

Recs. 18 & 19

Round-trip time (RTT) is a key parameter in transport protocols

RTT = Transmission time

to work through
the length of the data
packet and the ACK
packet = packet length (in bits), divided
by the transmission
rate (in bits/sec)

+ Propagation time

characteristic of the channel,
and can be comparable with
other components of RTT

= length of link (metres), divided
by speed of propagation (metres/sec)

+ Processing time

computation at switches
to run CRCs on headers,
read headers, etc.

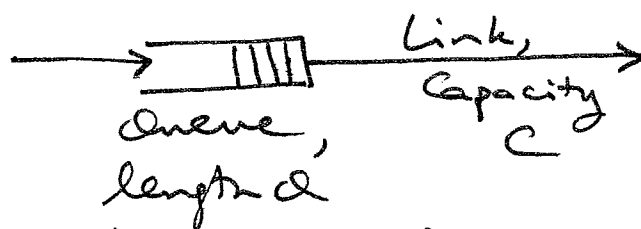
+ Queueing delay

associated with packets waiting
in queue to be processed and
sent on

averages $\rightarrow \left\{ \begin{array}{l} \text{Queue length (bits), divided} \\ \text{by processing rate (bits/sec)} \end{array} \right\}$ Little's law!

e.g. path with one bottleneck link
modeled as

(average)



By Little's law, delay in queue when link is transmitting at capacity is given by $\frac{Q}{C}$. So the delay across the link varies

with queue length; when $Q = 0$, the queue contributes no delay because packets are processed right away. Also, transmission rate $< C \Rightarrow Q = 0$.

$\therefore$ RTT on path (assuming just this one bottleneck link, i.e., no other links with capacity $\leq C$ & associated queues) is

$$RTT = \underbrace{RTT_{min}}_{\substack{\text{i.e., RTT} \\ \text{when } Q=0, \\ \text{which happens} \\ \text{when rate of} \\ \text{transmission is } < C}} + \underbrace{\frac{Q}{C}}_{\substack{\text{Queuing delay} \\ \text{associated with} \\ \text{transmission at} \\ \text{rate } C}}$$

Sending data to the input of the queue at rate $> C$ will cause queue to fill up and bits/packets to be lost, so input rate needs to be $\leq C$.

How does all this balance out when the sender uses a sliding-window protocol? $\Rightarrow$

Assuming no packets are lost, the sender's data rate is $\frac{W}{RTT}$ ← max. # of unacknowledged packets allowed in protocol.

So we need

$$\frac{W}{RTT_{\min} + \frac{Q}{C}} \leq C$$

$$\text{or } W \leq C \cdot RTT_{\min} + Q.$$

When $Q=0$, the round-trip time is $RTT_{\min}$, and the rate is $\frac{W}{RTT_{\min}} \leq C$.

When W crosses above $C \cdot RTT_{\min}$, the queue starts to be occupied. If $Q_{\max}$ is the maximum tolerable (average) queue length, then the largest tolerable window size is

$$W_{\max} \leq C \cdot RTT_{\min} + Q_{\max}$$

Above this, awkward/bad things start to happen!

(because we are using Little's law)

(to avoid bits being dropped from the queue)

Try Problems 9 & 10 in Chapter 20 of lecture notes.

EWMA → Filtering measured RTT to get a smoothed estimate sRTT, using an exponentially-weighted moving-average filter:

$$s[n+1] = (1-\alpha)s[n] + \alpha r[n] \quad (*)$$

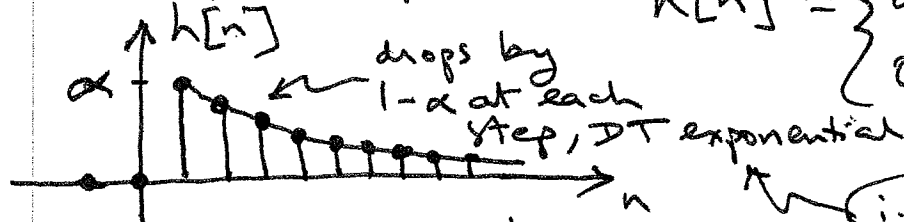
$$0 < \alpha < 1$$

Smoothed estimate
sRTT at time n

measured RTT
at time n

This is a causal LTI system, with unit sample response

$$h[n] = \begin{cases} \alpha (1-\alpha)^{n-1} & \text{for } n \geq 1 \\ 0 & \text{for } n < 1 \end{cases}$$



$$\text{So } s[n] = \alpha \sum_{m=-\infty}^{n-1} r[m] (1-\alpha)^{n-m-1}$$

$$= \alpha \left\{ r[n-1] + (1-\alpha)r[n-2] + (1-\alpha)^2 r[n-3] + \dots \right\}$$

i.e., EWMA.

System is stable. So if $r[n]$ goes to a value R and stays there, $s[n]$ asymptotically (exponentially) approaches a steady-state value S . What is S ? Determine by substitution in Eq. (*):

$$S = (1-\alpha)S + \alpha R \Rightarrow S = R, \text{ as desired.}$$

Getting a feel for $(1-\alpha)^n$:

$(1-\alpha)$ for $\alpha = \frac{1}{8}$ is $\frac{7}{8}$.

$(\frac{7}{8})^n < 0.5$ for $n \geq 6$, not before

$(\frac{7}{8})^n < 0.05$ for $n \geq 23$, not before

This explains the choice of the coefficient α in front of $r[n]$ in Eq. (*)