
Printf("maxRTCPInterval (%lld)\n", mediaInfo.maxRTCPInterval);
Printf("bufUnderflow (%lld)\n", mediaInfo.bufUnderflow);
Printf("bufOverflow (%lld)\n", mediaInfo.bufOverflow);
```

6.10 adamStrmResetMediaInfo

Prototype:

exCode_e adamStrmResetMediaInfo(dspId_e dsp, chanId_e ch, strmId_e strm);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID.
strm	Stream ID.

Description:

Reset voice quality statistics (MOS value) in DSP.

Example Code:

```
if (EXEC_SUCCESS != adamStrmResetMediaInfo(DSP_MTK, CH0, STRM0, & mediaInfo)) {
    printf("Error: adamStrmQueryMediaInfo failed! \n");
    return;
}
```

6.11 adamStrmPlayDtmf

Prototype:

exCode_e adamStrmPlayDtmf(dspld_e dsp, chanId_e ch, char digit, uint32 duration);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
dsp	DSP ID.
ch	Channel ID
digit	DTMF number
duration	DTMF tone duration (ms)

Description:

Generation DTMF on far-end direction.

Example Code:

```
chanId_e ch=0;
char digit=7;
uint32 dur=500;

if (EXEC_SUCCESS != adamStrmPlayDtmf(DSP_MTK, CH0, digit, dur)) {
    printf("Error: adamStrmPlayDtmf failed! \n");
    return;
}
```

7 Interface Control

7.1 adamInfcQuery

Prototype:

```
exCode_e adamInfcQuery(infcId_e infc, infcConfig_t *pInfcConf);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Infc	Interface ID.
pInfcConf	Pointer of an infcConfig_t instance to receive the interface configuration.

Description:

Query the interface configuration.

Example Code:

```

infCld_e infc = 0;
infConfig_t infcConf;

if (EXEC_SUCCESS != adamInfcQuery(infc, &infcConf)) {
    printf("Error: adamChanPlayCid failed! \n");
    return;
}

printf("Interface type: %d\n", infcConf.type);
printf("Line State: %d\n", infcConf.lineState);
printf("Hook State: %d\n", infcConf.hookState);
printf("Hook Threshold: Min_flashTs:%ld(ms) Max_flashTs:%ld(ms) Min_releaseTs:%ld(ms)\n",
infCConf.hookTs.flashMin, infCConf.hookTs.flashMax, infCConf.hookTs.releaseMin);
if (INFC_FXS == infcConf.type) {
    printf("Ring configuration: \n");
    for (i = 0 ; i < MAX_CADENCE ; i++) {
        printf("Cadence%d: %d(ms)\n", i, \
            infCConf.ring.cad[i].onTime, infCConf.ring.cad[i].offTime);
    }
    printf("Caller ID: number=[%s]\n", infCConf.ring.cid.number);
    printf("Caller ID generate at (%d)th break\n", infCConf.ring.cidAt);
}

```

7.2 adamInfcConfigLine

exCode_e adamInfcConfigLine(infCld_e infc, lineState_e state);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Infc	Interface ID.
State	Line state such as power down (LINE_DOWN), polarity reverse (LINE_ACTIVE_REV), ring (LINE_RING), etc. ** LINE_BUSY is a passive state that should only be triggered by the phone.

Description:

Change interface line state.

Example Code:

```
if (EXEC_SUCCESS != adamInfcConfigLine(INFC0, LINE_ACTIVE_FWD)) {
    printf("Error: adamInfcConfigLine failed! \n");
}
```

7.3 adamInfcConfigHook

Prototype:

exCode_e adamInfcConfigHook(infcId_e infc, hookState_e state);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Infc	Interface ID.
state	Hook state such as on-hook (HOOK_RELEASE), off-hook (HOOK_SEIZE), flash (HOOK_FLASH), etc.

Description:

Change interface hook state. Only works for FXO interface.

Example Code:

```
if (EXEC_SUCCESS != adamInfcConfigHook(INFC2, HOOK_FLASH)) {
    printf("Error: adamInfcConfigHook failed! \n");
}
```

7.4 adamInfcConfigRing

Prototype:

exCode_e adamInfcConfigRing(infcId_e infc, ringProfile_t *pRingProf);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport

DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
InfC	Interface ID.
pRingProf	Pointer of a ringProfile_t instance which contains ring configuration including cadence, duration, caller ID, and caller ID generation timing.

Description:

Change interface ring configuration.

Example Code:

```
infCConfig_t infCConf;

infCConf.ring.dur = 5000; /* ms */
infCConf.ring.cid = 5566; /*Call ID number */
infCConf.ring.cidAt = 1; /*Ring break after first ring */
infCConf.ring.cidWaitTime = 100; /*Wait 100ms to send CID at ring break */
infCConf.ring.cad[0].onTime = 500; /* ms */
infCConf.ring.cad[0].offTime = 1500; /* ms */
infCConf.ring.cad[1].onTime = 0; /* ms */
infCConf.ring.cad[1].offTime = 0; /* ms */
infCConf.ring.cad[2].onTime = 0; /* ms */
infCConf.ring.cad[2].offTime = 0; /* ms */

if (EXEC_SUCCESS != adamInfCConfigRing(INFC0, &(infCConf.ring))) {
    printf("Error: adamInfCConfigRing failed! \n");
    return;
}
```

7.5 adamInfCRing

Prototype:

```
exCode_e    adamInfCRing(infCId_e infC, uint32 dur, cid_t *pCid);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNsupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Infc	Interface ID.
dur	Ring duration (ms)
pCid	Pointer of a cid_t instance which contains caller ID information. Pass (NULL) if no caller ID to be presented.

Description:

Start ringing on an interface.

Example Code:

```
if(EXEC_SUCCESS != adamInfcRing(INFC0, 4000, NULL)){
    printf("Error: adamInfcRing failed! \n");
}
```

7.6 adamInfcStopRing

Prototype:

```
exCode_e    adamInfcStopRing(infcId_e infc);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM
FUNC_UNSupport
DEVICE_BUSY
UNKNOWN_ERROR

Arguments:

Name	Description
Infc	Interface ID.

Description:

Stop ringing on an interface.

Example Code:

```
if(EXEC_SUCCESS != adamInfcStopRing(INFC0)){
    printf("Error: adamInfcStopRing failed! \n");
}

/* Another alternative is call adamInfcRing with dur=0 */
if(EXEC_SUCCESS != adamInfcRing(INFC0, 0, NULL)){
    printf("Error: adamInfcRing failed! \n");
}
```

7.7 adamInfcConfigHookTs

Prototype:

exCode_e adamInfcConfigHookTs(infcId_e infc, hookThreshold_t *pHookTs);

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM

Arguments:

Name	Description
infc	Interface ID.
pHookTs	Pointer of a hookThreshold_t instance which contains hook threshold information.

Description:

Change hook state threshold. For user to correctly get a hook state event on FXS, such as hook-flash and hook-release, user may configure the hook threshold parameters through this API. Only works for FXS interface.

