

Scaling WebRTC deployments with multicast

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IETF 110 – MBONED WG
March 11th



A brief intro on our homemade WebRTC server



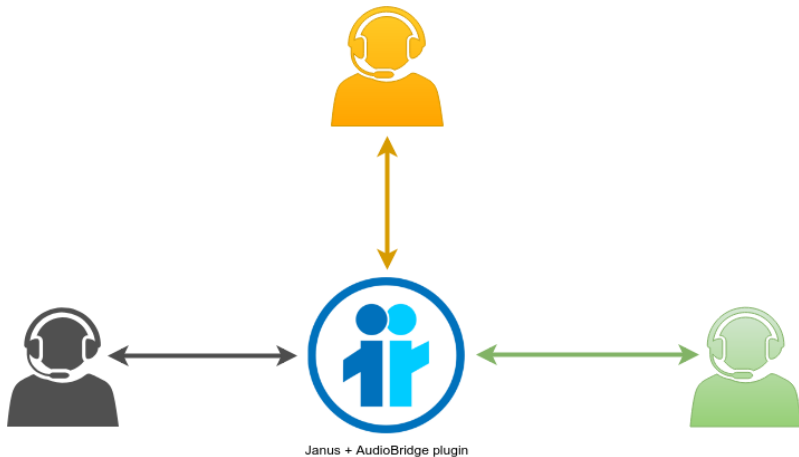
Janus

General purpose, open source WebRTC server

- <https://github.com/meetecho/janus-gateway>
- Demos and documentation: <https://janus.conf.meetecho.com>
- Community: <https://groups.google.com/forum/#!forum/meetecho-janus>



