

# **Videoconferencing Security and Privacy**

An Overview of Best Practices

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**WP8-T1** 

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## **Videoconferencing (VC) Systems**

- Not only video and audio
  - Screen sharing (for presentations)
  - Collaboration (editing of documents)
  - Text chat
  - File transfer
  - Integration into task management software, etc.
- Multitude of products
  - Zoom, Webex, Whereby, Pexip, Facebook, Google Meet, Microsoft Teams, ...
- Have become indispensable tools in pandemic lock-downs

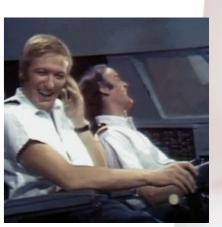




# Do you hate your Firewall Administrator?

- Prank: Lets play a round of "annoy-the-admin"
  - Go to your network administrator when they are a bit stressed
    - Just a little tense, but still willing to answer questions
  - Ask her/him: "I need your help with getting this videoconferencing app to work over our firewall/NAT-gateway"
    - Use a currently unsupported/unknown one, of course
  - Duck!
- Just kidding, of course
  - No responsibility taken for lost accounts, etc. :)
  - You owe them something next sysadmin appreciation day (at least)
- But if this sounds familiar, why is it so?





#### Videoconferencing as seen from the Network

- A multitude of protocols needed
  - Call setup, control: SDP, SIP(S), H.320, H.323, ...
  - Video/Audio/Data transfer: RTP(S), RTCP(S), (S)RTSP, RTMP(S), ...
  - Resource Reservation: RSVP, DCCP, ...
  - NAT/Firewall Traversal: ICE, STUN, TURN, H.460, ...
  - More proprietary protocols
- So far, time-consuming, but doable
  - Multiply by the number of VC systems in use







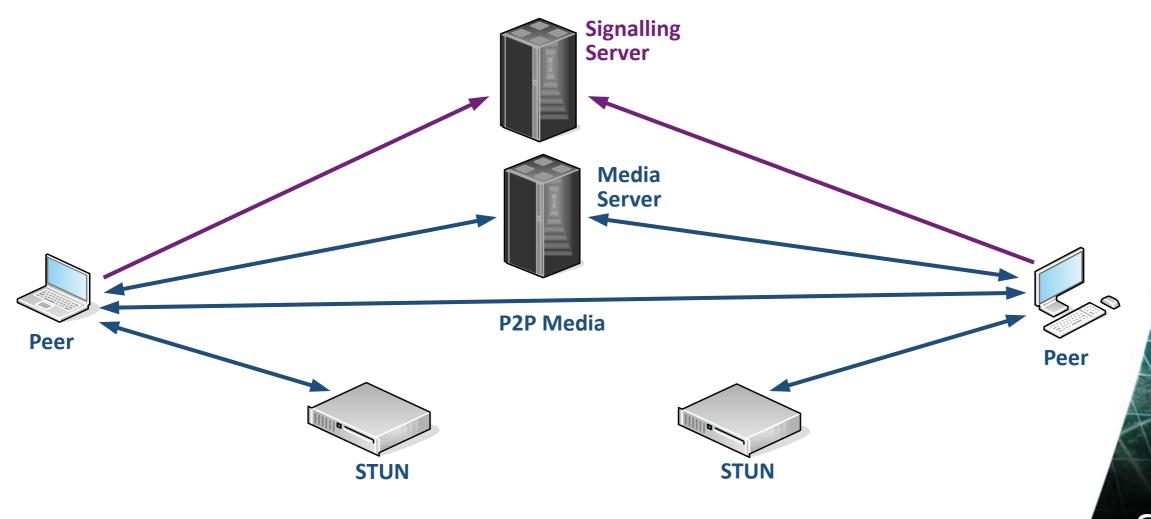
## **Videoconferencing & Security Headaches?**

- VC Data transfer protocols (RTP & Co)
  - Have to be reachable on random ports from the internet
  - No problem if participants are directly connected
- Firewalls and NAT gateways get in the way
  - Stateful Packet filters and NAT Gateways will block incoming packets
  - Unless the client has send a packet in the opposite direction before
  - But the client doesn't know from where a call will be coming
  - In other environments, firewall blocks all traffic, except to/from proxies
- Solution: NAT (and Firewall) traversal protocols
  - Interactive Connection Establishment (ICE)
  - Basically a combination of STUN and TURN