Example Code:

```
hookThreshold_t hookTs;

hookTs.flashMin = 350; /* msec*/
hookTs.flashMax = 700; /* msec*/
hookTs.releaseMin = 800; /* msec*/
hookTs.fxoFlashTime = 800; /* msec*/

if (EXEC_SUCCESS != adamInfcConfigHookTs(INFC2, &hookTs)) {
    printf("Error: adamInfcConfigHookTs failed! \n");
}
```

7.8 adamInfcLineTest

Prototype:

```
exCode_e adamInfcLineTest(infcLineTest_t* lineTest);
```

Return Values:

EXEC_SUCCESS

EXEC_FAIL

INVALID_PARAM

Arguments:

Name	Description
lineTest	Interface ID,linetest ID and lineTestData

Description:

Get Line Test information.

Example Code:

```
infcLineTest_t lineTest;

lineTest. infc = 0;
lineTest. lineTestId = 1;

if (EXEC_SUCCESS != adamInfcLineTest (&lineTest)) {
    printf("Error: adamInfcLineTest failed! \n");
}
```

7.9 adamInfcCULineTest

Prototype:

```
exCode_e adamInfcCULineTest(infcLineTest_t* lineTest, int slicType, infcUnicomLineTest_t* unicomLineTest)
```

Return Values:

EXEC_SUCCESS

EXEC_FAIL

INVALID_PARAM

Arguments:

Name	Description
lineTest	Interface ID,linetest ID and lineTestData
slicType	slicType
infcUnicomLineTest	UnicomLineTest results and params.

Description:

China Unicom inside and outside line test.

Example Code:

```
infcUnicomLineTest_t unicomLineTest;
infcLineTest_t lineTest;

if(EXEC_SUCCESS != adamInfcCULineTest (&lineTest, SlicType, &unicomLineTest))
    printf("Error: adamInfcCULineTest failed! \n");
```

7.10 adamInfcSlicTypeQuery

Prototype:

```
exCode_e adamInfcSlicTypeQuery(infcld_e infc, slicParams_t* slicParams);
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM

Arguments:

Name	Description
infc	Interface ID.
slicParams	Include slic fxsnum fxonum and slic type.

Description:

Get slic information: slic fxsnum fxonum and slic type.

Example Code:

```
slicParams_t slicParams;

if (EXEC_SUCCESS != adamInfcSlicTypeQuery (INFC0, & slicParams)) {
    printf("Error: adamInfcSlicTypeQuery failed! \n");
}
```

7.11 adamInfcRingParams

Prototype:

```
exCode_e adamInfcRingParams(infcld_e infc, infcRingParams_t* ringParams)
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM

Arguments:

Name	Description
infc	Interface ID.
ringParams	Include infcRingType_e frequency amplitude ,dcBias , ringTripThreshold and amplitudeSlab.

Description:

Set ringParams.

Example Code:

```
infcConfig_t infcConf;

if(EXEC_SUCCESS != adamInfcRingParams(infc, &infcConf.ringParams))
    printf("Error: adamInfcRingParams failed! \n");
```

7.12 adamInfcDcFeedParams

Prototype:

```
exCode_e adamInfcDcFeedParams(infcId_e infc, infcDcFeedParams_t* dcFeedParams)
```

Return Values:

EXEC_SUCCESS
EXEC_FAIL
INVALID_PARAM

Arguments:

Name	Description
infc	Interface ID.
dcFeedParams	Include ila and ilaSlab.

Description:

Set dcFeedParams.

Example Code:

```
infcConfig_t infcConf;

if(EXEC_SUCCESS != adamInfcDcFeedParams(infc, &infcConf.dcFeedParams))
    printf("Error: adamInfcDcFeedParams failed! \n");
```

7.13 adamInfcConfigHwProf

Prototype:

```
exCode_e adamInfcConfigHwProf(infcId_e infc,hwProf_e hwProfId)
```

Return Values:

INVALID_PARAM
FUNC_UN SUPPORT

Arguments:

Name	Description
infc	Interface ID.
hwProfId	Country code.

Description:

Change country code.

Example Code:

```
infclde infc = 0;
hwProf_e hwProf = 0;
hwProf = 2;

adamInfcConfigHwProf(infc,hwProf);
```

8 ADAM EVENT Processing

Event is used to notify user something happened, such as a signal has been detected which match the configured patterns, or user off-hook the phone. Developer should handle the event properly and do corresponding process for each event.

There are two modes of event handling process: callback mode and polling mode.

Example code of event handling process:

1. Callback mode:

In Callback mode, when a valid event is notified, it will be handled immediately by calling callback function. Callback function is set in function `adamPollEvent()`.

This mode is recommended.

```

event_t mEvent;
mEvent.eventNotify = _threadEventPolling;
if (EXEC_SUCCESS != adamPollEvent(BLOCKING, &mEvent)) {
    printf ("Critical error: Cannot create thread!\n");
    exit (1);
}
static void _threadEventPolling(void *argv) {
    event_t *mEvent = (event_t *)argv;
    switch(mEvent->evtCode){
    case EVENT_CODE_HOOK:
        switch(mEvent.context.hook.status) {
        case EVENT_CODE_CID:
        case HOOK_SEIZE:
            printf("\nInterface (%d) off-hooked.\n", mEvent.infclId);
            break;
        case HOOK_RELEASE:
            printf("\nInterface (%d) on-hooked.\n", mEvent.infclId);
            break;
        case HOOK_FLASH:
            printf("\nInterface (%d) hook-flashed.\n", mEvent.infclId);
            break;
        default:
            break;
        }
        break;
    case EVENT_CODE_TONE:
        printf("\nChannel (%d) tone[%s] detected.\n", \
            mEvent.chanId, etosTone(mEvent.context.tone.code));
        break;
    default:
        break;
    }
}

```

2. Polling mode:

In Polling mode, `adamPollEvent()` is called to poll DSP and Interface event. When `blockMode` is set to `BLOCK`, the function is queued until a valid event available then returns the event. When a valid event is notified, it will be put in a queue, event will be obtained from queue and handled, the recommended polling interval is 10ms.


```
pthread_t thread;
if (pthread_create(&thread, NULL, (void *) _threadEventPolling, NULL) != 0) {
    printf ("Critical error: Cannot create thread!\n");
    exit (1);
}
static void _threadEventPolling(void){
    event_t mEvent;
    while(1) {
        usleep(10000);
        memset(&mEvent, 0, sizeof(event_t));
        if (EXEC_SUCCESS == adamPollEvent(BLOCKING, &mEvent)) {
            printf("\n[T:%010u] %s: %s", (unsigned int)mEvent.dspTick, \
                etosEdge(mEvent.edge), etosEvent(mEvent.evtCode));
            switch(mEvent.evtCode) {
                case EVENT_CODE_HOOK:
                    switch(mEvent.context.hook.status) {
                        case HOOK_SEIZE:
                            printf("\nInterface (%d) off-hooked.\n", mEvent.infclId);
                            break;
                        case HOOK_RELEASE:
                            quitEvcom();
                            printf("\nInterface (%d) on-hooked.\n", mEvent.infclId);
                            break;
                        case HOOK_FLASH:
                            printf("\nInterface (%d) hook-flashed.\n", mEvent.infclId);
                            break;
                        default:
                            break;
                    }
                    break;
                case EVENT_CODE_TONE:
                    printf("\nChannel (%d) tone[%s] detected.\n", \
                        mEvent.chanId, etosTone(mEvent.context.tone.code));
                    break;
                default:
                    break;
            }
        }
    }
}
```

