

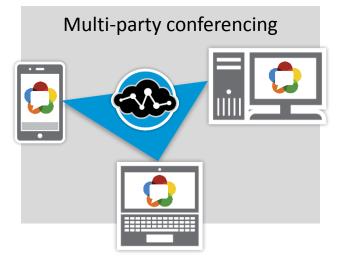
Dialogic.

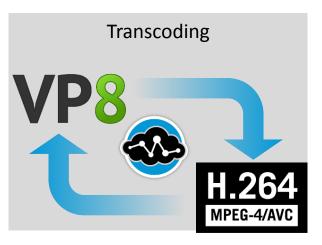
WebRTC Media in the Cloud

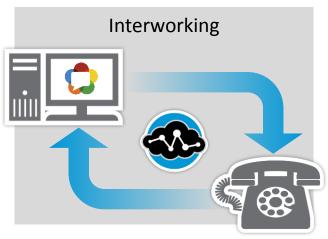
Chad Hart

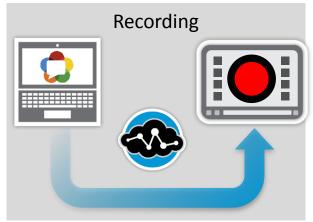
IIT RTC Conference 2014

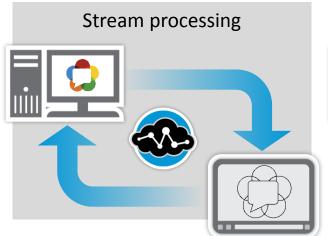
Why do you need a media server?

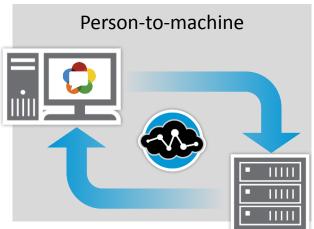












Client

Server

Client-side



Server-side

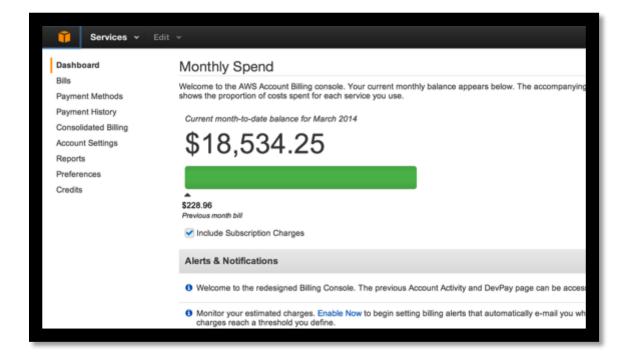
Network

- 5TB/month Outbound Public Bandwidth (with monthly billing)
- Unlimited Inbound Public Bandwidth
- Unlimited Private Network (Server-to-Server) Bandwidth
- 100Mbps Public and Private Network Ports (upgradable to 1Gt)
- Private VLAN via Public & Private Network
- Geographically Redundant DNS
- Dual-Stack IPv4 and IPv6 Capable

