# VXG Mobile Video Player SDK for Android Programmer's Guide

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# 1. Overview

Network Media Player SDK consists of a set of resources for fast and convenient development of mobile applications for viewing various media streams like RTMP, HLS, RTSP, RTP, MMS, WebM, FLV, MP4, TS, and other network video formats and playback files with following formats: AVI, MOV, MKV, FLV, AVI, 3GP, 3G2, ASF, WMV, MP4, M4V, TP, TS, MTP, M2T, etc. The core of the SDK is a library for application development.

# **Key Features:**

**Hardware acceleration** – a new hardware accelerated decoder for HD video.

**Multi-core decoding** – support of the multiple processor cores for decoding.

**Multi-channel support** – simultaneous connection to multiple resources or multiple video channels and simultaneous video decoding.

**Video integration with any Activity** – is based on SurfaceView and can be integrated with any Activity.

**Hardware pre and post video processing** – hardware de-interlacing and various pre and post video processing using OpenGL shaders.

**Custom and standard notifications** – notifies application about connection, disconnection and other events. It is possible to add custom events.

**Smart and online thumbnails** – quick and simple API to get thumbnails for local files and network streams.

**Low latency for network stream** – special API to control playback latency.

**Record streams** – special API to record streams into mp4 file.

**Audio and Subtitle control** – special API to control audio and subtitle tracks during playback.

**Audio pitch correction on changed rate** – the filter added for correcting the intonation of an audio signal without affecting other aspects of its sound when playback rate has been applied.

**Audio volume boost and volume detector** – special API to increase audio volume above system ability and to detect max/min volume to avoid any audio clipping on raising volume.

**Pre-buffering data in paused mode** – accumulation of media data in Paused mode to avoid clipping on further playback (audio mode only).

# 2. How to Use

#### 2.1 Android version

The SDK works with Android version 4.0 or newer. (Lower versions can be customized and provided by request as well).

#### 2.2 Folders and files

The SDK package consists of following files and folders:

**bin** (Sample application package)

- MediaPlayerSDKTest.apk
- MediaPlayerSDKTest.LowLatency.apk
- MediaPlayerSDKTest\_view2x2.apk

**libs** (*Library files to be linked to the application*)

- mediaplayersdk.jar
- librtspplr-xx.so
- librtstm-xx.so
- libSDL2-xx.so
- libxml2-xx.so
- libyuv\_shared-xx.so

where xx is one of supported platforms: ARM general, x86, ARV V7, ARM V7a.

**src** (Sample project to test Media Player SDK)

**doc** (Documentation including this document)

## 2.3 Development tools

Build environment is Eclipse. Please import the project to Eclipse for building the sample application.

### 2.4 Integration with an application

# 2.4.1 Integration using a resource file in 2 steps:

**Step1:** Add to layout xml for your activity as below:

```
<FrameLayout
    android:id="@+id/playerViewLayout"
    android:layout_width="fill_parent"
    android:layout_height="wrap_content" >
```

```
<veg.mediaplayer.sdk.MediaPlayer
         android:id="@+id/playerView"
         android:layout_width="fill_parent"
         android:layout_height="fill_parent"
         android:layout_gravity="center" />
   </FrameLayout>
Step 2: Change main activity
(MainActivity.java)
public
                                                           Activity
                                                                         implements
             class
                        MainActivity
                                            extends
MediaPlayer.MediaPlayerCallback
   // callback handler
   #override
   public int Status(int arg) {return 0;}
   @Override
  public void onCreate(Bundle savedInstanceState)
       // Create Player instance
      player = (MediaPlayer)findViewById(R.id.playerView);
// Get player instance
       // Connect or start playback
      player.Open(ConnectionUrl or File name,
    decoderType,
    rendererType,
    synchroEnable,
    synchroNeedDropVideoFrames,
                 rendererEnableColorVideo,
                 rendererEnableAspectRatio,
   DataReceiveTimeout,
```

```
decoderNumberOfCpuCores,
  this);
...
}

@Override
protected void onDestroy()
{
    // Destroy and close player
    if (player != null)
    {
        // Close connection to server
        player.close ();
        // Desroy player
        player.onDestroy();
}
    super.onDestroy();
}
...
}
```

# 2.4.2 Integration dynamically (without modifying resources)

# **Step 1**: Change main activity

```
public class MainActivity extends Activity implements

MediaPlayer.MediaPlayerCallback

{
...
    // callback handler
    #override
    public int Status(int arg) {return 0;}

@Override
    public void onCreate(Bundle savedInstanceState)
    {
```

```
// Create instance of Player
       boolean is_use_window = true;
       // boolean is_use_window = false; if audio only is to play in service
       player = new MediaPlayer(this, is_use_window);
// Set size and position for layout
      FrameLayout.LayoutParams params = new FrameLayout.LayoutParams(250,250,
Gravity.CENTER);
      player.setLayoutParams(params);
//
     // Add Player Instance to layout
      FrameLayout lp = (FrameLayout)findViewById(R.id.playerView);
      lp.addView(player);
// connect and start playback
      player.Open( ConnectionUrl,
                     decoderType,
rendererType,
synchroEnable,
synchroNeedDropVideoFrames,
                     rendererEnableColorVideo,
rendererEnableAspectRatio,
DataReceiveTimeout,
decoderNumberOfCpuCores,
this);
   }
  @Override
  protected void onDestroy()
        // Close network connection to server
           player.close ();
         // Desroy player
         player.onDestroy();
```

```
super.onDestroy();
}
```

# 2.4.3 Integration with Activity

The SDK is based on SurfaceView and can be integrated with any Activity using the code below:

```
<FrameLayout

android:id="@+id/playerViewLayout"

android:layout_width="fill_parent"

android:layout_height="wrap_content" >

<veg.mediaplayer.sdk.MediaPlayer

android:id="@+id/playerView"

android:layout_width="fill_parent"

android:layout_height="fill_parent"

android:layout_gravity="center" />
```

</FrameLayout>

# 3. Media Player

# 3.1 API Reference

There are following API providers in SDK: content provider, decoder provider and render provider:

Provider name	Provider acronym	Description	
Pipeline Provider	PLP_	Controls pipeline and	
		all components	
Content Provider	CP_	Connects to server,	
		downloads data and	
		controls connection	
Video Decoder Provider	VDP_	s/w or h/w video decoding	
Audio Decoder Provider	ADP_	s/w or h/w video decoding	
Video renderer Provider	VRP_	Video renderer	
Audio renderer Provider	ARP_	Audio renderer	

