

WebRTC 1.1 at 2015 ~~f2f~~TPAC?

First step after 1.0

Status of 1.0 (assuming lots of PRs go in)

- PeerConnection: has getSenders() and getReceivers()
- RtpSender
 - has settable parameters, active on/off, priority, maxBitrate, and payloadType
 - has read-only getCapabilities, parameters, codecs, header extensions, SSRCs,
 - has multiple encodings
- RtpReceiver: has readable parameters, same as RtpSender
- DtlsTransport: has state, certificates, all read-only
- IceTransport: has read-only state, selectedCandidatePair, role, component, gatheringState, local/remote parameters, local/remote candidates
- DataChannel: has SctpTransport
- SctpTransport: has DtlsTransport

All of the objects are there and read-only, with a little configuration possible. But we can't construct them or configure them fully.

Goal of first step after 1.0:

Construct and configure the WebRTC 1.0 objects without a PeerConnection and without SDP:

- IceTransport
- DtlsTransport
- RtpSender
- RtpReceiver
- SctpTransport
- DataChannel

Assumptions/Scope

- Don't need to support RTP/RTCP non-mux
- No initial need to support forking

IceTransport + DtlsTransport

[Constructor()]

```
partial interface RTCIceTransport {  
    void gather(RTCIceGatherOptions gatherOptions); // IceServers + IceTransportPolicy  
    void start(RTCIceParameters remoteParameters,  
              optional RTCIceRole role);  
    void addRemoteCandidate(RTCIceCandidate remoteCandidate);  
    void stop();  
    attribute EventHandler? onlocalcandidate;  
}
```

[Constructor(RTCIceTransport transport)]

```
partial interface RTCDtlsTransport {  
    void start(RTCDtlsParameters remoteParameters);  
    void stop();  
}
```

RtpSender + RtpReceiver

```
[Constructor(MediaStreamTrack track, RTCDtlsTransport transport)]  
partial interface RTCRtpSender {  
    void setTransport (RTCDtlsTransport transport);  
    void send (RTCRtpParameters parameters);  
    void stop ();  
};
```

```
[Constructor(RTCDtlsTransport transport)]  
partial interface RTCRtpReceiver {  
    void setTransport (RTCDtlsTransport transport);  
    void receive (RTCRtpParameters parameters);  
    void stop ();  
};
```

RTP Example

```
var ice = new IceTransport();
var dtls = new DtlsTransport(dtls);
var sender = new RtpSender(track, dtls);
var receiver = new RtpReceiver(dtls);
ice.gather({iceServers: [...]});
var rtpSendParameters = {codecs: [...], encodings: [...]};
// Signal out ice.getLocalParameters(), dtls.getLocalParameters(), rtpSendParameters
// Signal in remoteIceParameters, remoteDtlsParameters, rtpRecvParameters
ice.start(remoteIceParameters);
dtls.start(remoteDtlsParameters);
sender.send(rtpSendParameters);
receiver.receive(rtpReceiverParameters);
```

SctpTransport + DataChannel

```
[Constructor(RTCDtlsTransport transport)]
partial interface RTCSctpTransport : RTCDataTransport {
    static RTCSctpCapabilities getCapabilities ();
    void start (RTCSctpCapabilities remoteCaps);
    void stop ();
    attribute EventHandler ondatachannel;
}

[Constructor(RTCDataTransport transport, RTCDataChannelParameters parameters)]
partial interface RTCDataChannel {
}
```


Data Channel Example

```
var ice = new IceTransport();
var dtls = new DtlsTransport(dtls);
var sctp = new SctpTransport(dtls);
// Signal out ice.getLocalParameters(), dtls.getLocalParameters(), sctp.getCapabilities()
// Signal in remoteIceParameters, remoteDtlsParameters, remoteSctpCapabilities
ice.start(remoteIceParameters);
dtls.start(remoteDtlsParameters);
sctp.start(remoteSctpCapabilities);
sctp.ondatachannel = ...;
dc = new DataChannel(sctp, {label: "...", id: "..."});
dc.send("...");
```

Oh, and we need some more full dictionaries

```
dictionary RTCIceCandidate {
  DOMString          foundation;
  unsigned long      priority;
  DOMString          ip;
  RTCIceProtocol     protocol;
  unsigned short     port;
  RTCIceCandidateType type;
  DOMString          relatedAddress = "";
  unsigned short     relatedPort;
};
```

```
dictionary RTCRtpCodecCapability {
  // ...
  unsigned long      clockRate;
  unsigned long      channels;
  payloadtype        preferredPayloadType;
  unsigned long      maxptime;
  sequence<RTCRtcpFeedback> rtcpFeedback;
  Dictionary         parameters;
  unsigned short     maxTemporalLayers = 0;
  unsigned short     maxSpatialLayers = 0;
};
```

```
dictionary RTCRtpHeaderExtension {
  // ...
  unsigned short preferredId;
}
```

```
dictionary RTCRtpParameters {
  DOMString muxId = "";
  // ...
}
```

```
dictionary RTCRtpEncodingParameters {
  DOMString          encodingId;
  sequence<DOMString> dependencyEncodingIds;
};
```

```
dictionary RTCRtpFecParameters {
  // ...
  DOMString mechanism;
};
```

And few other things (needs examples)

```
Promise<RTCStatsReport> getStats(); // Goes on everything
```

```
attribute EventHandler onerror; // Goes on everything
```

```
[Constructor(RTCRtpSender sender)]  
interface RTCDtmfSender {}
```

```
[Constructor(RTCDtlsTransport transport)]  
interface RTCIdentity {  
    // ...  
    Promise<DOMString> getIdentityAssertion(DOMString provider, protocol, username);  
    Promise<RTCIdentityAssertion> setIdentityAssertion(DOMString assertion);  
};
```

```
// If you want proper ICE behavior with multiple IceTransports.
```

```
interface RTCIceTransportController {  
    void addTransport (RTCIceTransport transport, optional unsigned long index);  
};
```