9 Appendix: Constant and Enumeration

9.1 Constant

Constant	Value
MAX8	(0xff)
MAX16	(0xffff)
MAX32	(0xffffffff)
MAX_CID_CHAR_LEN	(32)
MAX_CADENCE	(3)
DUAL_TONE_FREQ	(2)
SIT_TONE_FREQ	(5)
MAX_TONE_FREQ	(4)
MAX_PACKET_SZ	(1024)
MAX_GAIN_AMP	(40)
MIN_GAIN_AMP	(-40)
MAX_CODEC_NUM	(CODEC_T38+1)
MASK_CODEC_G711A	(1 << CODEC_G711A)
MASK_CODEC_G711U	(1 << CODEC_G711U)
MASK_CODEC_G723	(1 << CODEC_G723)
MASK_CODEC_G722	(1 << CODEC_G722)
MASK_CODEC_G726	(1 << CODEC_G726)
MASK_CODEC_G729	(1 << CODEC_G729)
MASK_CODEC_SILCOMP	(1 << CODEC_SILCOMP)
MASK_CODEC_DTMFR	(1 << CODEC_DTMFR)
MASK_CODEC_T38	(1 << CODEC_T38)
DETECT_TONE_DTMF	(1 << 0)
DETECT_TONE_MODEM	(1 << 1)
DETECT_TONE_CPT	(1 << 2)
DETECT_CID	(1 << 3)
DETECT_DTMFR	(1 << 4)
JB_SZ_MIN	(0)
JB_SZ_MAX	(800)

9.2 activeState_e

```
typedef enum {
    STATE_INACTIVE,
    STATE_ACTIVE
} activeState_e;
```

9.3 blockMode_e

```
typedef enum {
    BLOCKING,
    NON_BLOCKING
} blockMode_e;
```

9.4 chanId_e

```
typedef enum {
    CH0,
    CH1,
    CH2,
    CH3,
    CH4,
    CH5,
    CH6,
    CH7
} chanId_e;
```

9.5 cidFormat_e

```
typedef enum {
    CID_FORMAT_BELLCORE_FSK,
    CID_FORMAT_ETSI_DTMF,
    CID_FORMAT_NTT,
    CID_FORMAT_ETSI_RPAS,
    CID_FORMAT_ETSI_DTAS
} cidFormat_e;
```

9.6 codec_e

```
typedef enum {
    CODEC_G711A,
    CODEC_G711U,
    CODEC_G722,
    CODEC_G723,
    CODEC_G726,
    CODEC_G729,
    CODEC_SILCOMP,
    CODEC_DTMFR,
    CODEC_T38,
    CODEC_INVALID
} codec_e;
```

9.7 dspIId_e

```
typedef enum {
    DSP_VIKING,
    DSP_MTK
} dspIId_e;
```

9.8 dtmf_e

```
typedef enum {
    DTMF_0,
    DTMF_1,
    DTMF_2,
    DTMF_3,
    DTMF_4,
    DTMF_5,
    DTMF_6,
    DTMF_7,
    DTMF_8,
    DTMF_9,
    DTMF_STAR,
    DTMF_POUND,
    DTMF_A,
    DTMF_B,
    DTMF_C,
    DTMF_D,
    RFC2833_DIAL_TONE=255,
    RFC2833_BUSY_TONE,
    RFC2833_CONGESTION_TONE,
    RFC2833_RINGBACK_TONE,
    RFC2833_ERORDER_TONE,
    RFC2833_CALLWAITING_TONE,
    RFC2833_IDENTIFICATION_TONE,
    RFC2833_NEGATIVE_IND_TONE,
    RFC2833_POSITIVE_IND_TONE,
    RFC2833_STUTTER_DIAL_TONE,
    RFC2833_HORNING_TONE,
    RFC2833_SECOND_DIAL_TONE,
    RFC2833_CAS_TONE,
} dtmf_e;
```

9.9 ecTail_e

```
typedef enum {
```

```

        TAIL_16MS,
        TAIL_32MS,
        TAIL_48MS,
        TAIL_64MS,
        TAIL_128MS
    } ecTail_e;

```

9.10 enableControl_e

```

typedef enum {
    CONTROL_DISABLE,
    CONTROL_ENABLE
} enableControl_e;

```

9.11 evaBool_e

```

typedef enum {
    EVA_FALSE,
    EVA_TRUE
} evaBool_e;

```

9.12 eventCode_e

```

typedef enum {
    EVENT_CODE_INVALID,
    EVENT_CODE_TONE,
    EVENT_CODE_CID,
    EVENT_CODE_LINE,
    EVENT_CODE_HOOK,
    EVENT_CODE_T38,
    EVENT_CODE_JB_UPDATE,
    EVENT_CODE_NON_RTP_RECVD,
    EVENT_CODE_RTCP_SEND,
    EVENT_CODE_RTCP_RECVD,
    EVENT_CODE_STREAM_UPDATE,
    EVENT_CODE_TIMER,
    EVENT_CODE_PERFORMANCE,
    EVENT_CODE_ERROR
} eventCode_e;

```

9.13 eventEdge_e

```

typedef enum {
    EDGE_ONCE,
    EDGE_BEGIN,

```

```
EDGE_END
} eventEdge_e;
```

9.14 exCode_e

```
typedef enum {
    EXEC_SUCCESS = 1,
    EXEC_FAIL = -1,
    INVALID_PARAM = -2,
    FUNC_UNSupport = -3,
    DEVICE_BUSY = -4,
    UNKNOWN_ERROR = -5
} exCode_e;
```

9.15 hookState_e

```
typedef enum {
    HOOK_FLASH,
    HOOK_RELEASE,
    HOOK_SEIZE,
    HOOK_PULSE1,
    HOOK_PULSE2,
    HOOK_PULSE3,
    HOOK_PULSE4,
    HOOK_PULSE5,
    HOOK_PULSE6,
    HOOK_PULSE7,
    HOOK_PULSE8,
    HOOK_PULSE9,
    HOOK_PULSE10,
    HOOK_PULSE11,
    HOOK_PULSE12,
    HOOK_PULSE13,
    HOOK_PULSE14,
    HOOK_PULSE15,
    HOOK_PULSE16,
    HOOK_PULSE17,
    HOOK_PULSE18,
    HOOK_PULSE19,
    HOOK_PULSE20,
    HOOK_ERROR
} hookState_e;
```

9.16 **infcld_e**

```
typedef enum {  
    INFC0,  
    INFC1,  
    INFC2,  
    INFC3,  
    INFC4,  
    INFC5,  
    INFC6,  
    INFC7  
} infcld_e;
```

9.17 **infcType_e**

```
typedef enum {  
    INFC_FXS,  
    INFC_FXO,  
    INFC_AUDIO,  
    INFC_OTHER  
} infcType_e;
```

9.18 **ipVer_e**

```
typedef enum {  
    IPV4,  
    IPV6  
} ipVer_e;
```

9.19 **jbMode_e**

```
typedef enum {  
    JB_ADAPT,  
    JB_FIXED  
} jbMode_e;
```

9.20 **lineState_e**

```
typedef enum {  
    LINE_DOWN,  
    LINE_ACTIVE_FWD,  
    LINE_ACTIVE_REV,  
    LINE_RING,  
    LINE_RING_PAUSE,  
    LINE_BUSY,
```

```

        LINE_SLEEP,
        LINE_ERROR
    } lineState_e;

```

9.21 polDir_e

```

typedef enum {
    POL_FWD,
    POL_REV
} polDir_e;