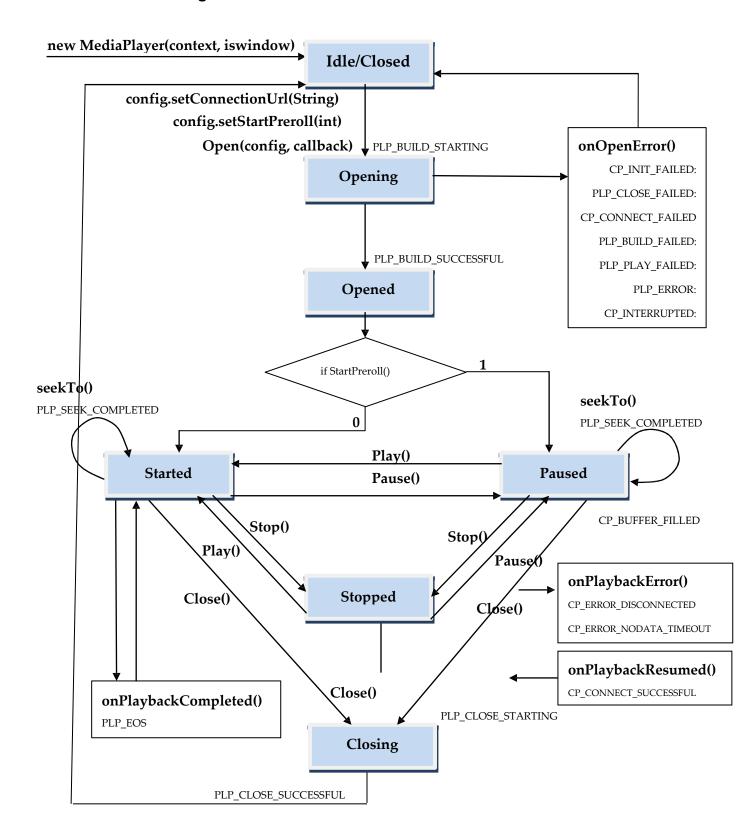
### 3.2 Notifications

Providers notifies about results, errors and notifications using "MediaPlayerCallback" callback. All messages are synchronous and provider waits until the application handles a message.

Valu	Name	Туре	Description
e			
1	PLP_BUILD_STARTING	NOTIFICATION	PLP notifies that pipeline is started to
			build
2	PLP_BUILD_SUCCESSFUL	RESULT	Pipeline has been built successfully
3	PLP_BUILD_FAILED	RESULT	Pipeline can not be built
4	PLP_PLAY_STARTING	NOTIFICATION	Pipeline is going to start
5	PLP_PLAY_SUCCESSFUL	RESULT	Pipeline has been ran successfully after
			Open (autostart)
6	PLP_PLAY_FAILED	RESULT	Error on pipeline starting
7	PLP_CLOSE_STARTING	NOTIFICATION	Pipeline is going to stop
8	PLP_CLOSE_SUCCESSFUL	RESULT	Pipeline has been closed successfully
9	PLP_CLOSE_FAILED	RESULT	Error on pipeline closing
10	PLP_ERROR	ERROR	Pipeline is disconnected due inner error
12	PLP_EOS	NOTIFICATION	End-of-stream notification
14	PLP_PLAY_PLAY	NOTIFICATION	Pipeline has been run successfully
15	PLP_PLAY_PAUSE	NOTIFICATION	Pipeline has been paused successfully

16	PLP PLAY STOP	NOTIFICATION	Pipeline has been stopped successfully
17	PLP SEEK COMPLETED	NOTIFICATION	Seek operation has been completed
101	CP CONNECT STARTING	NOTIFICATION	CP is initialized and is going to start
			connection
102	CP_CONNECT_SUCCESSFUL	RESULT	CP has been connected successfully
103	CP CONNECT FAILED	RESULT	CP notifies that connection is failed
104	CP_INTERRUPTED	RESULT	CP notifies that connection with server
	_		is interrupted by close function
105	CP_ERROR_DISCONNECTED	NOTIFICATION	CP notifies that connection with server
			is lost
106	CP_STOPPED	NOTIFICATION	CP has been stopped
107	CP_INIT_FAILED	RESULT	CP notifies that there is an error
			on initialization
108	CP_RECORD_STARTED	NOTIFICATION	CP notifies that recording started and
			new file has been created. Call
			player.RecordGetFileName(1) to get
			name of file.
109	CP_RECORD_STOPPED	NOTIFICATION	CP notifies that recording has stopped
			and the file has been finished. Call
			player.RecordGetFileName(0) to get
			name of file.
110	CP_RECORD_CLOSED	NOTIFICATION	CP notifies that recording has closed.
111	CP_BUFFER_FILLED	NOTIFICATION	CP notifies about pre-buffering process
			has completed.
112	CP_ERROR_NODATA_TIMEO	NOTIFICATION	CP notifies that no data had came for
	UT		DataReceiveTimeout period.
201	VDP_STOPPED	NOTIFICATION	VDP has been stopped
202	VDP_INIT_FAILED	RESULT	VDP notifies that there is an error
			on initialization
300	VRP_STOPPED	NOTIFICATION	VRP has been stopped
301	VRP_INIT_FAILED	RESULT	VRP notifies that there is an error
202	AND VIEED CLIDE CE	NOTIFICATION	on initialization
302	VRP_NEED_SURFACE	NOTIFICATION	VRP notifies that it is going to allocate
205	VDD EIDCTED ANGE	NIOTIEIC A TION	surface
305	VRP_FIRSTFRAME	NOTIFICATION	VRP notifies that first frame is rendered
400	ADP_STOPPED	RESULT	ADP has been stopped
401	ADP_INIT_FAILED	RESULT	ADP notifies that there is an error
			on initialization
500	ARP_STOPPED	NOTIFICATION	ARP has been stopped
501	ARP_INIT_FAILED	NOTIFICATION	ARP notifies that there is an error
			on initialization
503	ARP_VOLUME_DETECTED	NOTIFICATION	ARP notifies that volume detector has
			estimated audio volume values.
503	ARP_VOLUME_DETECTED	NOTIFICATION	finished and app can get min and max

# 3.3 State diagram



Application registers single **callback** function via **Open (config, callback)** call. State diagram separates notifications into 3 groups:

- onOpenError(). Occurs when error has happened in Open() function.
- onPlaybackError(). Occurs when error has happened in one of Paused/Started/Stopped states.
- onPlaybackCompleted(). Occurs in Started state only when end-of-stream has reached.

In case **onOpenError()** the closing procedure is processed automatically, i.e. MediaPlayer goes to **Closed** state.

