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WHAT IS WEBRTC?

Browser APIs

E2E Secure Connection between Peers

Multiple Audio/Video Tracks

Binary Data

- Can be lossy
- Can be unordered
- Multiple distinct DataChannels



What we need to solve

Connectivity

- Multiple Possible Routes
- Not in the same network!
- UDP?

Security

- E2E Secure

Latency

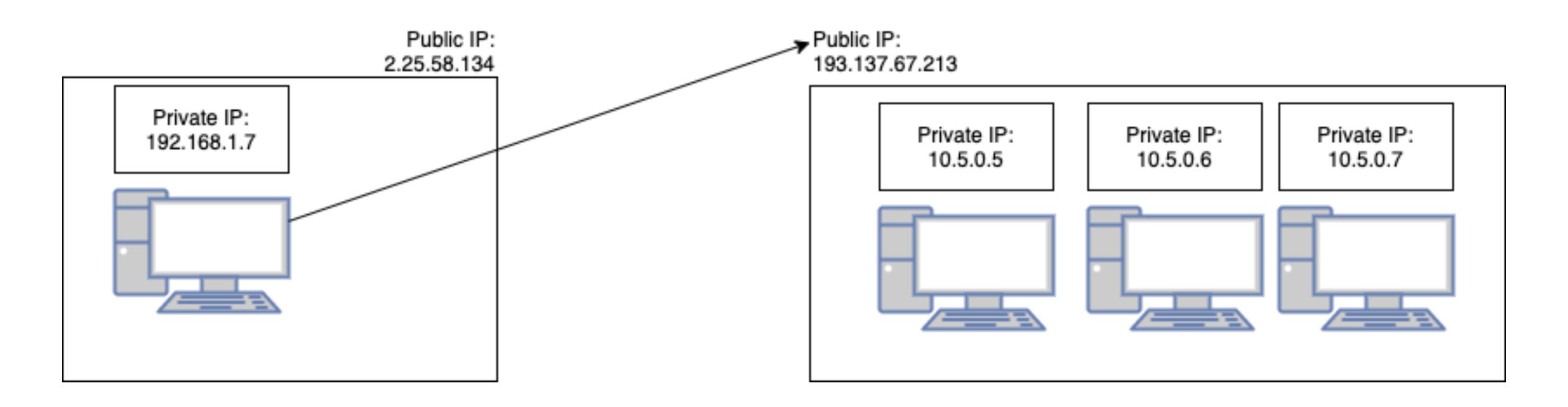
- 400 milliseconds max
- Unreliable Transport (UDP)

Ease of Use



WHAT DOES IT SOLVE

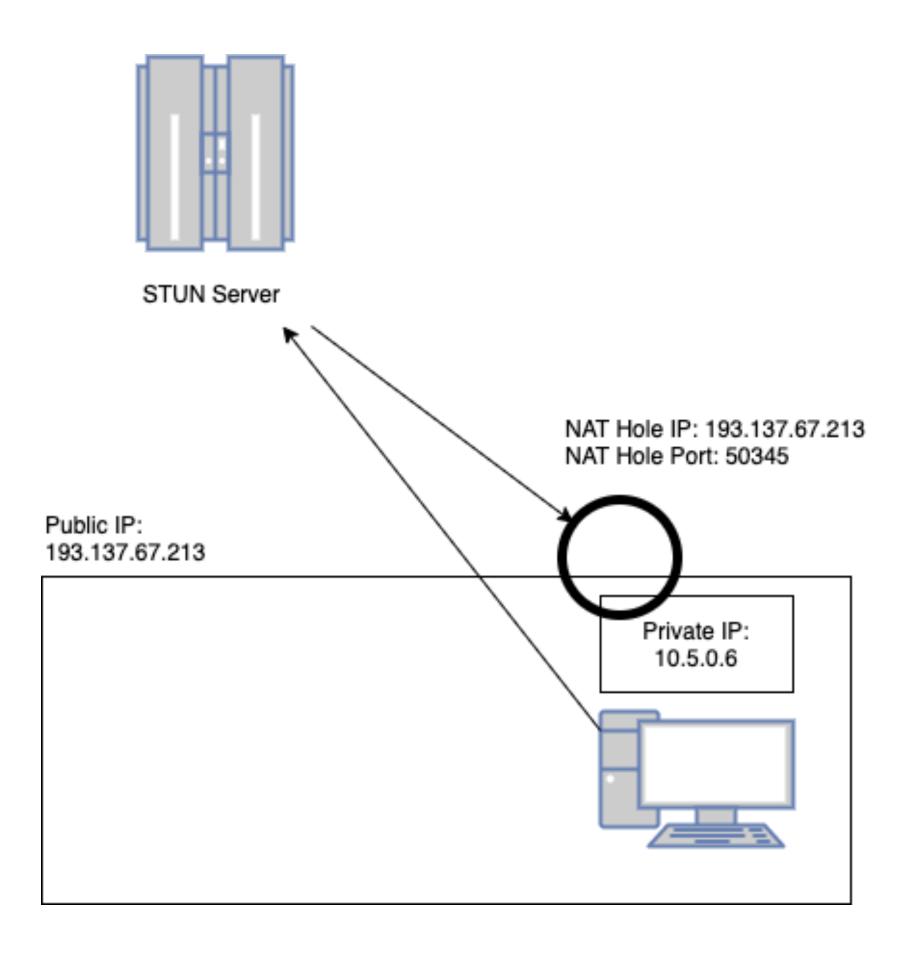
Problem: Connect two users with no Public IP