```

9.22 pTime_e

```

typedef enum {
    PTIME_10MS,
    PTIME_20MS,
    PTIME_30MS,
    PTIME_40MS,
    PTIME_50MS,
    PTIME_60MS
} pTime_e;

```

9.23 strmDir_e

```

typedef enum {
    STRM_INACTIVE,
    STRM_SENDFONLY,
    STRM_RECVONLY,
    STRM_SENDRECV
} strmDir_e;

```

9.24 strmlId_e

```

typedef enum {
    STRM0,
    STRM1,
    STRM2,
    STRM3
} strmlId_e;

```

9.25 t38State_e

```

typedef enum {
    T38_DISCONN

```



```
} t38State_e;
```

9.26 toneCode_e

```
typedef enum {
    TONE_DTMF_1 = 1,
    TONE_DTMF_2,
    TONE_DTMF_3,
    TONE_DTMF_4,
    TONE_DTMF_5,
    TONE_DTMF_6,
    TONE_DTMF_7,
    TONE_DTMF_8,
    TONE_DTMF_9,
    TONE_DTMF_0,
    TONE_DTMF_STAR,
    TONE_DTMF_POUND,
    TONE_DTMF_A,
    TONE_DTMF_B,
    TONE_DTMF_C,
    TONE_DTMF_D,
    TONE_DIAL,
    TONE_RINGBACK,
    TONE_BUSY,
    TONE_REORDER,
    TONE_SIT,
    TONE_CUSTOM_1,
    TONE_CUSTOM_2,
    TONE_CUSTOM_3,
    TONE_CUSTOM_4,
    TONE_CNG,
    TONE_CED,
    TONE_ANS,
    TONE_ANSAM,
    TONE_V21PREAMBLE,
    TONE_FAX_CM,
    TONE_FAX_TIMEOUT,
    TONE_INVALID
} toneCode_e;
```

9.27 toneDir_e

```
typedef enum {
    TONE_DIR_LOCAL,
    TONE_DIR_REMOTE
}
```

```
} toneDir_e;
```

9.28 toneType_e

```
typedef enum {
    TONE_REGULAR,
    TONE_MODULATE
} toneType_e;
```

9.29 MWIValue_e

```
typedef enum {
    MTK_EVA_MWI_VALUE_NONE,
    MTK_EVA_MWI_VALUE_SET,
    MTK_EVA_MWI_VALUE_CLEAR,
    MTK_EVA_MWI_VALUE_MAX
} MWIValue_e;
```

9.30 hwProf_e

```
typedef enum{
    C_DEF = 0, /*default*/
    C_AU,      /*Australia*/
    C_AT,      /*Austria*/
    C_BE,      /*Belgium*/
    C_BR,      /*Brazil*/
    C_CA,      /*Canada*/
    C_CN,      /*China*/
    C_CZ,      /*Czech*/
    C_DK,      /*Denmark*/
    C_EU,      /*ETSI*/
    C_FI,      /*FinLand*/
    C_FR,      /*France*/
    C_DE,      /*German*/
    C_GR,      /*Greece*/
    C_HK,      /*HongKong*/
    C_HU,      /*Hungary*/
    C_IE,      /*Ireland*/
    C_IL,      /*Israel*/
    C_IN,      /*India*/
    C_IT,      /*Italy*/
    C_JP,      /*Japan*/
    C_SG,      /*Singapore*/
    C_KR,      /*S.Korea*/
    C_MX,      /*Mexico*/
```

```

C_NL,      /*Netherlands*/
C_NZ,      /*NewZealand*/
C_NO,      /*Norway*/
C_PL,      /*Poland*/
C_PT,      /*Portugal*/
C_SI,      /*Slovenia*/
C_TW,      /*Taiwan*/
C_ES,      /*Spain*/
C_SE,      /*Sweden*/
C_GB,      /*U.K.*/
C_US,      /*U.S.*/
C_ZA,      /*S.Africa*/
C_VN,      /*Vietnam*/
C_RSV1,    /*Reserved Country #1*/
C_RSV2,    /*Reserved Country #2*/
C_RSV3,    /*Reserved Country #3*/
C_NUM
}hwProf_e;

```

10 Appendix: Data Structure

10.1 cadence_t

```
typedef struct {
    uint16 onTime;
    uint16 offTime;
} cadence_t;
```

Description:

Cadence holds the time information a signal on-off duration.

Attribute	Type	Valid Value Range	Description
onTime	uint16	0 ~ 65535(ms)	Time of signal on.
offTime	uint16	0 ~ 65535(ms)	Time of signal off.

10.2 chanConfig_t

```
typedef struct {
    uint16          detectMask;
    int8            ampTx;
    int8            ampRx;
    enableControl_e ecEnable;
}chanConfig_t;
```

Description:

Channel configuration holds the configuration information per channel, including the signal detector controller on/off, Tx/RX gain, and echo canceller on/off.

Attribute	Type	Valid Value Range	Description
detectMask	uint16	DETECT_TONE_DTMF DETECT_TONE_MODEM DETECT_TONE_CPT DETECT_CID DETECT_DTMFR	Bit mask configuration to enable/disable DSP detectors.
ampTx	int8	-40 ~ 40 (0.5db)	Adjust Tx (output raw PCM) gain to the interface within +/-20db range, step by 0.5db.
ampRx	int8	-40 ~ 40 (0.5db)	Adjust Rx (input raw PCM) gain from the interface within +/-20db range, step by 0.5db.
ecEnable	enableControl_e	CONTROL_DISABLE CONTROL_ENABLE	Enable or disable echo cancellation.

10.3 cid_t

```
typedef struct {
    cidFormat_e format;
    char number[MAX_CID_CHAR_LEN];
    char name[MAX_CID_CHAR_LEN];
    char dateTime[MAX_CID_CHAR_LEN];
    char blockInfo[MAX_CID_CHAR_LEN];
    MWIValue_e mwi;
    int fskPower;
    int dtmfPower;
    uint16 makeTime;
    uint16 breakTime;
    uint32 dtmfT3;
    struct{
        uint32 T9; /*unit:10ms*/
        uint32 T10;
        uint32 T11;
        uint32 T12;
        uint32 T13;
        uint32 T14;
    }cid2timing;
    struct{
        uint32 T2;
        uint32 T3;
        uint32 T4;
    }cid1timing;
}cid_t;
```

Description:

Caller ID holds caller ID information such as number, user name, and date-time.

Attribute	Type	Valid Value Range	Description
format	cidFormat_e	CID_FORMAT_BELLCORE_FSK, CID_FORMAT_ETSI_FSK, CID_FORMAT_NTT, CID_FORMAT_ETSI_RPAS, CID_FORMAT_ETSI_DTAS, CID_FORMAT_DTMF	Caller ID transmission format.
number	char[]		Caller ID display number.
name	char[]		Caller ID display name.
dateTime	char[]	MMddHHmm	Caller ID display date and time.
blockInfo	Char[]		Reason for absence of Caller ID
mwi	MWIValue_e	MTK_EVA_MWI_VALUE_NONE,	Related to mailbox value.

