

TCP

Transmission Control protocol

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TCP header:

4 bytes	source port (16 bit)	Destination port (16 bit)
4B	Sequence numbers (32 bit)	
4B	Acknowledgement number (32 bit)	
4B	Header length (4 bits) reserved (4 bits)	Window size (16 bit) adv window (16 bit)
4 bit	URG ECN ACK SYN FIN	
4B	checksum (16 bit)	Urgent pointer (16 bit)
	options (0 - 40 Byte)	
	Data (minimum)	

Total length of the header = $(4+4+4+4+4)+0 \Rightarrow 20 \text{ B}$

maximum
$= (4+4+4+4+4)+40 \Rightarrow 60 \text{ B}$

Source port (SP)

Destination port (DP)



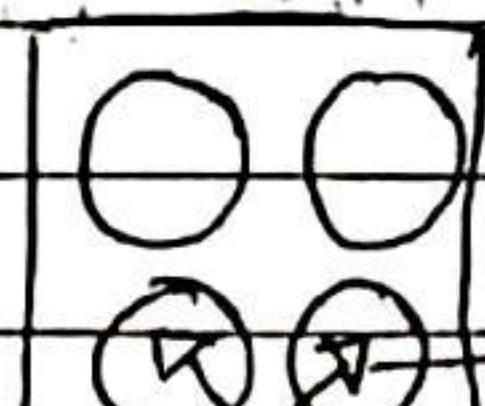
16-bit

$$2^{16} = (0 \text{ to } 2^{16}-1)$$

$= (0 \text{ to } 65,535) \rightarrow$ port number possible.

→ TCP is end-to-end protocol.

(because of the port number)



→ host

→ process

→ TCP could do multiplexing and

Demultiplexing. (Take the packet and distribute properly b/w host process)

HTTP - port number (80) → (website)

FTP - (21) → (To download a file)

Telnet - (23) → (Remote login)

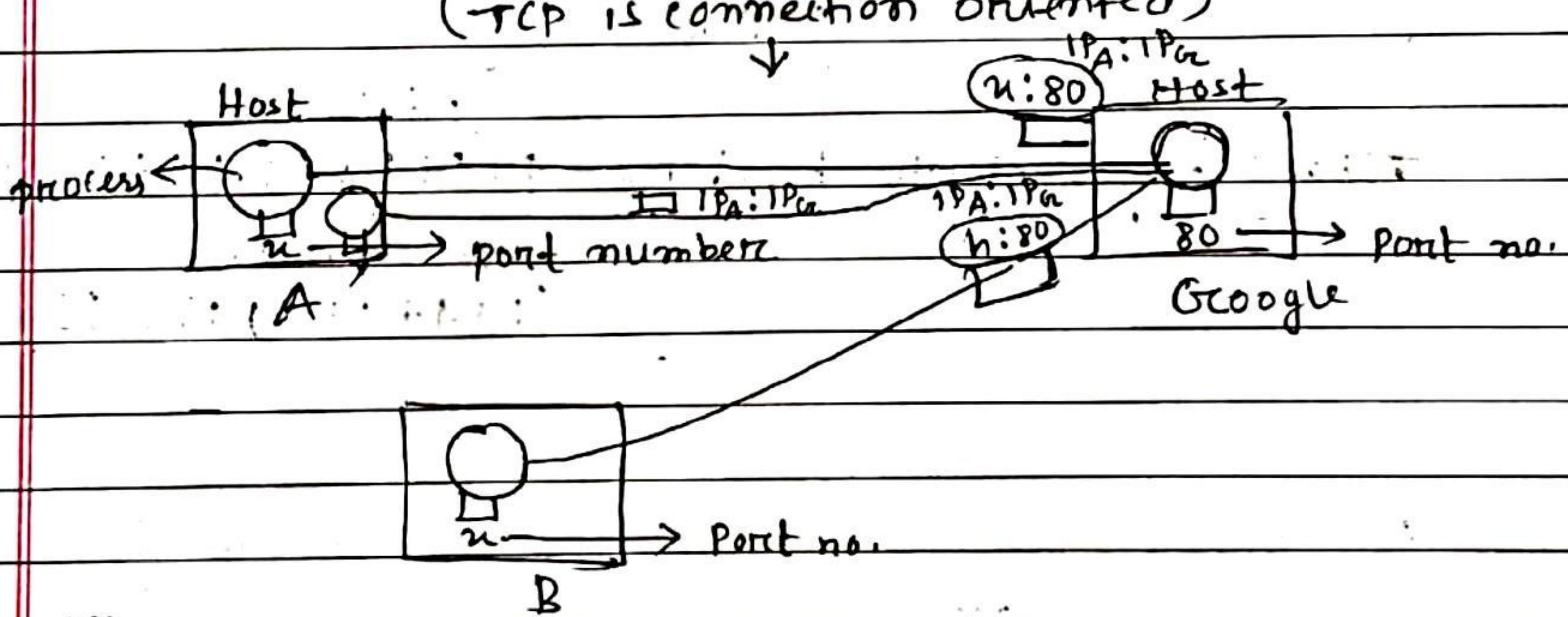
SMTP - (25) → (email transfer)

0 }
1 } well known
1023 } Internet

1024 }
1 } Reserved
49,151 } IANA (Organization)
Internet Assigned Numbers Authority

49,152 }
1 } used by general public.
65,535 }

(TCP is connection oriented)



$\boxed{\text{Socket}} = \boxed{\text{IP} + \text{port}}$ = 48 bit.

→ Combination of IP and port number is call socket.

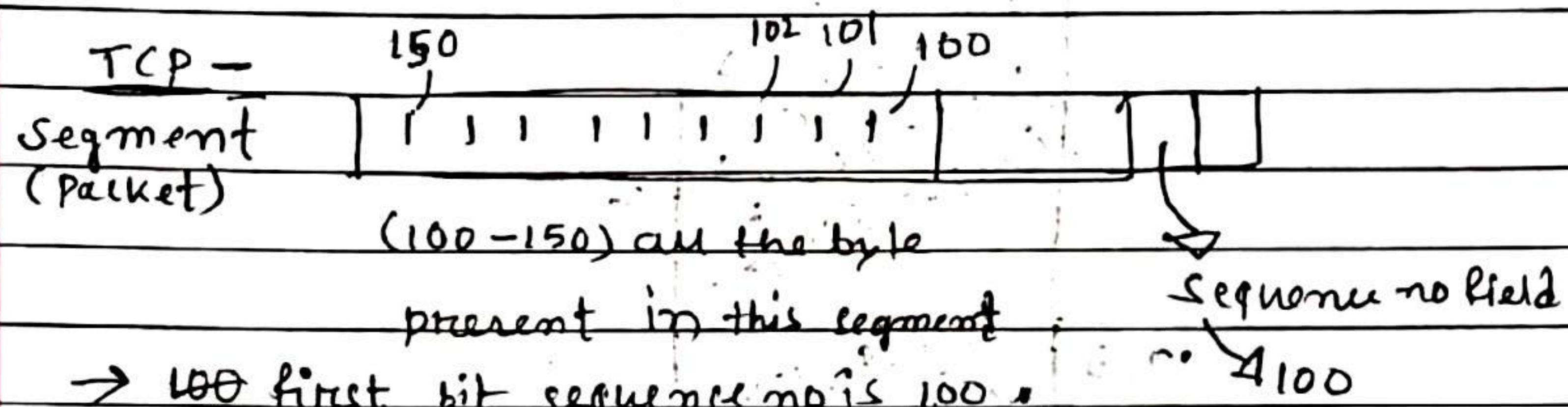
→ Using socket we can implement the connection.

