

# Player

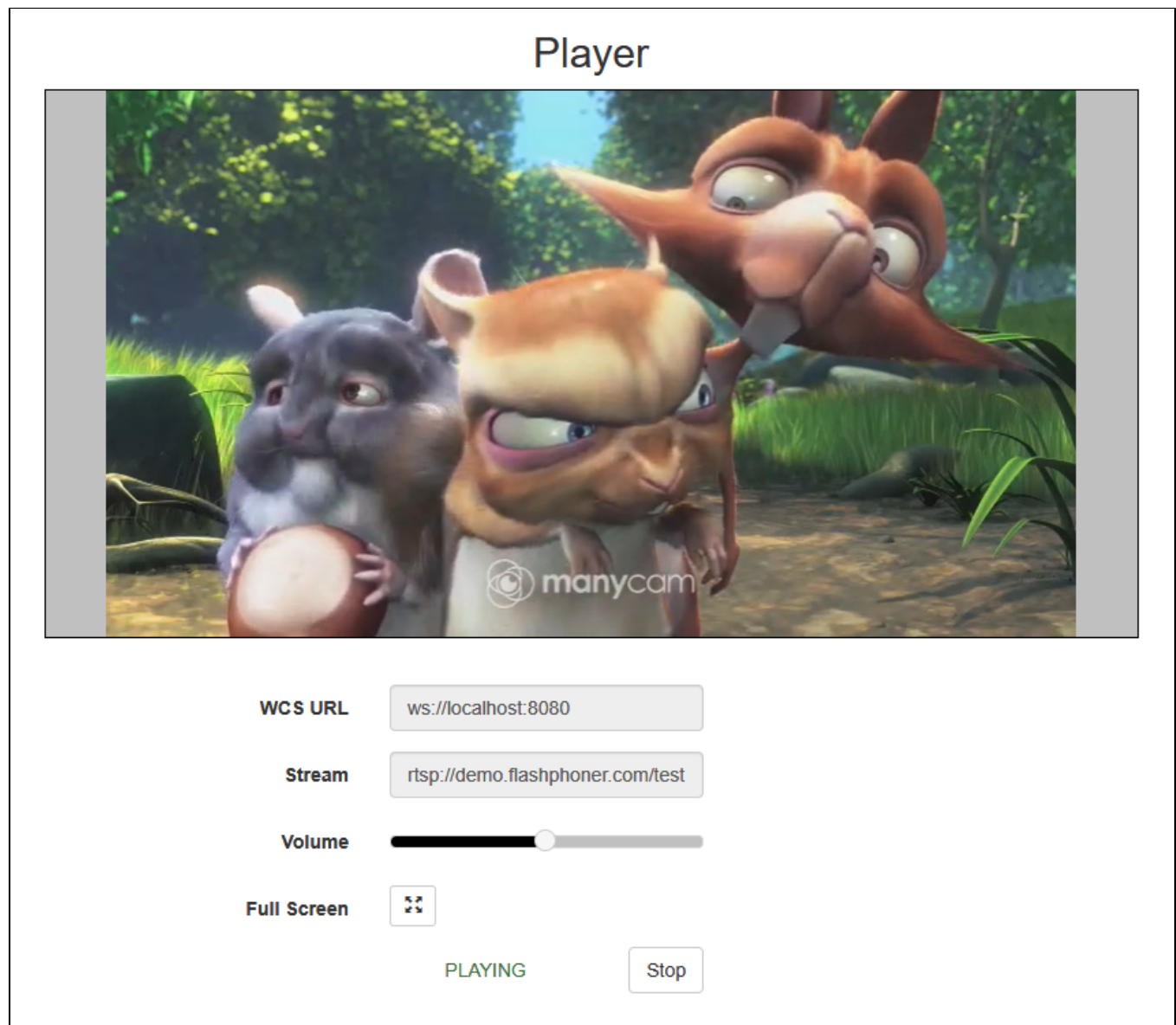
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## Player example

This player can be used to play any type of stream on Web Call Server

- RTSP
- WebRTC
- RTMP

On the screenshot below an RTSP stream is being playing in the web player.



## Code of the example

The path to the source code of the example on WCS server is:

/usr/local/FlashphonerWebCallServer/client2/examples/demo/streaming/player

player.css - file with styles  
player.html - page of the player  
player.js - script providing functionality for the player

This example can be tested using the following address:

<https://host:8888/client2/examples/demo/streaming/player/player.html>

Here host is the address of the WCS server.

## Analyzing the code

To analyze the code, let's take the version of file player.js with hash 7fff01f, which is available [here](#) and can be downloaded with corresponding build [2.0.219](#).