		MTK_EVA_MWI_VALUE_SET, MTK_EVA_MWI_VALUE_CLEAR, MTK_EVA_MWI_VALUE_MAX	
fskPower	int		Fsk call id power
dtmfPower	int	-25~-45(dbm)	DTMF call id power
makeTime	uint16		DTMF call id make timer
breakTime	uint16		DTMF call id break timer
dtmfT3	uint32		Timer between DTMF cid and the first ring
cid2timing.T9	uint32	0~15(10ms)	
cid2timing.T10	uint32	0~15(10ms)	
cid2timing.T11	uint32	4~5(10ms)	
cid2timing.T12	uint32	6~20(10ms)	
cid2timing.T13	uint32	4~12(10ms)	
cid2timing.T14	uint32	15~16(10ms)	
cid1timing.T2	uint32	200~500(ms)	
cid1timing.T3	uint32	500~800(ms)(RPAS timing)	
cid1timing.T4	uint32	45~500(ms)(DTAS timing)	

**** NOTE:** For some telephones do not support display name or date-time, they might not be able to show number when name and/or dateTime field is presented.

$T9+T10+T14 \leq 32(10ms)$

10.4 cpt_t

```
typedef struct {
    toneCode_e type;

    union {
        struct{
            int16    toneFreq[DUAL_TONE_FREQ];
            int16    toneDev[DUAL_TONE_FREQ];
            int16    minMake[MAX_CADENCE];
            int16    maxMake[MAX_CADENCE];
            int16    minBreak[MAX_CADENCE];
            int16    maxBreak [MAX_CADENCE];
            int16    power
        }dual;
        struct{
            int16    toneFreq[SIT_TONE_FREQ];
            int16    toneDev[SIT_TONE_FREQ];
            int16    minShortDur;
            int16    maxShortDur;
            int16    minLongDur;
            int16    maxLongDur;
            int16    power
        }
    }
};
```

```

        }sit;

    };
}cpt_t;

```

Description:

Call progress tone is used to configure the frequency, frequency deviation, power, etc., for tone detection.

Attribute	Type	Valid Value Range	Description
type	toneCode_e	TONE_DIAL TONE_RINGBACK TONE_BUSY TONE_REORDER TONE_SIT TONE_CUSTOM_1 TONE_CUSTOM_2 TONE_CUSTOM_3 TONE_CUSTOM_4	Type of CPT Note: For SIT tone detection configuration, set with sit struct . For others tone detection configuration, set with dual struct.
dual.toneFreq	int16[]	0 ~ 4000(Hz)	Tone frequency(s) to be detected.
dual.toneDev	int16[]	0 ~ 4000(Hz)	Tolerated frequency deviation for each tone frequency detection.
dual.minMake	int16[]	0 ~ 65535(ms)	Minimum make (signal-on) time of a cadence.
dual.maxMake	int16[]	0 ~ 65535(ms)	Maximum make (signal-on) time of a cadence. Note: For continuous tone detection configuration, set maxMake/minBreak/maxBreak to (0).
dual.minBreak	int16[]	0 ~ 65535(ms)	Minimum break (signal-off) time of a cadence. Note: For continuous tone detection configuration, set maxMake/minBreak/maxBreak to (0).
dual.maxBreak	int16[]	0 ~ 65535(ms)	Maximum break (signal-off) time of a cadence. Note: For continuous tone detection configuration, set maxMake/minBreak/maxBreak to (0).
dual.power	int16	-40 ~ 0(db)	Tone power threshold for

Attribute	Type	Valid Value Range	Description
			detection.
sit.toneFreq	int16[]	0 ~ 4000(Hz)	Tone frequency(s) to be detected.
sit.toneDev	int16[]	0 ~ 4000(Hz)	Tolerated frequency deviation for each tone frequency detection.
minShortDur	int16	0 ~ 65535(ms)	Minimum make (signal-on) time of the short tone cadence
maxShortDur	int16	0 ~ 65535(ms)	Maximum make (signal-on) time of the short tone cadence
minLongDur	int16	0 ~ 65535(ms)	Minimum make (signal-on) time of the long tone cadence
maxLongDur	int16	0 ~ 65535(ms)	Maximum make (signal-on) time of the long tone cadence
sit.power	int16	-40 ~ 0(db)	Tone power threshold for detection.

10.5 dspFeature_t

```
typedef struct {
    uint16  dspld;
    uint8   numOfChan;
    uint8   strmsPerChan;
    uint32  codecSupport;
    uint32  ptimeSupport;
    uint8   rtpRedundancy;
    uint8   maxEcTailLength;
}dspFeature_t;
```

Description:

DSP feature is used to acquire DSP capability information.

Attribute	Type	Valid Value Range	Description
dspld	uint16	Read-Only	Provide DSP ID information.
numOfChan	uint8	Read-Only	Number of channel supports on the DSP.
strmsPerChan	uint8	Read-Only	Number of streams supports on each channel.
codecSupport	uint32	Read-Only	CODECs supports on the DSP. Use MASK_CODEC_X to check.
ptimeSupport	uint32	Read-Only	PTime supports on the DSP.
rtpRedundancy	uint8	Read-Only	Check if DSP supports RTP redundancy (RFC2198).
maxEcTailLength	uint8	Read-Only	Maximum echo cancellation tail length that DSP supports.

10.6 eventContext_u

```
typedef union {
    struct{
        lineState_e    status;
        int            ringCount;
    }line; /* EVENT_CODE_LINE */

    struct{
        hookState_e    status;
        int            pulseCount;
    }hook; /*EVENT_CODE_HOOK*/

    struct{
        uint32    dspExecTimes;
        uint32    averageMhz;
    }performanceIdx; /*EVENT_CODE_PERFORMANCE*/

    struct{
        int8    number[MAX_CID_CHAR_LEN];
        int8    name[MAX_CID_CHAR_LEN];
        int8    dateTime[MAX_CID_CHAR_LEN];
    }cidData; /*EVENT_CODE_CID_DETECTED*/

    struct{
        toneCode_e    code;
        toneDir_e    dir;
    }tone; /*EVENT_CODE_TONE_DETECTED*/

    struct{
        uint8 streamId;
        t38State_e status;
    }t38;

    struct{
        uint8    streamId;
        uint32    total;
        uint32    drop;
        uint32    plc;
        uint32    jbSize;
        uint32    avgJitter;
    }jb; /*EVENT_CODE_JB_UPDATE*/

    struct{
        netAddr_t    srcAddr;
        netAddr_t    dstAddr;
    }
}
```

```

uint8      payload[MAX_PACKET_SZ];
}packet; /*EVENT_CODE_NON_RTP_RECVD*/
}eventContext_u;

```

Description:

Event context provide the detail information/data for particular events.

Attribute	Type	Valid Value Range	Description
line.status	lineState_e	LINE_DOWN LINE_ACTIVE_FWD LINE_ACTIVE_REV LINE_RING LINE_RING_PAUSE LINE_BUSY LINE_SLEEP LINE_ERROR	Interface line state.
line.ringCount	int		Ring times counter for RING event.
hook.status	hookState_e	HOOK_FLASH HOOK_RELEASE HOOK_SEIZE HOOK_PULSE1 HOOK_PULSE2 HOOK_PULSE3 HOOK_PULSE4 HOOK_PULSE5 HOOK_PULSE6 HOOK_PULSE7 HOOK_PULSE8 HOOK_PULSE9 HOOK_PULSE10 HOOK_PULSE11 HOOK_PULSE12 HOOK_PULSE13 HOOK_PULSE14 HOOK_PULSE15 HOOK_PULSE16 HOOK_PULSE17 HOOK_PULSE18 HOOK_PULSE19 HOOK_PULSE20 HOOK_ERROR	Interface hook state.
hook.pulseCount	int		Pulse time counter for PULSE event. ** Not used now.
performanceldx.dspExecTimes	uint32		Not used now.