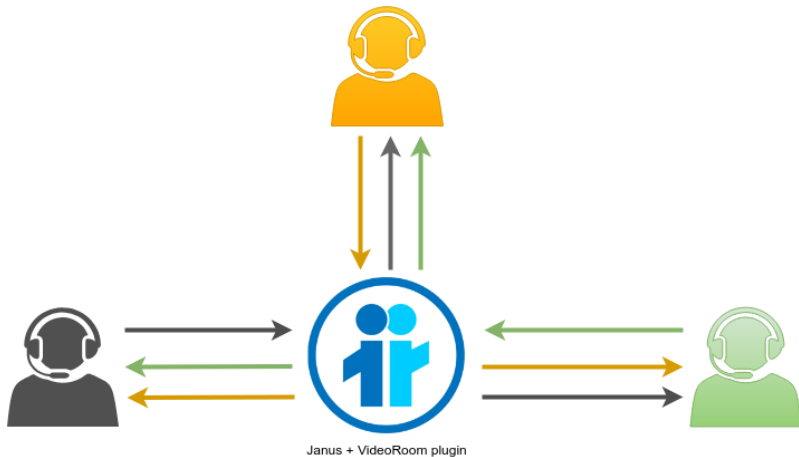
WebRTC-only MCU: AudioBridge plugin



<https://janus.conf.meetecho.com/docs/audiobridge>



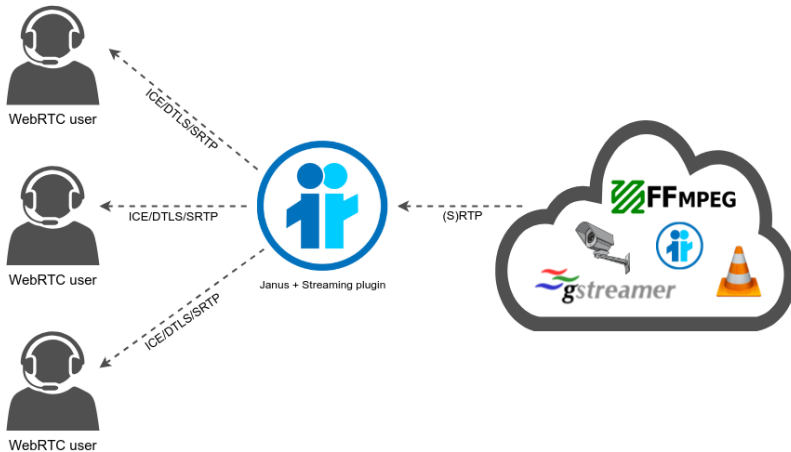
WebRTC-only SFU: VideoRoom plugin



<https://janus.conf.meetecho.com/docs/videoroom>



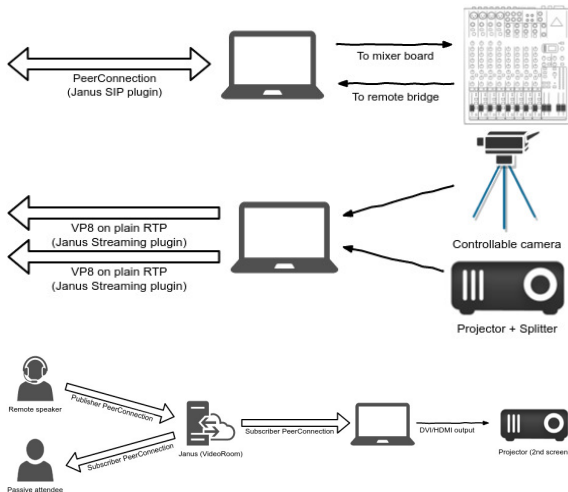
Re-broadcasting RTP: Streaming plugin



<https://janus.conf.meetecho.com/docs/streaming>

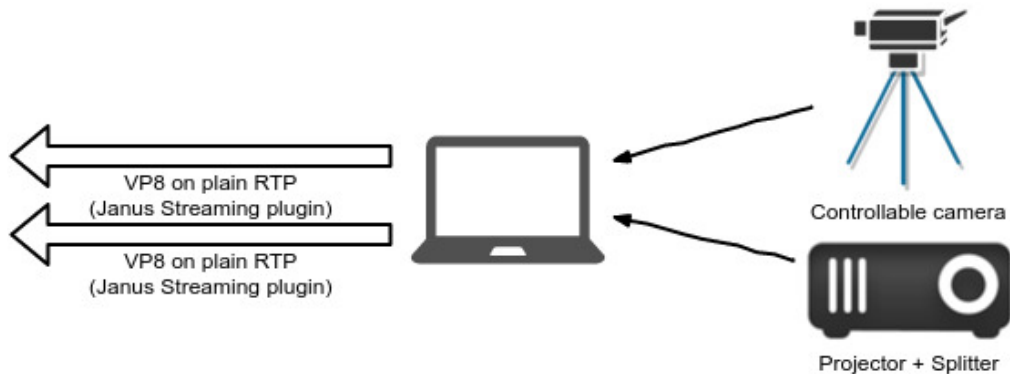


How we handled face-to-face IETF meetings



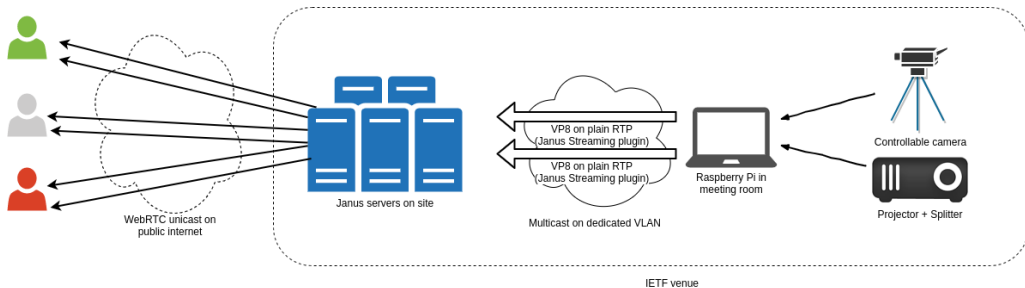
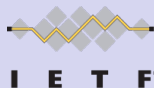


We used multicast a lot here!





We used multicast a lot here!