### 1. Initialization of the API

Flashphoner.init() [code](#)

```
Flashphoner.init({
  receiverLocation: '../dependencies/websocket-player/WSReceiver2.js',
  decoderLocation: '../dependencies/websocket-player/video-worker2.js',
  preferredMediaProvider: mediaProvider
});
```

### 2. Connection to server

Flashphoner.createSession() [code](#)

```
Flashphoner.createSession({urlServer: url}).on(SESSION_STATUS.ESTABLISHED, function(session){
  ...
}).on(SESSION_STATUS.DISCONNECTED, function(){
  ...
}).on(SESSION_STATUS.FAILED, function(){
  ...
});
```

### 3. Receiving the event confirming successful connection

SESSION\_STATUS.ESTABLISHED [code](#)

```
Flashphoner.createSession({urlServer: url}).on(SESSION_STATUS.ESTABLISHED, function(session){
  setStatus(session.status());
  //session connected, start playback
  playStream(session);
}).on(SESSION_STATUS.DISCONNECTED, function(){
  ...
}).on(SESSION_STATUS.FAILED, function(){
  ...
});
```

### 4. Playback of video stream

Session.createStream(), Stream.play() [code](#)

```
stream = session.createStream(options).on(STREAM_STATUS.PENDING, function (stream) {
  ...
});
stream.play();
```

When stream is created, the following parameters are passed

- streamName - name of the stream (rtsp:// address in this case)
- remoteVideo - div element, in which the video will be displayed
- picture resolution to transcode
- unmutePlayOnStart: false - disables automatic audio unmuting for autoplay to conform browsers requirements

[code](#)

```
var options = {
  name: streamName,
  display: remoteVideo,
  flashShowFullScreenButton: true
};
if (Flashphoner.getMediaProviders()[0] === "MSE" && mseCutByIFrameOnly) {
  options.mediaConnectionConstraints = {
    cutByIFrameOnly: mseCutByIFrameOnly
  }
}
if (resolution_for_wsplayer) {
  options.playWidth = resolution_for_wsplayer.playWidth;
  options.playHeight = resolution_for_wsplayer.playHeight;
} else if (resolution) {
  options.playWidth = resolution.split("x")[0];
  options.playHeight = resolution.split("x")[1];
}
if (autoplay) {
  options.unmutePlayOnStart = false;
}
```

#### 5. Receiving the event confirming successful stream playback

STREAM\_STATUS.PLAYING [code](#)

```
stream = session.createStream(options).on(STREAM_STATUS.PENDING, function (stream) {
  ...
}).on(STREAM_STATUS.PLAYING, function (stream) {
  $("#preloader").hide();
  setStatus(stream.status());
  onStarted(stream);
}).on(STREAM_STATUS.STOPPED, function() {
  ...
}).on(STREAM_STATUS.FAILED, function() {
  ...
});
stream.play();
```

#### 6. Handling the event about channel bandwidth

STREAM\_EVENT, STREAM\_EVENT\_TYPE.NOT\_ENOUGH\_BANDWIDTH [code](#)

```
}).on(STREAM_EVENT, function(streamEvent){
  if (STREAM_EVENT_TYPE.NOT_ENOUGH_BANDWIDTH === streamEvent.type) {
    var info = streamEvent.payload.info.split("/");
    var remoteBitrate = info[0];
    var networkBandwidth = info[1];
    console.log("Not enough bandwidth, consider using lower video resolution or bitrate. Bandwidth " +
      (Math.round(networkBandwidth / 1000)) + " bitrate " + (Math.round(remoteBitrate / 1000)));
  } else if (STREAM_EVENT_TYPE.RESIZE === streamEvent.type) {
    ...
  }
});
```

#### 7. Handling the event about changing stream picture size

STREAM\_EVENT, STREAM\_EVENT\_TYPE.RESIZE [code](#)

```

    }).on(STREAM_EVENT, function(streamEvent){
        ...
    } else if (STREAM_EVENT_TYPE.RESIZE === streamEvent.type) {
        console.log("New video size: " + streamEvent.payload.streamerVideoWidth + "x" + streamEvent.payload.streamerVideoHeight);
    }
    });

```

## 8. Playback stop

Stream.stop() [code](#)

```

function onStarted(stream) {
    $("#playBtn").text("Stop").off('click').click(function(){
        $(this).prop('disabled', true);
        stream.stop();
    }).prop('disabled', false);
    ...
}

```

## 9. Receiving the event confirming successful stream playback stop

STREAM\_STATUS.STOPPED [code](#)

```

stream = session.createStream(options).on(STREAM_STATUS.PENDING, function (stream) {
    ...
}).on(STREAM_STATUS.STOPPED, function () {
    $("#preloader").hide();
    setStatus(STREAM_STATUS.STOPPED);
    onStopped();
}).on(STREAM_STATUS.FAILED, function() {
    ...
}).play();

```

## 10. Playback volume control

Stream.unmuteRemoteAudio(), Stream.setVolume() [code](#)

```

$("#volumeControl").slider({
    range: "min",
    min: 0,
    max: 100,
    value: currentVolumeValue,
    step: 10,
    animate: true,
    slide: function(event, ui) {
        //WCS-2375. fixed autoplay in ios safari
        if (stream.isRemoteAudioMuted()) {
            stream.unmuteRemoteAudio();
        }
        currentVolumeValue = ui.value;
        stream.setVolume(currentVolumeValue);
    }
}).slider("disable");

```

## 11. Playback automatic start on page load

[code](#)

```

if (autoplay) {
    // We can start autoplay with muted audio only, so set volume slider to 0 #WCS-2425
    $("#volumeControl").slider('value', 0);
    $("#playBtn").click();
}

```