In case onPlaybackError() / CP\_ERROR\_DISCONNECTED / CP\_ERROR\_NODATA\_TIMEOUT notification has received the closing procedure is not processed automatically, pipeline state is not changed, further playback goes on automatically when network connection has been restored (CP\_CONNECT\_SUCCESSFUL received).

**onPlaybackResumed()** / **CP\_CONNECT\_SUCCESSFUL** occurs after successful restoring of network connection, playback continues **automatically** pipeline state is not changed.

**seekTo()** is processed by either **setStreamPosition()** or **setLivePosition()** in Started or Paused states. In case if result of setStreamPosition() or setLivePosition() is equal to 0, the notification **PLP\_SEEK\_COMPLETED** will be.

In case **onPlaybackCompleted()/PLP\_EOS** happened, the state of pipeline is not changed.

**CP\_BUFFER\_FILLED** notification received in Paused state indication that prebuffering has finished. getStreamPrebufferTime() function returns the time position of pre-buffered data. **seekTo()** up to the pre-buffered position doesn't require network connection.

# 3.4 Functions description

Following functions are members of MediaPlayer class. These functions should be used to playback network content and media files.

#### <u>Open</u>

Connect to network server or open media file, create pipeline and playback media data.

Definition
public void Open(
final String ConnectionUrl,
final int DataReceiveTimeout,
final MediaPlayerCallback callback)

public void Open( final String ConnectionUrl,

final int ConnectionNetworkProtocol,

final int ConnectionDetectionTime,

final int ConnectionBufferingTime,

final int DecoderType,

final int RendererType,

final int SynchroEnable,

final int SynchroNeedDropVideoFrames,

final int EnableColorVideo,

final int EnableAspectRatio,

final int DataReceiveTimeout,

final int NumberOfCPUCores,

final MediaPlayerCallback callback)

public void Open(final MediaPlayerConfig config, final MediaPlayerCallback callback)

Parameters:

ConnectionUrl URL to network resource (RTSP, HTTP, RTMP, HLS,

UDP and so on) or full path for media file

ConnectionNetworkProtoc

Network protocol or RTP/RTSP tunneling (0 – RTP by UDP, 1 – RTP by TCP, 2 – RTSP over http, 3 – RTSP

over https, -1 - AUTO)

ol

ConnectionDetectionTime Probing time to detect video and audio formats of

network stream (in milliseconds)

ConnectionBufferingTime Buffering on playback start to avoid network jitter (in

milliseconds)

DecoderType Select s/w or h/w video decoder RendererType Select SDL or openGL render

SynchroEnable Enable A/V synchronization, 1 - synchronization is on,

0 - is off

SynchroNeedDropVideoFr

ames

Drop video framer if frame is late, 1 - is on, 0 - is off

EnableColorVideo Enable grayscaled video

EnableAspectRatio Set video output mode (0 - stretch, 1 - fit to screen with

aspect ratio, 2 - crop, 3 - 100% size, 4 - zoom mode, 5 -

move mode)

DataReceiveTimeout Reconnect timeout, SDK does reconnect if there is no data

received during some time (milliseconds).

MediaPlayerCallback Notification callback, event is provided over this callback NumberOfCPUCores Number of CPU cores to decode video,  $\leq 0$  – autodetect

and set the number according device capability, positive

number sets number according application needs

#### Return Value

Upon successful completion **Open**() returns 0. Otherwise -1 is returned. All errors are provided in callback status.

#### Remarks

Connect to network resource or open local media file, create pipeline, allocate resource and start video playback.

### Examples

# Example #1

```
player.Open(
```

"http://example", // correct URL or full path for media file

2, // RTSP over http tunneling

500, // 500 ms on probing

500, // 500 ms buffer on start

0, // Decoder type: 0-S/W 1, - H/W

1, // Renderer Type : 0 - SDL, 1 - pure OpenGL

1, // A/V synchronization: 1- Enable , 0 - Disable

```
0,
                      // Drop Video frame if it is late : 1- Enable , 0 - Disable
              1,
                      // Color / Grayscale video output : 0 - grayscale, 1 - color
               1,
                      // Aspect ratio / Full size : 1 – aspect rate
               30000, // Reconnection timeout (milliseconds),
              0,
                      // Number Of Cpu Cores for decoding (1-6), 0 - autodetect
              This);
              Example #2
// Create config
MediaPlayerConfig conf = new MediaPlayerConfig();
conf.setConnectionUrl(<a href="http://example">http://example</a>); // correct URL or full path to media file
conf.setConnectionNetworkProtocol(2); // RTSP over http tunneling
conf.setConnectionDetectionTime(500); // Probing time – 500 ms
conf.setConnectionBufferingTime(500); // Buffering on start – 500 ms
conf.setDecodingType(1); // H/W decoder
conf.setRendererType(1); // pure OpenGL
conf.setSynchroEnable(1); // Audio and Video synchronization is ON
conf.setSynchroNeedDropVideoFrames(0); // Do not drop video if pts is later
conf.setEnableColorVideo(1); // Set color video
conf.setEnableAspectRatio(1); // Set aspect ratio
conf.setDataReceiveTimeout(30000); // Set timeout of connection, Disconnect event is
sent after(in milliseconds)
conf.setNumberOfCPUCores(0); // Number Of Cpu Cores for decoding (1-6), 0-
autodetect
conf.setStartPreroll(1); //Start player in Paused mode
//Recording options
Int record_flags =
       PlayerRecordFlags.PP_RECORD_AUTO_START | //auto start open
       PlayerRecordFlags.PP_RECORD_DISABLE_AUDIO | //video only
       PlayerRecordFlags.PP_RECORD_SPLIT_BY_TIME | //split by time
       PlayerRecordFlags.PP_RECORD_SPLIT_BY_SIZE; //split by size
conf.setRecordFlags(record_flags);
conf.setRecordSplitTime(30); //split by 30 sec
conf.setRecordSplitSize(20); //split by 20 megabytes
```

```
conf.setRecordPath("/sdcard/DCIM");
conf.setRecordPrefix("my_rec");

player.Open(conf, This);
```

All configuration parameters are described in the table below:

Name	Description	Values	Default value	Туре
setColorBackground	Set/Get Background color		Color.BLAC K	Int
setEnableAspectRatio setAspectRatioMode	Set/Get Video output modes	0 - stretch, 1 - fit to screen with aspect ratio 2 - crop by height 21 - crop by width, 3 - 100% size 4 - zoom mode 5 - move mode	1	Int
set A spect Ratio Zoom Mode Percent	Zoom value if video output mode is "Zoom mode"	25-300	100	Int
setAspectRatioMoveModeX setAspectRatioMoveModeY	Set position to top and left for video output if video output mode is "Move mode"	-500 - 500	-1,-1 – Center of the screen	Int
setStartOffest	Set start offset for HLS stream, real position is last segment – offset	Depends on the stream, in milliseconds	0x80000000 0000000L	long
setEnableAudio	Mute, unmute audio speaker	0-1	1	int
setSslKey	Set rtmp_token for RTMP TL			String
setExtStream	Set stream number for HLS stream with various channels	Depends on the stream	0	Int
setSelectedAudio	Select audio track on start if there is more than one	Depends on the stream or file	0 – first track	Int

	track in file or stream			
setSelectedSubtitle	Set subtitle track on stream opening	Depends on the stream or file, -1 - disabled	0 – first track	Int
setConnectionUrl	Set stream URL or file path			String
setDecodingType	Set video decoder type	0 - software, 1 - hardware	1	Int
setDecoderLatency	Control minimal latency on s/w decoder, This setting is for s/w decoder	1 - Low latency, frames are not buffered in decoder, 0 - frames are buffered in video decoder	0	Int
setRendererType	Obsolete parameter			
setSynchroEnable	Enable audio & video synchronization	0 - Disable synchronization 1 - Enable syncronization	1	Int
set Synchro Need Drop Video Frames	Drop video frames if they are late	0 - drop frames 1 - render frames	0	Int
setDropOnFastPlayback	Obsolete parameter			
setEnableColorVideo	Obsolete parameter			
setNumberOfCPUCores	Set number of CPU for video decoding	Depends on the system ≤ 0 – autodetection	1	Int
setConnectionNetworkProtocol	Select preferred transport protocol for RTSP	0 - udp, 1 - tcp, 2 - http, 3 - https, -1 - AUTO	-1	Int
setConnectionDetectionTime	Time on start for detection video and audio formats (in milliseconds)	100-10000	5000	Int
setConnectionBufferingType	Set buffering type 0 – by time 1 – by size	0 or 1	0	Int
setConnectionBufferingTime	Buffer size on start in milliseconds	0-25000	1000	Int
setConnectionBufferingSize	Buffer size on start in bytes	0 - Max buffer size	0	Int
setDataReceiveTimeout	Set max timeout Interrupt source if		60000	Int

	data is not received in defined time			
setConnectionTimeout	Interrupt source if connection is not passed in defined timeout		60000	Int
setStartPreroll	Enable Pause on start	0 - start immediately 1 - start - play 1 frame - pause	0	Int
setStartCookies	Set cookie in HTTP request			String
setFadeOnStart	Fade audio on stream start	0 - audio comes straight off 1 - audio is faded ~200ms	1	Int
setFadeOnSeek	Fade audio on change position	0 - audio comes straight off 1 - audio is faded ~200ms	1	Int
setFFRate	Fade audio on change rate	0 - audio comes straight off 1 - audio is faded ~200ms	1	Int
setVolumeDetectMaxSamples	Number of samples to detect middle volume		0	Int
setVolumeBoost	Set volume boost	0 - off min:-30dB, max:+30dB	0	Int
setFadeOnChangeFFSpeed	Fade audio on change speed	0 - audio comes straight off 1 - audio is faded ~200ms	1	Int
setRecordPath	Set path for recorded files			String
setRecordFlags	Set flag setting for recording	PP_RECORD_NO_ST ART(0x00000000) PP_RECORD_AUTO_ START(0x000000001) PP_RECORD_SPLIT_B Y_TIME(0x00000002) PP_RECORD_SPLIT_B Y_SIZE(0x00000004), PP_RECORD_DISABL E_VIDEO(0x00000008) PP_RECORD_DISABL E_AUDIO(0x00000010)	0	Int

setRecordSplitTime	Split stream on chunks by time if flags are PP_RECORD_SP LIT_BY_TIME, in seconds	0-100	0	Int
setRecordSplitSize	Split stream on chunks by size if flags are PP_RECORD_SP LIT_BY_ SIZE, in seconds		0	Int
setRecordPrefix	Prefix is added to name of recorded files			String
setRecordTrimPosStart	Start position for trim from file, in milliseconds			Int
setRecordTrimPosEnd	Stop position for trim file, in milliseconds			Int
setEnableABR	Set adaptive bitrate control, experimental version			Int

#### **OpenAsPreview**

Connect to network server or open media file, create pipeline and playback media data in Preview mode. Preview mode differs from normal: s/w decoding only key frames, real time render, no audio stream (only video).

#### Definition

public void <a href="OpenAsPreview">OpenAsPreview</a>(

final String ConnectionUrl, final int DataReceiveTimeout, final MediaPlayerCallback callback)

**Parameters** 

ConnectionUrl URL to network resource (RTSP, HTTP, RTMP, UPD) or full

path for media file

DataReceiveTimeout Reconnect timeout, SDK does reconnect if there is no received

data during some time (milliseconds)

MediaPlayerCallback Notification callback, event is provided over this callback

Return Value

Upon successful completion **OpenAsPreview**() returns 0. Otherwise -1 is returned. All errors are provided in callback status.

### Remarks

Connect to network resource or open local media file, create pipeline, allocate resource and start playback in Preview mode.

# Examples

```
player.OpenAsPreview(
```

"http://example", // correct URL or full path for media file 30000, // Connection timeout (milliseconds),
This);

### <u>Plav</u>

Resume play if player is in Pause state.

Definition

public void Play()

**Parameters** 

There are no parameters for this call

#### Return Value

Upon successful completion, **Play()** returns 0. Otherwise -1 is returned. All errors are provided in callback status.

Remarks

Resume play if player is in Pause state. This function can be used with playback from files only.

Examples

player.Play();

# **Pause**

Change playback state from Play to Pause.

Definition

public void Pause()

**Parameters** 

There are no parameters for this call

Return Value

Upon successful completion, Pause() returns 0. Otherwise -1 is returned. All errors

are provided in callback status. Remarks Pause playback if player is in Play state. This function can be used with playback from file only. Examples player.Pause (); <u>getState</u> Return player state. Definition public PlayerState getState() **Parameters** There are no parameters for this call Return Value Following states are provided: 0 - Opening 1 - Opened 2 - Started 3 - Paused 4 - Stopped 5 - Closing 6 - Closed Remarks Provide the current state of player. Examples

if (player.getState() == PlayerState.Closing);

#### **getStreamDuration**

Return duration of media file in milliseconds. This function works only in case file playback.

Definition

public long getStreamDuration()

**Parameters** 

There are no parameters for this call.

Return Value

Upon successful completion, getStreamDurarion() returns file duration in seconds.

Otherwise -1 is returned. All errors are provided in callback status.

Remarks

Provides duration of the file played.