... until we had to stop meeting in person



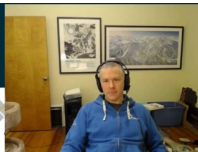
IETF109
MOPS



Virtual IETF meetings with WebRTC

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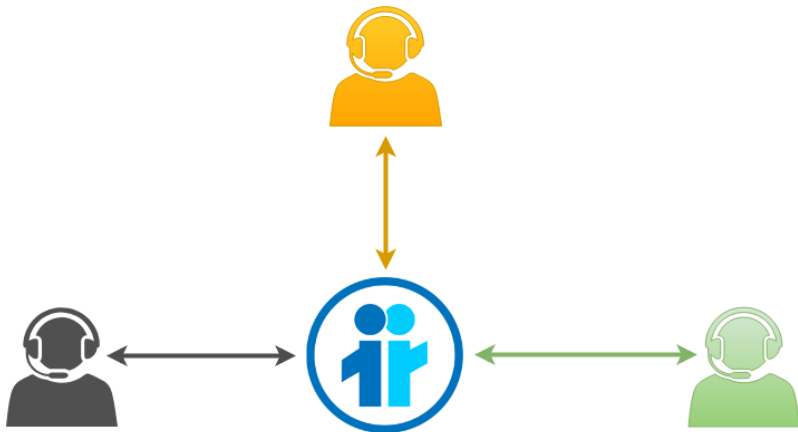
IETF 109 – MOPS WG
November 17th