● Sequence number:

→ TCP is Byte Stream protocol.

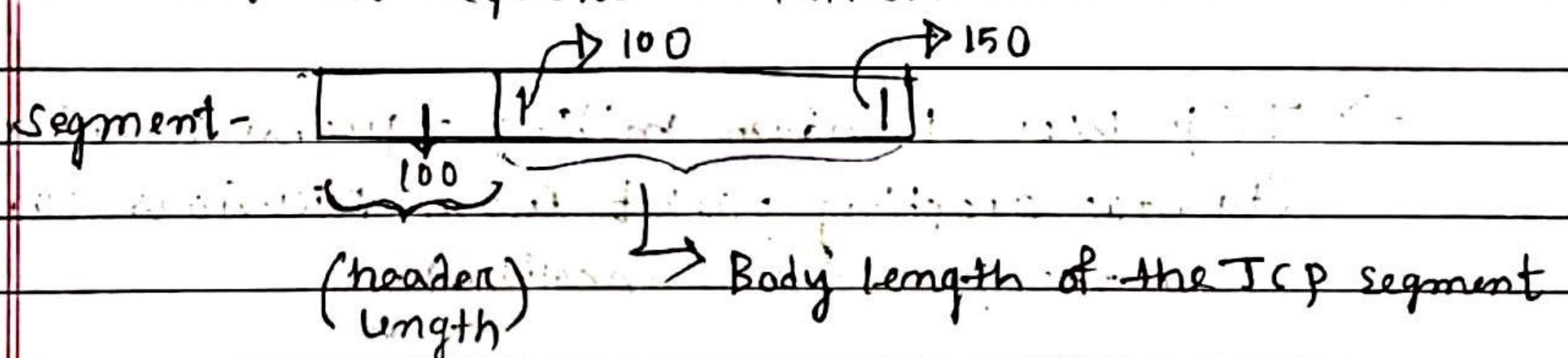
→ IP is packet stream

→ DLL (Data Link layer) - HDLC is bit stream.



● Acknowledgement number:

→ Last bit sequence number + 1.



● RISN:

Example -

IS

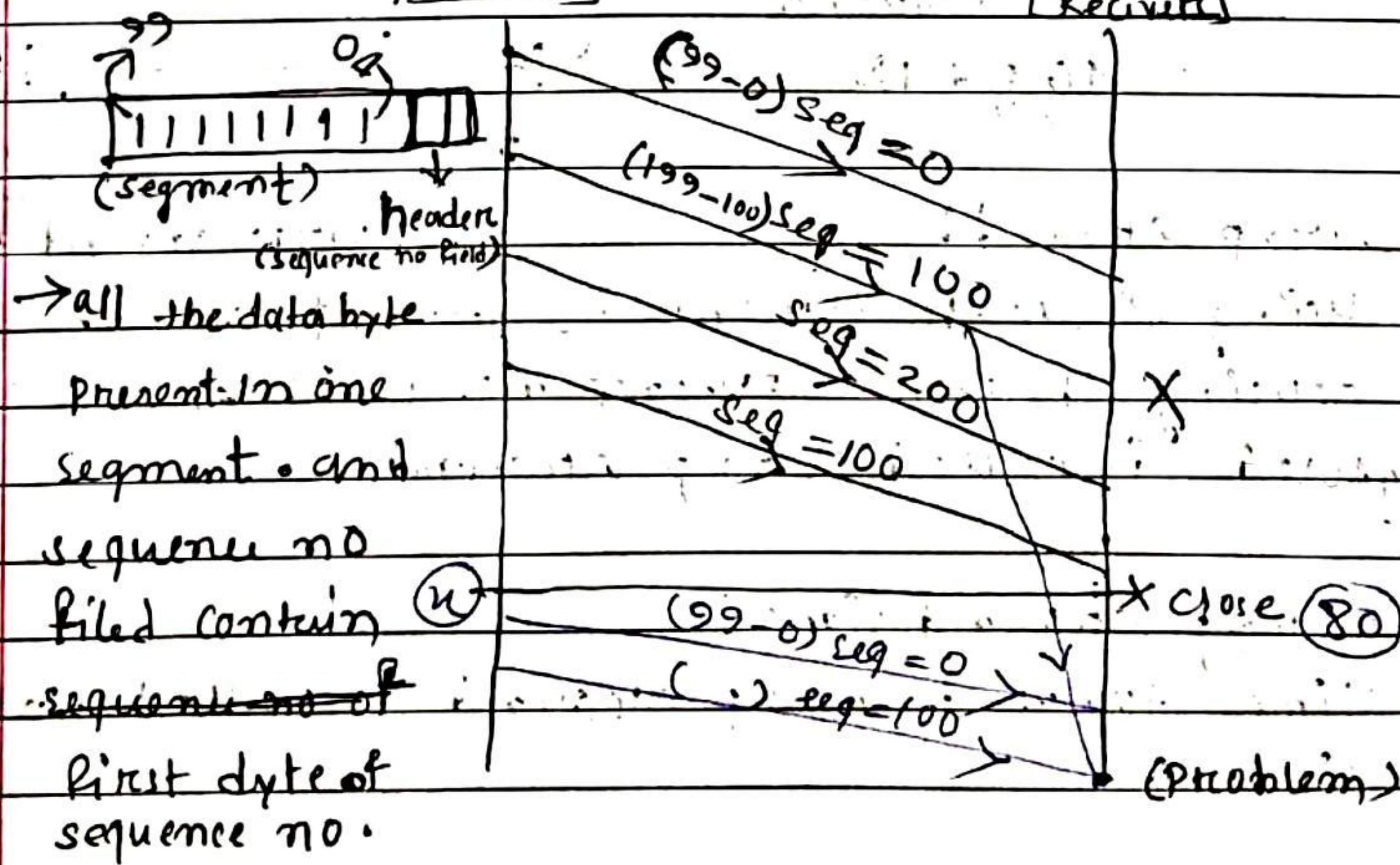
$$P.no = n$$

IR

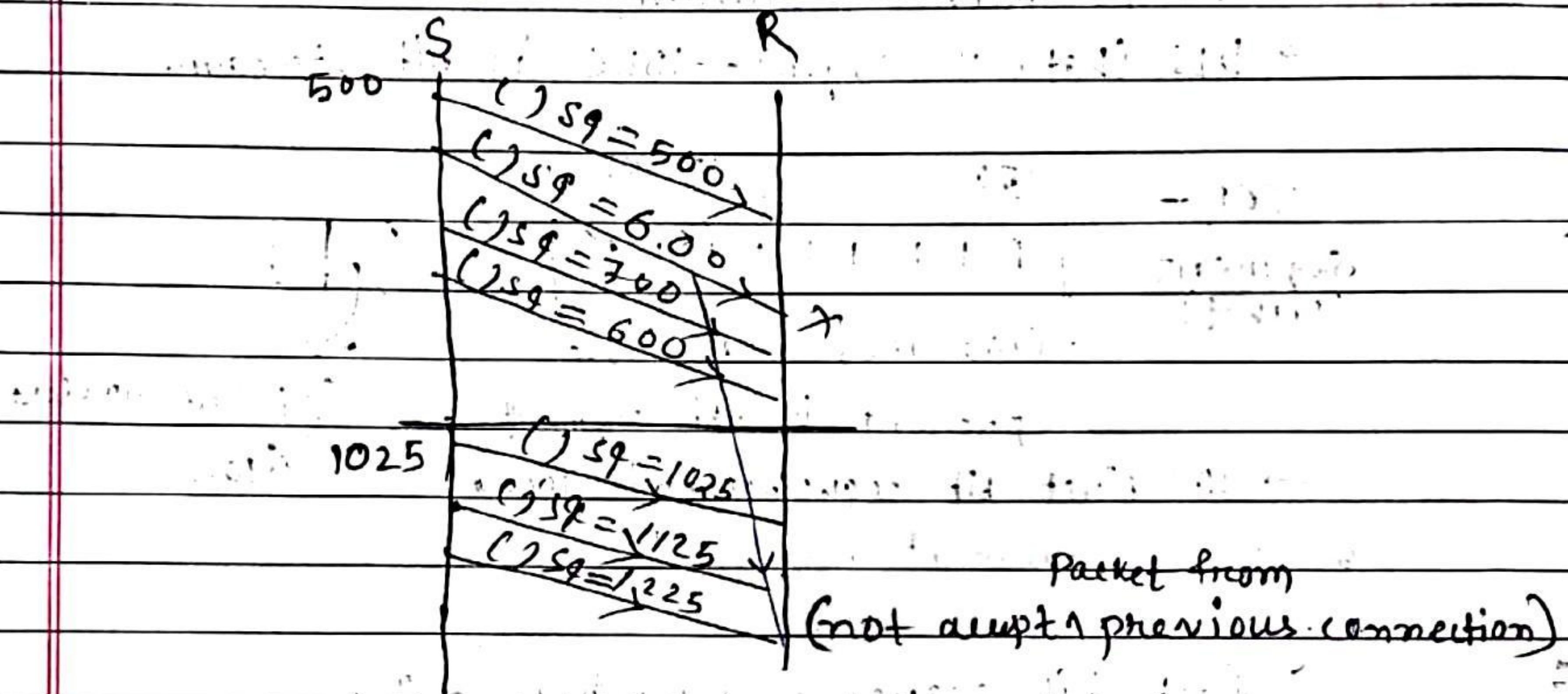
$$P.no = 80$$

[Sender]

[Receiver]



→ If you choose randomly sequence number, then chances are getting ~~not~~ collision, ~~is~~ very less.



→ TCP uses Random initial sequence numbers (RISN).
(to stop accepting packet from previous connection
in the present connection).

Wrap around time and problem on wrap around time -

$$2^{32} = (4GB) \text{ sequence} \\ = 4GB + 1B$$

Wrap around:
Using of all the sequence no present and then reuse
reusing the same sequence no. which have already
been used, is called wrap around.

What's around time?

Time taken to wrap around is called wrap around time.

Example -

Bandwidth (BW) = 1 MBPS

$$1 \text{ sec} = 1 \text{ MB}$$

$$1 \text{ M} = 1000000 \text{ B}$$

$$10^6 \text{ B} = 1 \text{ sec}$$

$$10^6 \text{ seq num} = 1 \text{ sec}$$

$$1 \text{ G} = 10^9 \text{ B}$$

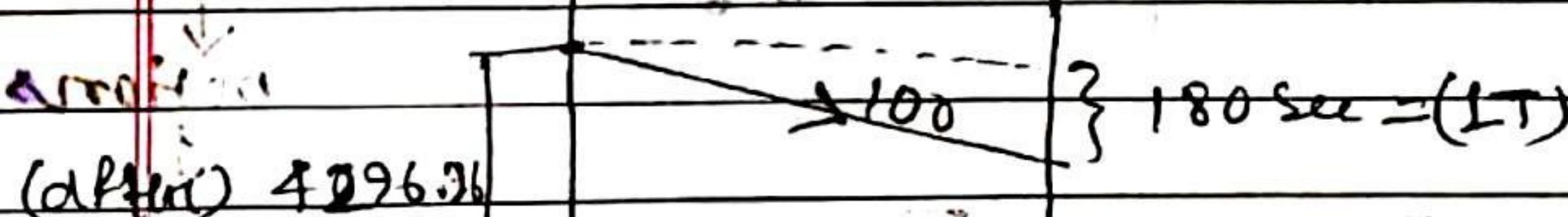
$$1 \text{ seq num} = \frac{1}{10^6} \text{ sec}$$

$$2^{32} \text{ seq num} = \frac{2^{32}}{10^6} \text{ sec}$$

$$W.A.T = 42.26 \text{ sec}$$

Lifetime (LT) = 3 min = 180 sec (worst case)
 (time taken to reach or from one destination to
 other destination)

(Sequence number format)



→ using same sequence number
 after the end of first, no
 problem

Example -

Bandwidth (BW) = 1 G.B.P.S.

$$10^9 \text{ B} = 1 \text{ sec}$$

$$1 \text{ sec} = 1 \text{ GB}$$

$$10^9 \text{ B} = 1 \text{ sec}$$

$$10^9 \text{ seq num} = 1 \text{ sec}$$

$$1 \text{ seq num} = \frac{1}{10^9} \text{ sec}$$

$$WAT = 2^{32} \text{ seq num} = \frac{2^{32}}{10^9} = 4.29 \text{ sec}$$

→ when Bandwidth time increase then W.A.T decreases.

