## MOQPOC IETF 116

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# Disclaimer

- Early stages Proof of concept (POC)
  - Just A way (NOT the way) of implementing MOQ

#### - Goal:

- Learn about proposed technologies
- Provide as much information / metrics possible

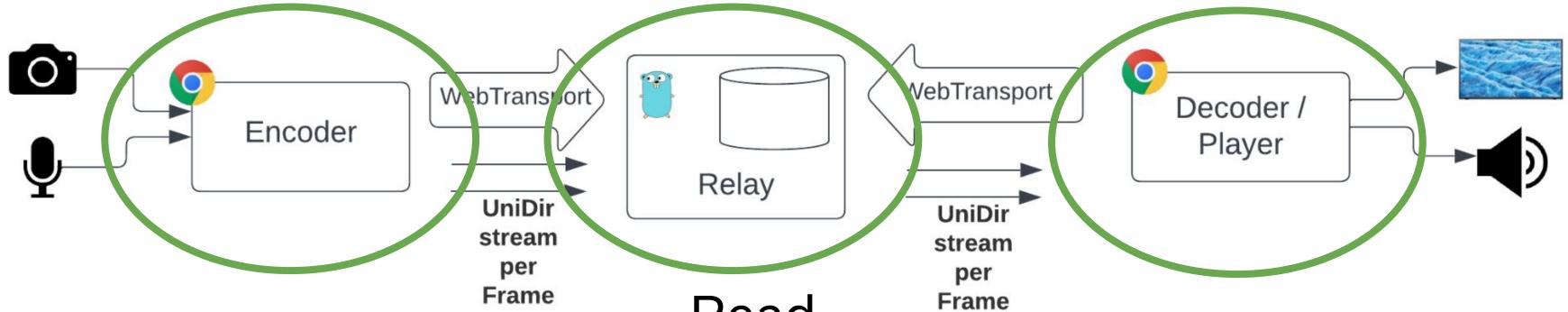
#### - Future:

- Hope could help us to have a better discussion about MOQ options
- Perhaps give some light over some trade-offs to make

## Use cases

- Ingest
  - ULL live
- Egress
  - ULL live (live edge)
  - Live rewind
  - Highlights / VOD

- Not yet
  - Priorities (sendOrder)
  - ABR



#### **Test Ultra low latency with Webcodecs + WebTransport: PLAYER**

(Encoder audio sampling frequency should be the same than audioContext (player) sampling frequency, this is almost guaranteed if you use same brov

server -> Demux -> Decode -> Play

Data needed
WT server: https://moq-test.oregon.jordicenzano.dev:4433/moqdelivery
Stream type: Live edge StreamID: streamtest
Packager type: v2 - Binary v
Player buffer (ms): 100 (it waits until audio buffers this amount to start playback)
Audio jitter buffer buffer for this player (ms): 200 Video jitter buffer buffer for this player (ms): 100
Start Stop

Read while write

#### **Test Ultra low latency with Webcodecs: ENCODER**

WebCam(v+a) -> Encode -> Mux -> Send -> Server

WT server: https://moq-test.oregon.jordicenzano.dev:4433/moqingest  StreamID: 20230328042624 Old StreamID: -  Packager type: v2 - Binary v  Max audio sending buffer allowed (ms): 300  Max video sending buffer allowed (ms): 150  Max inflight audio requests: 100  Max inflight video requests: 50  Expiration time for media chunks (except init) (in secs): 120  Start Stop	Data needed
Packager type: v2 - Binary v  Max audio sending buffer allowed (ms): 300  Max video sending buffer allowed (ms): 150  Max inflight audio requests: 100  Max inflight video requests: 50  Expiration time for media chunks (except init) (in secs): 120	WT server: https://moq-test.oregon.jordicenzano.dev:4433/moqingest
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Expiration time for media chunks (except init) (in secs): 120	Max inflight audio requests: 100
	Max inflight video requests: 50
Start Stop	Expiration time for media chunks (except init) (in secs): 120
	Start Stop