# **Videoconferencing Connection Diagram (w/o NAT)**







# **Session Traversal Utilities for NAT (STUN)**

- Simplified: An elaborate way of knowing your external IP-address
  - Think of whatsmyip.com, but without http(s)
- Tells also if you can be reached at a given port number
- Insider attack scenario
  - Spoofing STUN messages to expose ports of other internal systems
    - Using IP-addresses and ports of other internal systems
  - Should not happen if STUN is authenticated properly
- Information disclosure
  - STUN server knows about your internal IP-Addresses
  - Can you trust external operators (VC provider, Google, Cloud\*)?





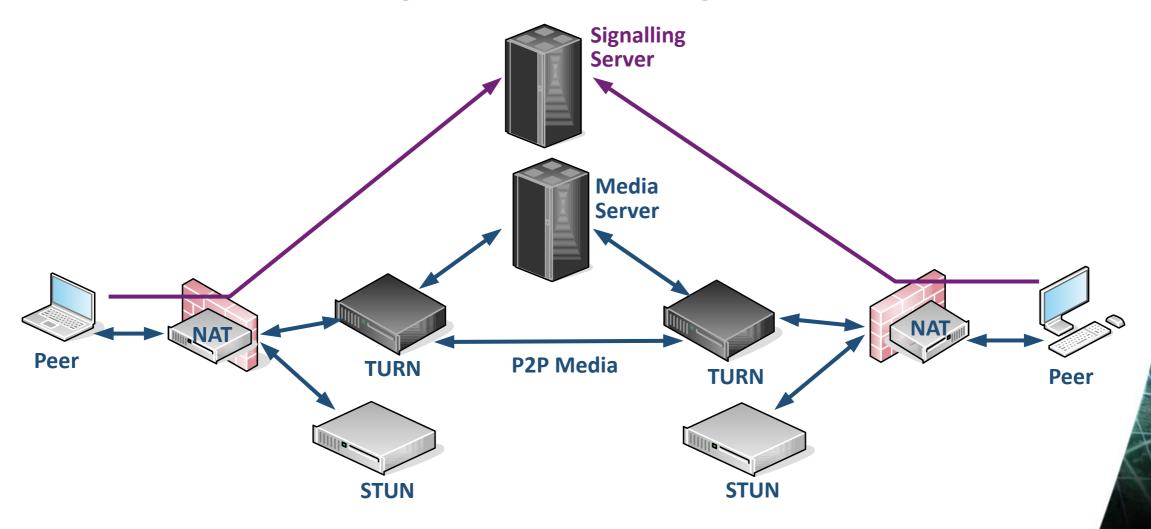
# **Traversal Using Relays around NAT (TURN)**

- Think of it as a sort of reverse proxy for VC clients
  - Client connects/registers with (local) TURN server
  - TURN server has public IP-address, can thus accept incoming calls
  - Often from other TURN servers
  - Knows where to relay the call to
- Usually run by the videoconferencing provider
- If run locally
  - Can be used to simplify/tighten firewall rules
    - Random ports/connections only to the TURN server
  - Internal IP-addresses stay undisclosed
  - TURN server needs to be in an unprotected network or DMZ





# Videoconferencing Connection Diagram (w/ NAT)







#### WebRTC

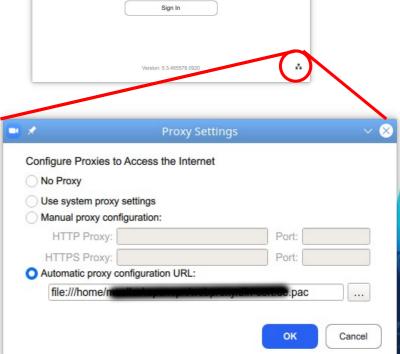
- Do all the VC client stuff from the browser
- May leak internal IP-addresses through STUN
  - See browserleaks.com WebRTC leak test
- Has the same connectivity problems as a VC appliance
  - Will work on public networks (i.e. your mobile phone/tablet)
  - Or behind simple NAT gateways (i.e. your home router)
- Not needed for many VC systems
  - If so, can turn WebRTC off in the browser
  - WebRTC Control Extension (all browsers)
  - about:config in Firefox-based browsers (see References)





