# JSEP Update

Justin Uberti IETF 83

# Topics

- Activity since IETF 82
- Implementation Status
- Issues Raised

## **Activity since IETF 82**

- Long discussion at IETF 82.5
- Minor changes made to JSEP draft, to allow ROAP to be implemented atop JSEP
- draft-ietf-rtcweb-jsep-00 adopted as WG document; fixed several issues

### Implementation Status

- Chrome API renamed for JSEP
  - old API renamed webkitDeprecatedPeerConnection
  - new API named webkitJsep00PeerConnection
- Targeting Chrome 19 as first version to support JSEP PeerConnection
  - Preliminary (working!) builds available now
  - May show up on Canary this week
- ROAP-on-JSEP JS library written by Harald Alvestrand, open-sourced, works with Chrome JSEP builds

### **W3C Status**

- API from uberti-ietf-rtcweb-jsep-00 included by Cullen Jennings in <u>branch</u> of W3C WebRTC spec
- Adam Bergqvist published an alternate branched version that keeps more of the existing API
- Discussions between authors are ongoing; expecting to reach consensus

### Issues Raised

- Provisional and final answers
  - Control of media transmission
  - Resource management
  - Multiple 2xx answers
- ICE Candidates
  - Associating candidates with m= lines
  - Release of unneeded candidates
  - Restarts
- What state sequences are legal?
- What SDP can application change?

### **Provisional Answers**

- Provisional answers (SDP\_PRANSWER)
  - Originally meaning was to allow SDP to be returned without starting media
  - Turns out in some cases, media is desired
- Replaced by direction attribute in SDP
  - Callee that wants to warm ICE and DTLS without starting media can send a PRANSWER with a=inactive; "final" ANSWER is sendrecv
  - PRANSWER could be sendrecv, if desired

### PRANSWER Use Case 1

#### A places a call to B

#### OFFER

[...] m=audio 11111 RTP/AVPF 0 a=candidate:1 1 udp 1694498815 11.11.11.11 11111 typ host a=sendrecv

### B's browser immediately responds to get ICE warmed up, no media exchanged PRANSWER

[...] m=audio 22222 RTP/AVPF 0 a=candidate:1 1 udp 1694498815 22.22.22 22222 typ host a=inactive

#### B accepts call, media flows in both directions

#### **ANSWER**

[...] m=audio 22222 RTP/AVPF 0 a=candidate:1 1 udp 1694498815 22.22.22 22222 typ host a=sendrecv

### PRANSWER Use Case 2

#### 1-800-GO-FEDEX scenario

#### **OFFER**

[...] m=audio 11111 RTP/AVPF 0 a=candidate:1 1 udp 1694498815 11.11.11.11 11111 typ host a=sendrecy

### 180 with 2-way early media

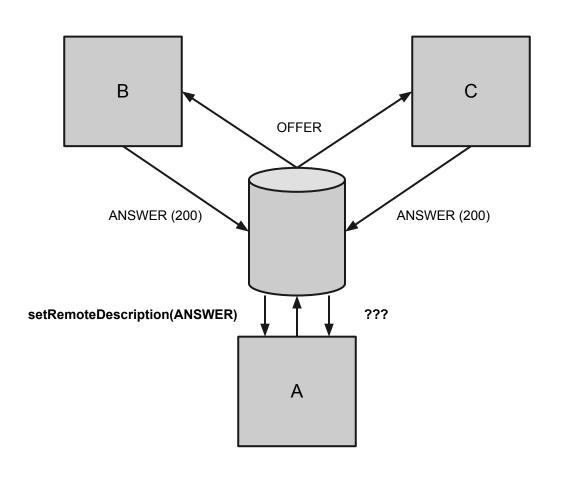
#### **PRANSWER**

[...] m=audio 22222 RTP/AVPF 0 a=candidate:1 1 udp 1694498815 22.22.22 22222 typ host a=sendrecy

### **PRANSWER Semantics**

- No implicit notion of media direction with PRANSWER
- Sole meaning is "non-final ANSWER"
- Allocated resources (e.g. codecs) are not released on PRANSWER
- Allocated resources are released on ANSWER

# Multiple 2xx Answers



### Multiple 2xx Answers

- Since resources are released on the first ANSWER, an ANSWER cannot directly follow an ANSWER
- Client can ACK + BYE subsequent ANSWERs (i.e. first-one-wins)
- Or can treat all ANSWERs as PRANSWER, with timer to ANSWER and clean up (i.e. last-one-wins)
- No mixing occurs

### **Associating Candidates**

```
interface IceCandidate {
   // the m= line index for this candidate
   readonly int mLineIndex;
   // the mid for the m= line for this candidate
   readonly DOMString mLineId;
   // creates a SDP-ized form of this candidate
   DOMString toSdp();
};
```

- JSEP candidates are currently identified by a string corresponding to SDP a=mid
  - optional in SDP
  - required for Jingle
- API change to allow either index or id string to be used

### Release of Candidates

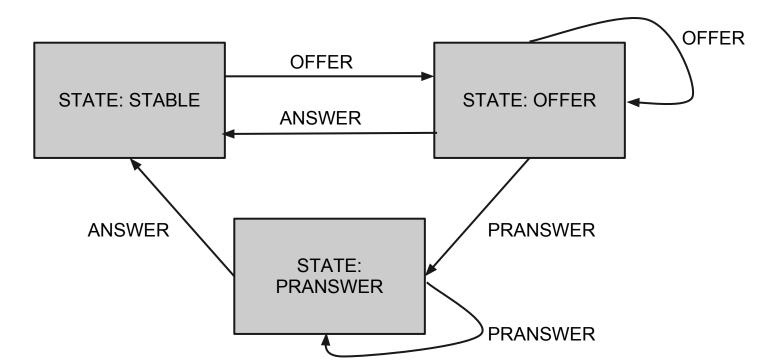
- JSEP allows adding candidates, separately from the offer/answer exchange, via processIceMessage (i.e. "trickle" candidates)
- ICE Agent determines optimal media path via RFC5245 nomination process
- "Worse" candidates are released after nomination completes
  - e.g. If STUN works, TURN candidates are released
  - local/peer-reflexive candidates kept for later use

### ICE Restarts

- Calls should survive network blips
  - ISP NAT resets
  - Call "rehydration" after reload
- Existing candidates can redo connectivity checks
- Or a full re-ICE can be performed by calling startice again
  - Requires new offer-answer to exchange new ICE credentials and candidates, but call survives

### **JSEP States**

- Clarity requested on state changes
  - Legal state transitions shown by arrows below
  - OFFER-OFFER can occur when adding streams



### **JSEP Attributes**

### Some attribs that the app may wish to tweak

- remove or reorder codecs (m=)
- change codec attributes (a=fmtp; ptime)
- enable/disable BUNDLE (a=group)
- enable/disable RTCP mux (a=rtcp-mux)
- reorder SRTP crypto-suites (a=crypto)
- change SRTP parameters or keys (a=crypto)
- change send resolution or framerate (TBD)
- change desired recv resolution or framerate (TBD)
- change total bandwidth (b=)
- remove desired AVPF mechanisms (a=rtcp-fb)
- remove RTP header extensions (a=rtphdr-ext)
- add/change SSRC grouping (e.g. FID, RTX, etc) (a=ssrc-group)
- add SSRC attributes (a=ssrc)
- change ICE ufrag/password (a=ice-ufrag/pwd)
- change media send/recv state (a=sendonly/recvonly/inactive)

## Questions