

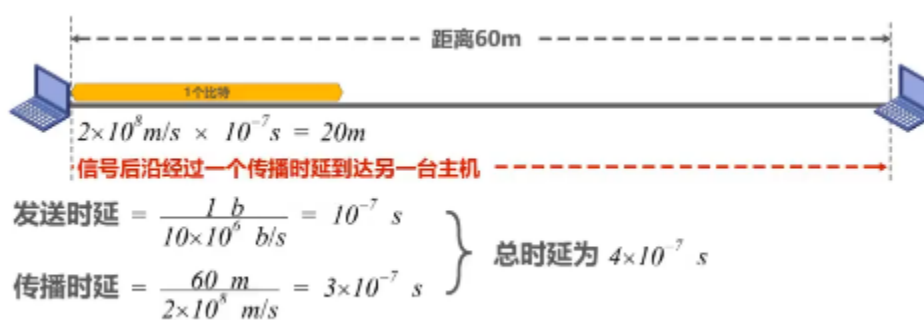
Chapter 1

P36 How long does it take to send a file of 640,000 bits from host A to host B over a circuit-switched network? All links are 1.536 Mbps, Each link uses TDM with 24 slots, 500 msec to establish end-to-end circuit.

Solution: $1536000/24=64000\text{bps}$ $640000/64000 = 10\text{s}$ $10+0.5 = 10.5\text{s}$

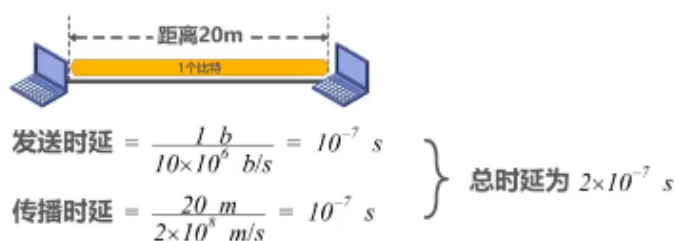
P58 Assuming that the link length d between two hosts is 60m, the link bandwidth R is 10Mbps, and the signal propagation rate s is 2×10^8 m / s. Now host A sends 1-bit packet to host B.

1) How long does it take for host B to receive this packet? If host A starts transmitting the packet at time $t = 0$, how far does the front end of the signal of the packet spread at time $t = d_{\text{trans}}$?



$t = d_{\text{trans}}$ 时, 信号前端在 20m 处

2) What if the link length d is 20 meters?



3) If the link length d is 10 meters, what is the specific transmission situation?

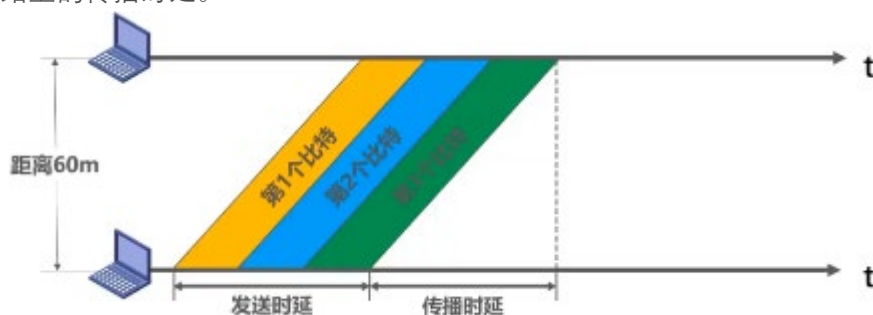
每个 bit 没发送完之前就会到达 B, 总延时: $1.5 \times 10^{-7} \text{ s}$

4) If host A sends n -bit packet to host B, how long does it take for host B to receive the packet?