<https://youtu.be/WeoKtWfmnxnU?t=285>



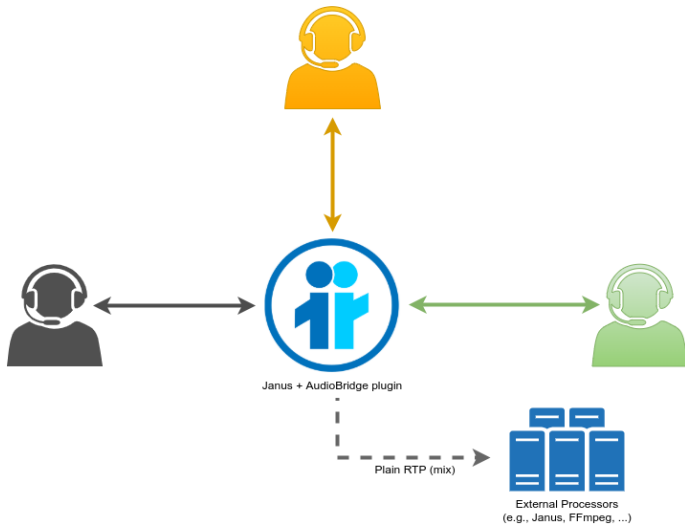
AudioBridge + RTP forwarding



Janus + AudioBridge plugin

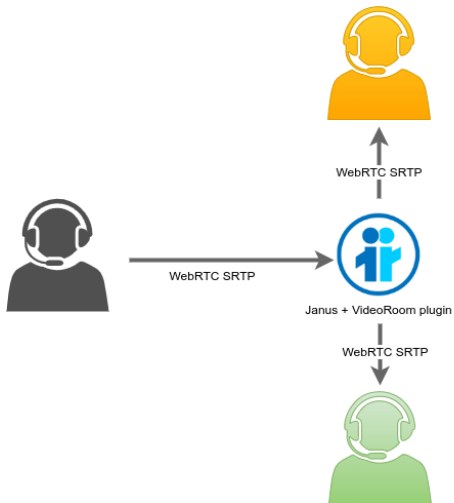


AudioBridge + RTP forwarding



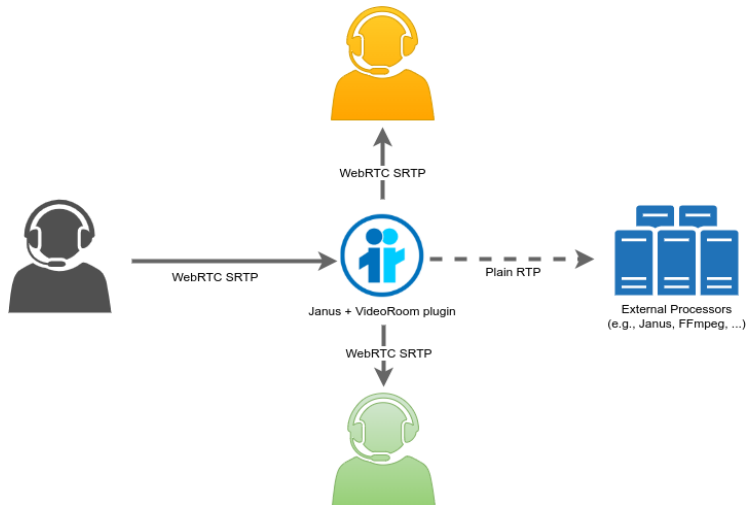


VideoRoom + RTP Forwarding



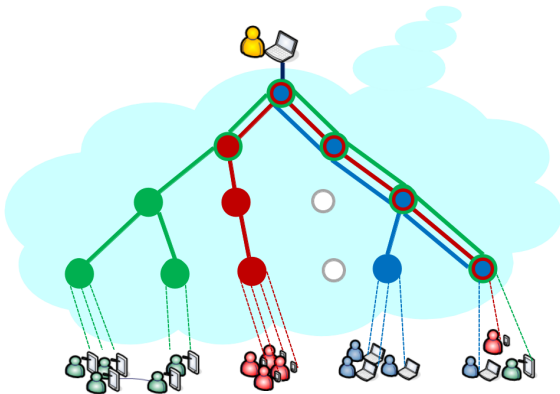


VideoRoom + RTP Forwarding





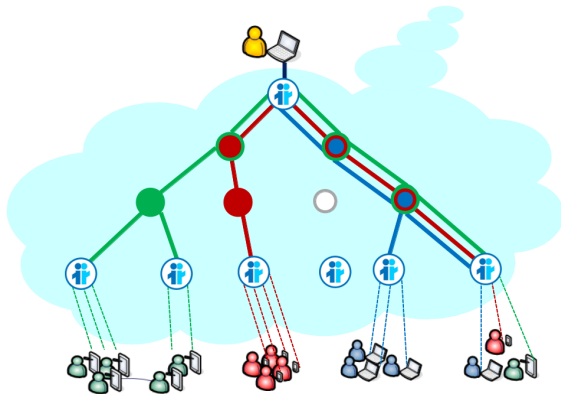
All bricks for a “WebRTC CDN” (SOLEIL)



http://www.fedoa.unina.it/10403/1/miniero_lorenzo_27.pdf



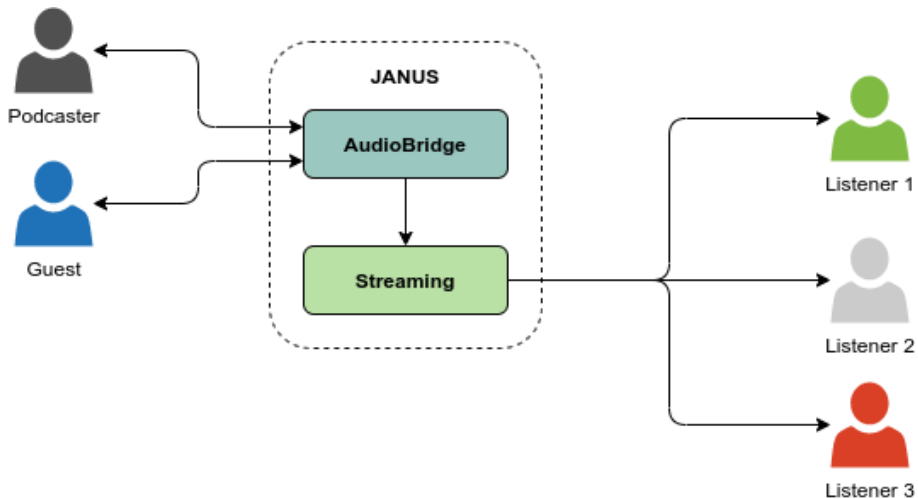
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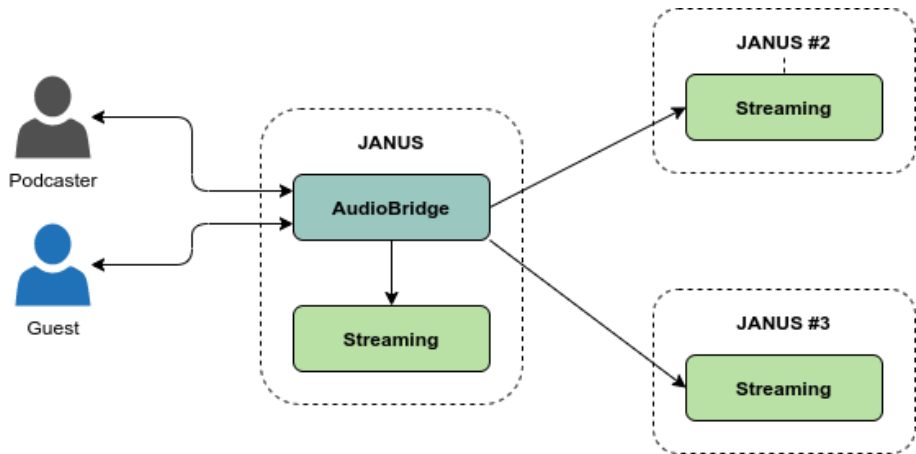