Examples

int duration = getStreamDuration();

# **getStreamPosition**

Get position in played media file. This function works only in case of file playback.

Definition

public long getStreamPosition()

Parameters

There are no parameters for this call.

Return Value

Upon successful completion, getStreamPosition() returns current position of played file (in milliseconds).

Remarks

Provides the played file position.

# Examples

int position = getStreamPosition();

### **setStreamPosition**

Set position of played media file. This function works only in case of file playback.

### Definition

public void setStreamPosition (final long lTime)

#### **Parameters**

lTime - new position in file (in milliseconds)

#### Return Value

Integer value is returned. 0 - on successful completion, PLP\_SEEK\_COMPLETED notification will send on success. Otherwise – error result.

#### Remarks

Provides the position of played file.

### Examples

long position;

setStreamPosition(position);

# **getLiveStreamPosition**

Function provides current, first, and last positions for live stream. This function works only in case of live stream playback (HLS).

#### Definition

Position getLiveStreamPosition()

#### **Parameters**

There are no parameters for this call.

#### Return Value

```
Upon successful completion, getLiveStreamPosition returns Position object.
public class Position
       private long first = 0;
       private long current = 0;
       private long last = 0;
       private long duration = 0;
}
first
               - dts of first segment in m3u8 list.
last
               - dts of last segment in m3u8 list.
               - dts of last downloaded packet in HLS stream.
current
Time base is milliseconds.
Remarks
Provide the current, first, last positions in played stream.
Examples
Position pos
                      player.getLiveStreamPosition();
if (pos == null)
       long duration = pos.getDuration();
                      = pos.getFirst();
       long first
       long current = pos.getCurrent();
       long last
                      = pos.getLast();
}
```

# <u>getStreamInfo</u>

Function returns the Meta data of media file or network stream.

```
Definition
public String getStreamInfo()
Parameters
There are no parameters for this call.
Return Value
Type – String, Format - xml.
Remarks
Provide the information about media file or network stream.
<?<u>xml</u> version="1.0"?>
<StreamInfoname="MediaPlayerSDK" version="1.0">
      <Metadata>
             <SHOUTCastMetadata>
             string1
             string2
             </SHOUTCastMetadata>
             <ContextMetadata>
             string
             </ContextMetadata>
             <ContextChapterMetadataid="0">
             string
             </ContextChapterMetadata>
```

```
<ContextChapterMetadataid="N">
      string
      </ContextChapterMetadata>
      <ContextProgramMetadataid="0">
      string
      </ContextProgramMetadata>
      <ContextProgramMetadataid="N">
      string
      </ContextProgramMetadata>
      <StreamMetadataid="0" type="AV_MEDIA_TYPE">
      string
      </StreamMetadata>
      <StreamMetadataid="N" type="AV_MEDIA_TYPE">
      string
      </StreamMetadata>
</Metadata>
<Stream name="stream name" duration="value in ms">
      <VideoStreams>
             <VideoStream id="0">
                   <title value="name"/>
                                value="AV_VIDEO_FORMAT_TYPE" />
                   <duration value="in ms"/>
                   <width value=""/>
                   <height
                                value=""/>
                   <fps value=""/>
             </VideoStream>
             <VideoStream id="N">
                   <title value="name"/>
                                value="AV_VIDEO_FORMAT_TYPE" />
                   <format
```

```
<duration value="in ms" />
             <width value=""/>
             <height
                          value=""/>
             <fps value=""/>
      </VideoStream>
</VideoStreams>
<AudioStreams>
      <AudioStream id="0">
             <title value="name"/>
             <format value="AV_AUDIO_FORMAT_TYPE"/>
             <duration value="in ms"/>
             <samplerate value="in Hz"/>
             <channels value="number of channels" />
      </AudioStream>
      <AudioStream id="N">
             <title value="name"/>
             <format value="AV_AUDIO_FORMAT_TYPE"/>
             <duration value="in ms"/>
             <samplerate value="in Hz"/>
             <channels value="number of channels"/>
      </AudioStream>
</AudioStreams>
<SubtitleStreams>
      <SubtitleStream id="0">
             <title value="name"/>
                          value="AV_SUBTITLE_FORMAT_TYPE" />
             <format
      </SubtitleStream>
      <SubtitleStream id="N">
             <title value="name"/>
                          value="AV_SUBTITLE_FORMAT_TYPE" />
             <format
      </SubtitleStream>
```

```
</SubtitleStreams>
       </Stream>
</StreamInfo>
All possible string values for tag: SHOUTCastMetadata
More
        info:http://www.indexcom.com/streaming/player/SHOUTcast.html
All possible string values for tags:
ContextMetadata, ContextChapterMetadata, ContextProgramMetadata,
StreamMetadata:
album -- name of the set this work belongs to
album_artist -- main creator of the set/album, if different from artist.
         e.g. "Various Artists" for compilation albums.
         -- main creator of the work
artist
comment
             -- any additional description of the file.
composer
            -- who composed the work, if different from artist.
            -- name of copyright holder.
creation_time-- date when the file was created, preferably in ISO 8601.
         -- date when the work was created, preferably in ISO 8601.
date
disc
         -- number of a subset, e.g. disc in a multi-disc collection.
            -- name/settings of the software/hardware that produced the file.
encoder
encoded_by -- person/group who created the file.
            -- original name of the file.
filename
          -- <self-evident>.
genre
language -- main language in which the work is performed, preferably in
         ISO 639-2 format. Multiple languages can be specified by
         separating them with commas.
            -- artist who performed the work, if different from artist.
performer
         E.g for "Also sprach Zarathustra", artist would be "Richard
         Strauss" and performer "London Philharmonic Orchestra".
           -- name of the label/publisher.
publisher
                 -- name of the service in broadcasting (channel name).
service name
service_provider -- name of the service provider in broadcasting.
title
         -- name of the work.
track
          -- number of this work in the set, can be in form current/total.
```

variant\_bitrate -- the total bitrate of the bitrate variant that the current stream is part of

### Examples

String info = player.getStreamInfo();

### <u>setLiveStreamPosition</u>

Change position of played live stream. This function works only in case of live stream.

### Definition

public void setLiveStreamPosition(final long lTime)

#### **Parameters**

lTime - new position in live stream (milliseconds)

# Return Value

Integer value is returned. 0 - on successful completion, PLP\_SEEK\_COMPLETED notification will send on success. Otherwise – error result.

#### Remarks

Change the position of life stream played.

# Examples

setStreamPosition(1000000);

# <u>qetStreamPrebufferTime</u>

Get pre-buffering position in played media file. This function works only in case of file playback.

#### Definition

public long getStreamPrebufferTime()

# Parameters

There are no parameters for this call.

