

Lecture 18 - Voice Over IP

Exercise 1: Why can't the speech and video stream be transmitted using only UDP? Why might we want to avoid transmitting them over TCP?

- UDP does not guarantee in-order delivery or indication of missing packets
- TCP - too much overhead
 - TCP adds delay to adjust for network congestion (flow control)

Exercise 2: How many bytes of header overhead are added to each packet assuming the smallest possible IP, UDP and RTP headers? If 64 kb/s PCM is being transmitted in 20 ms frames, what is the total data rate, including both headers and speech data? What fraction of that is for headers?

IP : 4(5) : $5 \times 4 = 20$ bytes for IP
UDP : 2×32 bits : 8 bytes for UDP
RTP : " : 8 bytes for RTP

36 bytes = headers

$20\text{ms} \times 64\text{ kb/s} = 160$ bytes = payload

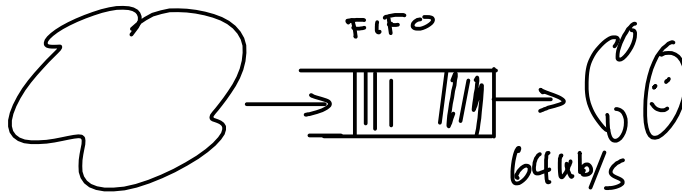
each packet is 196 bytes ; rate $\frac{196 \times 8}{20 \times 10^{-3}} = 78\text{ kb/s}$.

overhead : $\frac{78-64}{78} = \frac{14}{78} \approx 18\%$ overhead.

Exercise 3: Assuming the minimum header lengths, which has less overhead, TCP or RTP?

$$\begin{aligned} \text{difference} &= 5 \times 4 \text{ bytes for TCP} - 2 \times 2 \text{ bytes for RTP} \\ &= 3 \times 4 = 12 \text{ bytes} \end{aligned}$$

for a 196 byte packet.



Exercise 4: If the sample rate is 8 kHz and each sample is quantized with 8 bits, what is the bit rate in each direction?

$$8 \text{ bits/sample} \times 8000 \text{ samples/second} = 64,000 \text{ bits/second}$$

Exercise 5: What is the maximum bandwidth and the bit rate if the sampling rate is 16 kHz and there are 10 bits per sample?

$$\text{Nyquist rate} = 2 \times \text{bandwidth}$$

$$\text{max bandwidth} = 8 \text{ kHz}$$

$$\text{bit rate} = 16 \text{ kHz} \times 10 \text{ bits/sample} = 160 \text{ kb/s}$$

Exercise 6: Why can't trunks be bidirectional?

trunks are long & high loss & need to amplify signal in both directions

(this is difficult with full-duplex links)

Exercise 7: Which customer-facing interfaces provide which of the three services?