## Videoconferencing Systems that don't use WebRTC

- These are the most common today
  - Zoom, WebEx, Whereby, ...
- Admins like them because
  - Use HTTPs for everything
  - All we need here are proper proxy settings
  - All traffic goes through their servers (no P2P)
- Things just work
- And we can pretend it's secure
  - Because HTTPS
  - And VC provider says so



zoom

Join a Meeting





# **Example: GÉANT Zoom Session**

As seen from a client

Signalling/Call setup

```
# ss -antp
State
            Recv-Q
                    Send-Q
                              Local Address:Port
                                                     Peer Address:Port
                                                                            users:(("zoom",pid=12161,fd=93))
ESTAB
                              10.0.6.4:59544
                                                     185.174.117.137:443
                                                                            users:(("zoom",pid=12161,fd=34))
CLOSE-WAIT
                              10.0.6.4:47010
                                                       52.202.62.244:443
                                                                            users:(("zoom",pid=12161,fd=91))
ESTAB
                              10.0.6.4:59540
                                                     185.174.117.137:443
                                                                            users:(("zoom",pid=12161,fd=92))
ESTAB
                              10.0.6.4:59542
                                                     185.174.117.137:443
ESTAB
            0
                                                                                     "zoom",pid=12161,fd=90)
                              10.0.6.4:59538
                                                     185.174.117.137:443
ESTAB
                              10.0.6.4:59536
                                                     185.174.117.137:443
                                                                                     "zoom",pid=12161,fd=70))
                                                                            users:(("zoom",pid=12161,fd=67))
ESTAB
                              10.0.6.4:59534
                                                     185.174.117.137:443
ESTAB
                                                                            users:(("zoom",pid=12161,fd=65))
                              10.0.6.4:59530
                                                     185.174.117.137:443
# host 185.174.117.137
137.117.174.185.in-addr.arpa domain name pointer mmr137-117.zoom.nordu.net.
# host 52.202.62.244
244.62.202.52.in-addr.arpa domain name pointer ec2-52-202-62-244.compute-1.amazonaws.com.
```

Relay/Streaming)





#### **Encryption**

- Protects the confidentiality and integrity of exchanged data
- Most desirable: End-to-End Encryption (E2EE)
  - Streams get encrypted at the sender and decrypted by the receiver
- Second best: Transport Encryption (most often: TLS)
  - Stream is encrypted on the way to the server, but unencrypted there
- Watch out, however
  - Is encryption used by default?
  - And for what streams?
  - What crypt-algorithms/key lengths are used?
  - Forward Secrecy?
  - Key management?







#### What Meeting Attendees can do

- Keep your Videoconferencing software Up to Date
- Configure your audio/video settings
- Beware of your surroundings
- Don't share Invites
- Don't make recordings



## **Keep the VC Software Up-to-Date**

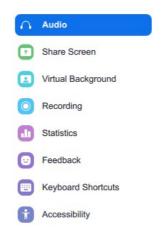
- On self-administered systems
  - Lookout for Updates regularly (at least once/week)
  - Use features from client SW for auto-updates
    - Like checking automatically when the client is started
  - Use Update channels from the OS Vendor
  - Use Update Manager if both of the above not available
    - I.e. "SUMo", "Patch My PC", etc.
  - Alternatively (if possible): Use Web-Client (i.e. web-browser)
    - Pro: Browsers get timely updates (~ 1/month)
    - Con: Browsers huge attack surface
- When told to patch or update: Do it as soon as possible!
- Don't forget to restart the software after updating

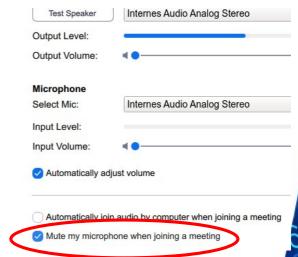




# Configure the Audio/Video Settings appropriately

- When joining a video conference
- Client Preferences should be set to
  - No Video
  - Audio Mute
- Enable only when you speak
- You may accidentally leak sensitive/private information
- If you want to be extra secure
  - Cover up camera (tape/shutter)
  - Use a microphone with a true hardware OFF switch
  - Use headset instead of speaker











