The RTP bleed and what can we do?

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Peter Lemenkov Software Engineer, Red Hat

What is it and how it affects us?



- The RTP bleed (https://www.rtpbleed.com) intially made public by Sandro Gauci (https://twitter.com/sandrogauci) and Klaus-Peter Junghanns (https://github.com/kapejod)
- Allows almost anyone to redirect media streams.
- Cannot be mitigated in 100% cases

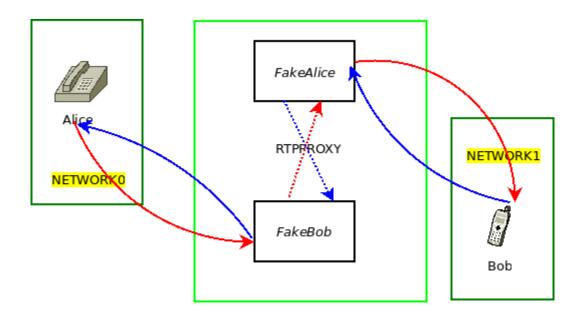
POTS RTP bleed



Why it was introduced in the first place?

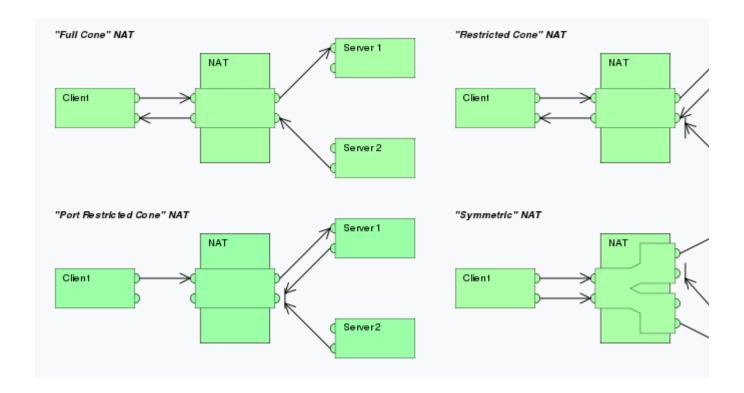
- Lots of legacy with poor security
- We have to deal with NAT somehow

RTP proxying



- Multiple implementations
- Most simple case there are more (transcoding, etc).
- RTP proxy is visible to both callers
- RTP proxy allocateis ports and resends data to calling parties
- Who actually sends data to these ports? The same question as in POTS.

Avoid RTP proxying with STUN?



• No.

RTP structure

• That's all we have.

SRTP structure

Extends RTP with authentication

What can we do to mitigate it using current architecture?

- In short we can't do much
- Remembering last IP? Doesn't work with 4G networks
- SSRC also could change although this looks to be rare
- No signature of any kind
- Even if possible to "sign" (SRTP), then what to do with legacy devices?

Will future Voice / Video over IP systems be affected?

- IPv6
- *Encryption* (SRTP + MIKEY, ZRTP, SDES, DTLS-SRTP)

Thank you

Peter Lemenkov Software Engineer, Red Hat lemenkov@gmail.com(mailto:lemenkov@gmail.com) https://github.com/lemenkov(https://github.com/lemenkov)