

Implementation Experience WHIP in Janus (and GStreamer)

Lorenzo Miniero
lorenzo@meetecho.com

IETF 113, WISH WG
March 21st, 2022



<https://www.meetecho.com/blog/whip-janus-part-ii/>



WISH-a-WHIP: WebRTC ingest for broadcasting

Lorenzo Miniero
 @elminiero

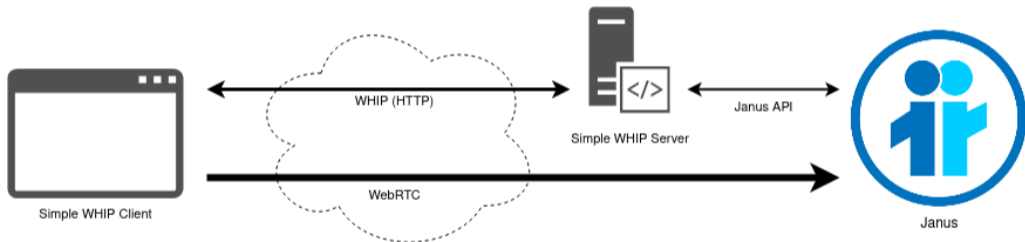
IIT Real-Time Communication 2021 – WebRTC Track
October 13th 2021, Chicago, IL, USA

https://www.youtube.com/watch?v=b_QBd3WnGgY

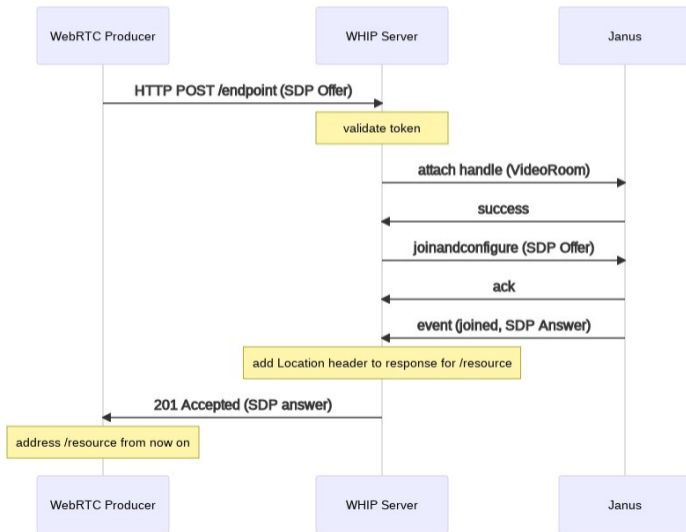
- Janus is a popular WebRTC server, so a good option for WHIP
 - It implements its own JSON-based API, though (Janus API)
- Simple (and transparent) solution: basic API translator in front of Janus
 - WHIP API maps quite simply to set of Janus API primitives
 - No need to change anything in the WebRTC stack
- Implemented simple prototype using node.js and Express
 - REST server that implements the WHIP API, and talks to Janus accordingly
 - Only takes care of ingest: distribution out of scope