AudioBridge + RTP forwarding + Streaming



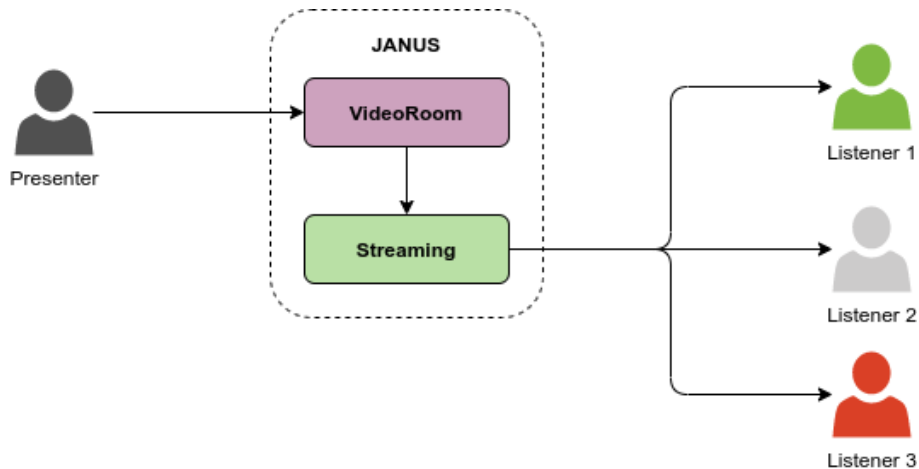


AudioBridge + RTP forwarding + Streaming



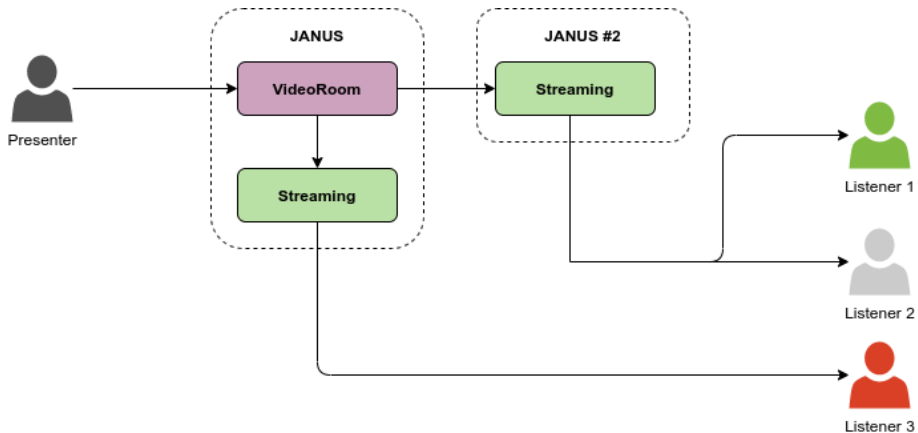


VideoRoom + RTP forwarding + Streaming



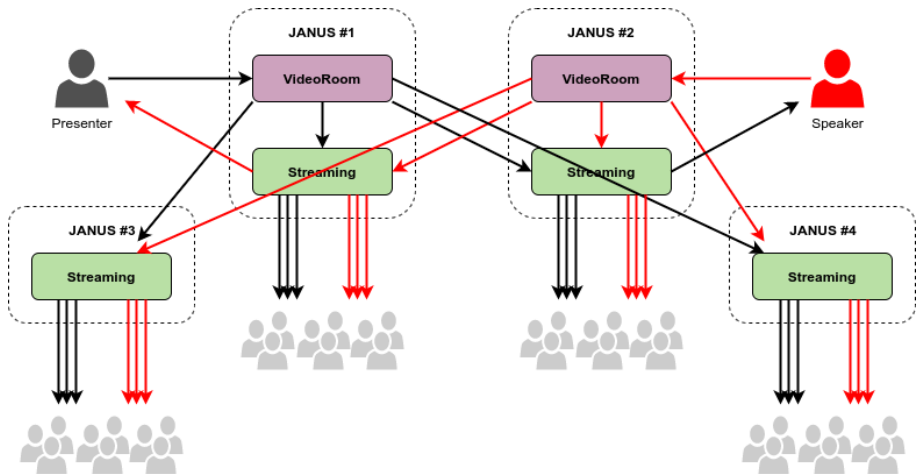