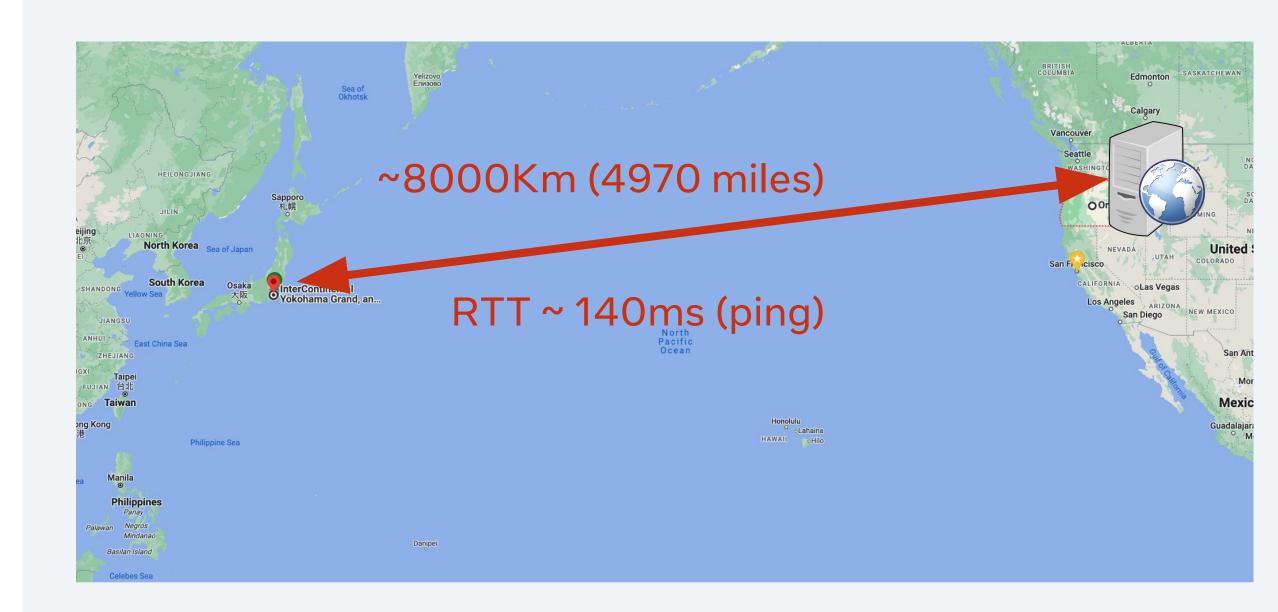
# Main features

- Uses QUIC streams over WebTransport
   (WT)
- QUIC stream per frame

- Encoder
  - Opens WT session against relay
  - Pushes data to relay

- Player
  - Opens WT session against relay
