# WebRTC: Коммуникации в реальном времени.

Обзор и демонстрация возможностей WebRTC - работа с видео и peer-to-peer соединения.
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#### WebRTC API's

#### MediaStream

```
navigator.getUserMedia(constraints, function (stream) {
    // handle stream
}, function (err) {
    // handle error
});
```

#### PeerConnection

```
var pc = new RTCPeerConnection();

// Signalling related API
pc.createOffer(function (offer) { /* process offer */ });
pc.createAnswer(function (answer) { /* process answer */ });
pc.setLocalDescription(description, successCallback, errorCallback);
pc.setRemoteDescription(description, successCallback, errorCallback);
pc.onicecandidate = function (event) { /* process candidate */ };

// Media API
pc.addStream(stream);
pc.onaddstream = function (stream) { /* process stream */ };
```

#### MediaStream Demo

Hello, World! (https://github.com/DavidKlassen/webrtc-demos/tree/master/hello-world/)

Canvas (https://github.com/DavidKlassen/webrtc-demos/tree/master/canvas/)

CSS (https://github.com/DavidKlassen/webrtc-demos/tree/master/css/)

#### PeerConnection и его протоколы

SDP (Session Description Protocol)

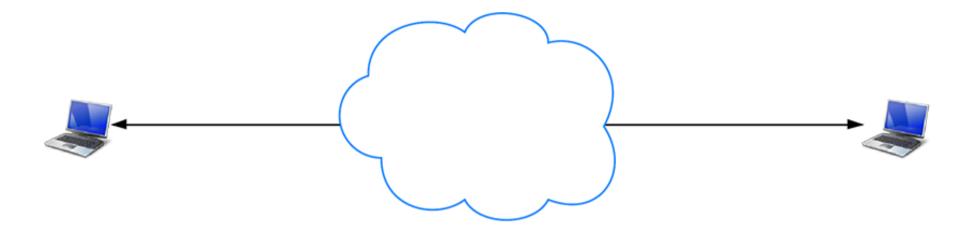
RTP (Real-time Transport Protocol)

RTCP (RTP Control Protocol)

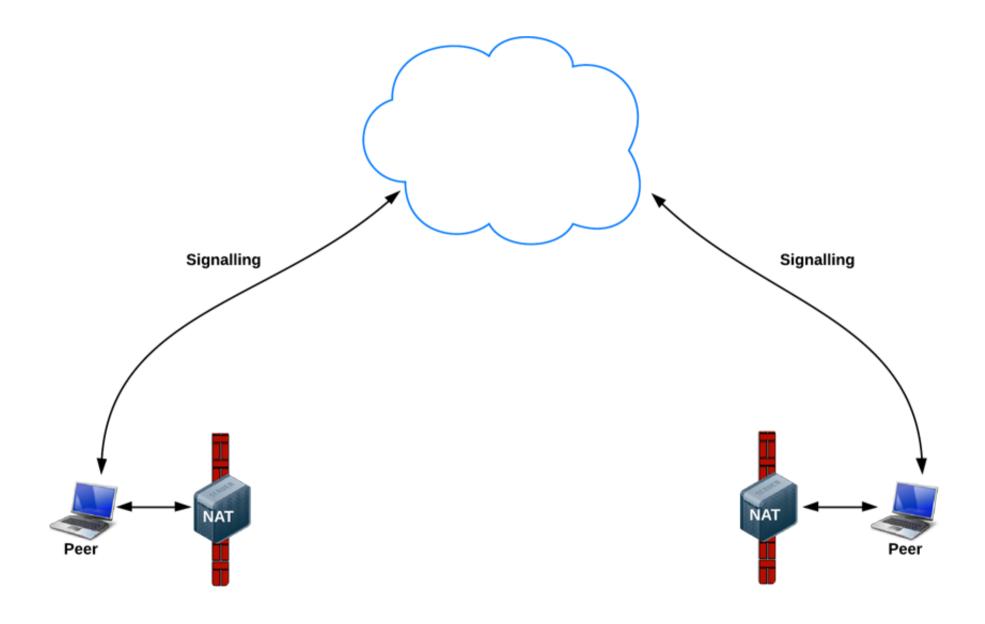
### **PeerConnection Demo**

Hello, World! (https://github.com/DavidKlassen/webrtc-demos/tree/master/peer-to-peer/)