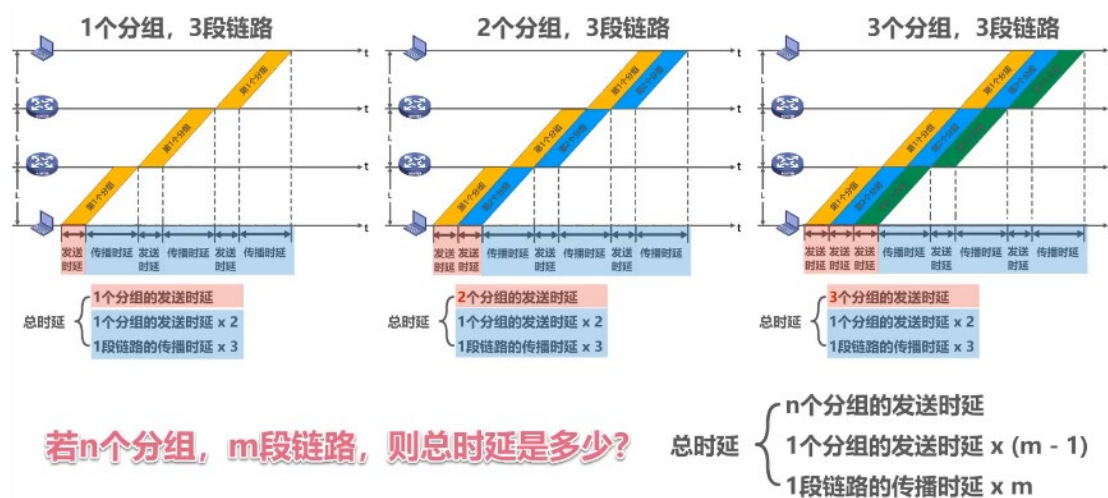
$$\begin{array}{lcl}
 \left. \begin{array}{l} \text{发送时延} = \frac{1 \text{ b}}{10 \times 10^6 \text{ b/s}} = 10^{-7} \text{ s} \\ \text{传播时延} = \frac{60 \text{ m}}{2 \times 10^8 \text{ m/s}} = 3 \times 10^{-7} \text{ s} \end{array} \right\} \text{总时延为 } 4 \times 10^{-7} \text{ s} & & \left. \begin{array}{l} \text{发送时延} = \frac{n \text{ b}}{10 \times 10^6 \text{ b/s}} = n \times 10^{-7} \text{ s} \\ \text{传播时延} = \frac{60 \text{ m}}{2 \times 10^8 \text{ m/s}} = 3 \times 10^{-7} \text{ s} \end{array} \right\} \text{总时延为 } (n+3) \times 10^{-7} \text{ s} \\
 \text{接收完 } n \text{ 比特的总时延为 } (4 \times 10^{-7} \text{ s}) \times n & \text{✗} & \text{✓}
 \end{array}$$

第一种计算方法是错误的。比特信号是一个跟着一个向前同时传播的, 因此, 无论有多少比特, 在总时延中只包含一个传播时延。增加比特数量, 只是增大了发送时延, 而传播时延数量不变。

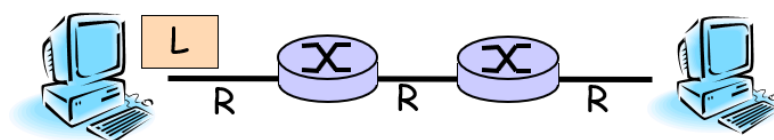
结论：若主机在一段链路上连续发送分组，则总时延为所有分组的发送时延加上信号在这一段链路上的传播时延。



假设：分组等长，各链路长度相同、带宽也相同，忽略路由器的处理时延。



P60



$L = 7.5 \text{ Mbits}$

$R = 1.5 \text{ Mbps}$

t_{prop} : propagation delay of each link

Divide the message into 5000 small packets: $l = 1.5 \text{ kbits}$

Calculate the end-to-end delay of message switching and packet switching

message switching 报文交换, 数据不分组, 按单个 7.5M 数据发送

总延时: 每次发送时间 $\times 3$ + 每段传播延时 $\times 3$ $L/R \times 3 + 3 \times t_{\text{prop}}$

packet switching 分组交换, 数据分组, 按每个分组大小进行数据发送

总延时: 分组数量 $n = 7.5\text{M}/1.5\text{K} = 5000$ $l/R \times (n+2) + 3 \times t_{\text{prop}}$

P61

The message to be sent is x (bits) in total. There are k links from the source to the destination. The propagation delay of each link is d (s) and the bandwidth is R (bps). In circuit switching, the circuit establishment time is s (s). In packet switching, the message can be divided into several packets with p (bits) length. It is assumed that the processing time and queuing time of the message/packets at each node are ignored.

Q: under what conditions is the end-to-end delay of circuit switching greater than that of packet switching?