Attribute	Type	Valid Value Range	Description
performanceldx.averageMhz	uint32		Not used now.
cidData.number	int8		CID display number information.
cidData.name	int8		CID display name information.
cidData.dateTime	int8		CID date-time information.
tone.code	toneCode_e		Tone code information.
tone.dir	toneDir_e		Tone direction information.
t38.streamId	uint8		Stream ID information
t38.status	t38State_e		T38 transmission status
jb.streamId	uint8		Stream ID information.
jb.total	uint32		Number of total packets.
jb.drop	uint32		Number of dropped packets.
jb.plc	uint32		Number of compensated packets.
jb.jbSize	uint32		Current jitter buffer size.
jb.avgJitter	uint32		Average jitter.
packet.srcAddr	netAddr_t		Not used now.
packet.dstAddr	netAddr_t		Not used now.
packet.payload	uint8[]		Not used now.

10.7 event_t

```
typedef struct {
    eventEdge_e      edge;
    eventCode_e      evtCode;
    uint32            dspTick;
    infcld_e         infcld;
    chanId_e         chanId;
    eventContext_u    context;
    void (*eventNotify) (void *argv);
} event_t;
```

Description:

Event report provide event category, event time, interface/channel ID, and event context information.

Attribute	Type	Valid Value Range	Description
edge	eventEdge_e	EDGE_ONCE EDGE_BEGIN EDGE_END	Edge information of an event report. For some events that will last for a while, such as tone or ring, etc., the event reports once at the beginning with

Attribute	Type	Valid Value Range	Description
			EDGE_BEGIN and again at the end with EDGE_END. User may use dspTick of both events to get the duration of the event last. For other events represented a state change, such as on-hook, off-hook, EDGE_ONCE is used.
evtCode	eventCode_e	EVENT_CODE_INVALID EVENT_CODE_TONE EVENT_CODE_CID EVENT_CODE_LINE EVENT_CODE_HOOK EVENT_CODE_JB_UPDATE EVENT_CODE_NON_RTP_RECVD EVENT_CODE_RTCP_SEND EVENT_CODE_RTCP_RECVD EVENT_CODE_STREAM_UPDATE EVENT_CODE_TIMER EVENT_CODE_PERFORMANCE EVENT_CODE_ERROR	Event message categorization information.
dspTick	uint32	0x0~0xFFFFFFFF(ms)	DSP (or CPU) tick to indicate the time information of an event.
infclId	infclId_e		Interface ID, presented when an event is interface related.
chanId	chanId_e		Channel ID, presented when an event is channel related.
Context	eventContext_u		Event context contains detail information of the event if applicable.
eventNotify	void		Call back function point for register. If user hook function on it, it need not polling event periodically. If there is an event coming, this registered function will be called.

10.8 hookThreshold_t

typedef struct {

```
uint32    flashMin;
uint32    flashMax;
uint32    releaseMin;
uint32    fxoFlashTime;
} hookThreshold_t;
```

Description:

Flash and release threshold information

Attribute	Type	Valid Value Range	Description
flashMin	uint32	0x0~0xFFFFFFFF(ms)	Minimum hook release time for hook-flash state.
flashMax	uint32	0x0~0xFFFFFFFF(ms)	Maximum hook release time for hook-flash state.
releaseMin	uint32	0x0~0xFFFFFFFF(ms)	Minimum hook release time for hook-release state.
fxoFlashTime	uint32	0x0~0xFFFFFFFF(ms)	Once hook-flash time for FXO

10.9 infcConfig_t

```
typedef struct {
    infcType_e    type;
    lineState_e   lineState;
    hookState_e   hookState;
    polDir_e      pol;

    ringProfile_t ring;
    hookThreshold_t hookTs;
} infcConfig_t;
```

Description:

Interface configuration provide the line, hook, and/or ring configuration information of an interface.

Attribute	Type	Valid Value Range	Description
Type	infcType_e	INFC_FXS INFC_FXO INFC_AUDIO INFC_OTHER	Read-only, device type of the interface.
lineState	lineState_e	LINE_DOWN LINE_ACTIVE_FWD LINE_ACTIVE_REV LINE_RING LINE_RING_PAUSE	Interface line state. Read-Writable for FXS interface and Read-only for FXO interface.

		LINE_BUSY LINE_SLEEP LINE_ERROR	
hookState	hookState_e	HOOK_FLASH HOOK_RELEASE HOOK_SEIZE HOOK_PULSE1 HOOK_PULSE2 HOOK_PULSE3 HOOK_PULSE4 HOOK_PULSE5 HOOK_PULSE6 HOOK_PULSE7 HOOK_PULSE8 HOOK_PULSE9 HOOK_PULSE10 HOOK_PULSE11 HOOK_PULSE12 HOOK_PULSE13 HOOK_PULSE14 HOOK_PULSE15 HOOK_PULSE16 HOOK_PULSE17 HOOK_PULSE18 HOOK_PULSE19 HOOK_PULSE20 HOOK_ERROR	Interface hook state. Read-Writable for FXO interface and Read-only for FXS interface. ** HOOK_PULSEXX is not used now.
pol	polDir_e	POL_FWD POL_REV	Line power feed polarity direction.
ring	ringProfile_t		Ring configuration per interface. Only for FXS.
hookTs	hookThreshold_t		Hook threshold configuration for hook states.,

10.10 jbConfig_t

```
typedef struct {
    jbMode_e          mode;
    uint32            szJbMax;
    uint32            szJbInit;
}strmConfig_t;
```

Description:

Jitter buffer configuration holds the mode, jitter buffer size information.

Attribute	Type	Valid Value Range	Description
-----------	------	-------------------	-------------

Attribute	Type	Valid Value Range	Description
mode	jbMode_e	JB_ADAPT JB_FIXED	Jitter buffer mode.
szJbMax	uint32	0 ~ 800	Maximum size of the jitter buffer.
szJbInit	uint32	0 ~ 800	Initial size of the jitter buffer.

10.11 medialInfo_t

```
typedef struct {
    uint64      rtpError;
    uint64      packetRecv;
    uint64      packetLoss;
    uint64      packetLossRate;
    uint64      maxJitter;
    uint64      maxRTCPInterval;
    uint64      bufUnderflow;
    uint64      bufOverflow;
} medialInfo_t;
```

Description:

Currently media information for user query.

Attribute	Type	Valid Value Range	Description
rtpError	uint64		Received error packet number, including invalid size, wrong SSRC number and wrong RTP version.
packetRecv	uint64		Received packet number.
packetLoss	uint64		Received packet loss number.
packetLossRate	uint64		Received packet loss rate.
maxJitter	uint64		Maximum packet jitter time in (ms).
maxRTCPInterval	uint64		Maximum received RTCP packet interval in (ms).
bufUnderflow	uint64		The number of DSP buffer underflow.
bufOverflow	uint64		The number of DSP buffer overflow.

10.12 netAddr_t

```
typedef struct {
    uint32      addrV4;
    uint16      addrV6[8];
    ipVer_e     ver;
    uint16      port;
    uint16      rtcpPort;
} netAddr_t;
```

Description:

Network address provide the IP address and data port information.