  - Relay pushes data to player

# Demo setup



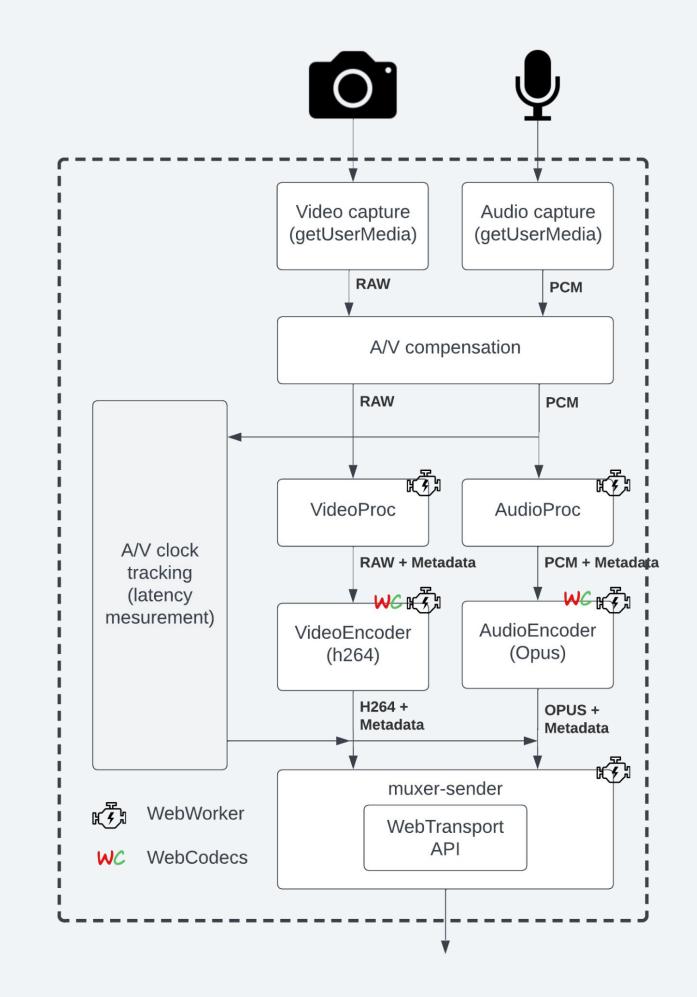
# DEMO

# DEMO Results

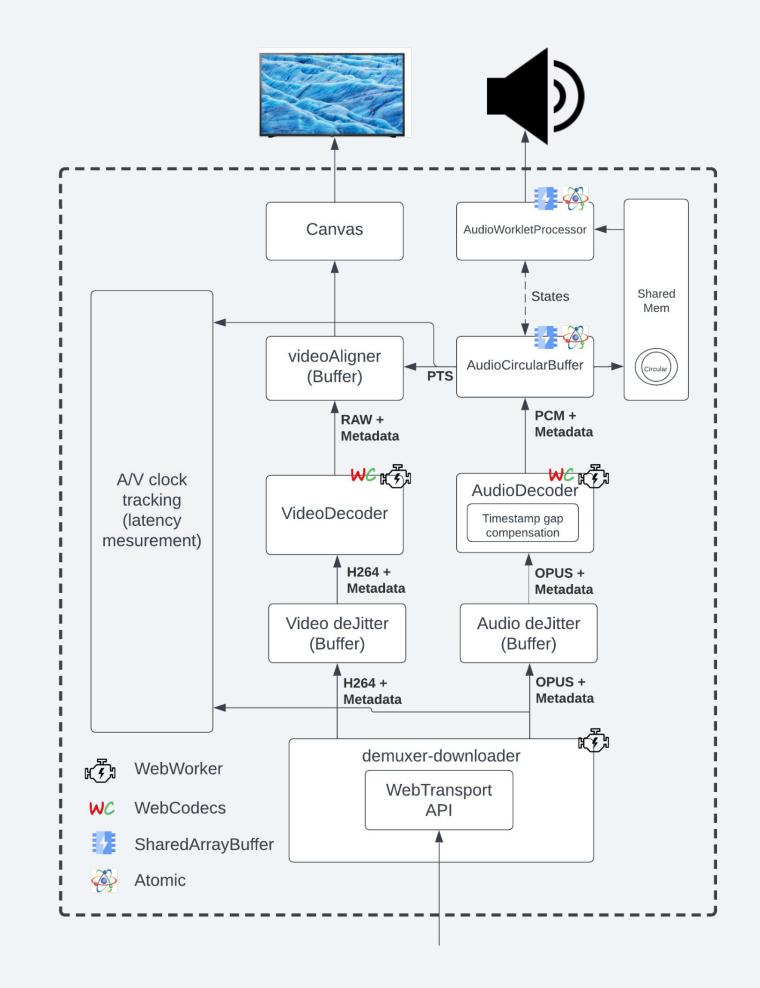
- H264 500Kbps + AAC 32Kbps
- Jitter buffer: 200ms

- E2E latency: ~ 400ms
- QOE: VC+ (IMHO)