[WAT < LT] → Problem

Example:

$BW = 1 \text{ Gbps}$ } given & calculate what are the
 $LT = 180 \text{ sec}$ } minimum no. of sequence no required,

$$1 \text{ sec} = 1 \text{ Gb}$$

$$1 \text{ sec} = 1 \text{ seq or Seq.}$$

$$180 \text{ sec} = 180 \times 1 \text{ Gb Seq}$$

∴ minimum seq num required to avoid WA
 within $LT = 180 \times 1 \text{ Gb seq num.}$

$$\left\lceil \log_2 (180 \times 10^9) \right\rceil = 42 \text{ bit} \quad (42 - 32 = 10 \text{ bit reqd})$$

options

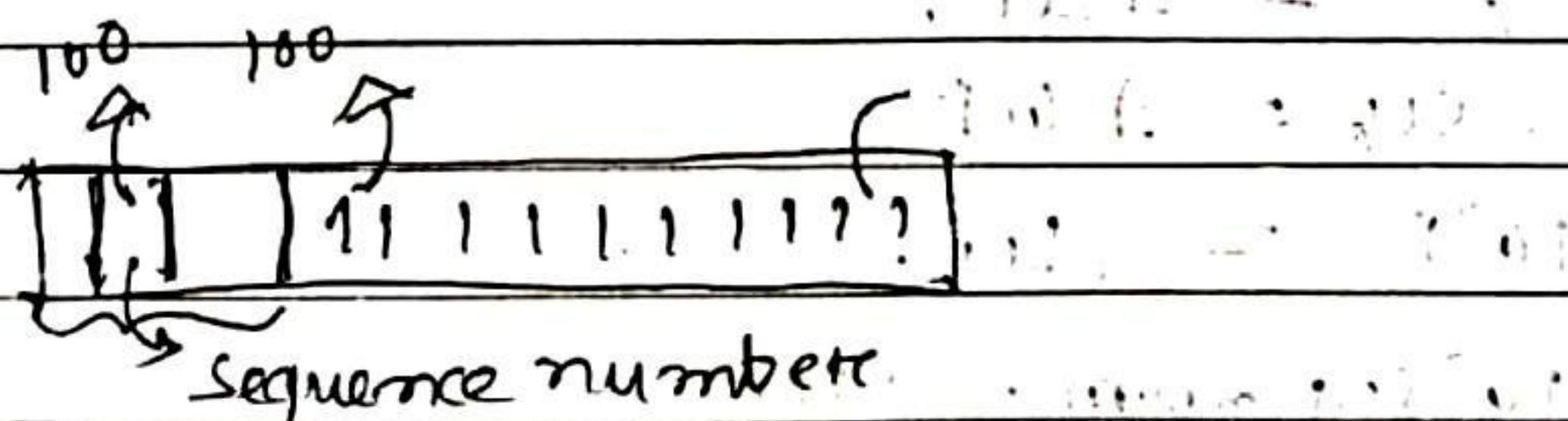
formula is $\left\lceil \log_2 (LT \times BW) \right\rceil$

Time stamp

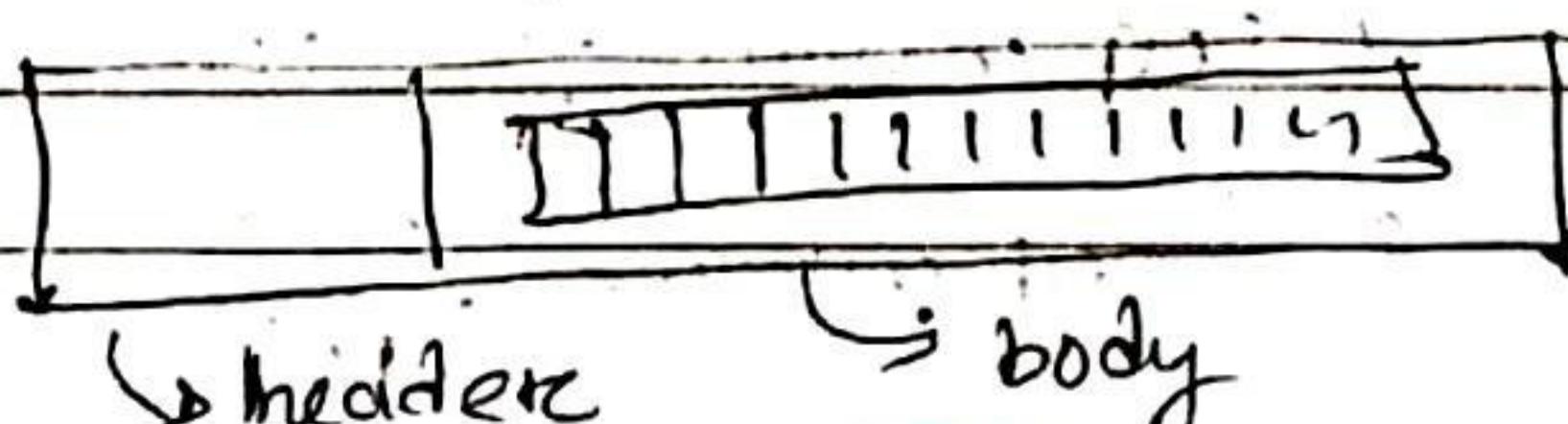
Header length and calculation of acknowledgement number.

Seq num: seq number of 1st byte.

Ack num: Seq number ex of byte expected next.