Attribute	Type	Valid Value Range	Description
addrV4	uint32		IPv4 address
addV6	uint16[]		IPv6 address
Ver	ipVer_e	IPv4 IPv6	Network address type
Port	uint16	0 ~ 65535	RTP port
rtcpPort	uint16	0 ~ 65535	RTCP port

10.13 ringProfile_t

```
typedef struct {
    cadence_t    cad[MAX_CADENCE];
    uint32       dur;
    cid_t        cid;
    uint8        cidAt;
    uint32       cidWaitTime;
} ringProfile_t;
```

Description:

Ring profile provide the configuration of a ring.

Attribute	Type	Valid Value Range	Description
Cad	cadence_t[]		Ring cadences.
Dur	uint32	0x0~0xFFFFFFFF(ms)	Ring duration.
cid	cid_t		Caller ID information.
cidAt	uint8	0~255	N-th ring-breaks for CID transmission.
cidWaitTime	uint32	0x0~0xFFFFFFFF(ms)	Waiting time to send CID after ring break.

10.14 session_t

```
typedef struct {
    ipVer_e      ver;
    netAddr_t    srcAddr;
    netAddr_t    dstAddr;
    uint8        encrypt; /*T/F*/ /*reserved for user to select the encrypt type*/
    uint32       dur; /*session time*/ /*reserved for update the session duration*/
    uint32       rtpTosVal; /*RTP packet tos value*/
    uint32       rtcpTosVal; /*RTCP packet tos value*/
} session_t;
```

Description:

Session holds the source and destination network address and other session configuration information.

Attribute	Type	Valid Value Range	Description
Ver	ipVer_e	IPV4 IPV6	IP version of the session.
srcAddr	netAddr_t		Source address of the session.
dstAddr	netAddr_t		Destination address of the session.
Encrypt	uint8		Not used now.
Dur	uint32		Not used now.
rtpTosVal	uint32	0~0xffffffff	Tos value in RTP packet header
rtcpTosVal	uint32`	0~0xffffffff	Tos value in RTCP packet header

10.15 strmAttr_t

```
typedef struct {
    codec_e          payloadSelect;
    pTime_e          ulPtime;
    pTime_e          dlPtime;
    enableControl_e   dtmfRelay;
    enableControl_e   dtmfRemove;
    enableControl_e   silenceComp;
    strmDir_e         direction;
    uint32            jbUpdateTime;
    uint8             tevCtlFlag;
}strmAttr_t;
```

Description:

Stream attribute holds the attribute configurations of a stream.

Attribute	Type	Valid Value Range	Description
payloadSelect	codec_e	CODEC_G711A CODEC_G711U CODEC_G722 CODEC_G723 CODEC_G726 CODEC_G729 CODEC_SILCOMP CODEC_DTMFR CODEC_T38	CODEC used for streaming.
ulPtime	pTime_e	PTIME_10MS PTIME_20MS PTIME_30MS PTIME_40MS PTIME_50MS	Uplink Stream P-time (P-rate) configuration.

Attribute	Type	Valid Value Range	Description
		PTIME_60MS	
dlPtime	pTime_e	PTIME_10MS PTIME_20MS PTIME_30MS PTIME_40MS PTIME_50MS PTIME_60MS	Downlink Stream P-time (P-rate) configuration.
dtmfRelay	enableControl_e	CONTROL_DISABLE CONTROL_ENABLE	Enable/disable DTMF relay (RFC2833/4733)
dtmfRemove	enableControl_e	CONTROL_DISABLE CONTROL_ENABLE	Remove DTMF in inband mode. Note : DTMF remove configuration is valid only when DTMF reley is disable.
silenceComp	enableControl_e	CONTROL_DISABLE CONTROL_ENABLE	Enable/disable silence compression (CN).
Direction	strmDir_e	STRM_INACTIVE STRM_SENDOONLY STRM_RECVONLY STRM_SENDECV	Stream transmission direction.
jbUpdateTime	uint32	0x0 ~ 0xFFFFFFFF(ms)	Not used now.
tevCtlFlag	Uin8	0x0 0x07	0x01:piggyback events enable 0x02:tone events enable 0x04:CAS tone events enable

10.16 strmConfig_t

```
typedef struct {
    session_t          session;
    strmAttr_t         strmAttr;
    uint8              payloadType[MAX_CODEC_NUM];
    jbConfig_t         jbConf;
    t38Ctrl_t          t38Ctrl;
}strmConfig_t;
```

Description:

Stream configuration holds the session, stream attribute, and payload type number information.

Attribute	Type	Valid Value Range	Description
Session	session_t		Stream session information.
strmAttr	strmAttr_t		Stream attribute configuration.
payloadType	uint8[]	0 ~ 127	Payload type number for each CODEC.
jbConf	jbConfig_t		Stream jitter buffer configuration.
t38Ctrl	t38Ctrl_t		T38 configuration

10.17 t38Ctrl_t

```
typedef struct {
    int32      version;
    int32      maxRate;
    int32      eccType;
    int32      rateMgnt;
    int32      opMode;
}t38Ctrl_t;
```

Description:

T38 configuration for FAX.

Attribute	Type	Valid Value Range	Description
version	int32	0 : T38 version 0 1 : T38 version 1 (not support yet) 2 : T38 version 2 (not support yet) 3 : T38 version 3 (not support yet)	T38 version
maxRate	int32	0 : 2400 bps 1 : 4800 bps 2 : 7200 bps 3 : 9600 bps 4 : 12000 bps 5 : 14400 bps	Maximum Fax transmission rate .
eccType	int32	0 : None 1 : UDP redundancy 2 : UDP FEC (not support yet)	T38 error correct type.
rateMgnt	int32	1 : T38 local TCF 2 : T38 transferred TCF	T38 Fax rate management type.
opMode	int32	0 : unknown 1 : caller 2 : callee	Faxer type. If user configure unknown, DSP will auto detect faxer type.

10.18 tone_t

```
typedef struct {
    toneType_e toneType;

    struct{
        uint16 toneFreq[MAX_TONE_FREQ];
        int16  tonePwr[MAX_TONE_FREQ];
    }regular;

    struct{
```

```

        int16    baseFreq;
        int16    modFreq;
        int16    modPwr;
        int16    modDepth;
    }modulate;

    int16 makeTime[MAX_CADENCE];
    int16 breakTime[MAX_CADENCE];
    int16 repeat[MAX_CADENCE];
}tone_t;

```

Description:

Tone is used to configure the frequency, power, cadence, repeat, etc., for tone generation.

Attribute	Type	Valid Value Range	Description
toneType	toneType_e	TONE_REGULAR TONE_MODULATE	Type of Tone
regular.toneFreq	uint16[]	0 ~ 4000(Hz)	Up to 4 tones frequency can be configured for generation.
regular.tonePwr	int16[]	-40 ~ 0(db)	Tone power for each tone frequency.
modulate.baseFreq	int16	0 ~ 4000(Hz)	Base frequency for amplitude modulation.
modulate.modFreq	int16	0 ~ 4000(Hz)	Amplitude modulation frequency.
modulate.modPwr	int16	-40 ~ 0(db)	Modulation power.
modulate.modDepth	int16	0 ~ 65535	Modulation depth.
makeTime	int16[]	0 ~ 65535(ms)	Up to 3 cadence configuration for tone. Time of signal on for each cadence.
breakTime	int16[]	0 ~ 65535(ms)	Up to 3 cadence configuration for tone. Time of signal off for each cadence.
Repeat	int16[]	0 ~ 65535(time)	Up to 3 cadence configuration for tone. Time of repeat for each cadence.