### Encoder

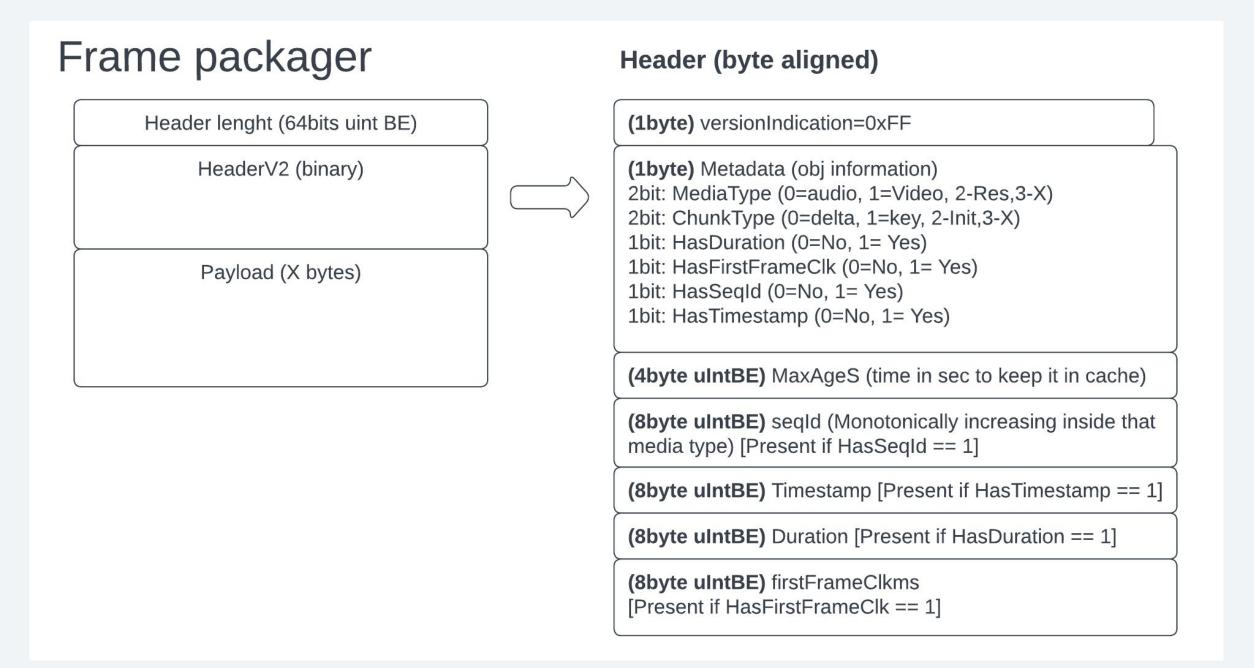


# Player



# Media Packager

VERY EXPERIMENTAL!!!!



#### Efficiency: 92%+

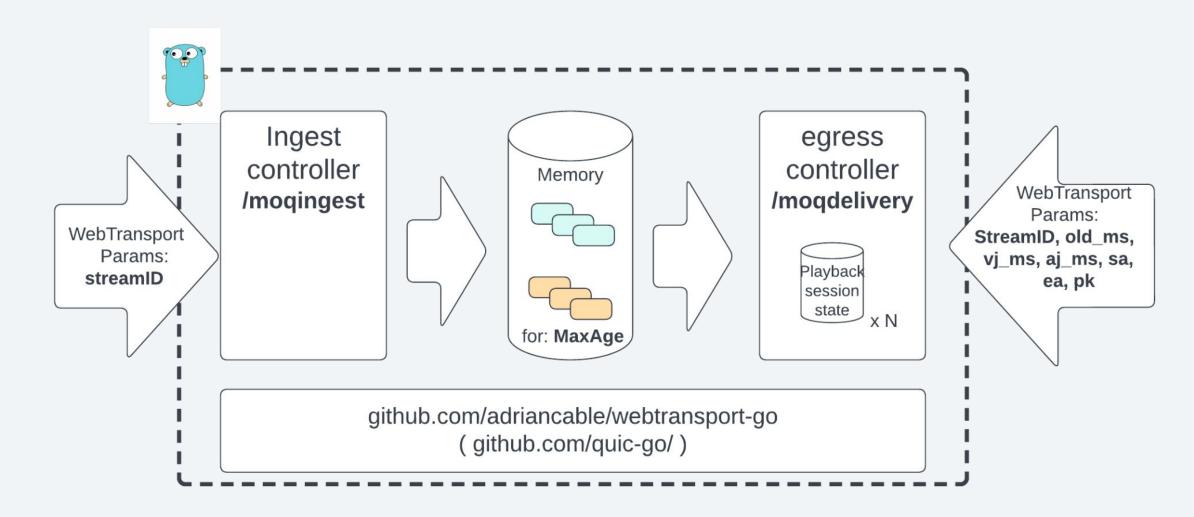
- Efficiency definition:

Total payload bytes vs packager bytes sent to transport

- Test conditions:

Video at 1Mbps, and Audio at 32Kbps

## Relay



From WT session

From Obj header

Cache key: streamID media-type/seqld

#### Examples:

- mystreamabc/video/123
- mystreamabc/video/-1 (init)

# References (C)

#### - Player & encoder:

https://github.com/facebookexperimental/webcodecs-capture-play

#### - Relay

https://github.com/facebookexperimental/go-media-webtransport-server

# QSA