IP data gram -



Example-

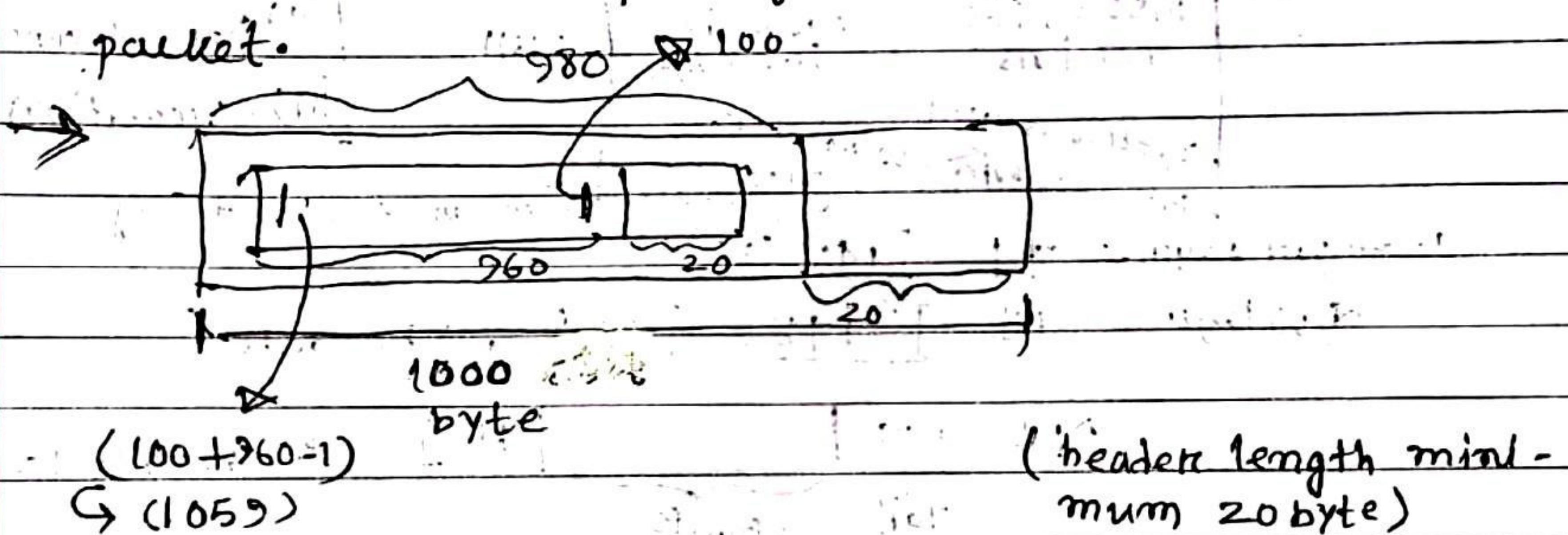
Total length = 1000 bytes } give for IP datagram.

Header length = 5 }

Inside IP datagram there is a TCP segment and
the H.L of TCP segment = 5

Sequence number of TCP segment = 100.

Find the acknowledgement numbers for this packet.



Hence, Total length of datagram = 1000 byte

Header length of datagram = $5 \times 4 = 20$

Header length of TCP segment = $5 \times 4 = 20$

Total length of TCP segment + body = $(1000 - 20) - 20$
= 960 bytes

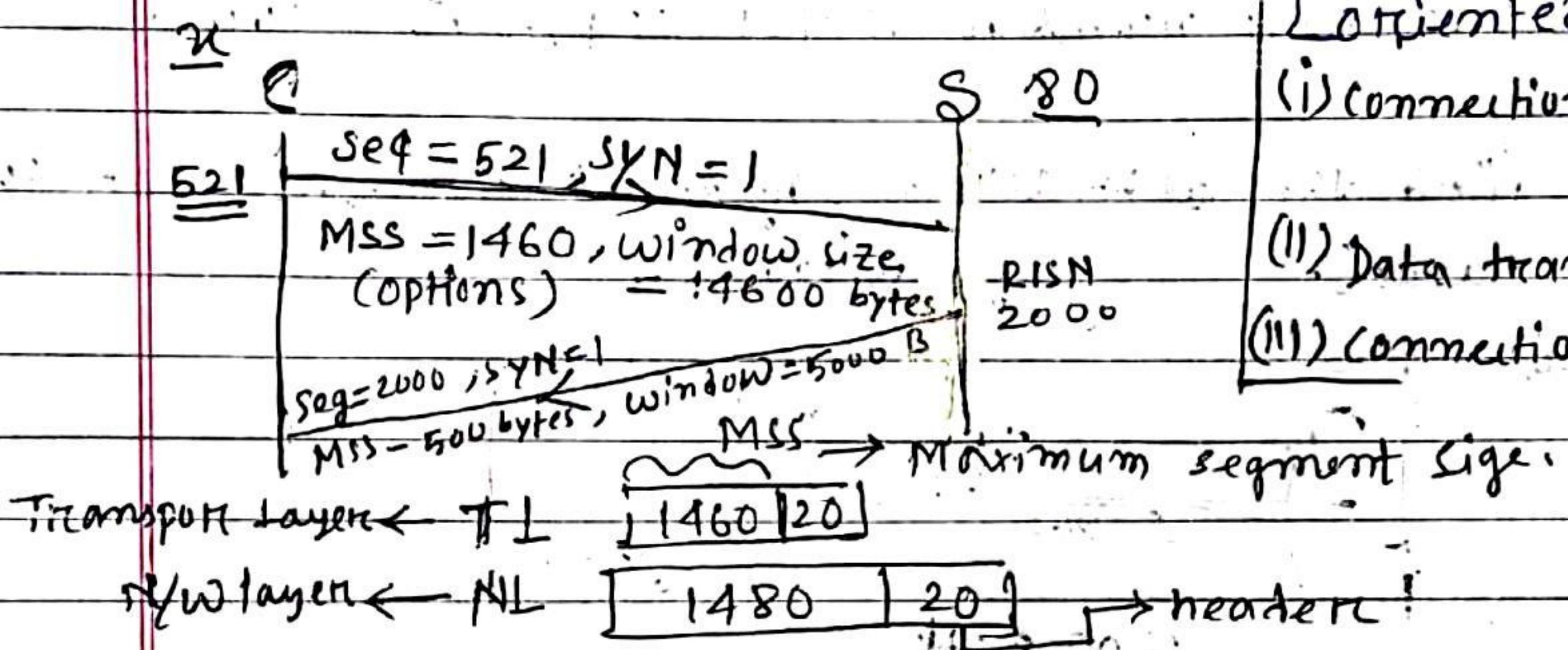
In TCP segment 1st bit sequence no = 100

Last bit sequence no = $(100 + 960 - 1)$
= 1059

So acknowledgement numbers for this packet
is 1060.