Client view

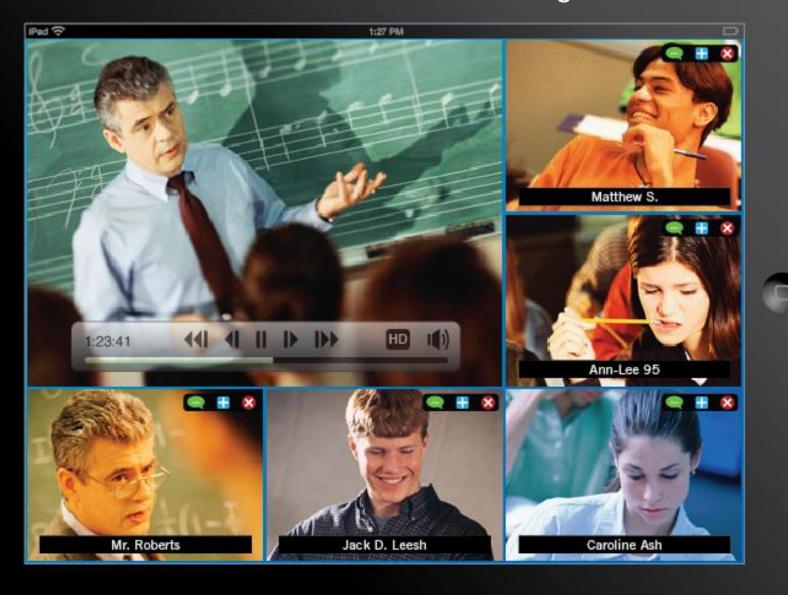


Provider View





Multi-PARTY! Video Conferencing

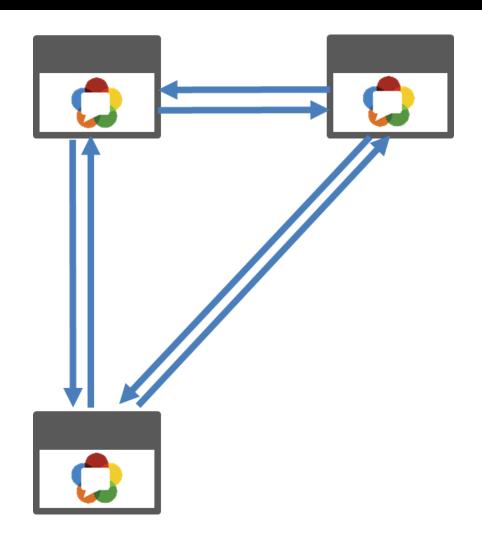


Easy & cheap approach to multi-party: Mesh

Dialogic

Full Mesh

- Works for a few parties
- No server cost
- Lowest latency

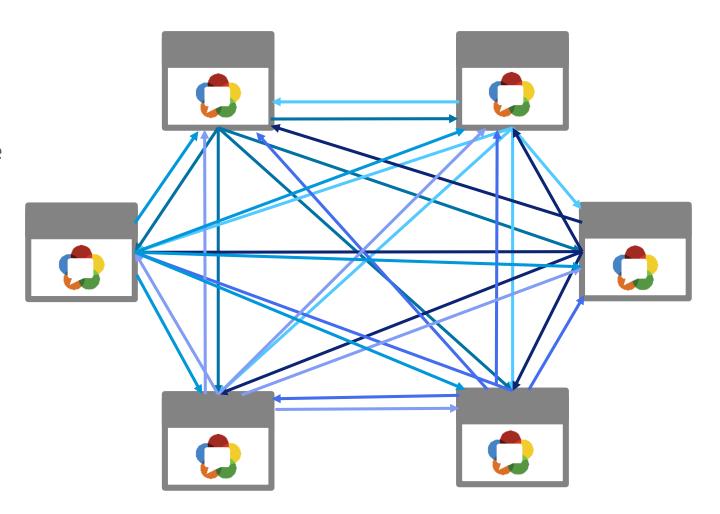


Mesh does not scale for many video conferees

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Full Mesh

- Clients get overloaded
 - Encode costs more than decode
 - Limited uplink bandwidth
- Inconsistent performance across participants