# Return Value

Upon successful completion, getStreamPrebufferTime() returns pre-buffered position of played file/stream (in milliseconds).

# Remarks

Provide the pre-buffered position that is played by player.

# Examples

int position = getStreamPrebufferTime ();

# setFFRate

Change speed of playback for local file and network stream.

### Definition

public void setFFRate(final int rate)

# **Parameters**

rate - rate value ().

Correct values:

Rate	Value
x0.1	100 – Min Value
x0.2	200
x0.5	500
x0.9	900
x1	1000
x2	2000
x3	3000
x4	4000
x16	16000 – Max Value

### Return Value

No value is returned by function setFFRate.

#### Remarks

Change speed of playback for local file and network stream.

Important note: Some data is skipped if rate is less or more than normal playback rate.

# Examples

setFFRate(2000); // Set playback rate to x2

### **aetRenderPosition**

Function provides last position in played media file. This function works only in case of file playback.

#### Definition

public long getRenderPosition()

#### **Parameters**

There are no parameters for this call.

#### Return Value

Upon successful completion, getStreamPosition() returns PTS of last video frame or audio sample (milliseconds).

#### Remarks

Provide the PTS of last played video frame or audio sample.

# Examples

long position = getRenderPosition();

#### Close

Disconnect from server and destroy pipeline.

### Definition

public void Close()

#### **Parameters**

There are no parameters for this call

#### Return Value

Upon successful completion, **Close()** returns 0. Otherwise -1 is returned. All errors are provided in callback status.

#### Remarks

Disconnect from network server, destroy pipeline, free all resources that were allocated on Open() call.

### Examples

player.Close();

# UpdateView

Set video output mode for current player instance.

# Definition

public int UpdateView(final boolean isAspectRatioEnabled)
public int UpdateView()

# Parameters

isAspectRatioEnabled – set aspect ratio that is set in network stream, 1 – set aspect ratio that is set in network stream, 0 – resize picture on full screen.

# Return Value

Upon successful completion, isAspectRatioEnabled() returns 0, otherwise -1 is returned. All errors are provided in callback status.

#### Remarks

UpdateView(1) sets aspect ratio or full screen mode. This function can be used during playback. UpdateView() function uses settings that are set in player config structure.

```
Video output mode of output picture player.getConfig().setAspectRatioMode(VideoOutputMode); VideoOutputMode can be: 0 – stretch
```

1 – fit to screen with aspect ratio

2 – crop video

3 – 100% size of pciture

4 – zoom mode

Zoom multiplier of output picture (in percent, 25-400%) is set in player config: player.getConfig().setAspectRatioZoomModePercent(ZoomMultiplier);

5 – move mode

X and Y positions are set in player config:

X position of output picture (in percent, 0-100%)

player.getConfig().setAspectRatioMoveModeX(X);

Y position of output picture (in percent, 0-100%)

player.getConfig().setAspectRatioMoveModeY(Y);

// zoom and move modes are experimental functions, they may cause some issues.

# Examples Example #1

player.UpdateView (0);

# Example #2

// Present video: picture size is 100% in the center of screen player.getConfig().setAspectRatioMoveModeX(50); // 50% center of screen player.getConfig().setAspectRatioMoveModeY(50); // 50% center of screen player.getConfig().setAspectRatioZoomModePercent(100);//size is 100% player.getConfig().setAspectRatioMode(5); // Zoom and move mode player. UpdateView();

# background Color

Set background color of player.

### Definition

public void backgroundColor(final int clr)

#### **Parameters**

clr – color in RGB format (ARGB is not supported).

#### Return Value

Upon successful completion, backgroundColor() returns 0, otherwise -1 is returned. All errors are provided in callback status.

#### Remarks

Set background color of player.

# Examples

backgroundColor(Color.BLACK);

### setVisibility

Set the enabled state of this view

#### Definition

public void setVisibility(int visibility)

#### **Parameters**

visibility – Controls the initial visibility of the view. Value of parameters are described in Android documentation.

http://developer.android.com/reference/android/view/View.html#attr\_android:visibility

# Return Value

No value is returned by function setStreamPosition.

# Examples

player. setVisibility (1);

### **qetVideoShot**

Capture video picture from video stream.

```
Definition
public VideoShot getVideoShot(
final int desiredWidth,
final int desiredHeight
)

Parameters
desiredWidth - width of returned picture
desiredHeight - height of returned picture

Return Value

Upon successful completion, getVideoShot() returns VideoShot object.
public class VideoShot
{
    public int getWidth();
    public int getHeight();
    public ByteBuffer getData();
}
```

#### Remarks

Provide the video shot of last render frame in format ARGB\_8888. This function works in Preview mode only.

### Example

```
VideoShot vs = player.getVideoShot(width, height);
Bitmap bm = Bitmap.createBitmap(width, height, Bitmap.Config.ARGB_8888);
bm.copyPixelsFromBuffer(vs.getData());
```

# **GetStatFPS**

Return frame rate of downloaded stream so application can estimate if network bandwidth is enough for defined stream.

#### Definition

public int GetStatFPS()

#### **Parameters**

There are no parameters for this call.

#### Return Value

Upon successful completion, **GetStatFPS()** returns fps of network stream. It is frame rate of stream that is downloaded from network, otherwise -1 is returned. All errors are provided in callback status

#### Remarks

Provide the frame rate of captured stream (download speed) to estimate if network speed is enough to playback stream in real time.

#### Example

Int fps = player.GetStatFPS();

#### **GetStatBitrate**

Return bitrate of downloaded stream so application can estimate if network bandwidth is enough for defined stream.

#### Definition

public int GetStatBitrate ()

#### **Parameters**

There are no parameters for this call.

#### Return Value

Upon successful completion, **GetStatBitrate()** returns bitrate of network stream. It is the bitrate of a stream that is downloaded from network, otherwise -1 is returned. All errors are provided in callback status

# Remarks

Provide the bitrate of captured stream (download speed) to estimate if network speed is enough to playback stream in real time.

# Example

Int bitrate = player.GetStatBitrate();

# **GetDroppedFrame**

Return number dropped frames on render.

Definition

public int GetDroppedFrame ()

**Parameters** 

There are no parameters for this call.

Return Value

Upon successful completion, **GetDroppedFrame()** returns number of dropped frame in render provider.

# Example

Int bitrate = player.GetDroppedFrame();

## <u>getInternalBuffersState</u>

Return fullness of inner buffers in pipeline so application. Detail information are provided for every buffer in video and audio pipeline.