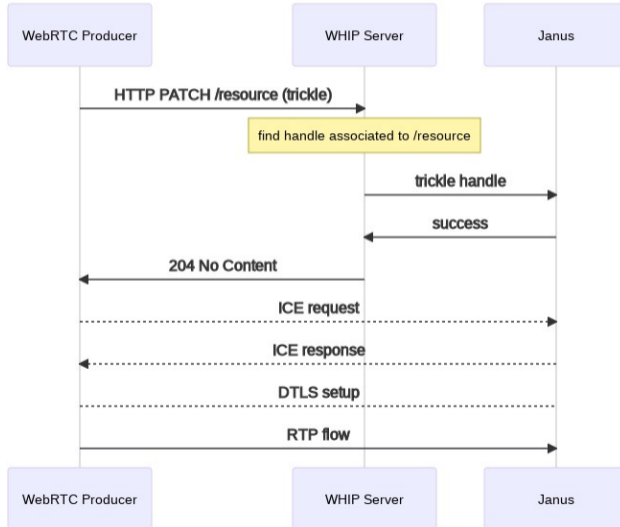
Simple WHIP Server

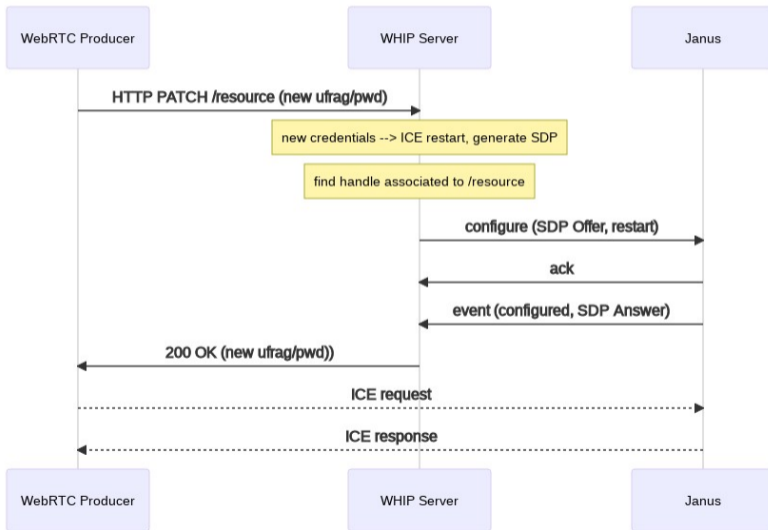
<https://github.com/meetecho/simple-whip-server/>

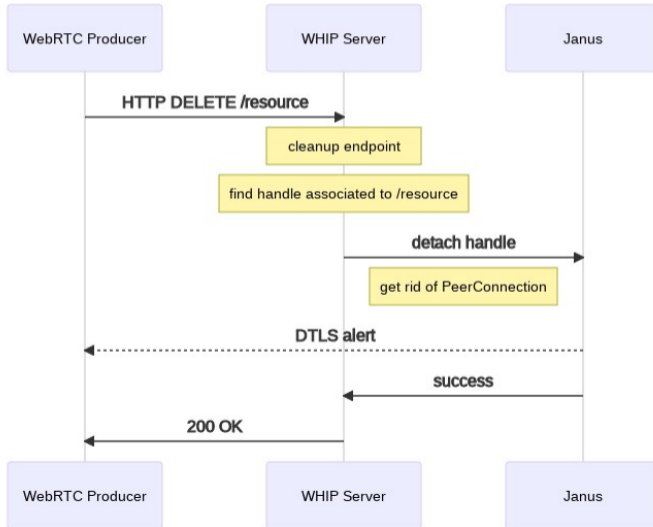


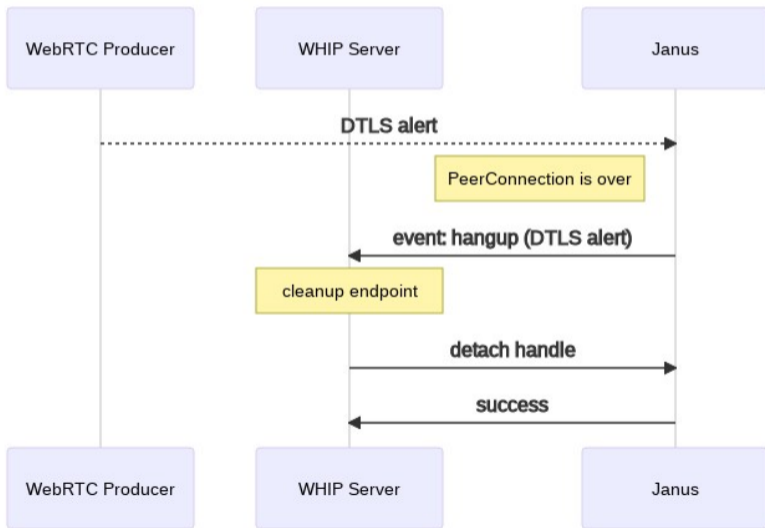
Mapping WHIP interactions to the Janus API