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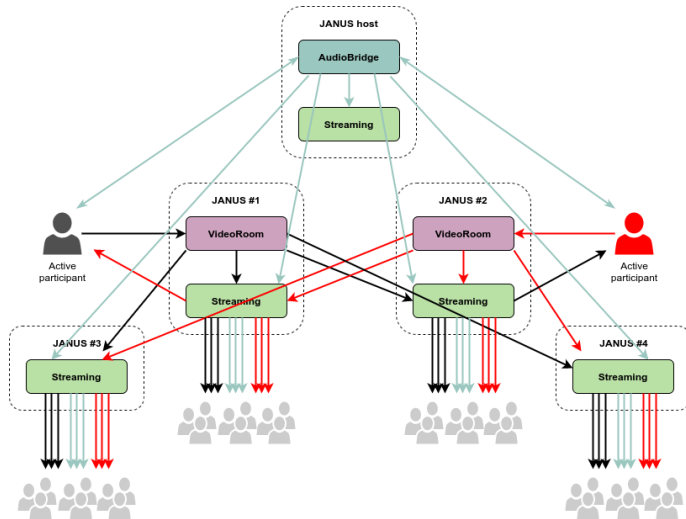


VideoRoom + RTP forwarding + Streaming



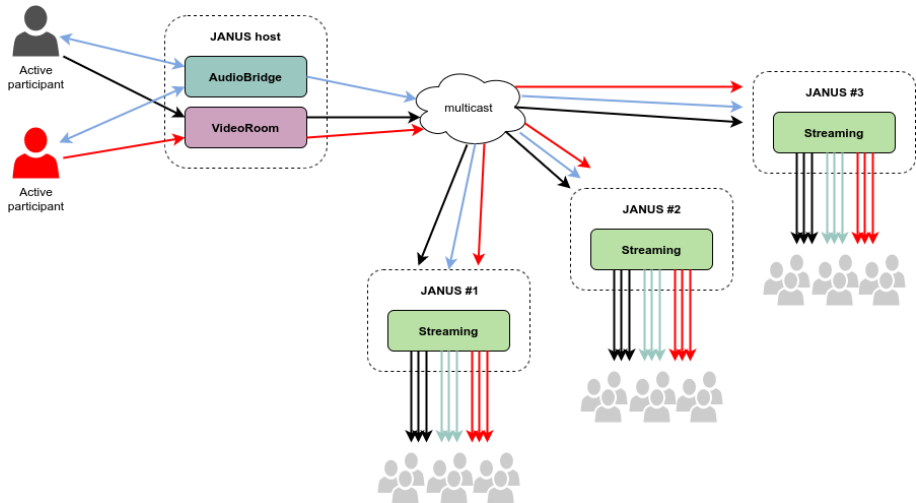


Bringing it all together for Virtual Events



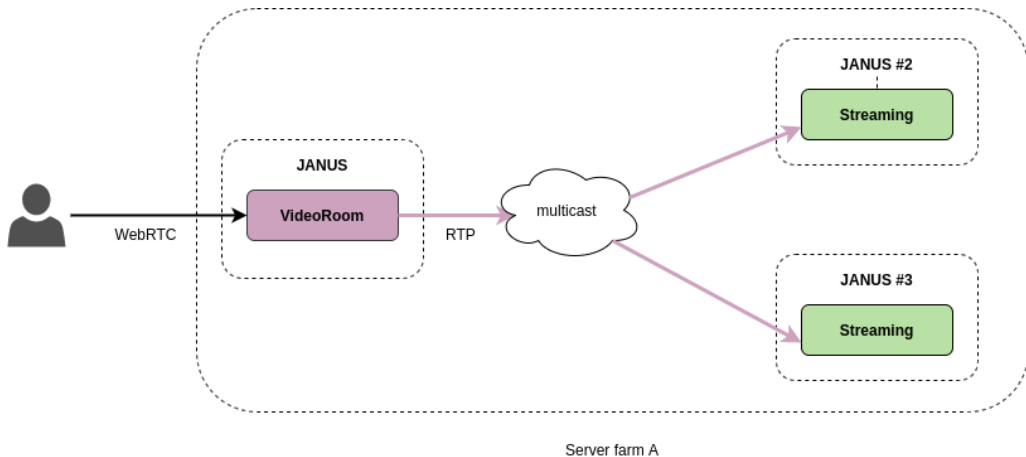


Multicast would be a perfect fit here!