TCP connection establishment:

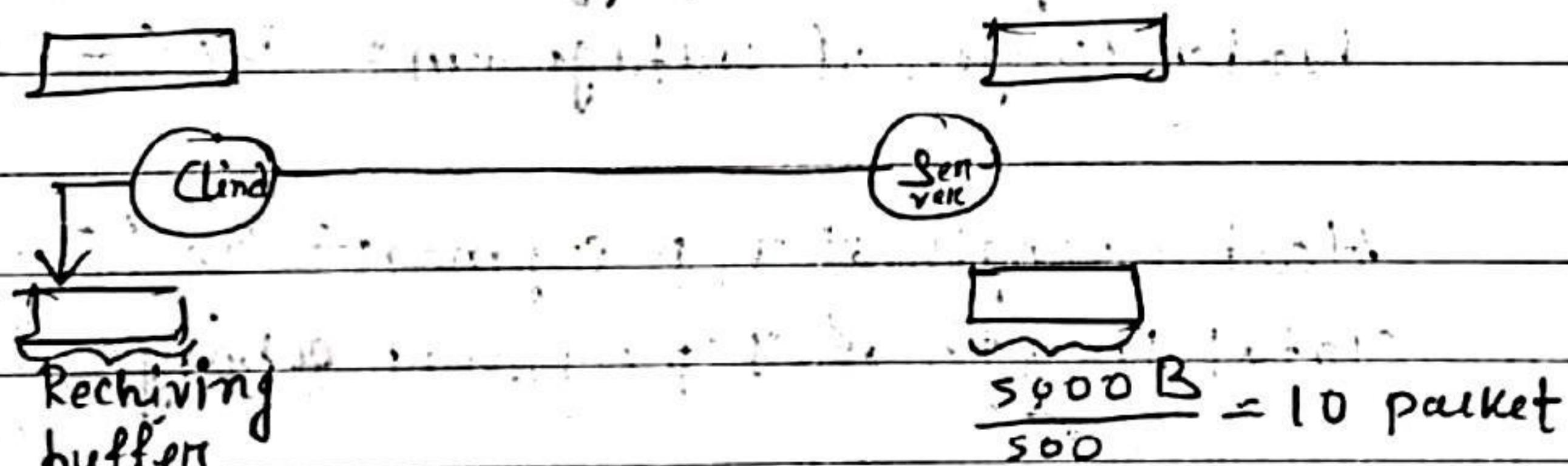
- SYN → Synchronization Flag
- ACK → Acknowledgement Flag



TCP is connection oriented - it has 3 phases

- (i) Connection establishment
- (ii) Data transfer
- (iii) Connection termination

→ TCP is Full duplex.



$$\frac{14600}{1460} = 10 \text{ packet}$$

the host as a window size means what is the capacity of a client (receiving window) is used to flow control.

$$\text{SYN} = 1 \text{ seq num.}$$

→ means whenever you send a packet with "SYN=1" whatever sequence number present in the packet that is consumed.

ACK<sub>1</sub>

n C

521

 $\text{Seq} = 521, \text{SYN} = 1$
 $\text{MSS} = 1460, \text{window} = 14600 \text{ B}$

S. 80

→ request packet (1)

2000 (Random initial sequence
number)
 $\text{Seq} = 2000, \text{SYN} = 1$
 $\text{MSS} = 500 \text{ B}, \text{window} = 5000 \text{ B}$
 $\text{ACK} = 1, \text{ACK} = 522$

→ reply packet (2)

 $\text{Seq} = 522, \text{ACK} = 1$
 $\text{ACK} = 2001, \text{ACK} = 522$

→ ACK packet (3)

fig. 1

→ SYN = 1, will consume 1 sequence number.

ACK = 1, will not consume any sequence number.

FIN = 1, consume 1 SN.

1 Data ≈ 1 sequence number
B

→ After sending this 3 packets connection will be established therefore it is also called as "3-way hand shake".

● Data transfer after connection establishment -

→ next page.

(7 - 22) remain

→ every packet
carries ACK. (ACK=1)

sending 100B of data

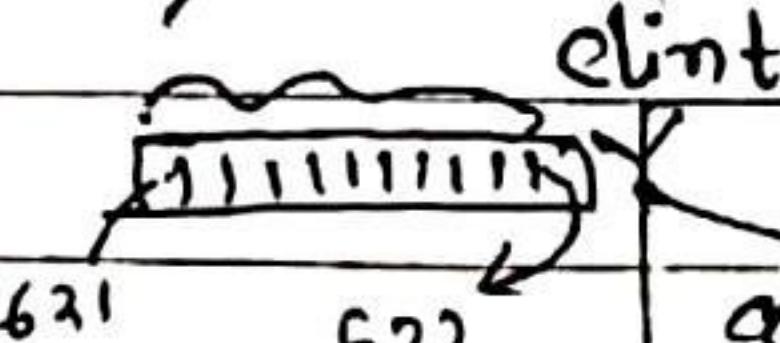
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Date _____

Page _____

after
previous

Fig. 1



client

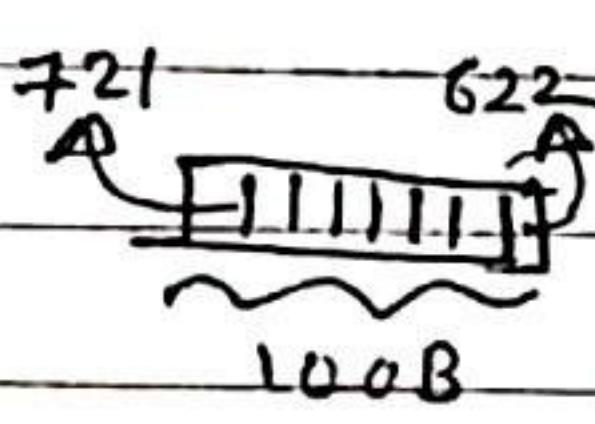
server

seq = 522
ack = 200, ACK = 1

2001 2100

100B

Pure ACK



seq = 622, ACK = 1

seq = 622, ack = 2101, ACK = 1

seq = 2101, ack = 2101, ACK = 1

Pure
ACK

SYN

ACK

0 — (I, segment / request)

1 — (II, segment / reply)

ACK { 0, 1 — only ACK is present in the header.

ACKnowledgement flag.

0, 0 — not possible.