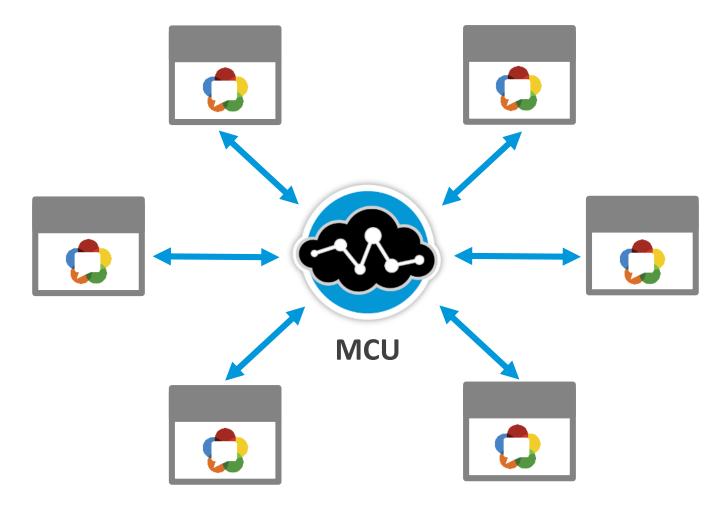


Traditional Telephony Approach: MCU

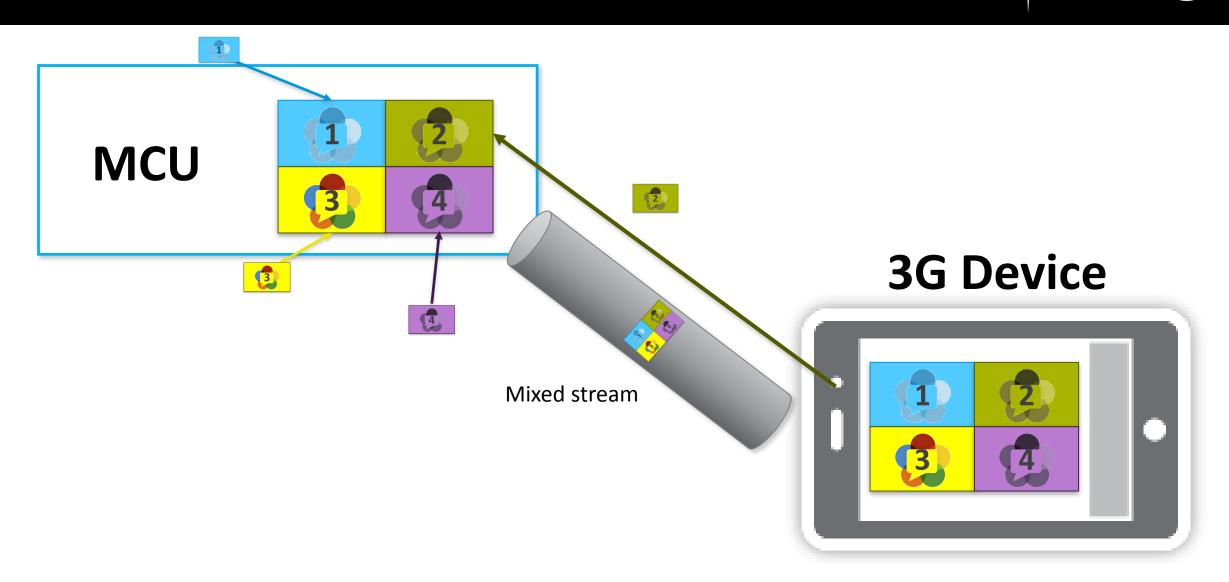
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Multipoint Control Unit (MCU)

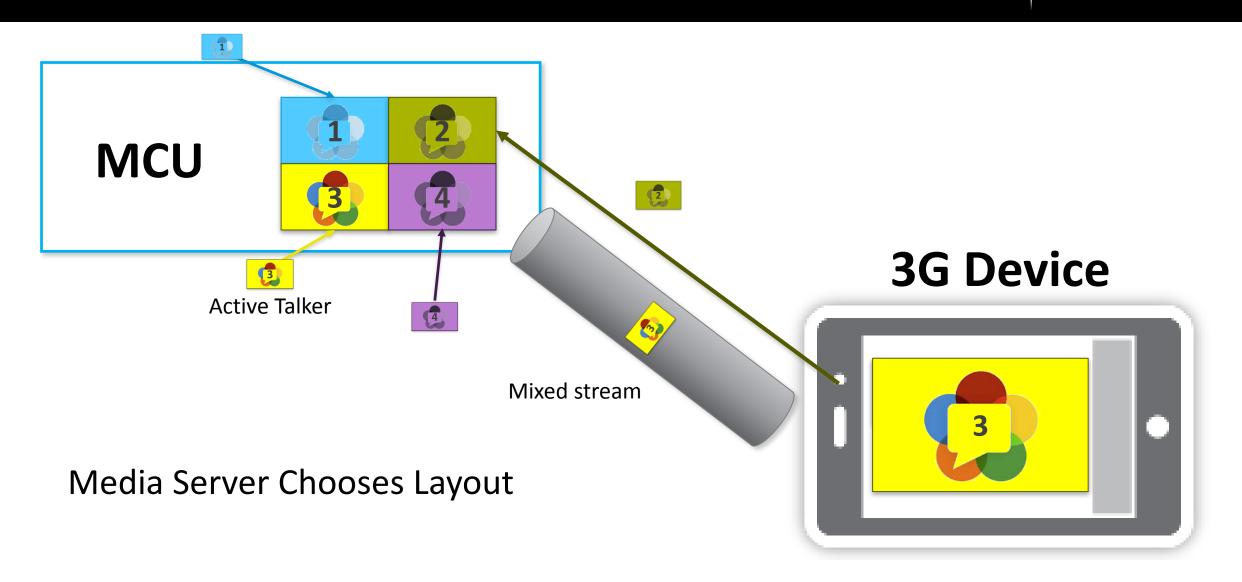
- Good at manipulating all media
 - Transcoding
 - Transizing
 - Transrating
 - Interworking
- Server-side CPU intensive
- Client can request different conference mixes
- Usually client friendly
 - Downsizing & mixing can reduce bandwidth
 - Fewer client streams to process