【答案】在电路交换的情况下，电路建立延迟等于 s 秒，发送延迟等于 x/b 秒，传播延迟等于 kd 秒。总延迟等于 $s + x/b + kd$ 秒。在分组交换的情况下，没有电路建立延迟，发送延迟（假定各个分组连续发送）也等于 x/b 秒，传播延迟等于 kd 秒，但增加了转发延迟 kp/b 。总延迟等于 $x/b + kd + kp/b$ 秒。

令 $x/b + kd + kp/b < s + x/b + kd$ ，得到 $kp/b < s$ 。所以在 s 大于 kp/b 的条件下，分组交换网的时延比电路交换网小。

P78-79

1. In the OSI model, the first bottom-up level to provide end-to-end services is (**B**)

A. data link layer B. transport layer C. session layer D. application layer

2. Among the following options, which is not described in the network architecture is (**C**)

- A. layers in the network
- B. protocols used by each layer
- C. internal implementation details of the protocols
- D. functions that must be completed by each layer

3. In the TCP / IP reference model, what directly provides services for the application layer is (**B**)

A. presentation layer B. transmission layer C. network layer D. network interface layer

4. Assuming that the application layer in the **ISO/OSI** reference model wants to send 400B data (without splitting), except the physical layer and application layer, all other layers introduce 20B additional overhead when packaging PDU, the transmission efficiency of the application layer is about (**A**)

- A.80%
- B.83%
- C.87%
- D.91%

解析：OSI 参考模型共 7 层（应用层、表示层、会话层、传输层、网络层、数据链路层、物理层），除去物理层和应用层，剩 5 层。它们会向 PDU 引入 $20 \times 5 = 100\text{B}$ 的额外开销。应用层是最顶层，因此数据传输效率为 $400\text{B} / 500\text{B} = 0.8$

Chapter 2

P62-65

1. In the physical layer interface characteristics, the time sequence used to describe the completion of each function is (**D**)

- A. Mechanical Characteristics
- B. Electrical Characteristics
- C. Functional Characteristics
- D. Procedural Characteristics

2. Among the following options, that do not belong to the definition scope of physical layer interface specification is (**C**)

- A. interface shape B. pin function
- C. physical address D. signal level

3. Under the condition of no noise, if the frequency bandwidth of a communication link is 3kHz and QAM modulation technology with four phases and four amplitudes in each phase is adopted, the maximum data transmission rate of the communication link is (**B**)

- A. 12kbps B. 24kbps C. 48kbps D. 96kbps

波特率 6K, 每个波特表示 4bit。 $C=2W\log_2 N=2 * 3K * \log_2 (4*4) =24Kbps$

4. If the data transmission rate of a communication link is 2400bps and 4-phase modulation is adopted, the baud rate of the link is (**B**)

- A. 600 baud B. 1200 baud
- C. 4800 baud D. 9600 baud

2. (11 年考研 34 题) 若某通信链路的数据传输速率为 2400bps, 采用 4 相位调制, 则该链路的波特率是

- A. 600 波特 B. 1200 波特 C. 4800 波特 D. 9600 波特

【解答】选 B。有 4 种相位, 则一个码元需要由 $\log_2 4=2$ 个 bit 表示, 则波特率=比特率/2=1200 波特。

【考查知识点】考察波特与比特率的关系: 比特率和波特率之间的换算关系如下: 比特率 = 波特率 * $\log_2 n$ 。比特率是数字信号的传输速率——单位时间内所传输的二进制代码的有效位数。单位——比特/秒(bps)或千比特/秒(kbps)。波特率是调制速率(又称波形速率)——线路中每秒传送的波形的个数。单位——波特(band)。

5. If the frequency bandwidth connecting R2 and R3 link is 8kHz and the signal-to-noise ratio is 30dB, and the actual data transmission rate of the link is about 50% of the maximum data transmission rate, the actual data transmission rate of the link is about (**C**)

- A. 8kbps B. 20kbps C. 40kbps D. 80kbps

根据香农定理:

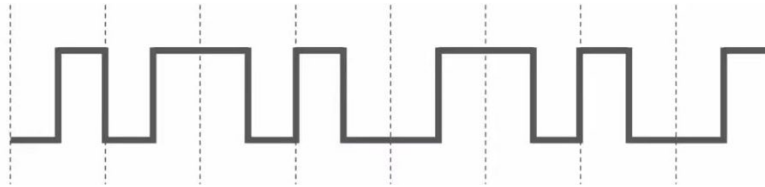
理论最大: $8k\log_2 (1+1000) =8k10=80kbps$

实际最大: $0.5*理论最大=40kbps$

6. Among the following factors, which will not affect the channel data transmission rate is (**D**)

- A. signal to noise ratio B. frequency bandwidth
- C. symbol speed D. signal propagation speed ($2-3 \times 10^8$)

7. If the following figure shows the signal waveform received by the 10Base-T network interface card, the bit string received by the network interface card is (A)



A. 00110110 B. 10101101 C. 01010010 D. 11000101

从波形看出是曼彻斯特编码，所以是 00110110 或 11001001

Chapter 3

P18

1. HDLC protocol sets the bit stream 01111100 01111110 after framing as (A)

- A. 01111100 00111110 10
- B. 01111100 01111101 01111110
- C. 01111100 01111101 0
- D. 01111100 01111110 01111101

P24

1. If the information received by an Ethernet adapter is 101101001 and the generator is $G(x) = x^3 + x^2 + 1$, judge whether the transmission has bit error?

$$\begin{array}{r}
 110010010 \\
 1101 \overline{) 101101001000} \\
 \underline{1101} \\
 1100 \\
 \underline{1101} \\
 1100 \\
 \underline{1101} \\
 1100 \\
 \underline{1101} \\
 10 \neq 0 \text{ (有误差)}
 \end{array}$$

P43-44

The data link layer uses the GBN protocol, and the sender has sent frames numbered 0 ~ 7. When the timer expires, if the sender only receives the acknowledgement of frames 0, 2 and 3, the number of frames to be retransmitted by the sender is (C)

- A. 2
- B. 3
- C. 4
- D. 5

解析：已经收到 3 号帧的确认，表示 0—3 号帧都已经正确接收，超时发生就重发 4—7 帧。GBN 不只是超时重传的特点，它还有累计确认的功能，即：发一个确认 N，表示前 N 帧都已正确收到。超时重传一旦发生，就重发最后确认帧之后的所有帧。不要和选择重传 (SR) 弄混，选择重传可以乱序接收，只发送单个帧的确认。如果是选择重传，还要再多传 1 号帧。

Considering GBN protocol and SR protocol, assuming that the length of sequence number space is n, what is the maximum allowed sender window?

$n/2$

解析：如果发送方发送一整组数据，接收方接受数据并发出 ACK 消息，发送方此时没有接收到 ACK 消息；对于发送方而言，窗口为 $[m-w, m-1]$ ；对于接收方而言，窗口为 $[m, m+w-1]$ ；其中 m 表示 `send_base`， w 表示 window 大小；

问题：为了区分是第一个分组的重传还是下一组的初次传送

QUESTION: 0 1 2 3 0 1 2 3 ; 序号 k 为 4; 若此时窗口大小为 3

0 1 2 **3** 0 1 2 3 ; $[0, 1, 2]$ 表示发送方窗口; $[3, 0, 1]$ 表示接收方窗口; 若此时接收方接受到 0 分组，它为第一个分组的 0 的重传? 还是下一个分组的 **0** 的初次传送?

QUESTION: 0 1 2 3 0 1 2 3; 序号 k 为 4; 若此时窗口大小为 2

0 1 **2** 3 0 1 2 3 ; $[0, 1]$ 表示发送方窗口; $[2, 3]$ 表示接收方窗口; 此时无论如何都不会出现相同序号的分组的情况。

即分组序号 $k \geq 2w$ (窗口) .

P68

A LAN adopts CSMA/CD protocol to realize the media access control. The data transmission rate is 10Mbps, the distance between host A and host B is 2km, and the signal propagation speed is 200000km/s. If there is a conflict when two hosts send data, how long will it take from the time when they start sending data to the time when both hosts detect the conflict? (What is the minimum and maximum duration?)