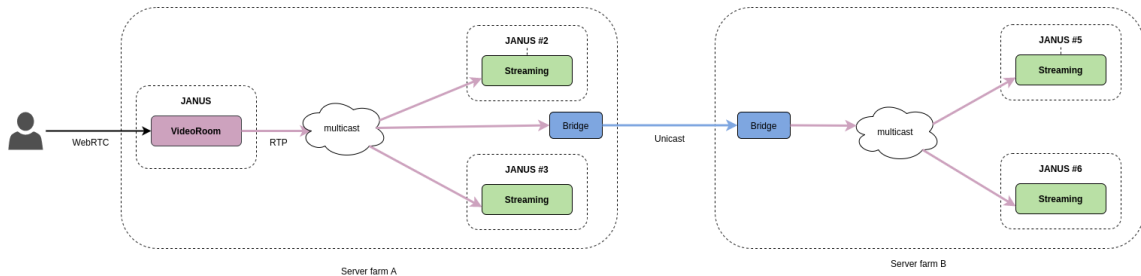


Pushing it even further





Pushing it even further





Why not multicast everywhere?



- WebRTC is ubiquitous → obvious choice
 - Simple way to allow bidirectional communication in browsers
 - Supported in all major browsers, now, so very low barrier to entry
- WebRTC doesn't support multicast, though
 - Conceived as peer-to-peer technology (servers are peers)
 - Each WebRTC PeerConnection has its own encrypted session
- There's a bit of multicast in there anyway!
 - <https://tools.ietf.org/html/draft-ietf-rtcweb-mdns-ice-candidates>
- Maybe some multicast could leveraged when broadcasting?
 - Automatic Multicast Tunneling may be helpful here



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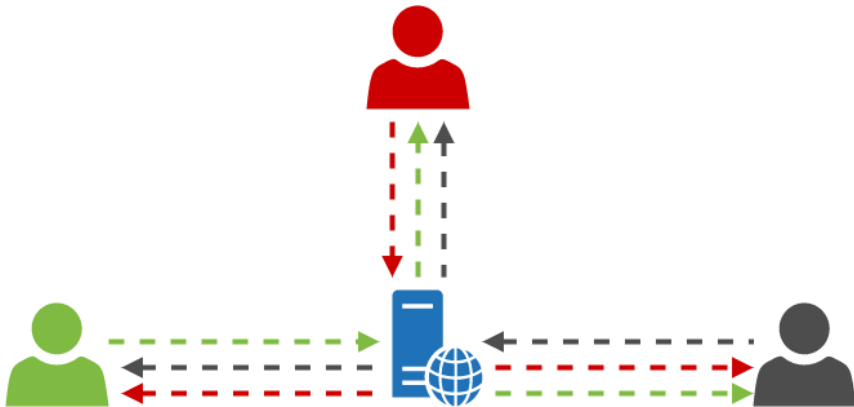
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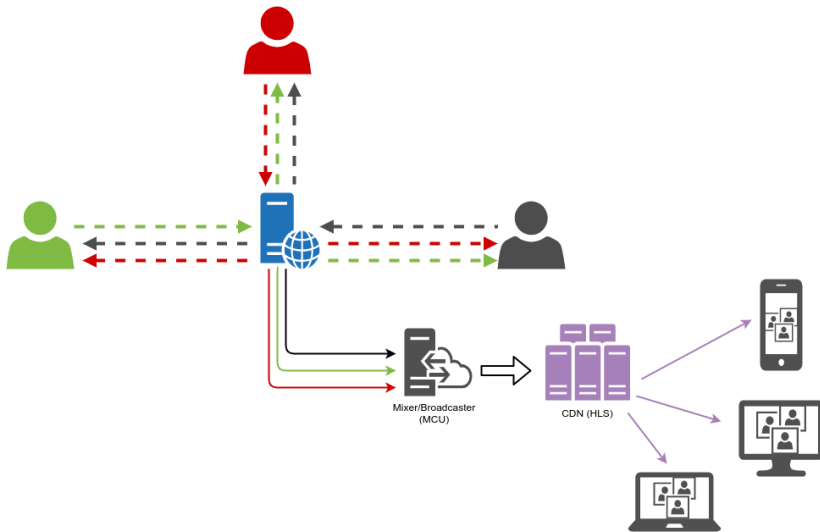