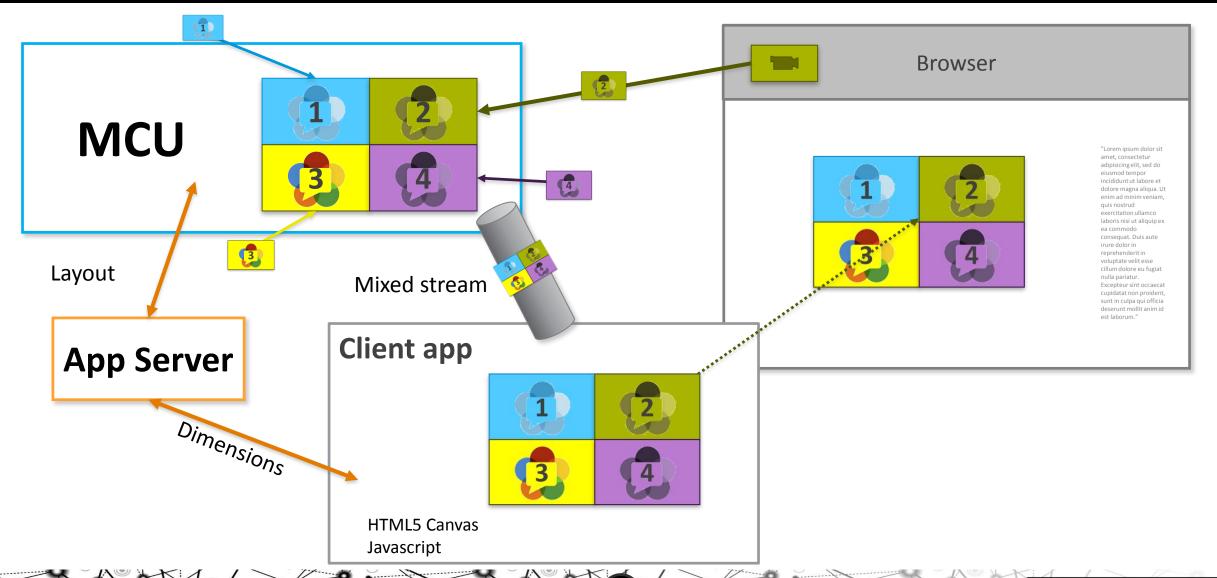
Video UX: Traditional method – mixed screen



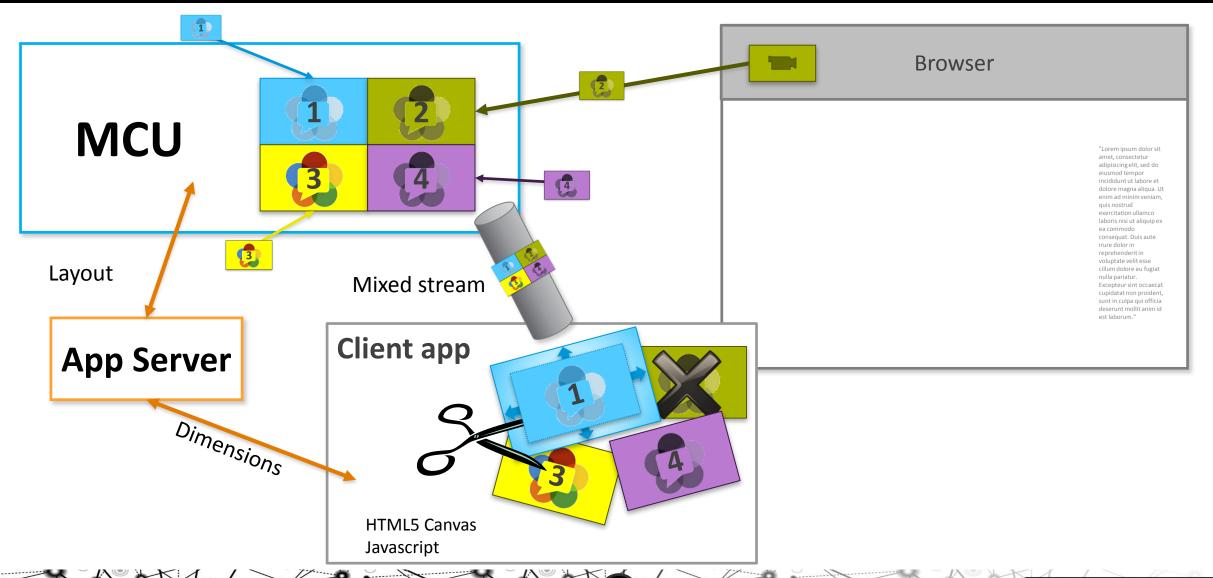
Video UX: Traditional method – active talker