- Needs to support HTTP (WHIP API) and have a WebRTC stack
 - Browsers are the obvious choice, but what about a native solution?
 - Many broadcasters today use custom tools (e.g., OBS)
- Unfortunately OBS-WebRTC is not currently an option
 - Used legacy WHIP API, and currently only supports Millicast ingestion
- Chose GStreamer's `webrtcbin`¹ for the purpose
 - Used it already with success in other applications (e.g., JamRTC)
 - Modular and very powerful, so easy to feed with external sources

Simple WHIP Client

<https://github.com/meetecho/simple-whip-client/>

¹<https://gstreamer.freedesktop.org/documentation/webrtc/>

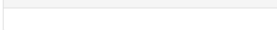
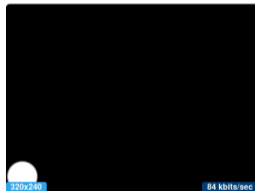
- Almost everything supported
 - Trickle (PATCH), STUN/TURN (via `Link` too), tokens, DELETE, etc.
 - Added support for non-trickle too (manual addition of candidates to SDP)
 - Option to force TURN (`iceTransportPolicy: "relay"` equivalent)
- Customizable audio/video pipelines
 - Easy to experiment with different sources and codecs
- A couple of things not supported in `webrtcbin` yet, though
 - ICE restarts (there seems to be a PR, though)
 - `Link` support in POST (we currently only do it via OPTIONS in the client)
- Only supports Linux at the moment (help to port to Windows/MacOS welcome!)

```
./whip-client -u
  https://mercury.conf.meetecho.com:8443/whip/endpoint/test \
-t hackathon \
-A "audiotestsrc is-live=true wave=red-noise ! audioconvert !
  audioresample ! queue ! opusenc ! rtpopuspay pt=100 ! queue !
  application/x-rtp,media=audio,encoding-name=OPUS,payload=100" \
-V "videotestsrc is-live=true pattern=ball ! videoconvert ! queue !
  vp8enc deadline=1 ! rtpvp8pay pt=96 ! queue !
  application/x-rtp,media=video,encoding-name=VP8,payload=96" \
-S stun.l.google.com:19302
```

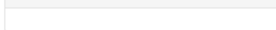
[Janus](#)[Home](#)[Demos ▾](#)[Documentation](#)[Papers](#)[Need help?](#)[JanusCon!](#)Meetecho

Plugin Demo: Video Room

Local Video

Remote Video #1 **WHIP Publisher 4321**

Remote Video #2



Remote Video #3



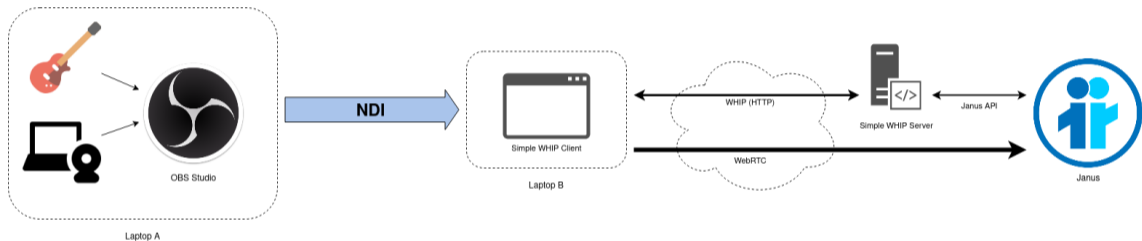
Remote Video #4



Remote Video #5



- Implemented in both server and client as pull requests
 - <https://github.com/meetecho/simple-whip-server/pull/4>
 - <https://github.com/meetecho/simple-whip-client/pull/9>
- **Server**
 - Generates (and returns) random ETag in response to POST and PATCH restarts
 - Expects ETag, and right one, when receiving trickle candidates (412 otherwise)
 - Expects "*" when receiving restarts, returns error otherwise (is this wrong?)
- **Client**
 - If an ETag is found in response to POST, it's used for PATCH requests
 - Client doesn't support restarts yet, so related ETag part is missing



<https://2021.commcon.xyz/talks/whip-ndi-and-janus-genesis-of-a-broadcasting-demo>
https://fosdem.org/2022/schedule/event/rtc_whip/



<https://2021.commcon.xyz/talks/whip-ndi-and-janus-genesis-of-a-broadcasting-demo>
https://fosdem.org/2022/schedule/event/rtc_whip/