~~DATA~~

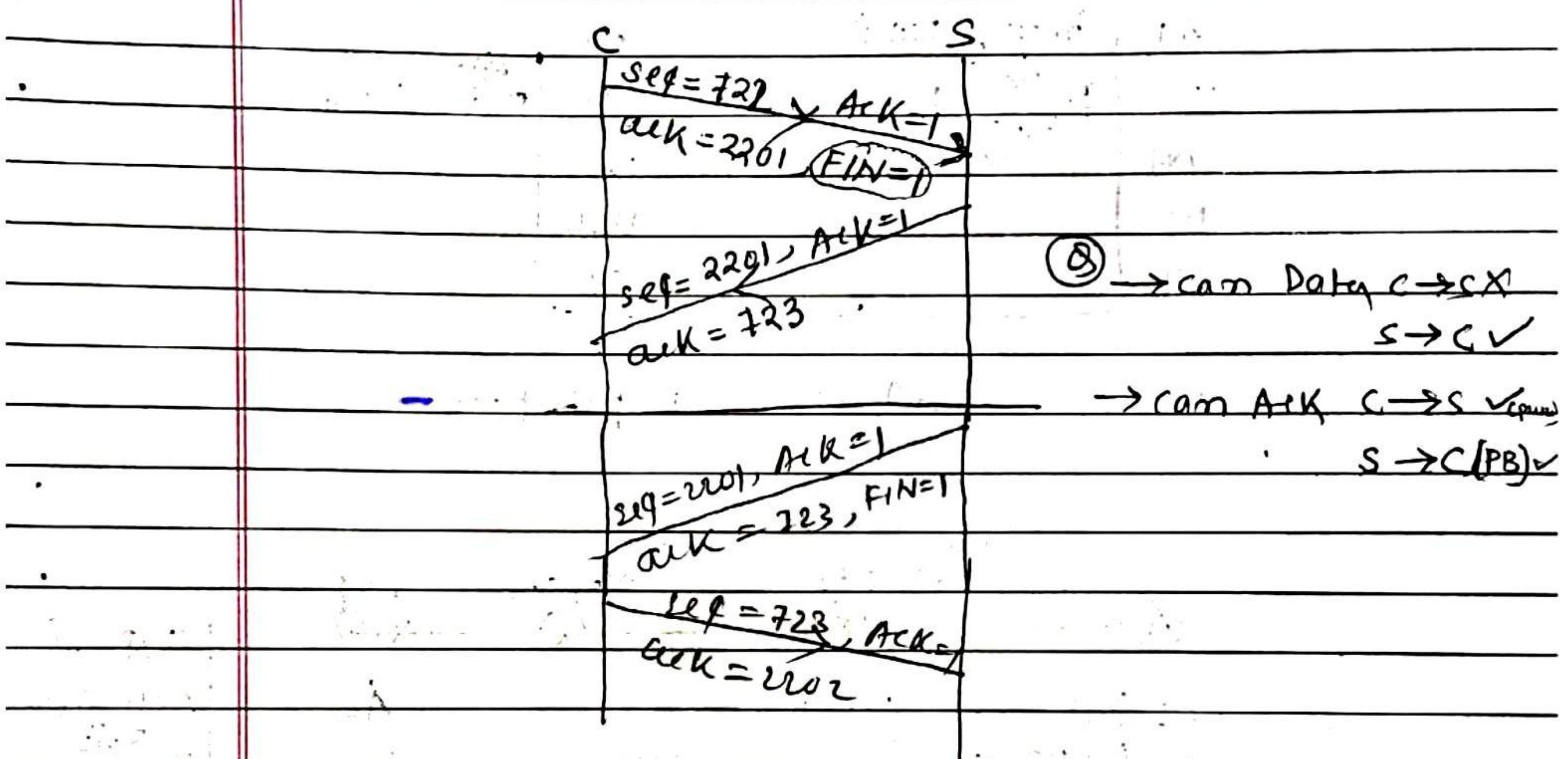
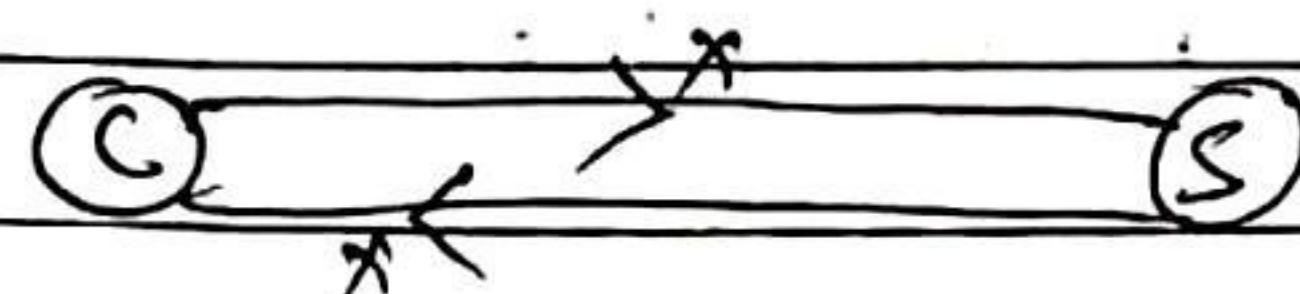
close connection.
FIN=1

- Connection termination — (close connection) b/w C & S

→ TCP is a full-duplex connection.

FIN=1

↳ finish



→ 3 packet for connection establishment,
→ 4 " " " for termination.

3 phases — Connection Establishment.



Data transfer



Termination.

3-Fig used -

1) SYN : Synchronizing seq number.

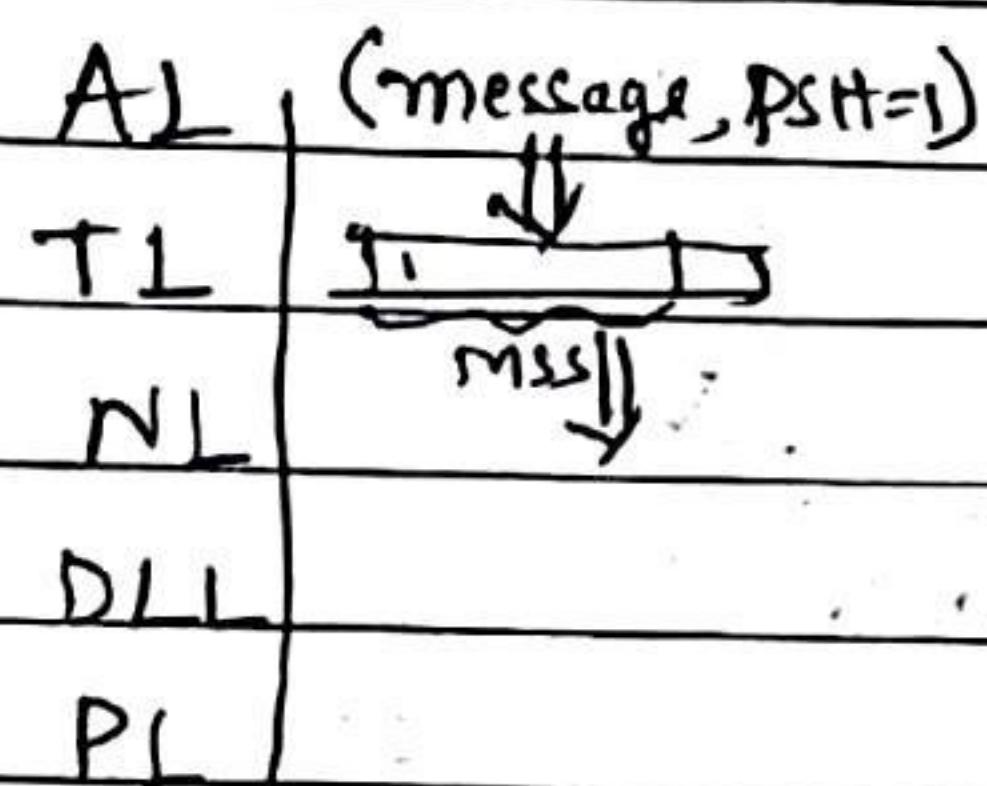
2) ACK : Indicate whether ACK num field is valid.