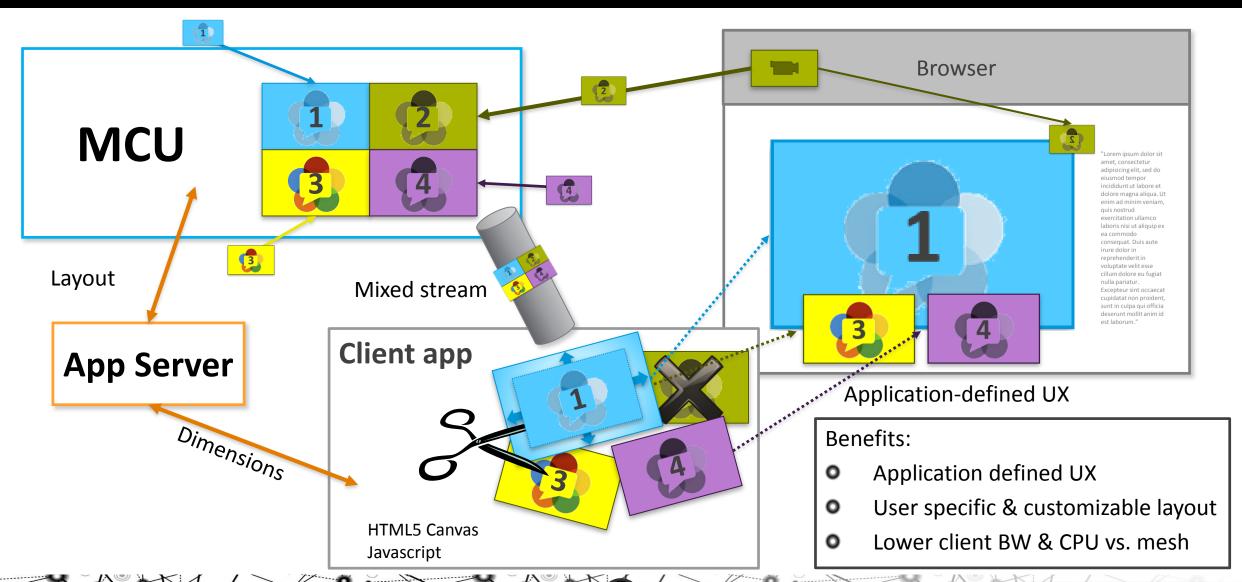
Modern UX with MCU mixing, HTML5 & JavaScript



Modern UX with MCU mixing, HTML5 & JavaScript



Modern UX with MCU mixing, HTML5 & JavaScript

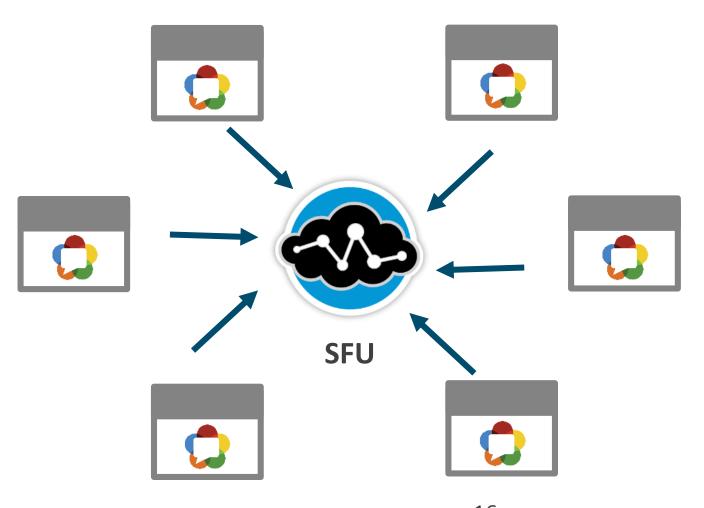


Newer approach: SFU

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Selective Forwarding Unit (SFU) routing