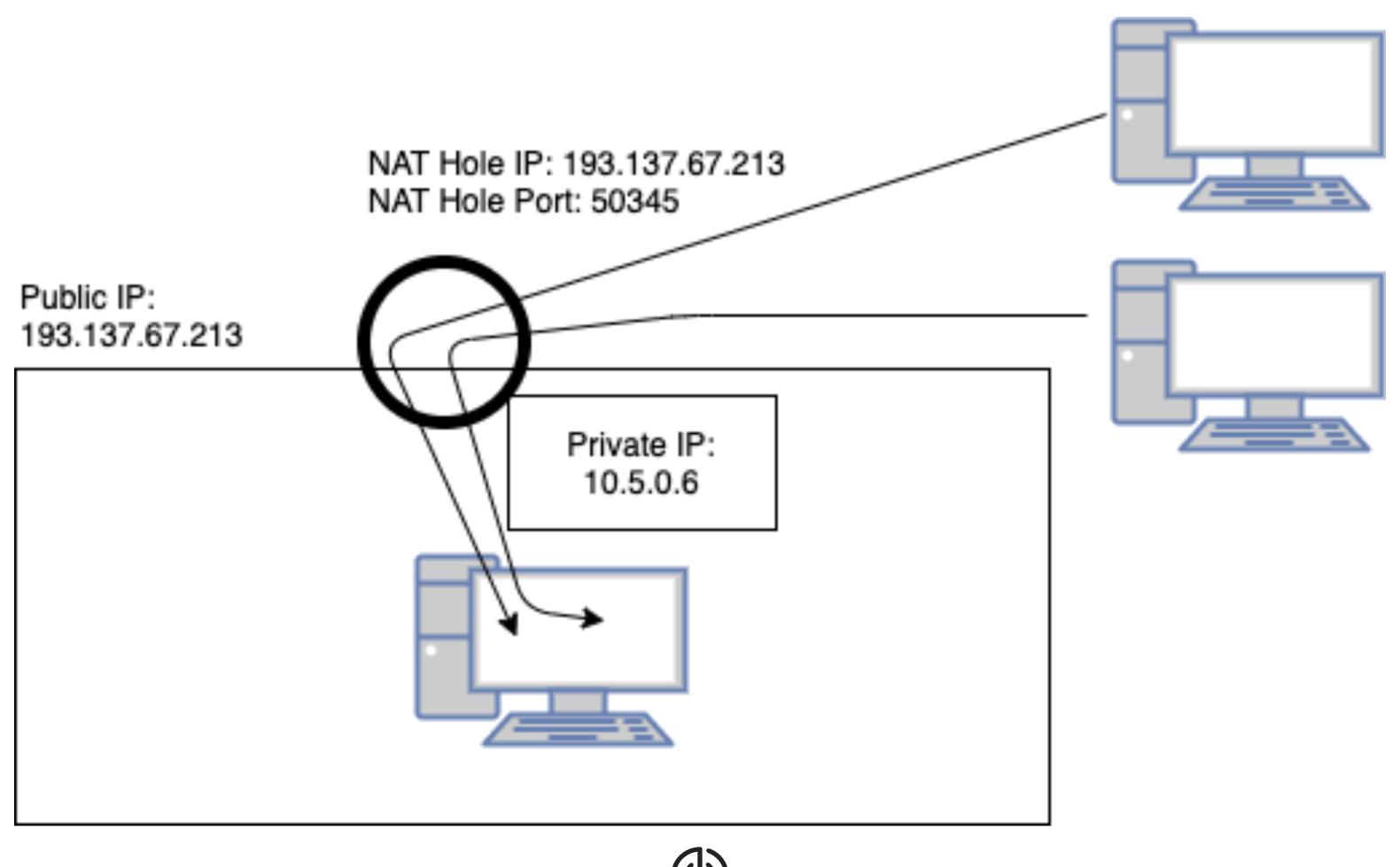


Solution: NAT Traversal





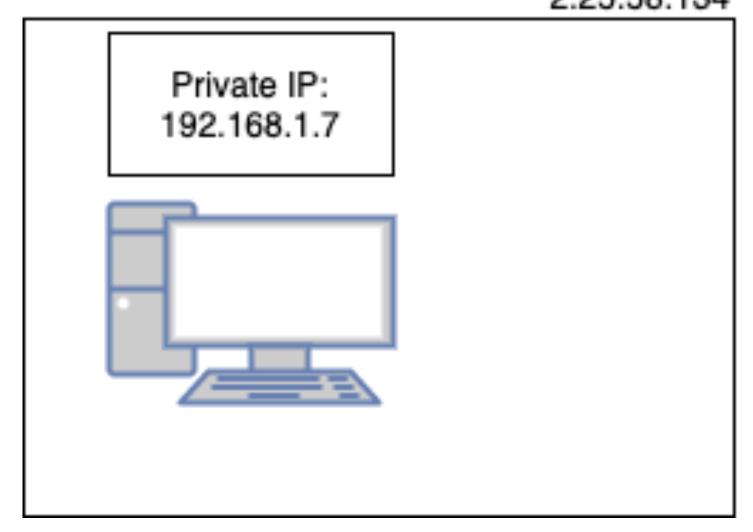
Solution: NAT Traversal





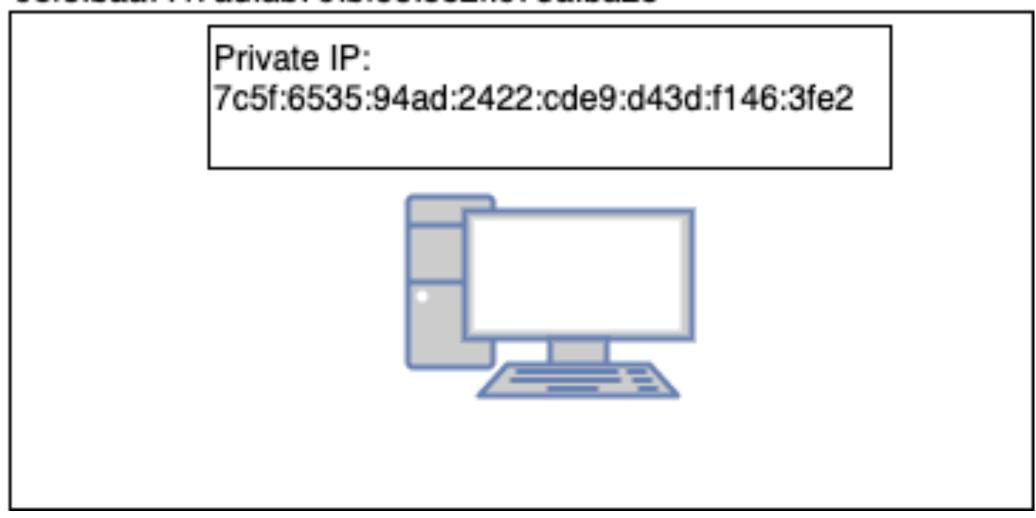
Problem: One is IPv4, One is IPv6

Public IP: 2.25.58.134



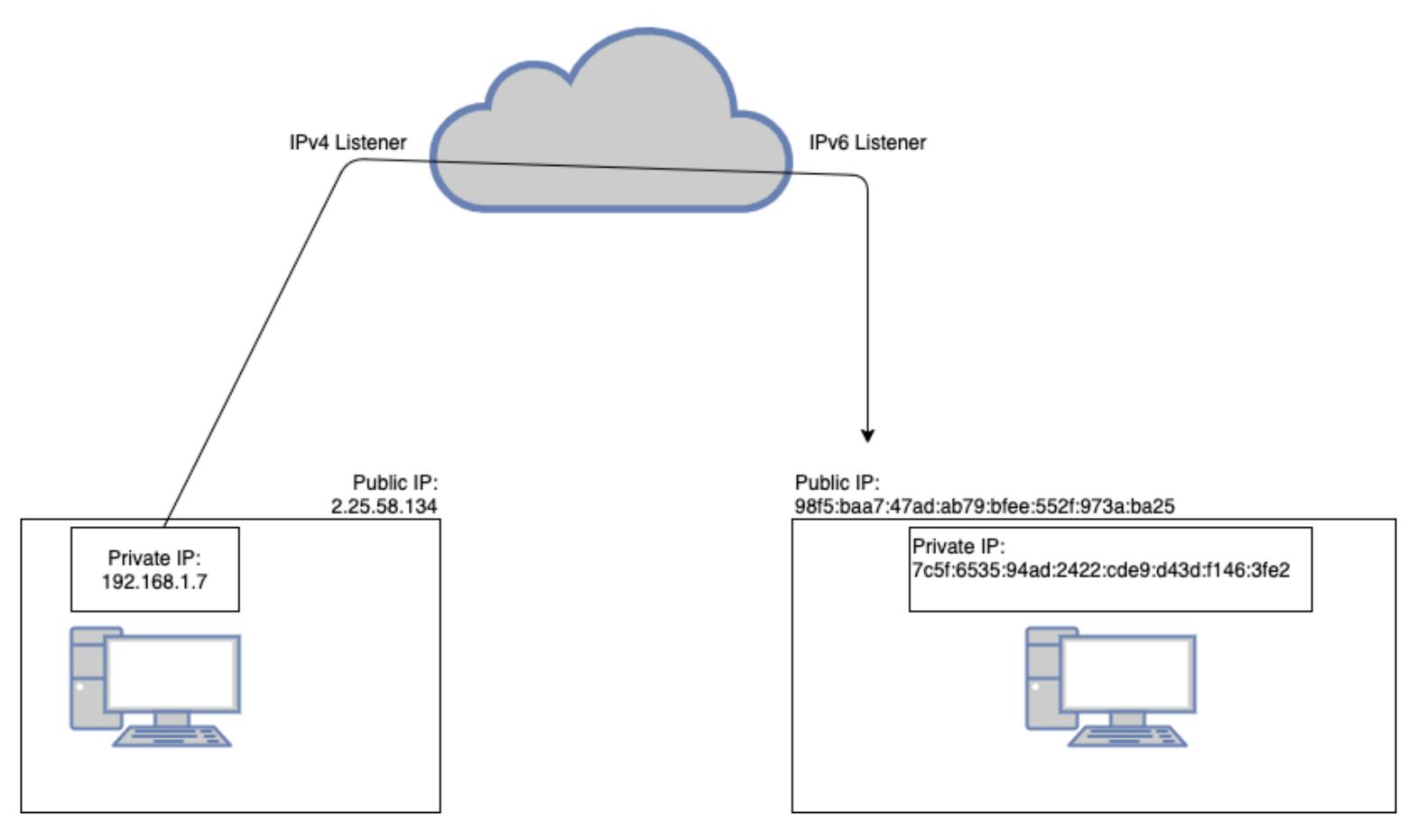
Public IP:

98f5:baa7:47ad:ab79:bfee:552f:973a:ba25



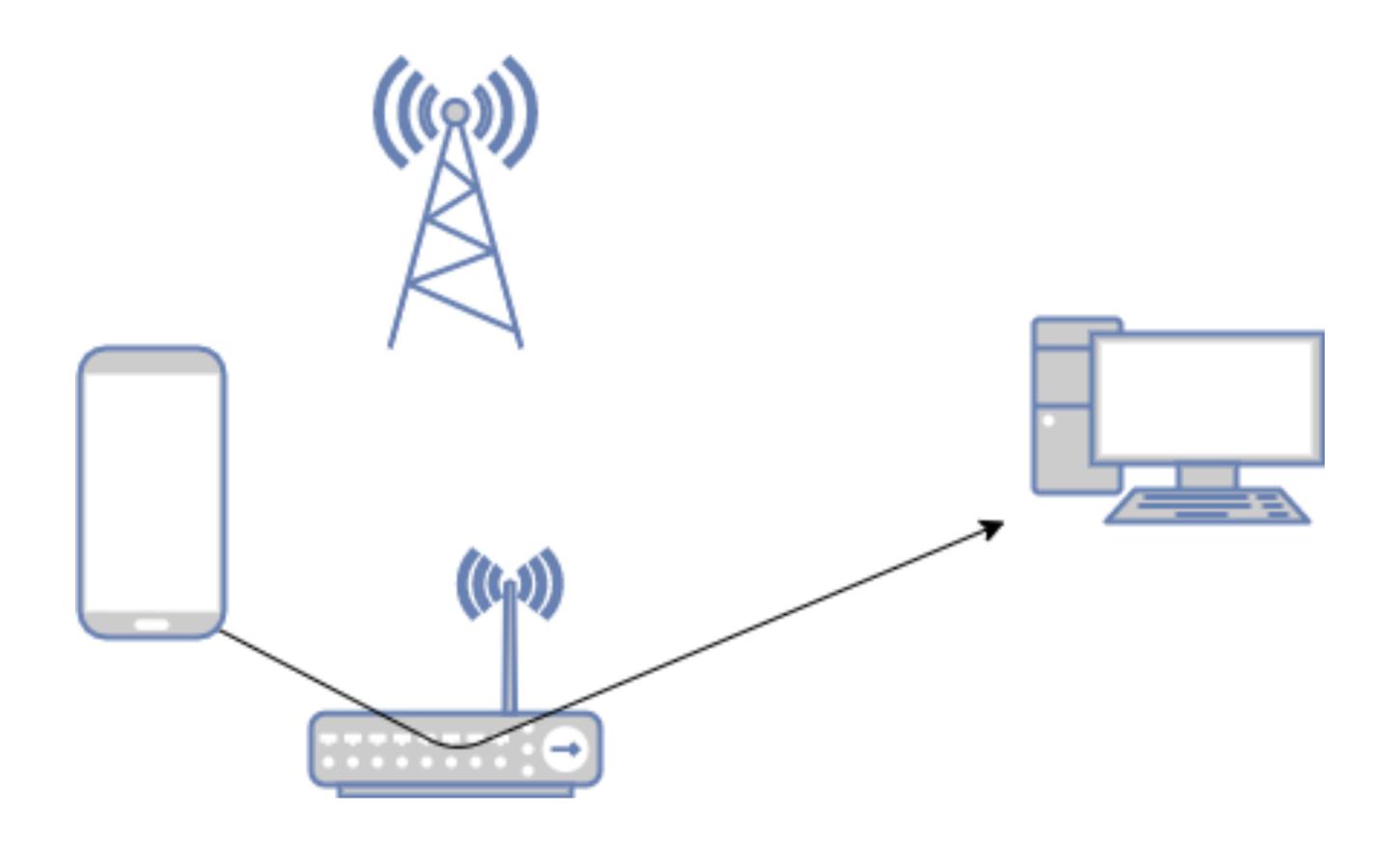


Solution: TURN



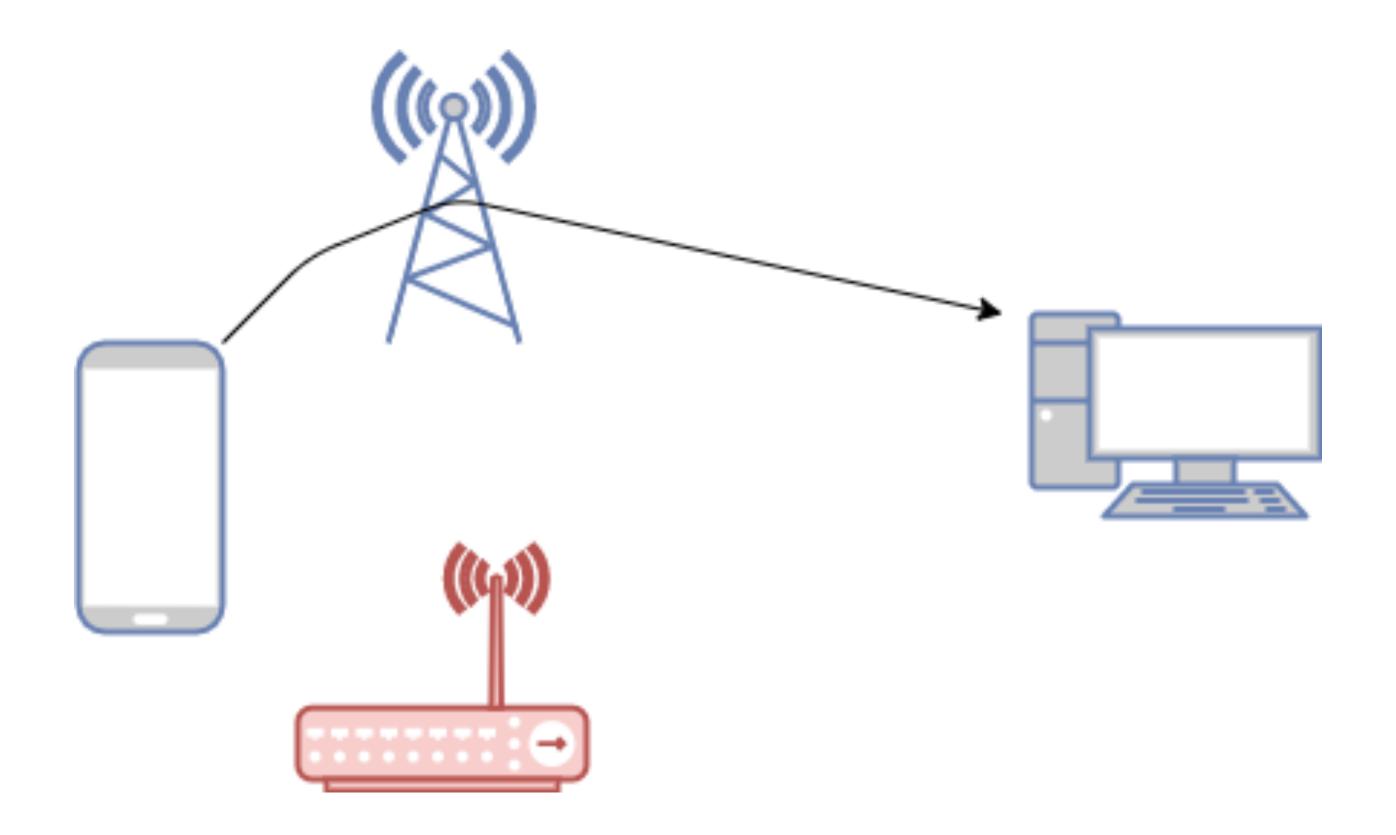


Problem: Wi-Fi on Fire



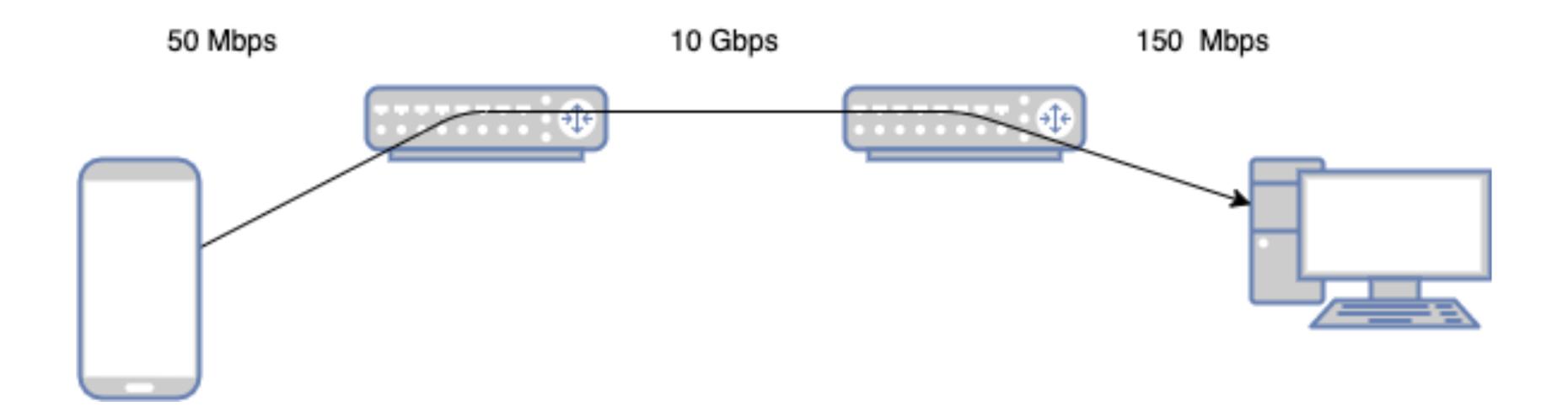


Solution: ICE Renomination