## Definition

public BuffersState getInternalBuffersState ()

**Parameters** 

There are no parameters for this call

# Return Value

Upon successful completion, **getInternalBuffersState()** returns buffers fullness and size with detail information about every buffer, otherwise null is returned. All errors are provided in callback status

#### Remarks

 $int\ getBufferSizeBetweenSourceAndVideoDecoder()-Size\ of\ buffer\ between\ Source\ and\ Video\ decoder$ 

int getBufferFilledSizeBetweenSourceAndVideoDecoder() – Fullness of buffer between Source and Video decoder

 $int\ getBufferSizeBetweenSourceAndAudioDecoder()-Size\ of\ buffer\ between\ Source\ and\ Audio\ decoder$ 

 $int\ get Buffer Filled Size Between Source And Audio Decoder ()-Fullness\ of\ buffer\ between\ Source\ and\ Audio\ decoder$ 

int getBufferSizeBetweenVideoDecoderAndVideoRenderer() – Size of buffer between Video decoder and Render

int getBufferFilledSizeBetweenVideoDecoderAndVideoRenderer() – Fullness of buffer between Video decoder and Render decoder

 $int\ get Buffer Size Between Audio Decoder And Audio Renderer() - Size\ of\ buffer\ between\ Audio\ decoder\ and\ Render$ 

int getBufferFilledSizeBetweenAudioDecoderAndAudioRenderer() – Fullness of buffer between Audio decoder and Render decoder

Example: BuffersState

buf\_state;

buf\_state = player.getInternalBuffersState();

if (buf\_state.getBufferSizeBetweenAudioDecoderAndAudioRenderer() != 0 && buf\_state.getBufferSizeBetweenAudioDecoderAndAudioRenderer()!=0)

Log.e("TEST", "buf\_level: Source: " + buf\_state.getBufferFilledSizeBetweenSourceAndAudioDecoder()\*100/buf\_state.getBuff

erSizeBetweenSourceAndAudioDecoder() + " Render: "

buf\_state.getBufferFilledSizeBetweenAudioDecoderAndAudioRenderer()\*100/buf\_state.getBufferSizeBetweenAudioDecoderAndAudioRenderer());

### **GetStatPercFree**

Return fullness of inner buffers in pipeline so application can estimate if device can playback data in real time or with latency (Video data only).

Definition public int GetStatPercFree()

**Parameters** 

There are no parameters for this call

### Return Value

Upon successful completion, **GetStatPerFree()** returns level of capacity for inner buffers, otherwise -1 is returned. All errors are provided in callback status

#### Remarks

Return fullness of inner buffers in pipeline so application can estimate if device can playback data in real time or latency.

# Example

Int buf\_level = player.GetStatPerFree();

# **GetDataDelavOnSource**

Return delay in milliseconds if input stream comes with some delay in case network bottleneck.

### Definition

public int GetDataDelaySource ()

## **Parameters**

There are no parameters for this call

# Return Value

Upon successful completion, **GetDataDelayOnSource()** returns the difference between when package is expected and when package comes

#### Remarks

### Example

int delay = player.GetDataDelayOnSource ();

# **IsHardwareDecoding**

Return what decoder (s/w or h/w) is used by player.

Definition

public boolean IsHardwareDecoding ()

**Parameters** 

There are no parameters for this call.

Return Value

Upon successful completion, IsHardwareDecoding returns true if h/w decoder is used and false for s/w decoder

Remarks

Shows if h/w or s/w video decoder is used in player.

Example

Boolean hw\_decoder = IsHardwareDecoding ();

# **RecordStart**

Start recording.

Definition

public void RecordStart();

**Parameters** 

There are no parameters for this call

# RecordStop

Stop recording.

Definition

public void RecordStop();

**Parameters** 

There are no parameters for this call

# **RecordGetFileName**

Retrieve the name of file has been recording.

Definition

public String RecordGetFileName(int param)

**Parameters** 

int param; //0: get last stopped file. 1: get last started file.

### Remarks

Notifications CP\_RECORD\_STARTED and CP\_RECORD\_STOPPED are received when recording activities took place. In order to ensure what is latest file name has been recorded we'd better call RecordGetFileName(0) at CP\_RECORD\_STOPPED event, and RecordGetFileName(1) at CP\_RECORD\_STARTED event happen.

# **AudioGetCount**

Retrieve a number of audio tracks.

Definition

public int AudioGetCount()

#### **Parameters**

none

#### Return Value

Returns a number of available audio tracks.

#### Remarks

AudioGetCount() retrieves a number of audio tracks. It can be used when player is in opened state only (PlayerState.Opened) or after notification PlayerNotifyCodes. PLP\_BUILD\_SUCCESSFUL.

# **AudioSelect**

Select an audio track to play.

## Definition

public int AudioGetCount(int track\_i)

## **Parameters**

track\_i – the number of selected track, the value must be in the range of value has been returned by AudioGetCount().

## Remarks

AudioSelect(track\_i) can be used also before opening of the player. track\_i value is saved automatically into MediaPlayerConfig.setSelectedAudio(). This track will actually used after player.Open().

### **AudioGetSelected**

Retrieve a selected audio track.

### Definition

public int AudioGetSelected()

### **Parameters**

none

Return Value

Returns selected audio track.

# Remarks

AudioGetSelected() returns actually selected audio tack. In case player is not Opened state, AudioGetSelected() returns MediaPlayerConfig.getSelectedAudio() value.

# **startVolumeDetect**

Starts audio volume detection process

Definition

public void startVolumeDetect(final int vd\_max\_samples)

**Parameters** 

int vd\_max\_samples - a number of audio samples has to be processed

Return Value

void

#### Remarks

When startVolumeDetect() has finished ARP\_VOLUME\_DETECTED sent. Application can retrieve maximum and mean estimated audio volumes through getting the properties getPropInt(PlayerProperties.PP\_PROPERTY\_AUDIO\_VOLUME\_MAX) and getPropInt(PlayerProperties.PP\_PROPERTY\_AUDIO\_VOLUME\_MEAN).

#### setVolumeBoost

Gain audio volume

### Definition

public void setVolumeBoost(final int volume\_boost)

#### **Parameters**

int volume\_boost - value of audio volume gain in range [-30, 30] dB

Return Value

void

# Remarks

Application should first retrieve volume statistics through startVolumeDetect() and max\_volume

getPropInt(PlayerProperties.PP\_PROPERTY\_AUDIO\_VOLUME\_MAX);

mean\_volume =

setPropInt(PlayerProperties.PP\_PROPERTY\_AUDIO\_VOLUME\_MEAN). After all, set corresponded value **volume\_boost** to avoid audio clipping in raising audio volume.

For instance app obtains mean\_volume=-27, max\_volume=-4.