- Clients send one & receive many
- Client can instruct SFU which streams to send
- High throughput
- Can be lots of downlink bandwidth
- Low latency

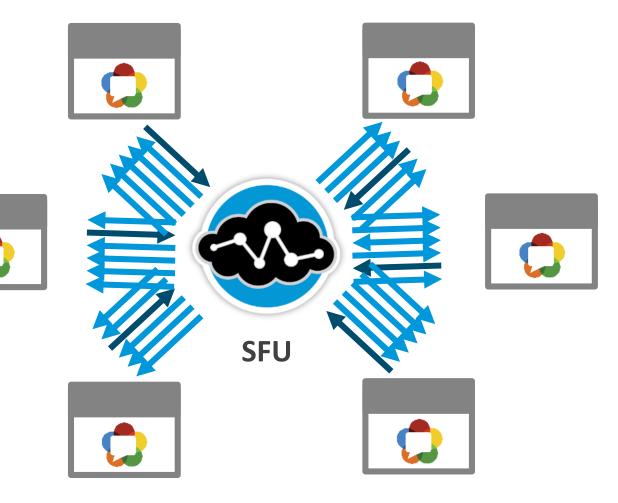


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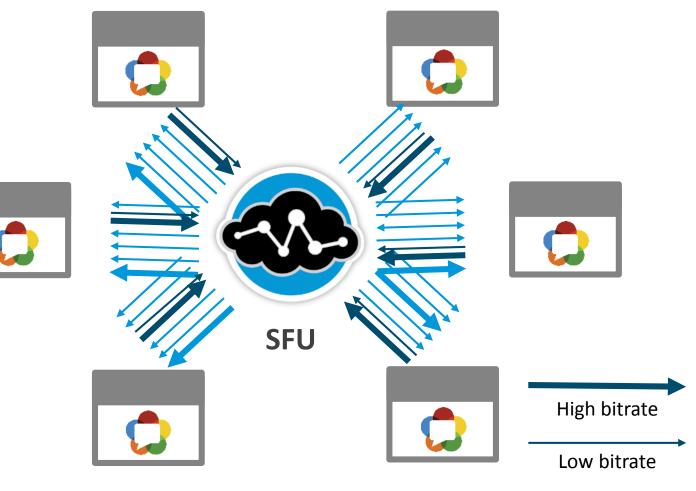


New WebRTC approach: Simulcast

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Selective Forwarding Unit (SFU) with Simulcast

- Clients send multiple streams to SFU
 - one high-bit rate
 - one or more lower-bit
- Client directs SFU which streams to receive
- Reduces bandwidth vs. SFU
- Simulcast in WebRTC coming

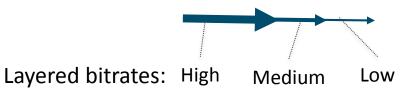


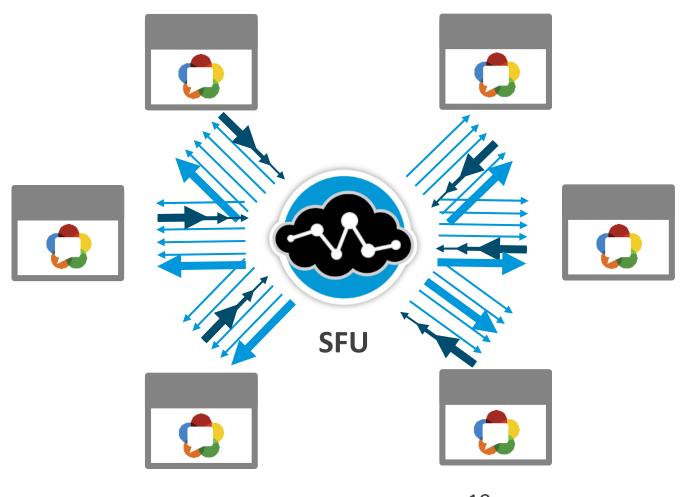
Future Approach with VP9 – SVC?