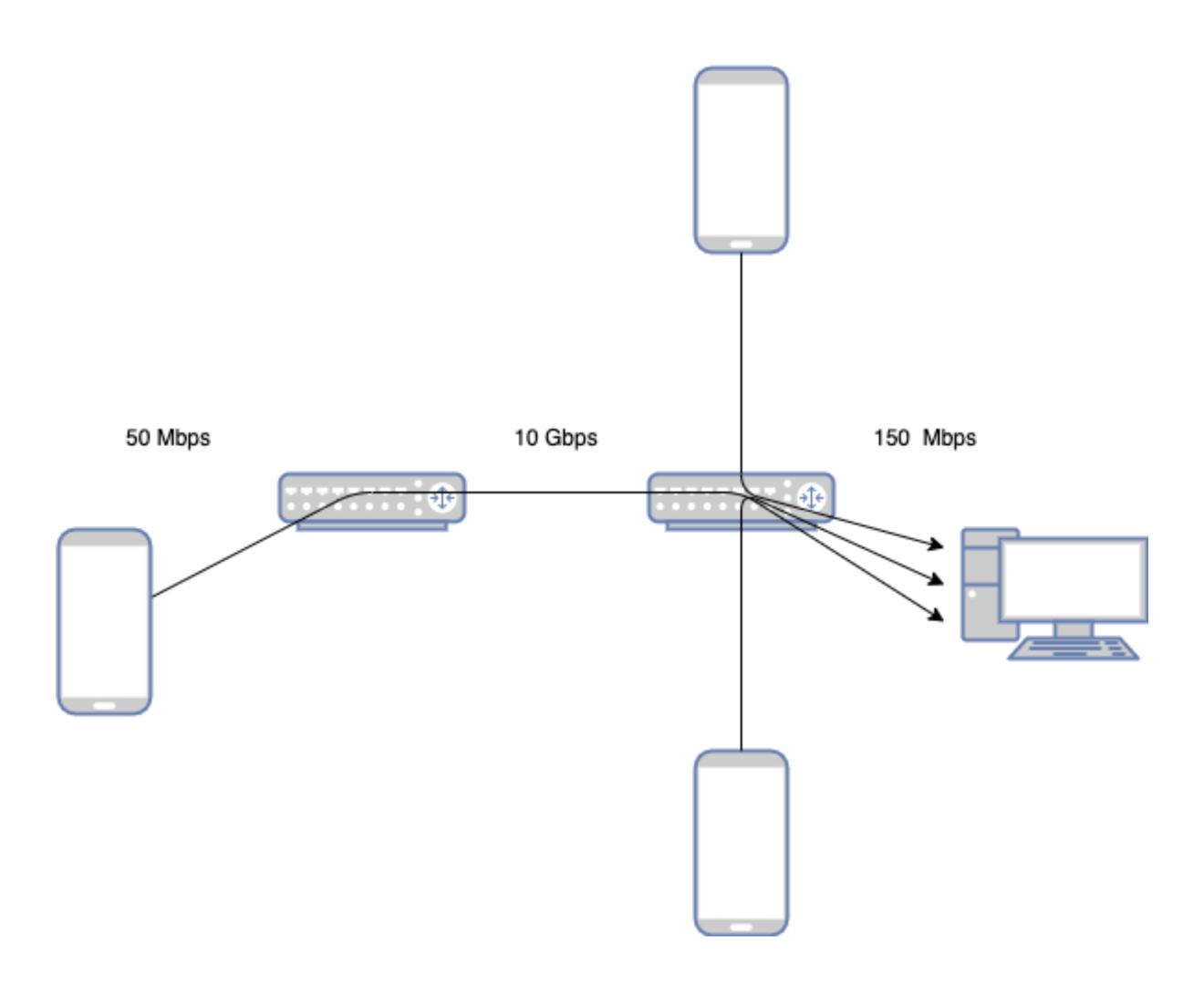


Problem: What bitrate do I upload?

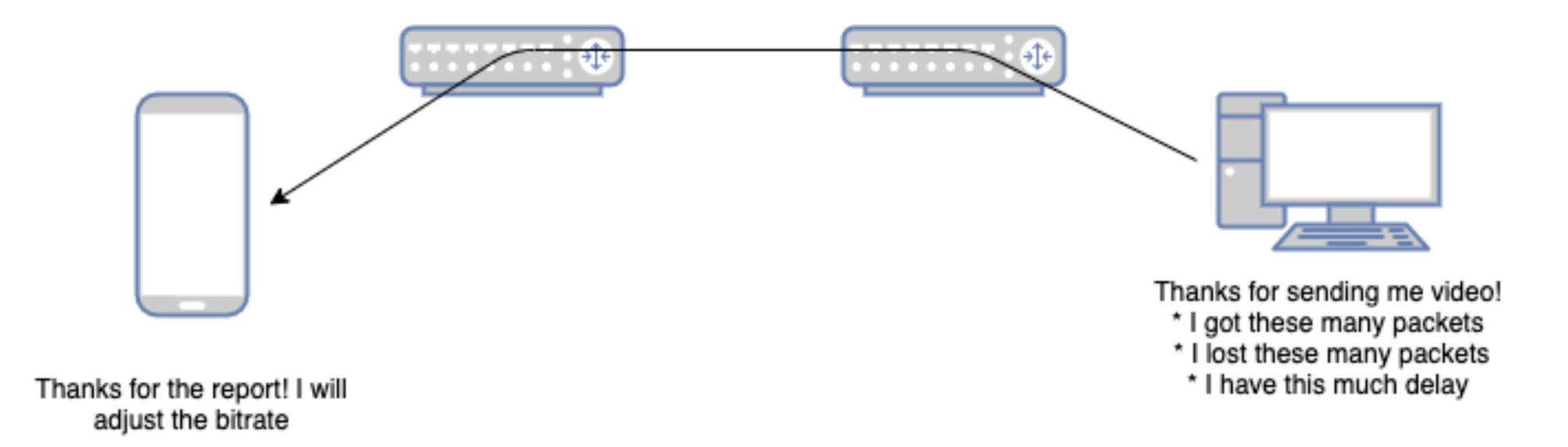




Problem: What bitrate do I upload?



Solution: Congestion Control





Problem: connect without knowing IP?