It means that:

The mean square energy is approximately -27 dB.

The largest sample is at -4 dB, or more precisely between -4 dB and -5 dB.

In other words, raising the volume by +4 dB (volume\_boost=4) does not cause any clipping.

## **SubtitleGetCount**

Retrieves a number of subtitle tracks.

Definition

public int SubtitleGetCount()

**Parameters** 

none

Return Value

Returns a number of available subtitle tracks.

### Remarks

SubtitleGetCount() retrieves a number of subtitle tracks. It can be used when player is in opened state only (PlayerState.Opened) or after notification PlayerNotifyCodes. PLP\_BUILD\_SUCCESSFUL.

# **SubtitleSelect**

Select a subtitle track to play.

Definition

public int SubtitleSelect(int track\_i)

# **Parameters**

track\_i – the number of selected track, the value must be in the range of value has been returned by SubtitleGetCount().

## Remarks

SubtitleSelect(track\_i) can be used also before opening of the player. track\_i value is saved automatically into MediaPlayerConfig.setSelectedSubtitle(). This track will actually used after player.Open().

# **SubtitleGetSelected**

Retrieves a selected subtitle track.

Definition

public int SubtitleGetSelected()

**Parameters** 

none

Return Value

Returns selected subtitle track.

Remarks

SubtitleGetSelected() returns actually selected subtitle track. In case player is not in Opened state, SubtitleGetSelected() returns MediaPlayerConfig.getSelectedSubtitle() value.

### **SubtitleSourceAdd**

Add an external subtitle source

Definition

public int SubtitleSourceAdd(String path2)

**Parameters** 

String path2 – path to subtitle source

Return Value

0 - OK. Otherwise error.

Remarks

SubtitleSourceAdd(String path2) adds a path to subtitle source. Application can set up multiple external subtitle sources. After adding subtitle source, the player will increase the count of subtitle tracks (SubtitleGetSelected()) and select required track by SubtitleSelect() call. SubtitleSourceAdd() can be called in both before and after Open() call. Also these paths to external sources can be added through MediaPlayerConfig. subtitlePaths string list.

## **SubtitleSourceRemove**

Removes an external subtitle source has been added by SubtitleSourceAdd() function.

Definition

public int SubtitleSourceRemove(String path2)

Parameters

String path2 – path to subtitle source

Return Value

0 - OK. Otherwise error.

# 4. Thumbnailer

Thumbnailer is Class that provides the functionality to make thumbnails and stream information for local files and network streams. Smart searching is used to make thumbnails with maximum informativity.

# 4.1 Functions description

Following functions are member of Thumbnailer class. These functions should be used to get a thumbnail for file or network stream.

### **Open**

Connect to network server or open local media file.

Definition

public Object Open(final String ConnectionUrl)
public Object Open(final ThumbnailerConfig config)

**Parameters** 

ConnectionUrl URL of network resource (RTSP, HLS, RTMP, MMS, UDP

and so on) or full path of local media file.

Class ThumbnailerConfig provides additional setting to open Thunbnailer.

public ThumbnailerConfig( String connectionUrl,

int connectionNetworkProtocol,

int dataReceiveTimeout, int numberOfCPUCores,

float bogoMIPS)

connectionUrl URL of network resource (RTSP, HLS, RTMP, MMS,

UDP and so on) or full path of local media file

connectionNetworkProtocol Protocol for RTP or RTSP tunneling, 0 – RTP by UDP, 1

- RTP by TCP, 2 - RTSP and HTTP tunneling, -1 -

AUTO mode

dataReceiveTimeout reconnect timeout, SDK does reconnect if there is no

received data during some time (milliseconds)

numberOfCPUCores

Number of CPU core to decode video, 0 – autodetect and set the number according device capability, positive number sets number according application needs

#### Return Value

Upon successful completion **Open**() returns 0. Otherwise - ERROR is returned.

### Remarks

Connect to network resource or open local media file, create pipeline, allocate resource. This function should be called before get Frame.

```
Example #1 thumbnailer.Open("http://example.com"); Example #2 thumbnailer.Open(ThumbnailerConfig);
```

# **getFrame**

Capture thumbnail frame.

### Definition

public ThumbnailFrame getFrame()

### **Parameters**

There are no parameters for this call.

# Return Value

```
Upon successful completion, getFrame returns ThumbnailFrame object.
public class ThumbnailFrame
{
    public int getWidth();
    public int getHeight();
    public ByteBuffer getData();
```

#### Remarks

}

Provide the thumbnail for local file or stream in format ARGB\_8888.

```
Example
```

```
ThumbnailFrame frame = thumbnailer.getFrame();
shot.getData().rewind();
Bitmap bmp = Bitmap.createBitmap(shot.getWidth(), shot.getHeight(),
Bitmap.Config.ARGB_8888);
bmp.copyPixelsFromBuffer(shot.getData());
```

### aetInfo

Function returns the information about media file or network stream.

```
Definition
```

public String getInfo()

## **Parameters**

There are no parameters for this call.

```
Return Value
```

```
Type – String, Format - xml.
```

#### Remarks

Provide the information about media file or network stream.

String is xml format like below:

```
<?xml version=1.0?>
<StreamInfoname="AVFileFormat" version="1.0">
             value="test.mp4"/>
<name
             value="100"/>
<duration
      <VideoStreams>
              <VideoStream id=0>
                    <format
                                  value="h264"/>
                                  value="100"
                    <duration
                    <width
                                  value="1920"/>
                    <height
                                  value="1080"/>
```

<fps

value="30"

```
</VideoStream>
       </VideoStreams>
       <AudioStreams>
              "<AudioStream id=1>
                     <format
                                   value="aac"
                                                  />
                                   value="100"
                                                  />
                     <duration
                     <samplerate
                                   value="48000"/>
                                   value="2"
                     <channels
                                                  />
              </AudioStream>
       </AudioStreams>
</StreamInfo>
Examples
String info = thumbnailer.getInfo();
GetState
Return thumbnailer state: .
Definition
public ThumbnailerState getState()
Parameters
There are no parameters for this call
Return Value
Following states are provided:
       0 - Opening,
       1 - Opened,
       2 - Closing,
       3 - Closed;
Remarks
Provide the current state of Thumbnailer.
Example
if (thumbnailer.getState() == ThumbnailerState.Opened);
```

# Close

Disconnect from server or close file and destroy all resources.

Definition

public void Close()

Parameters

There are no parameters for this call

Return Value

Upon successful completion, **Close()** returns 0. Otherwise – ERROR is returned.

Remarks

Disconnect from network server, destroy pipeline, free all resources that were allocated on Open() call.

Example

thumbnailer.Close();