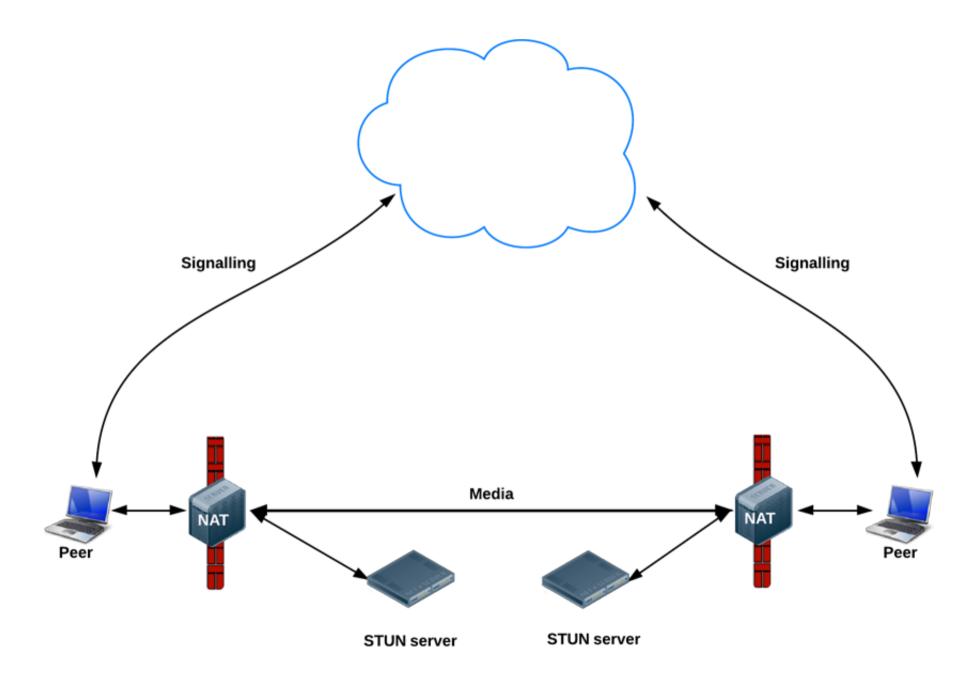
# Что такое peer-to-peer?



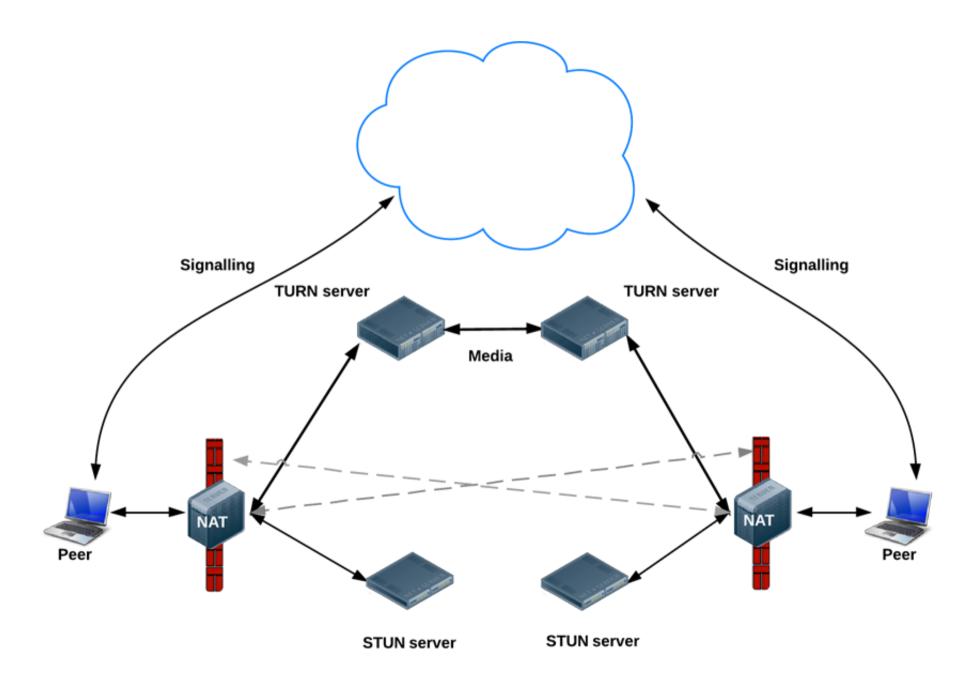
#### Реальность



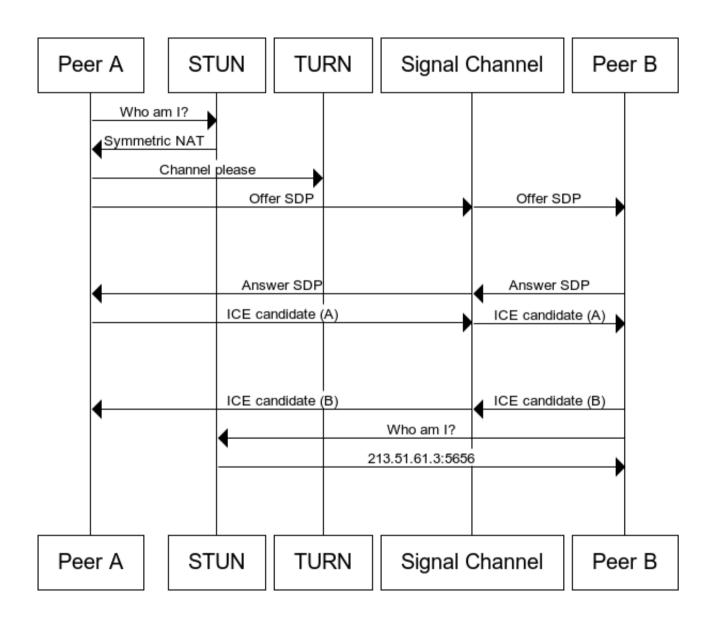
## STUN (Session Traversal Utilities for NAT)



# **TURN (Traversal Using Relays around NAT)**



## Signalling



# Потрогаем все это руками

Hello, World! (https://github.com/DavidKlassen/webrtc-demos/tree/master/signaling)

# Что еще?

DataChannel Media Servers (kurento)

# Thank you

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