I want to connect to Laptop A

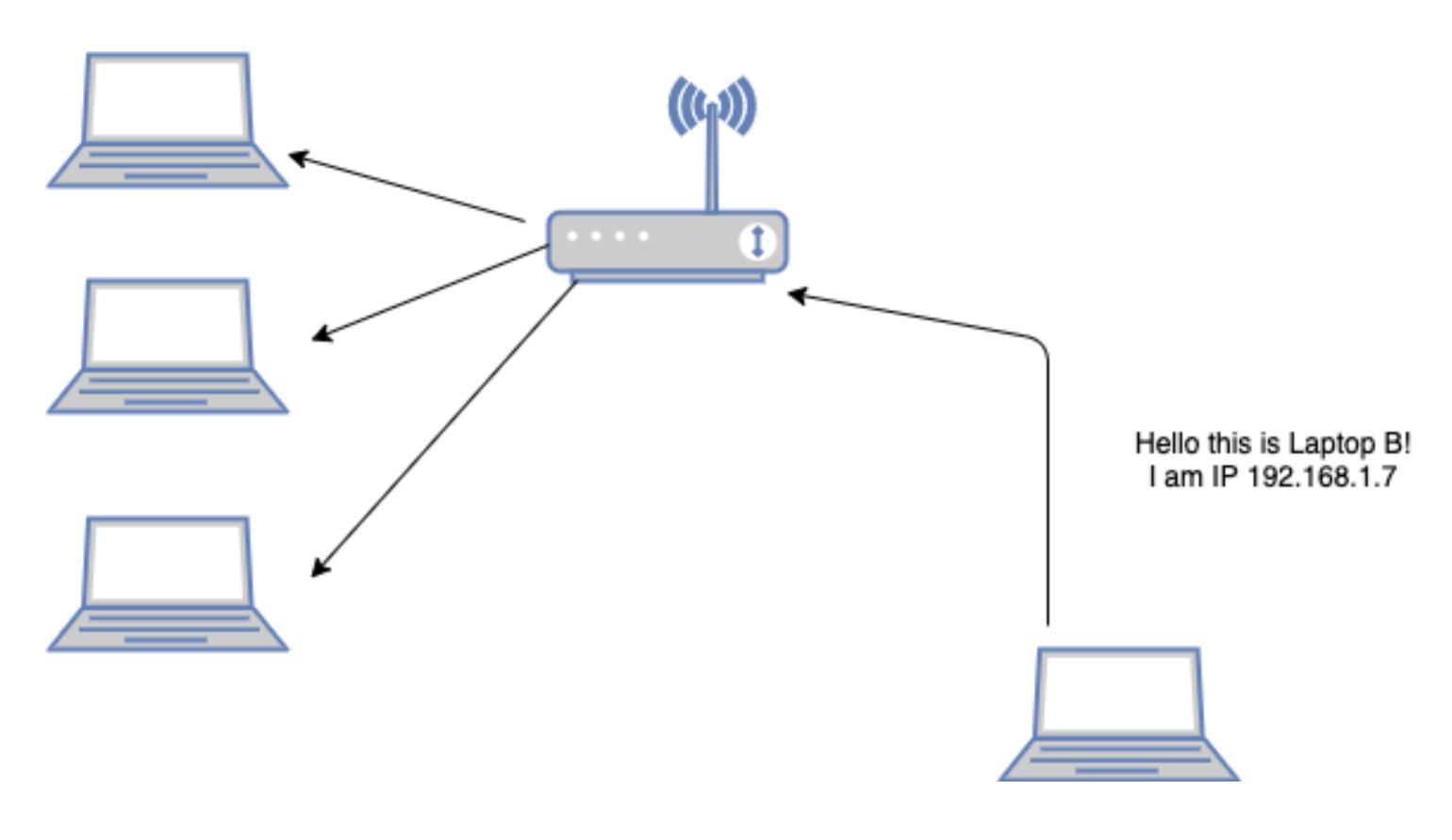


I want to connect to Laptop B





Solution: Multicast DNS







Connecting (Signaling Offer/Answer)

```
ackage main
import (
    "github.com/pion/webrtc/v2"
func main() {
   peerConnection, err := webrtc.NewPeerConnection(webrtc.Configuration{})
   if err != nil {
        panic(err)
   offer, err := peerConnection.CreateOffer(nil)
   if err != nil {
        panic(err)
   err = peerConnection.SetLocalDescription(offer)
   if err != nil {
        panic(err)
   // send Offer to remote PeerConnection via any protocol
    // receive Answer from remote PeerConnectian
   answer := webrtc.SessionDescription{}
   err = peerConnection.SetRemoteDescription(answer)
   if err != nil {
        panic(err)
    // You are connected!
```

Sending Data (DataChannels)

```
datachannel, err := peerConnection.CreateDataChannel("my-fun-channel", nil)
if err != nil {
    panic(err)
}

datachannel.OnOpen(func() {
    err = datachannel.SendText("Hello World!")
    if err != nil {
        panic(err)
    }
})
}
```

Receiving Data (DataChannels)

```
peerConnection.OnDataChannel(func(datachannel *webrtc.DataChannel) {
    datachannel.OnOpen(func() {
        fmt.Printf("New stream %s \n", datachannel.Label())
    })

    datachannel.OnMessage(func(msg webrtc.DataChannelMessage) {
        fmt.Printf("%s \n", msg.Data)
    })
})
```

Deploy to the browser!

Send Video

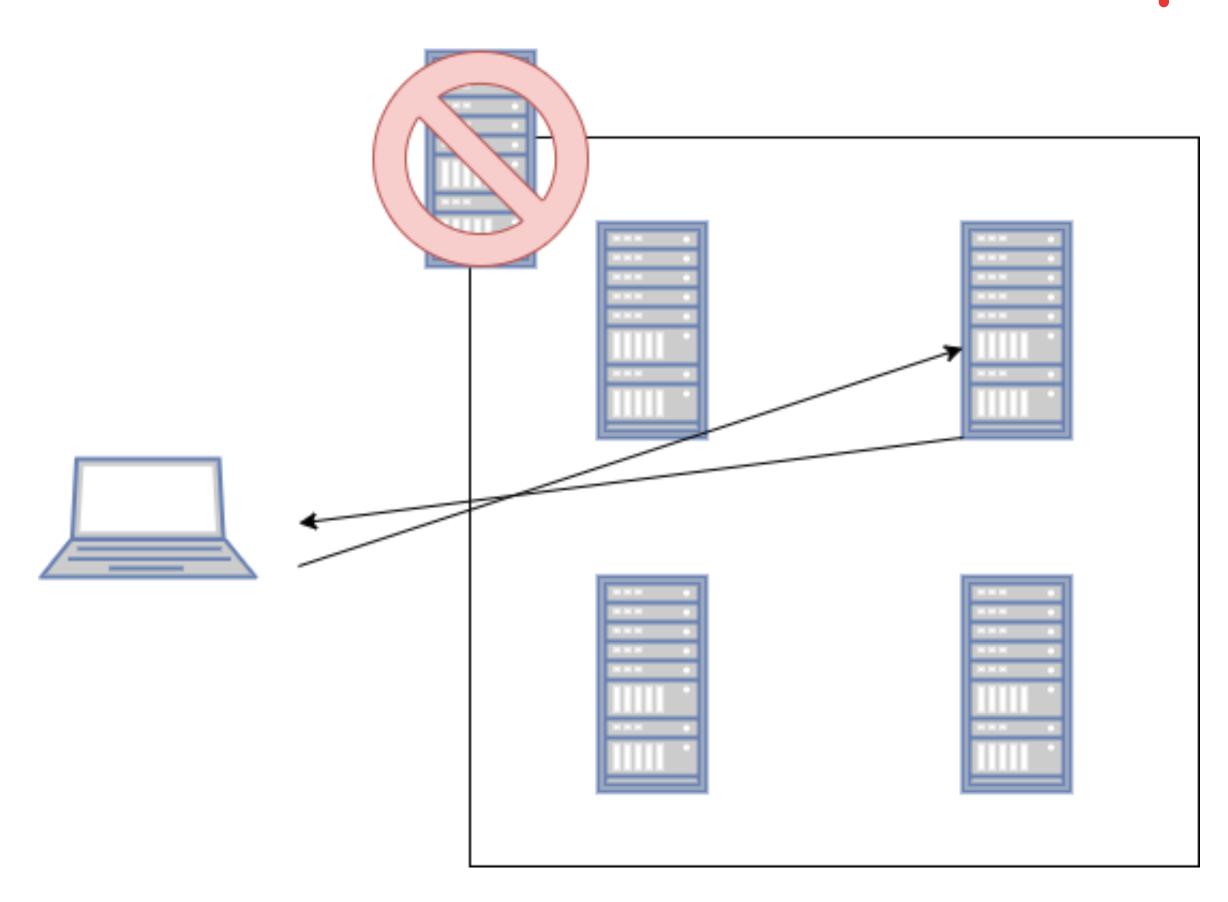
```
videoTrack, err := peerConnection.NewTrack(webrtc.DefaultPayloadTypeVP8, 50000, "video", "pion")
if err != nil {
    panic(err)
_, err = peerConnection.AddTrack(videoTrack)
if err != nil {
    panic(err)
for {
    frame, _, err := ivf.ParseNextFrame()
    if err != nil {
        panic(err)
    err = videoTrack.WriteSample(media.Sample{Data: frame, Samples: 900000})
    if err != nil {
        panic(err)
```

Receive Video

```
peerConnection.OnTrack(func(track *webrtc.Track, receiver *webrtc.RTPReceiver) {
    if track.Codec().Name == webrtc.Opus {
        for {
            packet, err = track.ReadRTP()
            if err != nil {
                panic(err)
            }
            // Playback using Media library of your choice
        }
    }
})
```