10.19 toneSeq_t

```

typedef struct {
    uint8 *toneIdSeq;
    uint8 numOfTone;
}toneSeq_t;

```

Description:

Tone sequence holds a series of tone to be generated.

Attribute	Type	Valid Value Range	Description
toneIdSeq	uint8*		Array of Tone ID to be played in sequence.
numOfTone	uint8	0 ~ 255	Number of Tone in the array.

10.20 infcLineTest_t

```
typedef struct {
    infcId_e infc;
    uint8 lineTestId;
    char lineTestData[5000];
}infcLineTest_t;
```

Description:

infcLineTest_t is used to configure the infc ,line testid and return line test data.

Attribute	Type	Valid Value Range	Description
infc	infcId_e	0~7	interface
lineTestId	uint8	zarlink:1~4,7,10~16 siliconlab:1~7	Line test id
lineTestData	Char		return line test data

10.21 SlicParams_t

```
typedef struct {
    uint8 fxsNum;
    uint8 fxoNum;
    char slicType[15];
}slicParams_t;
```

Description:

slicParams is used to return fxsnum,fxonum and slictype.

Attribute	Type	Valid Value Range	Description
fxsNum	uint8	0 ~ 255	Fxs number of slic
fxoNum	uint8	0 ~ 255	Fxo number of slic
slicType	char		slicType

10.22 InfcRingParams_t

```
typedef struct {
    infcRingType_e type;
```

```

int32 frequency;
int32 amplitude;
int32 dcBias;
int32 ringTripThreshold;
int32 amplitudeSlab;
}infcRingParams_t;

typedef enum {
    RING_SINE,
    RING_TRAP
} infcRingType_e;

```

Description:

infcRingParams is used to set and return infcRingType, frequency, amplitude and dcBias.

Attribute	Type	Valid Value Range	Description
type	infcRingType_e	RING_SINE, RING_TRAP	Ring wave form
frequency	int32	1000~100000	frequency
amplitude	int32	0~100000(mV)	amplitude
dcBias	int32	0~(100000- amplitude)	dcBias
ringTripThreshold	int32	0~62500(uA)	ringTripThreshold
amplitudeSlab	int32	47~65(V)	Amplitude of si32176

10.23 InfcDcFeedParams_t

```

typedef struct {
    int32 ila;
    int32 ilaSlab;
}infcDcFeedParams_t;

```

Description:

dcFeedParams is used to set ila.

Attribute	Type	Valid Value Range	Description
ila	Int32	18000~40000(uA)	Loop current.
ilaSlab	Int32	18,20,22,24,26,28(mA)	32176 Loop current.

10.24 infcUnicomLineTest_t

```

typedef struct {
    Outline_Param Outline;
    Inline_Param Inline;
    Outline_Status OutlineStatus;
}

```

```

    Inline_Status InlineDCVStatus;
    Inline_Status InlineRingVStatus;
    Inline_Status InlineCurrentStatus;
} infcUnicomLineTest_t;

typedef struct{
    int32 AGACVoltage;
    int32 BGACVoltage;
    int32 ABACVoltage;
    int32 AGDCVoltage;
    int32 BGDCVoltage;
    int32 ABDCVoltage;
    int32 AGInsulationResistance;
    int32 BGInsulationResistance;
    int32 ABInsulationResistance;
    int32 AGCapacitance;
    int32 BGCapacitance;
    int32 ABCapacitance;
} Outline_Param;

typedef struct {
    signed long DCFeed_Voltage;
    signed long Ringing_Voltage;
    signed long Loop_Current;
} Inline_Param;

typedef enum {
    OutLine_Normal,
    OutLine_PhoneDisconnet,
    OutLine_PhoneoffHook,
    OutLine_PowerLineContacted,
    OutLine_BothLineContacted,
    OutLine_ALineMixOther ,
    OutLine_BLineMixOther,
    OutLine_BothLineGrounding,
    OutLine_ALineGrounding,
    OutLine_BLineGrounding,
    OutLine_ABLinePoorInsulation,
    OutLine_ShortCircuit,
    OutLine_BothLineLeakageToGround,
    OutLine_ALineLeakageToGround,
    OutLine_BLineLeakageToGround
} Outline_Status;

typedef enum {
    Inline_Test_Item_Normal,
    Inline_Test_Item_UNNormal

```

```
} Inline_Status;
```

Description:

infComUnicomLineTest_t is used to that China Unicom inside and outside line test.

Attribute	Type	Valid Value Range	Description
Outline	Outline_Param		Outline Param
Inline	Inline_Param		Inline Param
OutlineStatus	Outline_Status	0~14	Outline Status
InlineDCVStatus	Inline_Status	0~1	Inline Status
InlineRingVStatus	Inline_Status	0~1	Inline Status
InlineCurrentStatus	Inline_Status	0~1	Inline Status

10.25 replaceData_t

```
typedef struct {
    int enable;
    dsp_dir dir;
    int stage;
    char filePath[128];
}replaceData_t;
```

Description:

replaceData_t is used to set DSP replace data information.

Attribute	Type	Valid Value Range	Description
enable	int	0/1	Func enable/disable
dir	dsp_dir	0/1/2	Direction,rx/tx
stage	int	0/2/4/.../16	DSP stage
filePath	char[]		Replace file path

11 Appendix: Default Call Progress Tone Profile

Call Progress Tone (CPT) profile configures the signal patterns to detect whether a signal contains a signal match the configured patterns and report a detection event. It is used on FXO interface to understand the line status by listening to distinctive tones.

Tone	Frequency	Cadence
CPT_DIALTONE	350Hz@-21db 440Hz@-21db	Cadence[0] = 1000(ms)/on, 0(ms)/off,
CPT_RINGBACK	440Hz@-18db 480Hz@-18db	Cadence[0] = 1000(ms)/on, 3000(ms)/off
CPT_BUSY	480Hz@-21db 620Hz@-21db	Cadence[0] = 500(ms)/on, 500(ms)/off
CPT_REORDER	480Hz@-18db 620Hz@-18db	Cadence[0] = 250(ms)/on, 250(ms)/off

Default tone setting is (refer to toneSetting.txt):

```
#<<Setting sequence>>
#f1,f2,f3,pwr1,pwr2,pwr3,mak1,brk1,mak2,brk2,mak3,brk3
#Dail tone
450,0,0,-10,-10,0,500,0,0,0,0,0
#Ring back tone
450,0,0,-10,-10,0,1000,4000,0,0,0,0
#Error tone
950,0,0,-10,-10,0,500,0,0,0,0,0
#Call waiting tone
950,0,0,-15,-15,0,400,10000,0,0,0,0
#Second dail tone
400,0,0,-10,-10,0,400,0,0,0,0,0
#Busy tone
450,0,0,-5,-5,0,350,350,0,0,0,0
#Warning tone
950,0,0,-5,-5,0,500,0,0,0,0,0
#stutter dial tone
400,0,0,-5,-5,0,100,100,0,0,0,0
#1K Hz tone
1000,0,0,-10,0,0,1000,0,0,0,0,0
#Call Waiting Ring back tone
450,0,0,-10,-10,0,400,10000,0,0,0,0
```