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Selective Forwarding Unit (SFU) with SVC

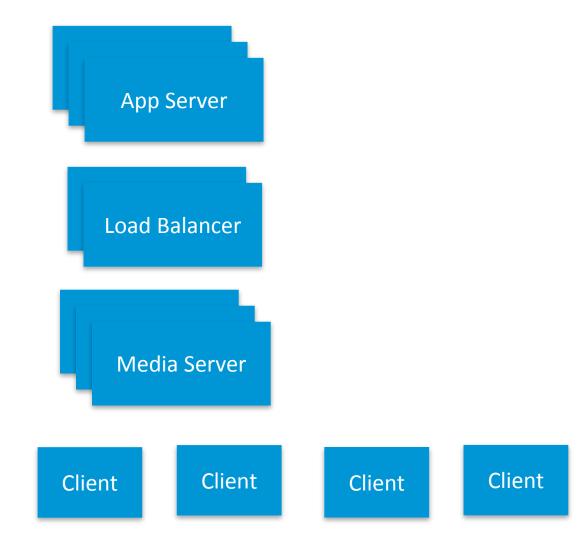
- •Clients send layered stream to SFU
 - Varying bitrates time, size,quality
- SFU directs who gets what
- Coming to WebRTC eventually





Typical media server network model

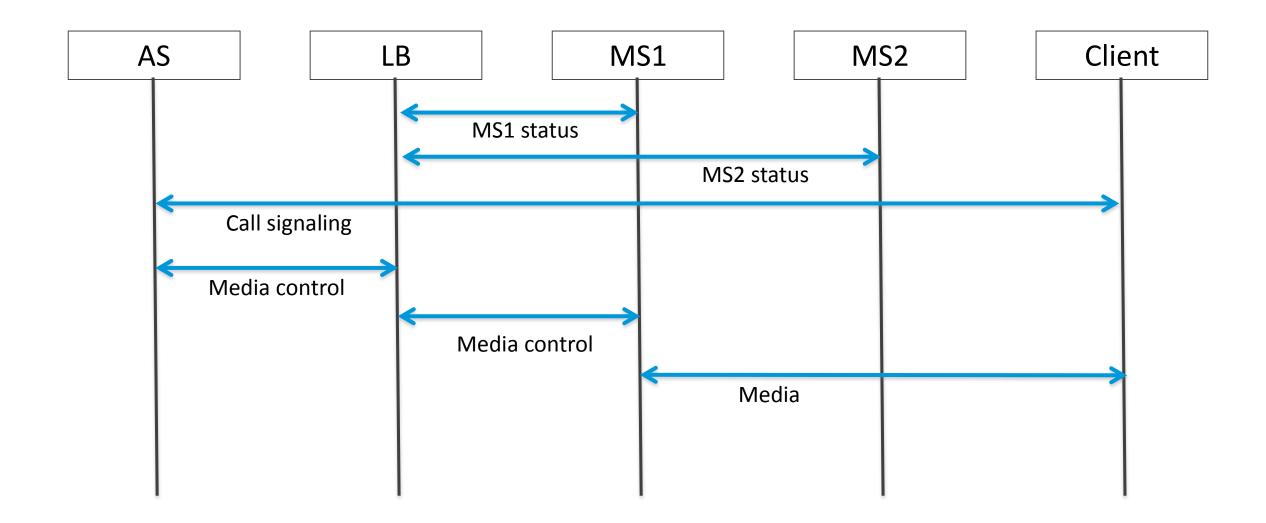
Dialogic.