PION IN ACTION

ssh-p2p github.com/nobonobo/ssh-p2p



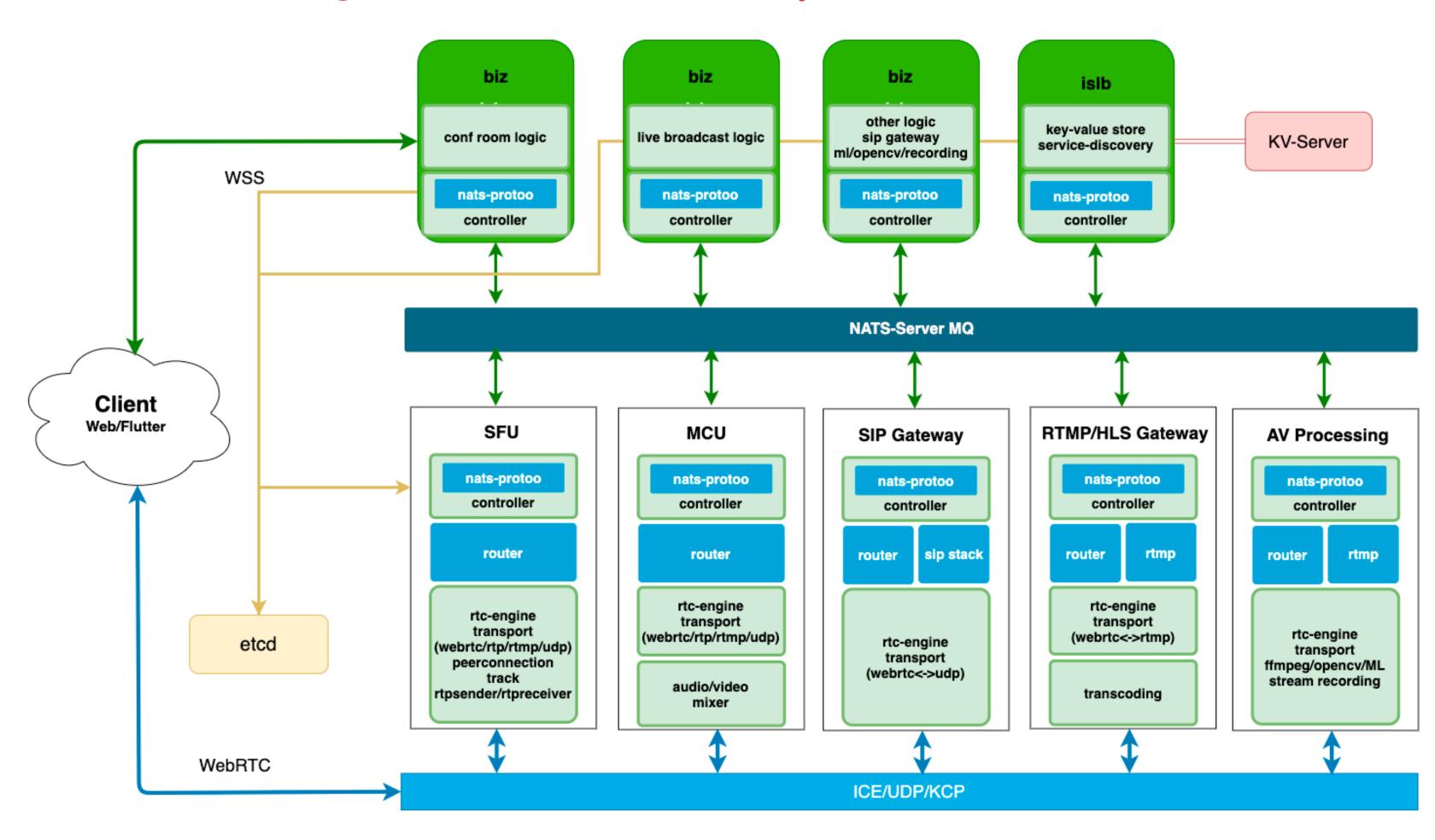
CloudRetro.io



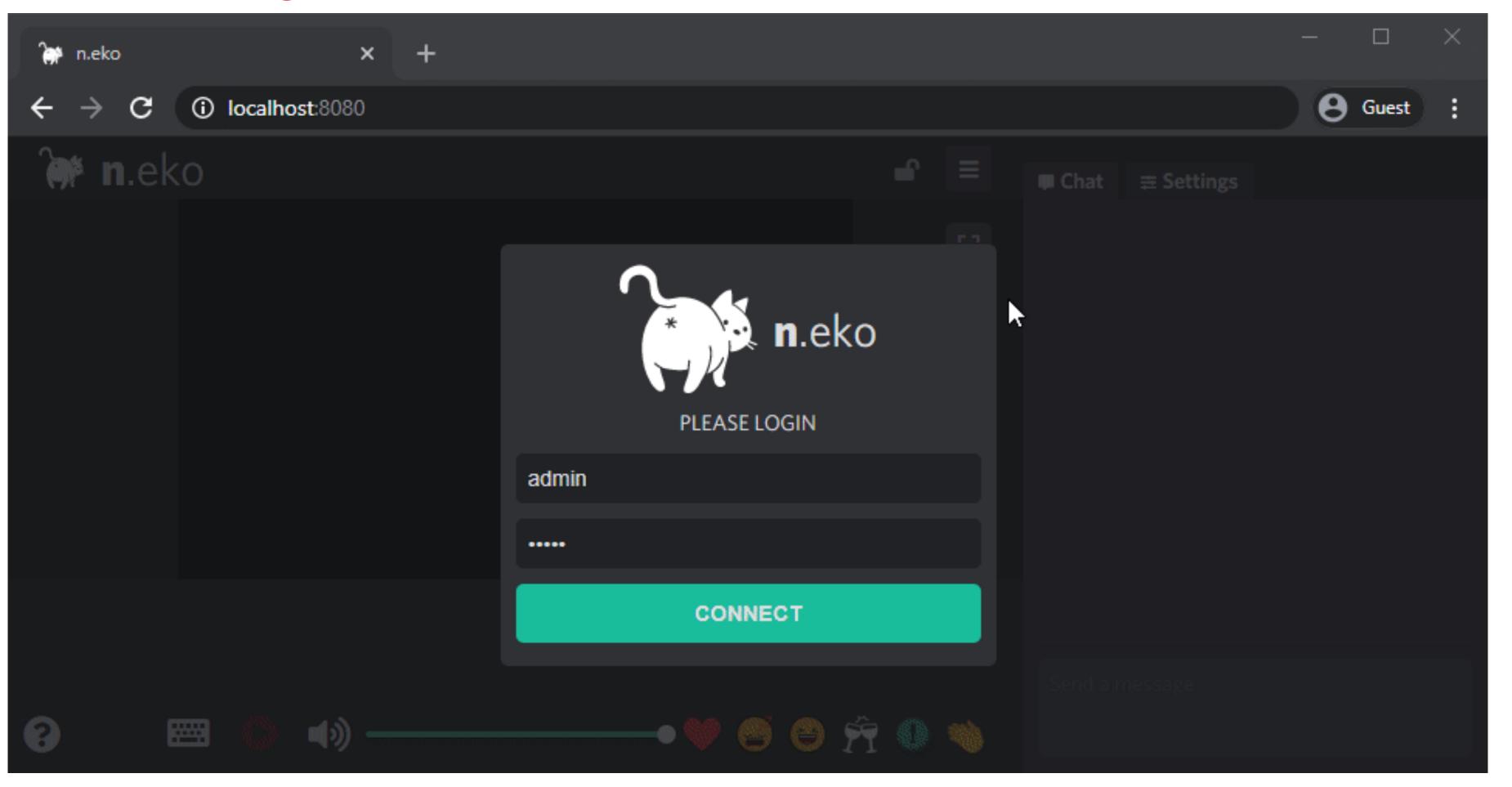
RTCTunnel



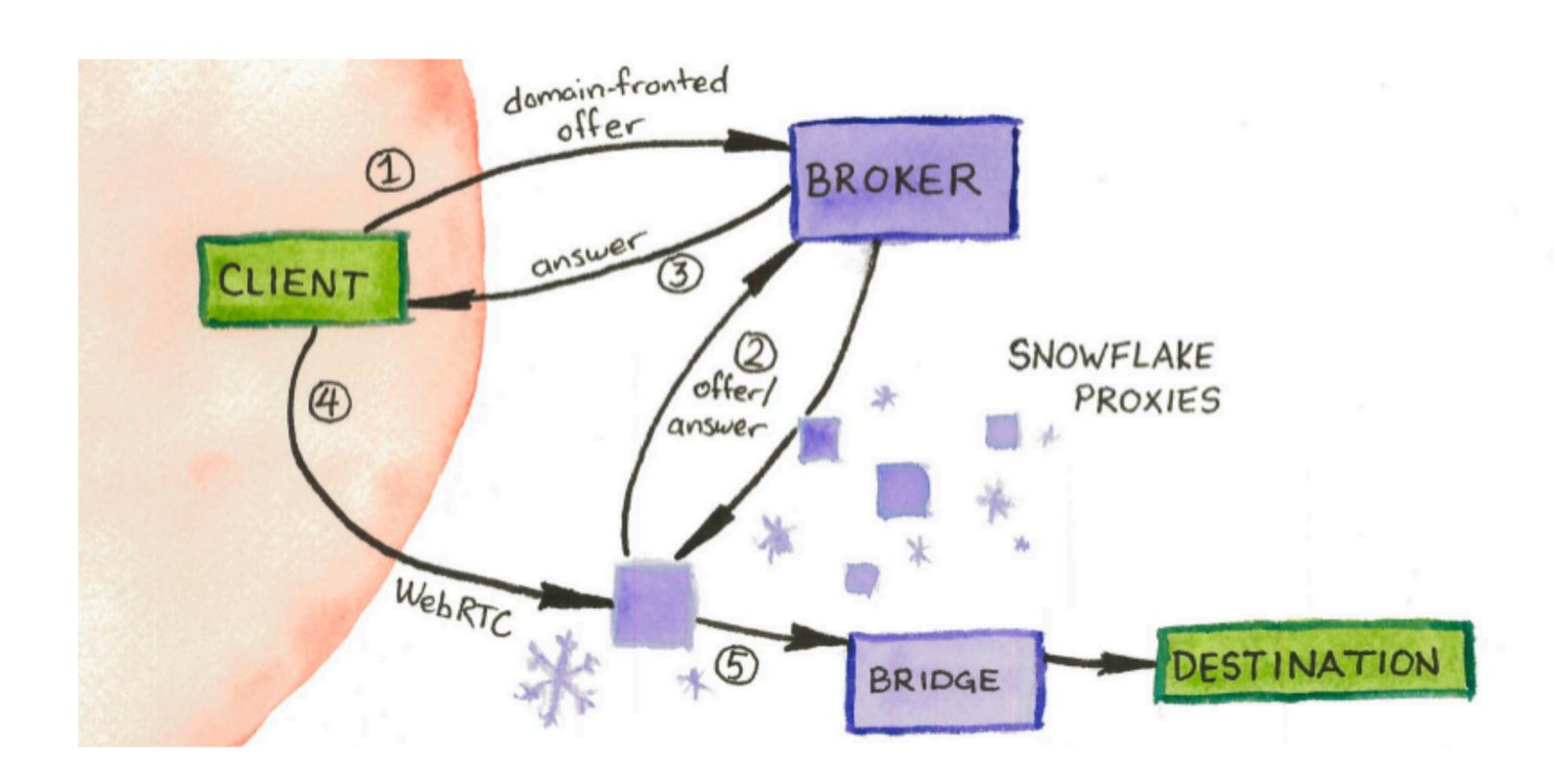
lon github.com/pion/ion



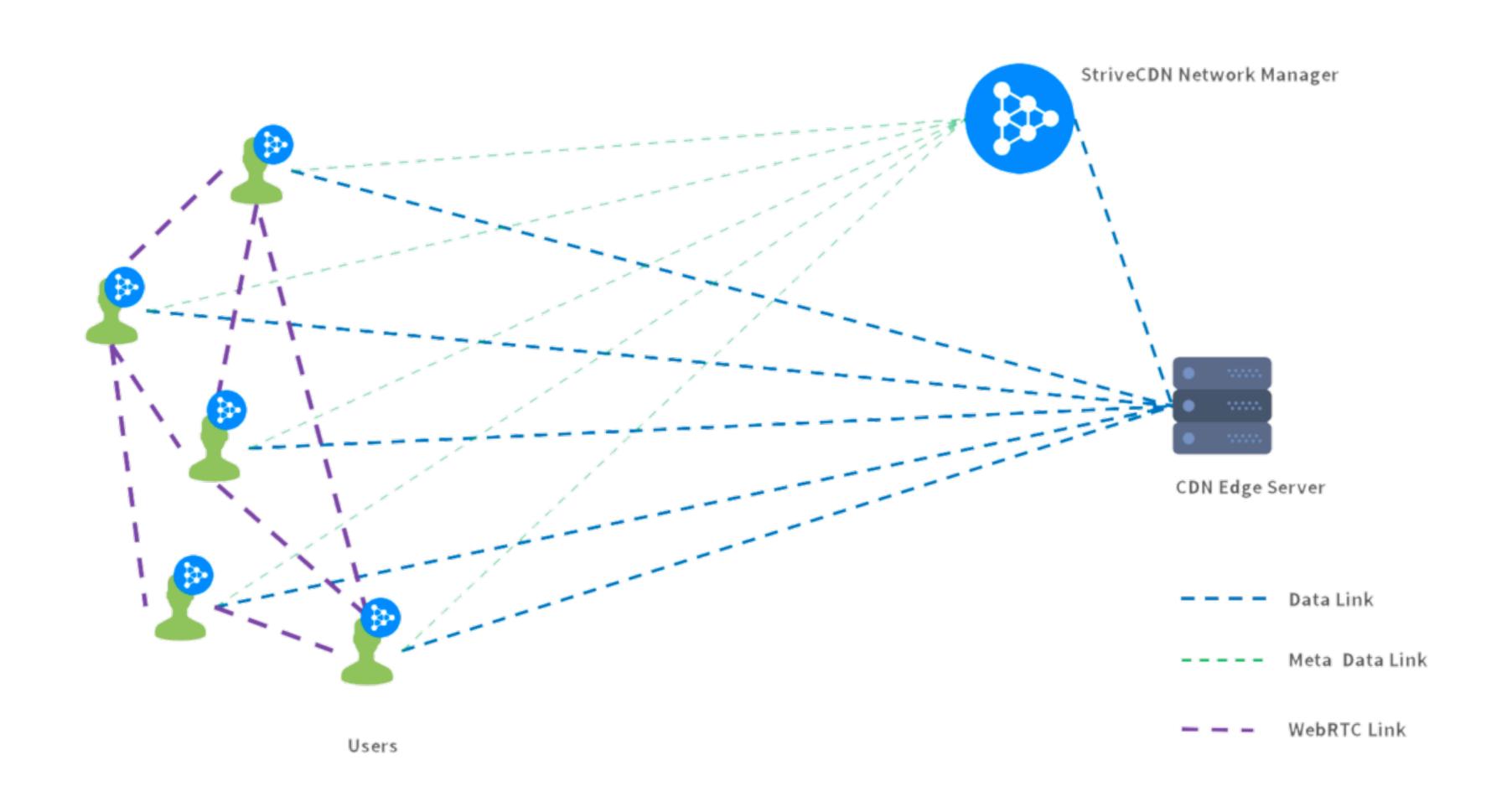