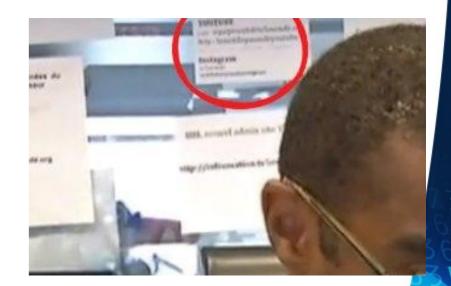
#### **Beware of your Surroundings**

- TV5Monde hack anyone?
  - French TV station got hacked in 2015
  - Password visible in background during broadcast (TV, not VC though)
  - Station was taken off the air (temporarily)



- Empty black-/whiteboard, pinwands, monitors, etc.
  - Plain color works best, esp. with virtual backgrounds
  - Otherwise, parts of the real background may shine through
- Beware of mirrors and mirroring surfaces behind you!
- This rule applies to Audio too!
  - Close the door/windows
  - Activate background noise reduction for the microphone (if possible)







# **Beware of your Screen(Background)**

- To prevent leaking information during screensharing
- Limit Sharing to one application window
- Share whole desktop only when needed
  - I.e. (full-screen) presentations
  - Multi window demonstrations
- Put all shared programs on one screen/desktop
  - When using multiple screens/virtual desktops
- Close all other programs not needed during the conference
  - You may accidentaly expose data on these windows
    - Including window titles
  - Taskbar shows info/preview on mouseover
    - Even for windows on other screens/desktops









#### **Don't share Invites**

- When sharing invites publicly, it's like inviting everybody
- OK, then everybody can come
  - Inevitably, people will come that don't belong here
  - C.f. "Zoombombing"
- Ask the organizer if you want to bring along others





#### Don't take screenshots

- Or make audio/video recordings, or record text chats
- Likely against the law if meeting is non-public
  - Consequences will depend on the legislation in your country
  - But this is usually not a petty offense
- Other laws and disciplinary action may also apply
  - Violation of privacy laws will probably be your least problem
- Yes it can be done without the others noticing
  - When you use other tools than the VC client
- Ask for permission (opt-in of all attendees)
  - One attendee who does not consent may mean you're not allowed to
- No streaming/recording without permission either





# What Meeting Organizers can do

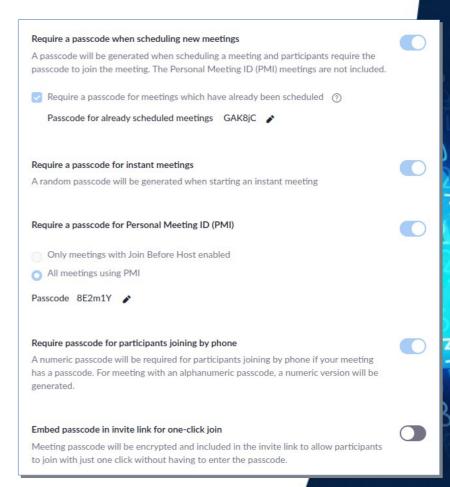
- Everything an Attendee should do
- Require authentication for joining meetings
- Review attendees
- Eliminate disruptors
- Lock the conference
- Inform if recording
- Disable screenshots/recordings





#### **Require Authentication for Conferences**

- When setting up a conference, require passwords or other authentication
- Room numbers may not be very random
  - And a name like "my meeting" certainly is guessable
  - Zoom meeting room numbers used to be 9 digits
  - Attackers have no trouble trying a couple of billion numbers
    - Scripts & botnets
- If the name/id/number is public, choose a good password
- Consider distributing conference link and password through different channels







#### **Review Attendees**

- Make sure that no one is on the list that is not invited
- Visually identify known persons (for sensitive matters)
- If there's someone your don't know ask what she's doing here
- Throw them out, unless they have reason to be here
- Careful though, esp. with shared meeting spaces
  - You had reserved this room for this time slot, right?