(1) 主机甲和主机乙之间单向传播延迟时间: $10\mu\text{s}$ 。两台主机均检测到冲突时，最短所需时间和最长所需时间对应下面两种极端情况：

①主机甲和主机乙同时各发送一个数据帧，信号在信道中发生冲突后，冲突信号继续向两个方向传播。因此，甲乙两台主机均检测到冲突时，最短需经过 $10\mu\text{s}$ 。

②主机甲(或主机乙)先发送一个数据帧，当该数据帧即将到达主机乙(或主机甲)时，主机乙(或主机甲)也开始发送一个数据帧。这时，主机乙(或主机甲)将立即检测到冲突;而主机甲(或主机乙)要检测到冲突，冲突信号还需要从主机乙(或主机甲)传播到主机甲(或主机乙)，因此，甲乙两台主机均检测到冲突时，最长需经过 $20\mu\text{s}$ 。

Chapter 4

P20

In a LAN using CSMA / CD protocol, the transmission medium is a complete cable, the transmission rate is 1Gbps, and the signal propagation rate in the cable is 200000 km / s. if the minimum data frame length is reduced by 800 bits, the distance between the farthest two stations needs to be at least (**D**)

- | | |
|---------------------|--------------------|
| A increased by 160m | B increased by 80m |
| C reduced by 160m | D reduced by 80m |

解析：

1. 传输时延 $> 2 \times$ 最远两个站点间的传播时延
2. 传输时延 = 数据包的大小 / 数据包的发送率
3. 传播时延 = 站点间距离 / 信号传播速率

1 式为最远站点发出的数据产生冲突后，能被远端站点感知的最短时间，保证了在远端站点发送完数据之前能够检测到冲突

由 2 式，数据帧减小 800bit，即传输时延减小 $800\text{bit}/1\text{Gbps}=8\times 10^{-7}\text{s}$

由 1 式，因此传播时延至少减小 $1/2 \times 8\times 10^{-7}=4\times 10^{-7}\text{s}$

由 3 式，因此站点间距离最少减少 $4\times 10^{-7}\times 200000=0.08\text{km}=80\text{m}$

最短帧长 = $2 \times \text{传播时延} \times \text{数据传输率}$

$$800\text{ bit} = 2 \times \frac{d}{200000\text{ km/s}} \times 1\text{Gb/s}$$

d 设 d 为减少的距离。

$$= 2 \times \frac{d}{2 \times 10^8\text{ m/s}} \times 10^9\text{ bit/s}$$

\Downarrow

$$10 d = 800\text{ m}$$
$$\underline{d = 80\text{ m}}$$

故需减少 80m 的站点距离。

Exercises-1 (PPT5-72 74 84

)

1. The protocols for directly encapsulating RIP、OSPF and BGP messages are

A. TCP、UDP、IP

B. TCP、IP、UDP

C. UDP、TCP、IP

D. UDP、IP、TCP

Exercise-2

0010 0000	32
0010 1000	40
0011 0000	48
0011 1000	56

1. In subnet 192.168.4.0/30, the maximum number of hosts that can accept IP packets with destination address 192.168.4.3 is

A.0 B.1 **C.2** D.4 主机位都是1广播地址,

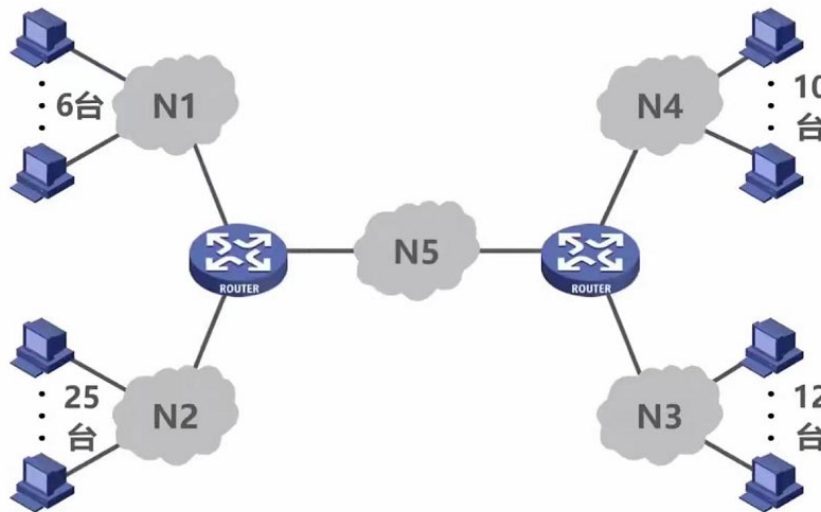
2. There are four routing table entries with the same forwarding interface in a routing table, and their destination network addresses are 35.230.32.0/21, 35.230.40.0/21, 35.230.48.0/21 and 35.230.56.0/21 respectively. After aggregating the four routes, the destination network address is

A.35.230.0.0/19 B. 35.230.0.0/20
C. 35.230.32.0/19 D. 35.230.32.0/20

Exercises-3

Assuming that the applied IP address block is 218.75.230.0/24, please assign IP addresses to the devices in the figure below using **fixed length** and **variable length** subnet masks respectively.

