

p3.

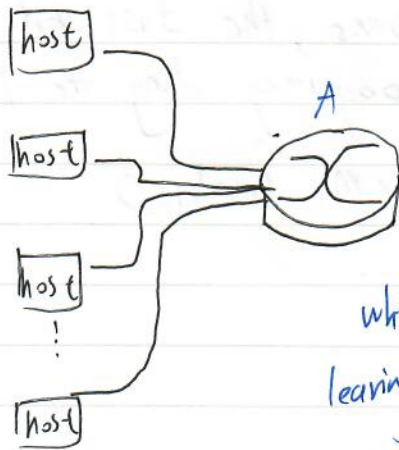
a. circuit-switched network would be more appropriate.

Reason: The application will continue running for a long period of time meanwhile need steady transmission.

When it use circuit-switched networks, the resources would be reserved for the users until the users release the resources, hence is suitable for users who need continuous network service.

On the other hand, packet-switched networks is suitable for users who need burst network service.

b. below depicts the worst case



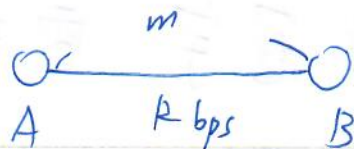
the rate of packets running into the router A would be $\sum_{i=1}^n \frac{N}{K}$ (bps)

while the rate of packets leaving router A is LR (bps)

if $LR > \sum_{i=1}^n \frac{N}{K}$, no congestion control is needed.

However, if process time can't be negligible, it would be another case.

p6.



$$p = s \text{ (m/s)}$$



$$(a) \quad d_{\text{prop}} = \frac{m}{s} \text{ (seconds)}$$

$$(b) \quad d_{\text{trans}} = \frac{L}{R} \text{ (seconds)}$$

$$(c) \quad \text{end-to-end delay} = d_{\text{prop}} + d_{\text{trans}} = \frac{m}{s} + \frac{L}{R}$$

(d) at time $t = d_{\text{trans}}$, the last bit just left host A and started to propagate along the link.

(e) $d_{\text{prop}} > d_{\text{trans}}$, when $t = d_{\text{trans}}$, the first bit of the packet is still propagating along the link and it is $\left(\frac{d_{\text{prop}} - d_{\text{trans}}}{d_{\text{prop}}} \right) \times m$ (meters)

away from host B

(f) $d_{\text{prop}} < d_{\text{trans}}$, when $t = d_{\text{trans}}$

the first bit of the packet would be at host B, waiting for the total bits to arrive so that the packet can be processed.

(g) $S = 2.5 \times 10^8$ $L = 120$ bits $R = 56$ kbps

$$d_{\text{prop}} = \frac{m}{S}, \quad d_{\text{trans}} = \frac{L}{R}$$

$$\frac{m}{2.5 \times 10^8} = \frac{120}{56 \times 10^3} \Rightarrow \underline{m \approx 535.7 \text{ (km)}}$$

P8. (a) $\frac{3M}{150k} = 20$ users can be supported.

.. (FDM - frequency division multiplexing can help 20 users use the Network simultaneously)

(b) 0.1 \Rightarrow a user use 10% of time transmitting
ie, at a moment, the probability of a user transmitting via network is also 10%

(c) 120 users

$p(n \text{ users transmitting})$

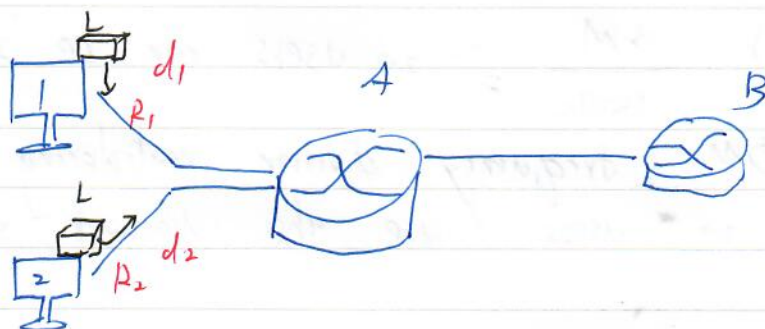
$$= C_n^{120} \times (10\%)^n (1-10\%)^{120-n}$$

$$(d) \sum_{i=21}^{120} C_i^{120} (10\%)^i (90\%)^{120-i}$$

$$= 1 - \sum_{i=0}^{20} C_i^{120} (10\%)^i (90\%)^{120-i}$$

$$\approx 0.00794119225 \quad (\text{By online binomial CDF calculator})$$

P11.



