

Flash Streaming

Example of a streaming player in a native Flash / Flex application

The following example shows how to play a video stream while simultaneously publishing another stream by using a client Flash application that can be run by a simple swf file. Streaming in this example can work via two protocols: `rtmp://` and `rtmfp://`

The screenshot displays an example Flash application that sends a video stream to the server and plays the video from the server via the RTMFP protocol.



The `Server` field contains the RTMFP address of the server to establish connection. The `Publish` field contains the name of the video stream to send to the server from the web camera. The `Play` field contains the name of the video stream to play from the server. Below you can specify additional parameters of capturing and sending a video stream:

- width and height of a frame
- fps
- quality
- key frames per second
- is audio or video present in the video stream being sent

Example files

This example is a compiled SWF file embedded to an HTML page using Flex / ActionScript3 and MXML and accessible at:

/usr/local/FlashphonerWebCallServer/client2/examples/demo/streaming/flash_client/streaming.html

- streaming.html - example page
- streaming/bin/streaming.swf - application file

Working with the source code of the example

To analyze the code, let's take this version of the `streaming.mxml` file with the hash `90eb5073687bbe63bbb7467de3f3be4f3fe33802` located [here](#). The result of compiling the `streaming.mxml` file is the example application `streaming.swf`. The compiled swf and the source code are available for download in the corresponding build [0.5.3.1894](#).

1. Accessing camera and microphone

Right after the application is loaded, we get access to the web camera and the microphone

[line 42](#)

[line 44](#)

```
cam = Camera.getCamera();
videoMy.attachCamera(cam);
mic = Microphone.getEnhancedMicrophone();
```

2. Applying camera and microphone settings

[line 76](#)

[line 84](#)

Recommended setting for the web camera:

- A motion sensibility threshold to send video: `cam.setMotionLevel(0,2000);`

Recommended setting for the microphone:

- Speex audio codec:mic.codec = SoundCodec.SPEEX;
- The number of frames per packet:mic.framesPerPacket=1;
- A sound threshold to send audio:mic.setSilenceLevel(0,2000);

```
private function initCam():void{
    cam.setMode(int(camWidth.text),int(camHeight.text),int(camFPS.text),true);
    cam.setQuality(0,int(camQuality.text));
    cam.setKeyFrameInterval(int(camKeyFrame.text));
    cam.setMotionLevel(0,2000);
    Logger.info("Cam initizlized "+cam.width+"x"+cam.height);
}

private function initMic():void{
    var options:MicrophoneEnhancedOptions = new MicrophoneEnhancedOptions();
    options.mode = MicrophoneEnhancedMode.FULL_DUPLEX;
    options.echoPath = 128;
    options.nonLinearProcessing = true;
    mic.codec = SoundCodec.SPEEX;
    mic.encodeQuality = 5;
    mic.framesPerPacket=1;
    mic.gain=50;
    mic.setSilenceLevel(0,2000);
    mic.enhancedOptions = options;
    Logger.info("Mic initialized");
}
```

3. Connecting to the server

[line 103](#)

Here we establish connection to the server and send `obj.appKey = "flashStreamingApp"`; This appKey tells the server, that it should deal with a Flash application, not with a WebSocket/WebRTC client

```
private function connect():void{
    trace("connect");
    var url:String = connectUrl.text;
    nc = new NetConnection();
    //if (url.indexOf("rtmp") == 0){
    nc.objectEncoding = ObjectEncoding.AMF0;
    //}
    nc.client = this;
    nc.addEventListener(NetStatusEvent.NET_STATUS, handleConnectionStatus);
    var obj:Object = new Object();
    obj.login = generateRandomString(20);
    obj.appKey = "flashStreamingApp";
    nc.connect(url,obj);
}
```

4. Stream publishing

Sending the stream to the server is performed in the `publish()` method of the example

[line 165](#)

```

if (publishAudio.selected){
    initMic();
    publishStream.attachAudio(mic);
    Logger.info("Init audio stream")
}
if (publishVideo.selected){
    initCam();
    publishStream.attachCamera(cam);
    addH264();
    Logger.info("Init video stream");
}
addListenerAndPublish

```

Directly before sending the stream, we set additional buffering parameters and parameters of the H.264 codec in `addH264()` and `addListenerAndPublish()` methods

[line 199](#)

[line 208](#)

```

private function addListenerAndPublish():void{
    publishStream.videoReliable=true;
    publishStream.audioReliable=false;
    publishStream.useHardwareDecoder=true;
    publishStream.addEventListener(NetStatusEvent.NET_STATUS, handleStreamStatus);
    publishStream.bufferTime=0;
    publishStream.publish(publishStreamName.text);
}

public function addH264():void{
    var videoStreamSettings:H264VideoStreamSettings = new
H264VideoStreamSettings();
    videoStreamSettings.setProfileLevel(H264Profile.MAIN,H264Level.LEVEL_3_1);
    publishStream.videoStreamSettings = videoStreamSettings;
}

```

5. Stream playback

Playing the stream starts after the `play()` method is invoked

[line 223](#)

```

private function play():void{
    if (playStreamName.text == "") {
        playStatus.text = "Empty stream name";
        playStatus.setStyle("color", "#ff0000");
        return;
    }
    playStatus.setStyle("color", "#000000");
    Logger.info("play");
    subscribeStream = new NetStream(nc);
    addListenerAndPlay();
}

```

All parameters and buffer sizes are set directly before playing in the `addListenerAndPlay()` method

line 244

```
private function addListenerAndPlay():void{
    subscribeStream.videoReliable=true;
    subscribeStream.audioReliable=false;
    subscribeStream.useHardwareDecoder=true;
    subscribeStream.addEventListener(NetStatusEvent.NET_STATUS,
handleSubscribeStreamStatus);
    subscribeStream.bufferTime=0;
    var soundTransform:SoundTransform = new SoundTransform();
    soundTransform.volume=0.7;
    subscribeStream.soundTransform = soundTransform;
    subscribeStreamObject = createStreamObject();
    subscribeStream.play(playStreamName.text);
    videoFarEnd.attachNetStream(subscribeStream);
    videoFarEnd.width = 320;
    videoFarEnd.height = 240;
    videoFarEnd.visible = true;
}
```