

Screen Sharing with Camera

Overview

The example shows a presentation case: webcam and screen sharing streams publishing from a single page with optional streams mixing at server side.

Screen sharing parameters:

- `FPS` - framerate per second
- `Width` - picture width
- `Height` - picture height

Server stream mixing parameters:

- `Add to mixer` - add streams automatically to mixer
- `Mixer` - mixer name

Connection parameters:

- Websocket URL of WCS sever

.

The code of the example

The code of the example is on WCS server by the following path:

```
/usr/local/FlashphonerWebCallServer/client2/examples/demo/streaming/screen-camera-mixer
```

- `screen-camera-mixer.css` - styles file
- `screen-camera-mixer.html` - example HTML page
- `screen-camera-mixer.js` - main example script

The example may be tested by the following address:

```
https://host:8888/client2/examples/demo/streaming/screen-camera-mixer/screen-camera-mixer.html
```

where `host` is WCS server address.

Analyzing the code

To analyze the code let's take a `screen-camera-mixer.js` version with hash 42b55d1, which is available [here](#) and may be downloaded in build 2.0.247.

1. API initialization.

`Flashphoner.init()` [code](#)

```
Flashphoner.init();
```

2. Connecting to the server

`Flashphoner.createSession()` [code](#)

```
const connect = function() {
  ...
  let url = field("url");
  ...
  console.log("Create new session with url " + url);
  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

`STREAM_STATUS.ESTABLISHED` [code](#)

```
const connect = function() {
  ...
  let url = field("url");
  ...
  console.log("Create new session with url " + url);
  Flashphoner.createSession({urlServer:
url}).on(SESSION_STATUS.ESTABLISHED, function(session){
  //session connected, start streaming
  setStatus("screen", SESSION_STATUS.ESTABLISHED);
  setStatus("camera", SESSION_STATUS.ESTABLISHED);
  onConnected(session);
  }).on(SESSION_STATUS.DISCONNECTED, function(){
  ...
  }).on(SESSION_STATUS.FAILED, function(){
  ...
  });
}
```

```
});  
}
```

4. Screen sharing stream publishing

`Session.createStream()`, `Stream.publish()` code

The following parameters are passed to `createStream()` method:

- `streamName` - stream name
- `constraints.video.width` - picture width
- `constraints.video.height` - picture height
- `constraints.video.frameRate` - publishing framerate
- `constraints.video.type: "screen"` - stream type: screen sharing
- `constraints.video.withoutExtension: true` - screen sharing from browser without a special extension
- `constraints.video.mediaSource: "screen"` - for screen sharing from Firefox browser only
- `localVideoScreen` - div tag to display a local video
- `disableConstraintsNormalization = true` - disable constraints normalization (for MacOS Safari only)

```
const startStreamingScreen = function(session) {  
  let streamName = getStreamName("screen", field("url"));  
  let constraints = {  
    video: {  
      width: parseInt($('#width').val()),  
      height: parseInt($('#height').val()),  
      frameRate: parseInt($('#fps').val()),  
      type: "screen",  
      withoutExtension: true  
    }  
  };  
  if (Browser.isFirefox()) {  
    constraints.video.mediaSource = "screen";  
  }  
  let options = {  
    name: streamName,  
    display: localVideoScreen,  
    constraints: constraints  
  }  
  if (isSafariMacOS()) {  
    options.disableConstraintsNormalization = true;  
  }  
  session.createStream(options).on(STREAM_STATUS.PUBLISHING,  
function(screenStream) {  
  ...  
}
```

```

    }).on(STREAM_STATUS.UNPUBLISHED, function() {
      ...
    }).on(STREAM_STATUS.FAILED, function(stream) {
      ...
    }).publish();
  }
}

```

5. Receiving the event confirming successful screen publishing

`STREAM_STATUS.PUBLISHING` code

Screen sharing stream WebRTC statistics collection and web camera stream publishing start on this event

```

const startStreamingScreen = function(session) {
  ...
  session.createStream(options).on(STREAM_STATUS.PUBLISHING,
function(screenStream) {
  /*
   * User can stop sharing screen capture using Chrome "stop" button.
   * Catch onended video track event and stop publishing.
   */
  document.getElementById(screenStream.id()).srcObject.getVideoTracks()
[0].onended = function (e) {
    screenStream.stop();
  };
  document.getElementById(screenStream.id()).addEventListener('resize',
function(event){
    resizeVideo(event.target);
  });
  setStatus("screen", STREAM_STATUS.PUBLISHING, screenStream);
  screenStats = StreamStats("screen", screenStream, STAT_INTERVAL);
  startStreamingCamera(session, screenStream);
}).on(STREAM_STATUS.UNPUBLISHED, function() {
  ...
}).on(STREAM_STATUS.FAILED, function(stream) {
  ...
}).publish();
}
}

```

6. Web camera stream publishing

`Session.createStream()`, `Stream.publish()` code

The following parameters are passed to `createStream()` method:

- `streamName` - stream name
- `localVideoCamera` - div tag to display a local video