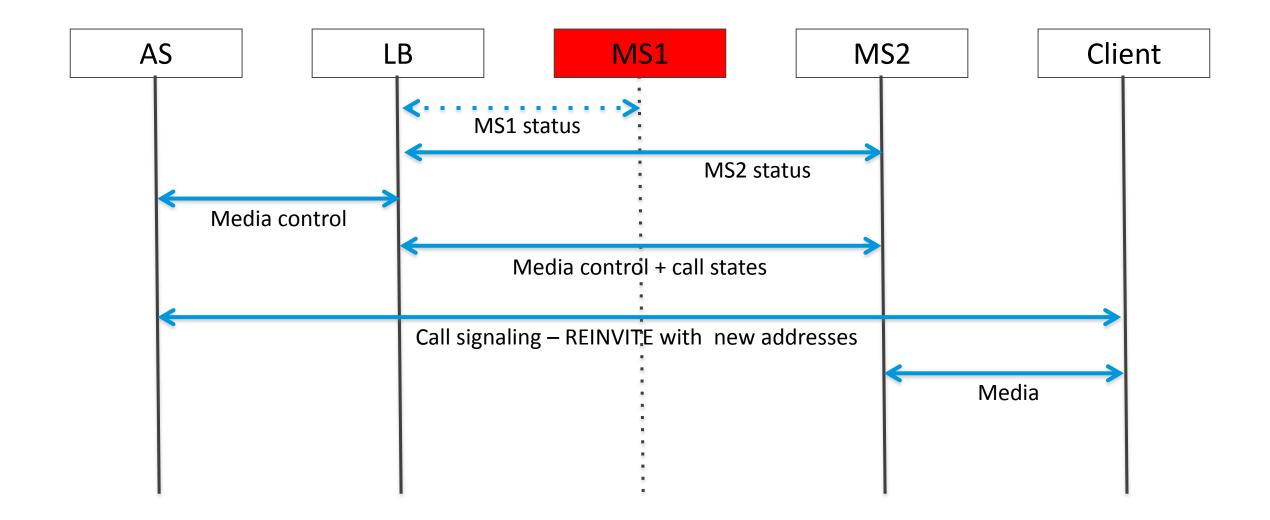
Client

Client

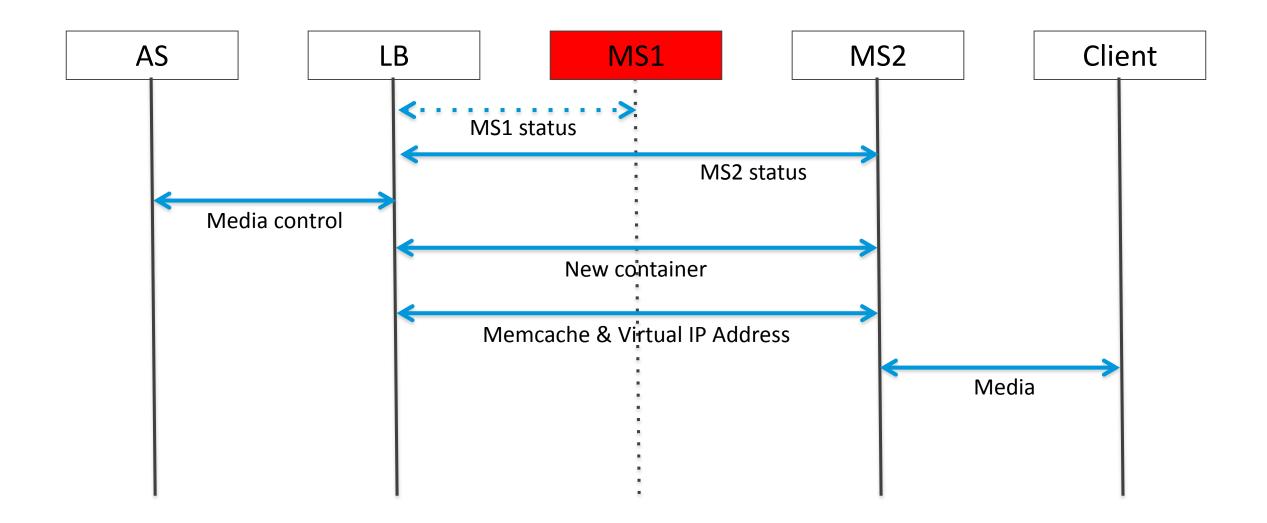
Pseudo Call-flow with a media server



Pseudo Call-flow – traditional fail-over with RE-INVITE



Pseudo Call-flow – fail-over the cloud way





- Declarative
- Concurrency
- Soft real-time
- Robustness
- Distribution
- Hot code loading
- External interfaces
- Portability

ICE, STUN, and TURN save the day



Image source: webrtcHacks- http://webrtchacks.com/an-intro-to-webrtcs-natfirewall-problem/

- Use of Public IP's mitigates need for TURN servers
- Open source TURN servers do reasonably well
- Running the TURN server coresident with the media server works well when you need it