00000000-00011111
0-31
00100000-00111111
32-63
01000000-01011111
64-95
01100000-01111111
96-127
10000000-10011111
128-159
10100000-10111111
160-191
11000000-11011111
192-223
11100000-11111111
193-255



00000000-00011111
0-31 分给N2 /27
00100000-00101111
32-47 分给N3 /28
00110000-00111111
48-63 分给N4 /28
01000000-01001111
64-79 分给N1 /28
01010000-01010011
80-83 分给N5 /30

84~255都没分配

Exercises-3

Assuming that the applied IP address block is 218.75.230.0/24, please assign IP addresses to the devices in the figure below using **fixed length** and variable length subnet masks respectively.

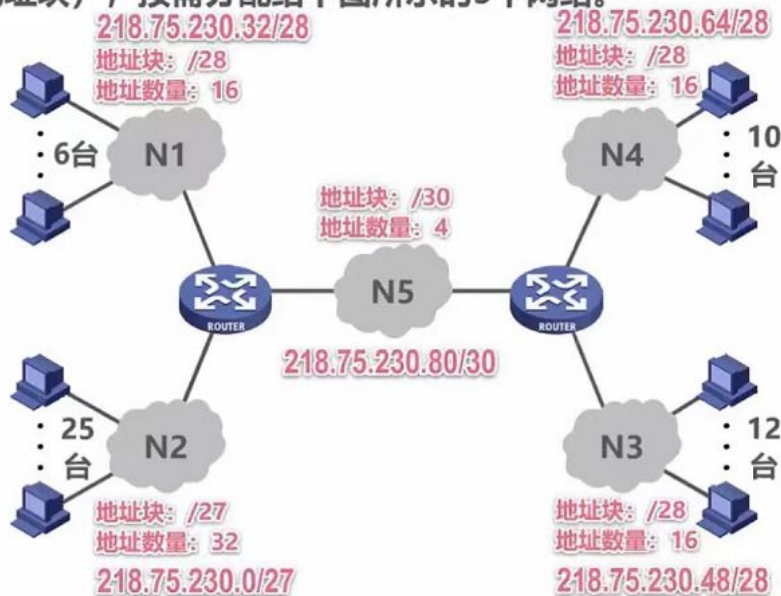
使用子网掩码255.255.255.224 对C类网络218.75.230.0进行子网划分的细节			
子网	网络地址	广播地址	该子网可分配的地址
1	218.75.230.0	218.75.230.31	218.75.230.1 ~ 218.75.230.30
2	218.75.230.32	218.75.230.63	218.75.230.33 ~ 218.75.230.62
3	218.75.230.64	218.75.230.95	218.75.230.65 ~ 218.75.230.94
4	218.75.230.96	218.75.230.127	218.75.230.97 ~ 218.75.230.126
5	218.75.230.128	218.75.230.159	218.75.230.129 ~ 218.75.230.158
6	218.75.230.160	218.75.230.191	218.75.230.161 ~ 218.75.230.190
7	218.75.230.192	218.75.230.223	218.75.230.193 ~ 218.75.230.222
8	218.75.230.224	218.75.230.255	218.75.230.225 ~ 218.75.230.254

从子网1~8中任选5个分配给左图中的N1~N5。

Exercises-3

Assuming that the applied IP address block is 218.75.230.0/24, please assign IP addresses to the devices in the figure below using fixed length and **variable length** subnet masks respectively.