## **Eliminate Disruptors / Lock the Conference**

- If people misbehave in a conference kick them out
  - Ban them from re-joining (if supported)
  - Or revoke their credentials
- Locking: Nobody can enter the conference anymore
  - Keeps out unwanted attendees
  - Prevents kicked/banned attendees from re-joining
  - However: late-comers can't join
  - Nor can those who had network problems and disconnected
  - You may want to watch your IM or E-mail to let them back in





# Inform beforehand if recording/streaming

- Otherwise, see attendees a few slides before
- Get consent!
- Good practice:
  - Tell why you are recording
  - What will be done with the recording
  - How long will the recording be kept
  - Where will it be kept
  - Who will have access to it
  - Etc.
- If the recording contains sensitive data, secure access/storage
  - Authentication, Encryption







# **Disable Screenshots/Recordings**

- Organizer can disable screenshots/recordings for attendees
- Affects only the client software,
- Attendees may record with other tools
- However, it makes the policy clear
- Also: wipe shared whiteboards, etc.
  - After the meeting ends
  - If on providers VC cloud space
- If third-party tools are required, inform beforehand about it
  - So needed programs/browser-extensions can be tested/reviewed







## What have you learned?

- Why configuring networks for videoconferencing may be hard
- What to look for when selecting a good VC software/provider
- What you have to do to keep video conferences secure
  - As an organizer
  - As an attendee







# Thank you

Any questions?

Next module: Office Security & Privacy

30<sup>th</sup> of September 2020

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#### References

- "A Norwegian school quit using video calls after a naked man 'guessed' the meeting link" https://techcrunch.com/2020/03/26/norwegian-schoolwhereby
- Tips for Secure Video Conferencing (Zip file to download)
   https://www.sans.org/sites/default/files/2020-04/Video
  %20Conference%20Tips%20Rev.zip
- SuMO Update Manager: http://kcsoftwares.com/index.php?sumo
- Patch My PC Update Manager: https://patchmypc.com/
- WebRTC settings for Firefox: https://wiki.mozilla.org/Media/WebRTC/Privacy
- WebRTC Control Extension: https://mybrowseraddon.com/webrtc-control.html
- Real-Time Messaging Protocol (RTMP)/ Secure RTMP (RTMPS) Specification, V1.0, https://www.adobe.com/devnet/rtmp.html







#### **RFCs**

- Session Initiation Protocol (SIP), Secure SIP(SIPS), RFC 3261: https://tools.ietf.org/html/rfc3261
- Session Description Protocol (SDP), RFC 4566: https://tools.ietf.org/html/rfc4566
- Real-time Transport Protocol (RTP), RFC 3550: https://tools.ietf.org/html/rfc3550
- Secure RTP (RTPS), RFC 3711: https://tools.ietf.org/html/rfc3711
- RTP Control Protocol (RTCP), RFC 3550: https://tools.ietf.org/html/rfc3550
- Secure RTCP (RTCPS), RFC 3711: https://tools.ietf.org/html/rfc3711
- Real Time Streaming Protocol (RTSP), RFC 7826: https://tools.ietf.org/html/rfc7826
- RTSP over TLS (RTSPS), RFC 7826: https://tools.ietf.org/html/rfc7826
- Resource ReSerVation Protocol (RSVP), RFC 2205: https://tools.ietf.org/html/rfc2205
- Datagram Congestion Control Protocol (DCCP), RFC 4340: https://tools.ietf.org/html/rfc4340
- Session Traversal Utilities for NAT (STUN), RFC 5389: https://tools.ietf.org/html/rfc5389
- Traversal Using Relays around NAT (TURN), RFC 5766: https://tools.ietf.org/html/rfc5766
- Interactive Connectivity Establishment (ICE), RFC 8445: https://tools.ietf.org/html/rfc8445





## Standard Port Numbers for SIP, STUN & TURN

- SIP: 5060 UDP & TCP
- SIPS: 5061 UDP & TCP
- STUN, TURN: 3478 UDP & TCP
- STUN, TURN over D(TLS): 5349 UDP & TCP