assume at $t=0$, host 1 & host 2 send packet simultaneously, the first packet arriving router A

$$\text{at } t = \min \left\{ d_1 + \frac{L}{R_1}, d_2 + \frac{L}{R_2} \right\}$$

assume the link rates between router A & router B is R_{AB} , then no queuing delay will happen if

$$\underbrace{\min \left\{ d_1 + \frac{L}{R_1}, d_2 + \frac{L}{R_2} \right\}}_{\substack{\text{the time when the} \\ \text{first packet arrives A}}} + \underbrace{\frac{L}{R_{AB}}}_{\substack{\text{the time spent} \\ \text{on sending first} \\ \text{packet on the AB link}}} < \underbrace{\max \left\{ d_1 + \frac{L}{R_1}, d_2 + \frac{L}{R_2} \right\}}_{\substack{\text{the time when the} \\ \text{second packet arrives A}}}$$

(d_1, d_2) which satisfy the above inequality expression are the answers ~~✗~~

p 22. packet loss probability = p

(1) $(1-p)^N$ (2) X denotes the random variable of ~~✗~~ times needed to transmit the packet

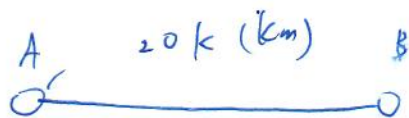
$$P(X = n) = (1-k)^{n-1} \cdot k \quad \text{where} \quad k = (1-p)^N$$

$$E[X] = \sum_{n=0}^{\infty} [1-k]^{n-1} \cdot k = \frac{1}{k} = (1-p)^{-N} \quad (\text{geometric distribution})$$

Therefore the average times of retransmission

$$\text{is } \boxed{(1-p)^{-N} - 1} \quad \text{✗}$$


p 25.
No.
Date



$$\text{speed} = 2.5 \times 10^8 \text{ (m/s)}$$

$$R = 2 \text{ Mbps}$$

$$(a) \quad 2 \text{ M} \times \frac{20 \text{ k} \times 10^3}{2.5 \times 10^8} = 160 \text{ k (bits)}$$

(b)  when the first arrived host B
the $d_{\text{prop}} \times R$ bits just left
host A

$$\Rightarrow \boxed{160 \text{ k bits}} \#$$

(c) the maximum number of bits in the link
at the same time

$$(d) \quad \frac{20 \text{ k (km)}}{160 \text{ k (bits)}} = \frac{125 \text{ m}}{\#} \text{ , longer than a football field}$$

$$(e) \quad \frac{m}{R \cdot d_{\text{prop}}} = \frac{m}{R \times \frac{m}{s}} = \boxed{\frac{s}{R}} \#$$

$8 \cdot 10^6$ bits // 2Mbps each link

a. (1) about $\frac{8 \cdot 10^6}{2M} = 4(s)$ (2) 12 (s)

b. 800 10000 bits long packets

(1) $\frac{10000}{2M} = 5 \text{ ms}$ (2) 10 (ms)

c. 15 ms later the first packet reach destination
20 ms later the second ...

$(15 + 5 \times 800) \text{ ms}$ " 800th "

Therefore it takes 4010 ms for message to reach the destination, which is much less than the result in part (a),

d. once packet loss happens, the source can retransmitted the lost packets instead of the whole message to save a large amount of time.

Furthermore, if the packet is too large, it may take too much space at routers, and hence cause higher probability of packet loss of other Internet users.

e. not every packet would follow the same route during transmitting, therefore the order of the arrival of packets may be out of the expectation and need more strategy to handle this issue.

Second, every packets carry not only messages but also headers, therefore the total size increase as packets number increase.

B04901110 林冠宇 電網導 p18

destination www.facebook.com

a.

13:32					
# hops					
13	26.082	25.908		average	8.602571
13	4.294	4.221	4.449	standard deviation	7.269977
13	5.834	5.809	8.717		
13	4.053	4.092	4.069		
13	7.521	7.617	7.77		
17:08					
13	5.612	5.613	5.609	average	5.215267
13	4.365	4.673	4.544	standard deviation	0.88469
13	4.794	4.869	4.842		
13	4.163	4.158	4.776		
13	6.745	6.74	6.726		
02:53					
13	8.799	9.316	8.572	average	5.9326
13	3.894	3.835	3.626	standard deviation	1.96023
13	4.215	5.472	5.451		
13	4.335	4.141	4.977		
13	7.306	7.297	7.753		

edge-star-mini-shv-01-tpe1.facebook.com (31.13.87.36) is the destination.

b. the number of routers remain 13 for 50 data, but the paths change from time to time.

c. Since I did this task in NTU, the ISP the packets pass through involved (TAnet – Taiwan academic network, TPIX-TW – PeeringDB – an IXP, and the facebook ISP located in US).

In this case, no large delay occur at the peering interfaces between adjacent ISPs.

d. –(a)

destination: www.ucla.com - gateway.lb.it.ucla.edu [164.67.228.152]

13:32:

delay

average – 155ms deviation- 5.664ms

17:08:

average – 122ms deviation- 13.664ms

02:53:

average – 134ms deviation- 7.664ms

(calculated via excel)

d-(b) No, the hops remain 20 in all data.

d-c about 4 ISPs are involved in the task, which are TAnet, ASnet, Indiana University ISP, and UCLA network. No apparent large delay occurred at peering interfaces.