应用需求：从地址块218.75.230.0/24中取出5个地址块（1个“/27”地址块，3个“/28”地址块，1个“/30”地址块），按需分配给下图所示的5个网络。



218.75.230.0/24地址块所包含的全部地址如下所示:

218.75.230.0	网络N2的网络地址
⋮	网络N2可分配地址
218.75.230.31	网络N2的广播地址
218.75.230.32	网络N1的网络地址
⋮	网络N1可分配地址
218.75.230.47	网络N1的广播地址
218.75.230.48	网络N3的网络地址
⋮	网络N3可分配地址
218.75.230.63	网络N3的广播地址
218.75.230.64	网络N4的网络地址
⋮	网络N4可分配地址
218.75.230.79	网络N4的广播地址
218.75.230.80	网络N5的网络地址
⋮	网络N5可分配地址
218.75.230.83	网络N5的广播地址
218.75.230.84	
⋮	剩
⋮	余
	待
	分
	配
218.75.230.255	

Exercises-4

Figure 1 shows the network topology, and Figure 2 shows the hex-content of the first 80 bytes of an Ethernet frame of the host for a Web request.

- (1) What is the IP address of the Web server? What is the MAC address of the default gateway of this host?
- (2) When the IP packet encapsulated in this frame is forwarded through router R, which fields in the IP packet need to be modified?

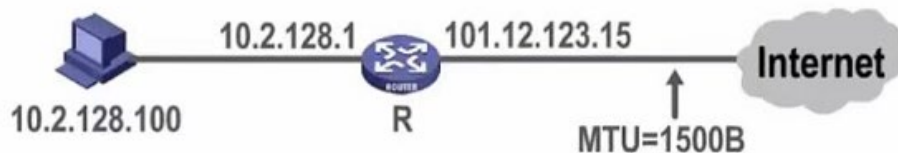


Fig. 1

00 21 27 21 51 ee 00 15	c5 c1 5e 28 08 00 45 00
01 ef 11 3b 40 00 80 06	ba 9d 0a 02 80 64 40 aa
62 20 04 ff 00 50 e0 e2	00 fa 7b f9 f8 05 50 18
fa f0 1a c4 00 00 47 45	54 20 2f 72 66 63 2e 68
74 6d 6c 20 48 54 54 50	2f 31 2e 31 0d 0a 41 63

Fig. 2

- (1) 40 aa 62 20, 64.170.98.32; 00-21-27-21-51-ee
- (2) TTL字段, 首部校验和重新计算, 总长字段若大于MTU, 则总长字段、标志字段、片偏移字段都需要修改

Exercises-1 (PPT 6-32)

1. Host A sends a TCP segment (SYN=1,seq=11220) to host B and expects to establish a TCP connection with host B. if host B accepts the connection request, the correct TCP segment sent by host B to host A may be (3)

(1) SYN=0,ACK=0,seq=11221,ack=11221

(2) SYN=1,ACK=1,seq=11220,ack=11220

(3) SYN=1,ACK=1,seq=11221,ack=11221

(4) SYN=0,ACK=0,seq=11220,ack=11220

Exercises-2 (PPT6-40)

1. A TCP connection is established between host A and host B, and the MSS length of TCP is 1000 bytes. If the current congestion window of host A is 4000 bytes, after host A sends two MSS segments to host B continuously, host A successfully receives the confirmation segment of the first segment sent by host B, and the size of the receiving window in the confirmation segment is 2000 bytes, then the maximum number of bytes that host A can also send to host B is (1)

(1)1000

(2)2000

(3)3000

(4)4000

Sender limits unACKed data to RcvWindow

Exercises-3 (PPT 6-82)

1. A TCP connection has been established between host A and host B. host A sends three consecutive TCP segments to host B, including 300 bytes, 400 bytes and 500 bytes payload respectively. The sequence number of the third segment is 900. If host B receives only the first and third segments correctly, the confirmation sequence number sent by host B to host A is(2)

- (1)300
- (2)500
- (3)1200
- (4)1400

考察**TCP**的序号的含义，以及累计确认的意义

Exercises-4

1. A TCP connection always sends TCP segments with 1KB MSS. The sender has enough data to send. A timeout event occurs when the congestion window is 16KB. If the transmission of TCP segments in the next four RTTS is successful, when all TCP segments sent in the fourth RTT receive positive responses, the size of the congestion window is(c)

- A. 7KB
- B. 8KB
- C. 9KB
- D. 16KB

慢启动（PPT6-101\102）、超时事件(PPT6-105)