Neko github.com/nurdism/neko



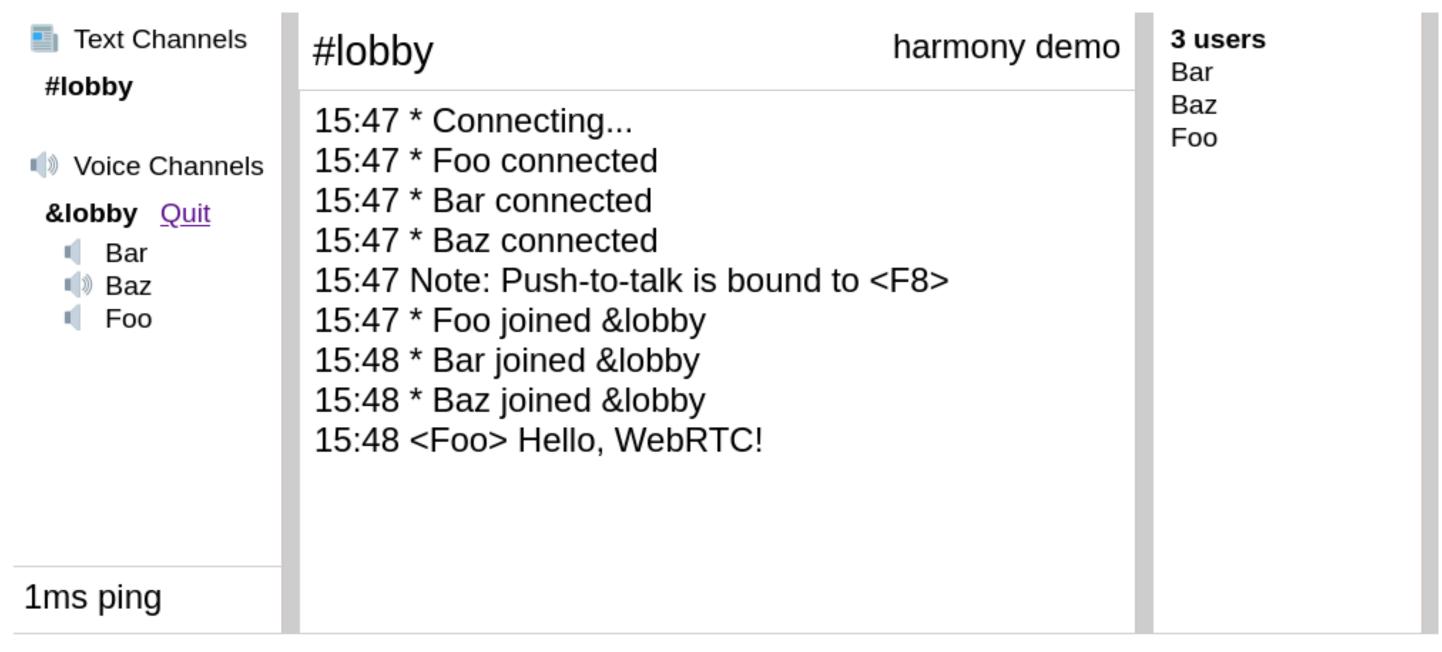
Snowflake snowflake.torproject.org



Strive CDN



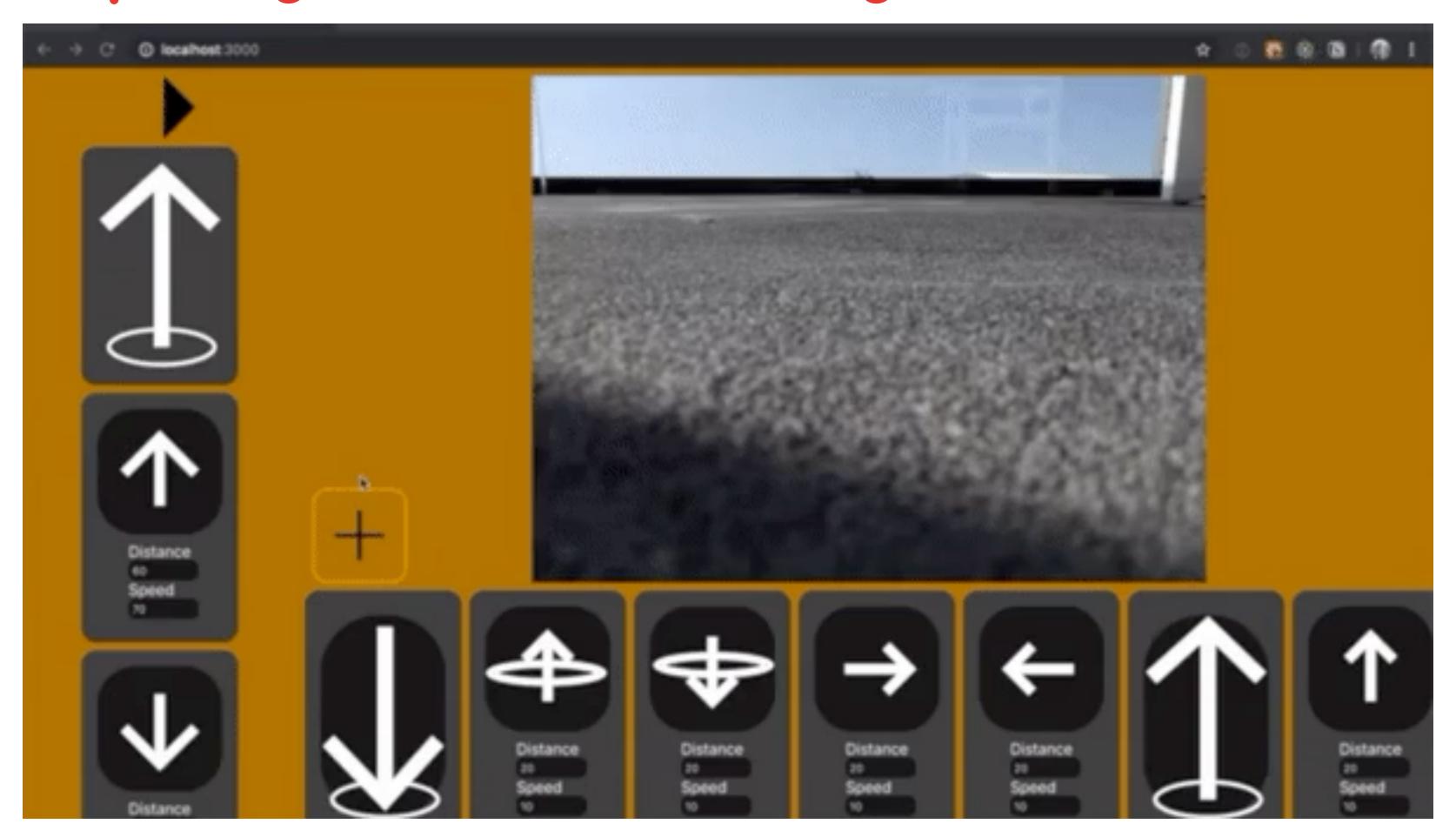
Harmony gitlab.com/tslocum/harmony



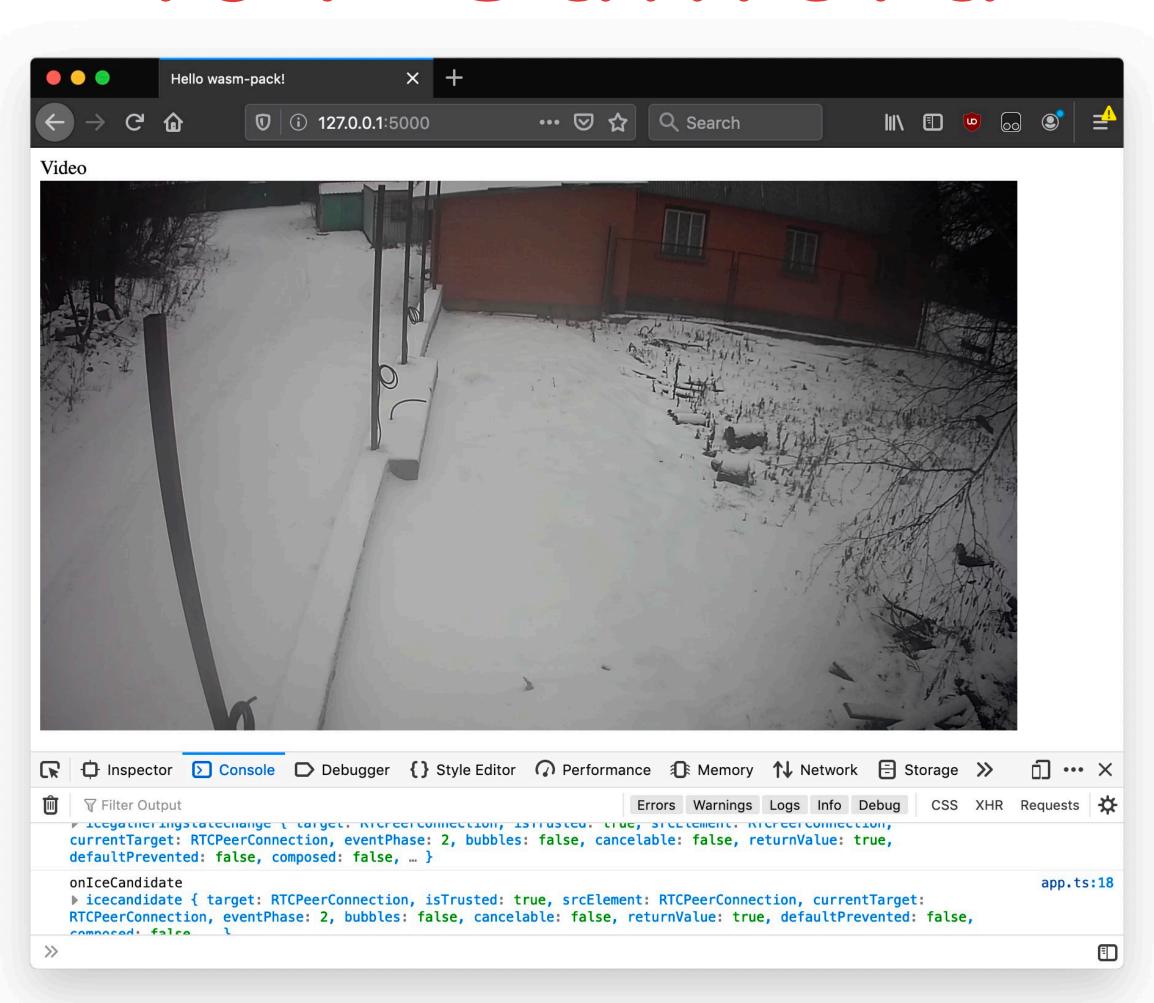
Message #lobby

TelloGo

https://github.com/Ragnar-H/TelloGo



loT Camera



GSCII

https://github.com/dialup-inc/ascii



Pion needs you

Empower those helping the internet!

Gain deep WebRTC knowledge

A fun challenge where you pick the goals



github.com/pion

pion.ly/slack

twitter.com/_pion