Broadcasting WGs to YouTube Live



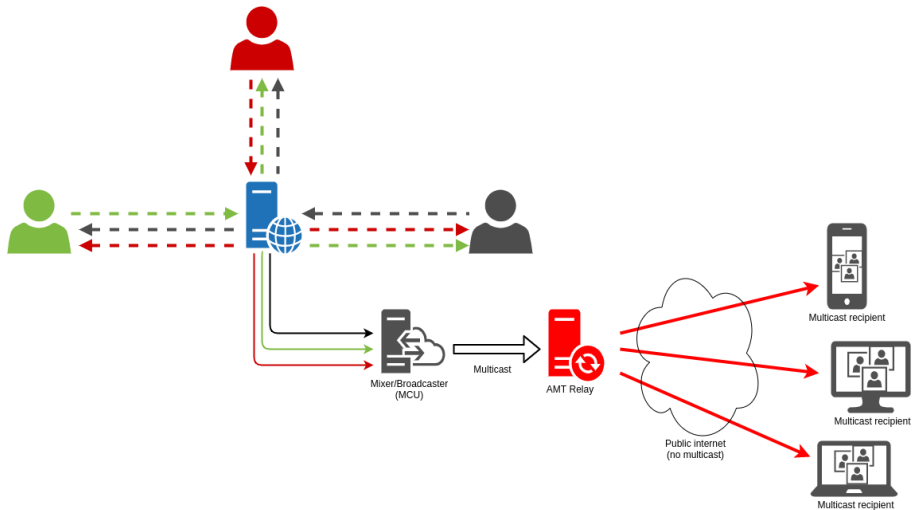


Broadcasting WGs to YouTube Live





Broadcasting WGs via AMT?





More interesting things happening @ UniNA/CINI!

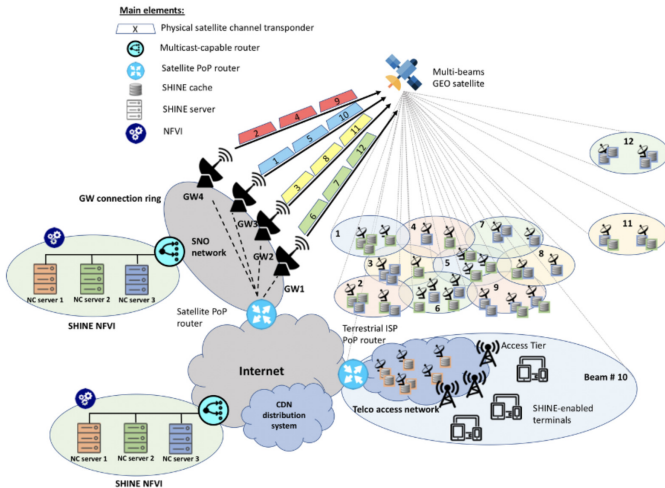


Figure 1: SHINE-enabling satellite multi-link service



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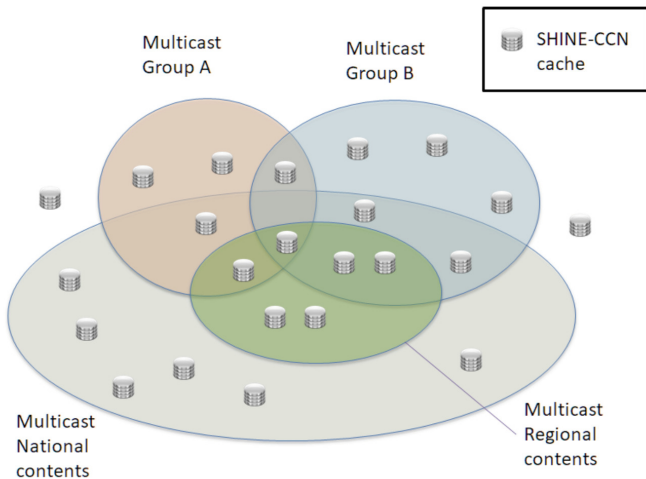


Figure 2: Multicast overlay



Thanks! Questions? Comments?