```

const startStreamingCamera = function(session, screenStream) {
  let streamName = getStreamName("camera", field("url"));
  let options = {

```

```

        name: streamName,
        display: localVideoCamera
    }
    session.createStream(options).on(STREAM_STATUS.PUBLISHING,
function(cameraStream) {
    ...
    }).on(STREAM_STATUS.UNPUBLISHED, function() {
    ...
    }).on(STREAM_STATUS.FAILED, function(stream) {
    ...
    }).publish();
}

```

7. Receiving the event confirming successful camera publishing

`STREAM_STATUS.PUBLISHING` code

Web camera stream WebRTC statistics collection starts on this event

```

const startStreamingCamera = function(session, screenStream) {
    ...
    session.createStream(options).on(STREAM_STATUS.PUBLISHING,
function(cameraStream) {
    document.getElementById(cameraStream.id()).addEventListener('resize',
function(event){
        resizeVideo(event.target);
    });
    setStatus("camera", STREAM_STATUS.PUBLISHING, cameraStream);
    cameraStats = StreamStats("camera", cameraStream, STAT_INTERVAL);
    onStart(cameraStream);
    }).on(STREAM_STATUS.UNPUBLISHED, function() {
    ...
    }).on(STREAM_STATUS.FAILED, function(stream) {
    ...
    }).publish();
}

```

8. WebRTC statistics collection and displaying

`Stream.getStats()`, `stats.outboundStream.video.bytesSent`,
`stats.otherStats.availableOutgoingBitrate` code

The following parameters are displayed:

- current publishing bitrate
- maximum available bitrate for the stream estimated by browser

```

const StreamStats = function(type, stream, interval) {
    const streamStats = {
        type: type,
        timer: null,
        stream: stream,
    }

```

```

        bytesSent: 0,
        start: function(interval) {
            if (!streamStats.timer) {
                streamStats.timer = setInterval(streamStats.displayStats,
interval);
            }
        },
        stop: function() {
            if (streamStats.timer) {
                clearInterval(streamStats.timer);
                streamStats.timer = null;
            }
            setBitrate(streamStats.type, "");
            setAvailableBitrate(streamStats.type, "");
        },
        displayStats: function() {
            if (streamStats.stream) {
                streamStats.stream.getStats((stats) => {
                    if (stats) {
                        if (stats.outboundStream &&
stats.outboundStream.video) {
                            let vBitrate =
(stats.outboundStream.video.bytesSent - streamStats.bytesSent) * 8;
                            setBitrate(streamStats.type, vBitrate);
                            streamStats.bytesSent =
stats.outboundStream.video.bytesSent;
                        }
                        if (stats.otherStats &&
stats.otherStats.availableOutgoingBitrate !== undefined) {
                            setAvailableBitrate(streamStats.type,
stats.otherStats.availableOutgoingBitrate);
                        }
                    }
                });
            }
        }
    };
    streamStats.start(interval);
    return streamStats;
}

```

9. Stop camera publishing

`Stream.stop()` code

```

const setPublishButton = function(action, session, cameraStream) {
    $("#publishBtn").text(action).off('click').click(function(){
        if (action == "Start") {
            ...
        } else if (action === "Stop") {
            $(this).prop('disabled', true);
            cameraStream.stop();
        }
    }).prop('disabled', false);
}

```

10. Stop screen publishing

`Stream.stop()` [code](#)

```
const startStreamingCamera = function(session, screenStream) {
  ...
  session.createStream(options).on(STREAM_STATUS.PUBLISHING,
function(cameraStream) {
  ...
  }).on(STREAM_STATUS.UNPUBLISHED, function() {
    setStatus("camera", STREAM_STATUS.UNPUBLISHED);
    screenStream.stop();
  }).on(STREAM_STATUS.FAILED, function(stream) {
    ...
  }).publish();
}
```

11. Stop WebRTC statistics collection

[code](#)

```
const onStopped = function(session) {
  if (cameraStats) {
    cameraStats.stop();
  }
  if (screenStats) {
    screenStats.stop();
  }
  ...
}
```