3) FIN : Request for connection termination.

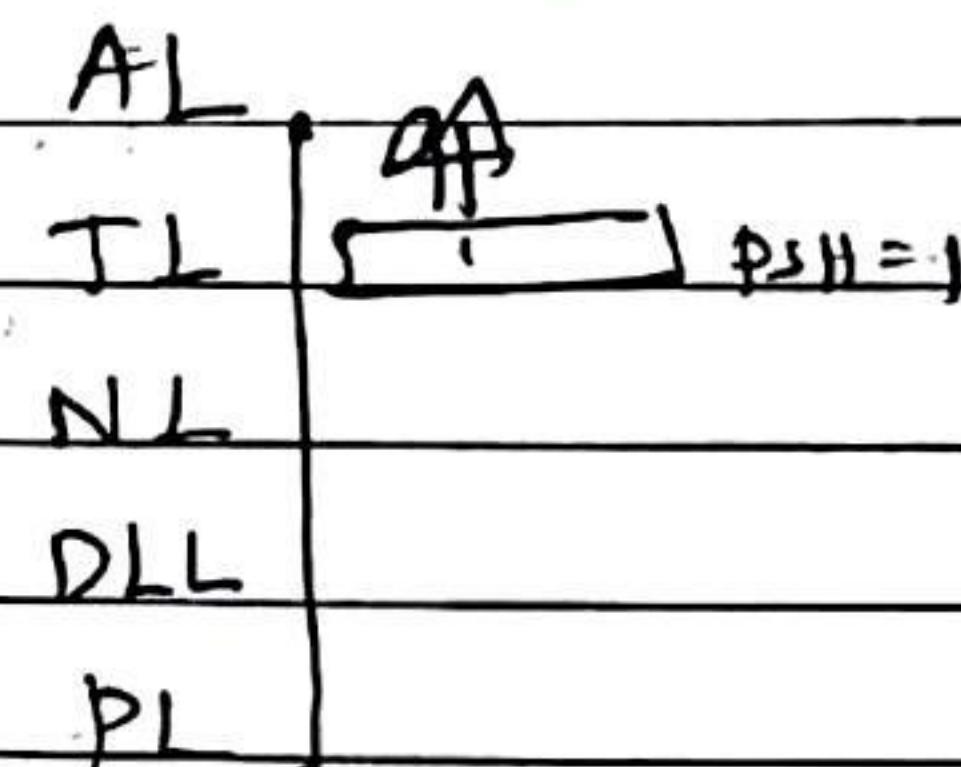
• PSH flag - (push flag)

→ PSH used to push the data immediately downward as well as upward.

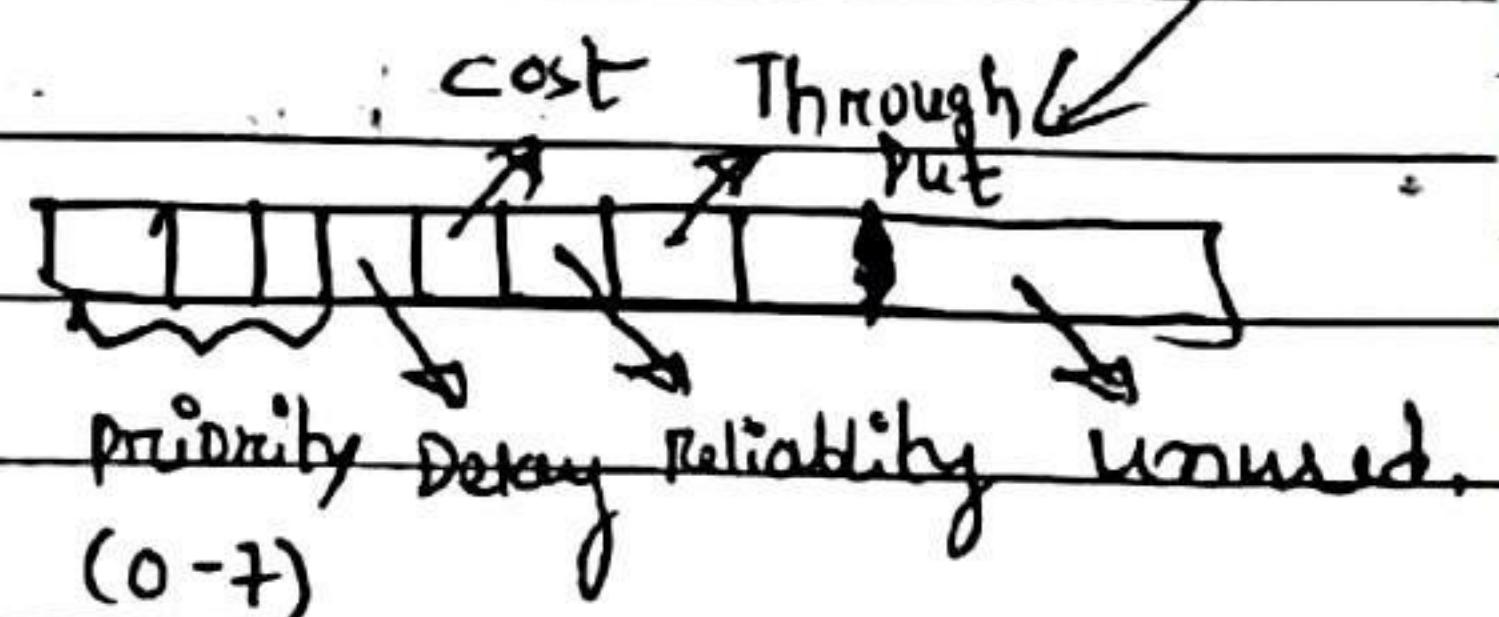
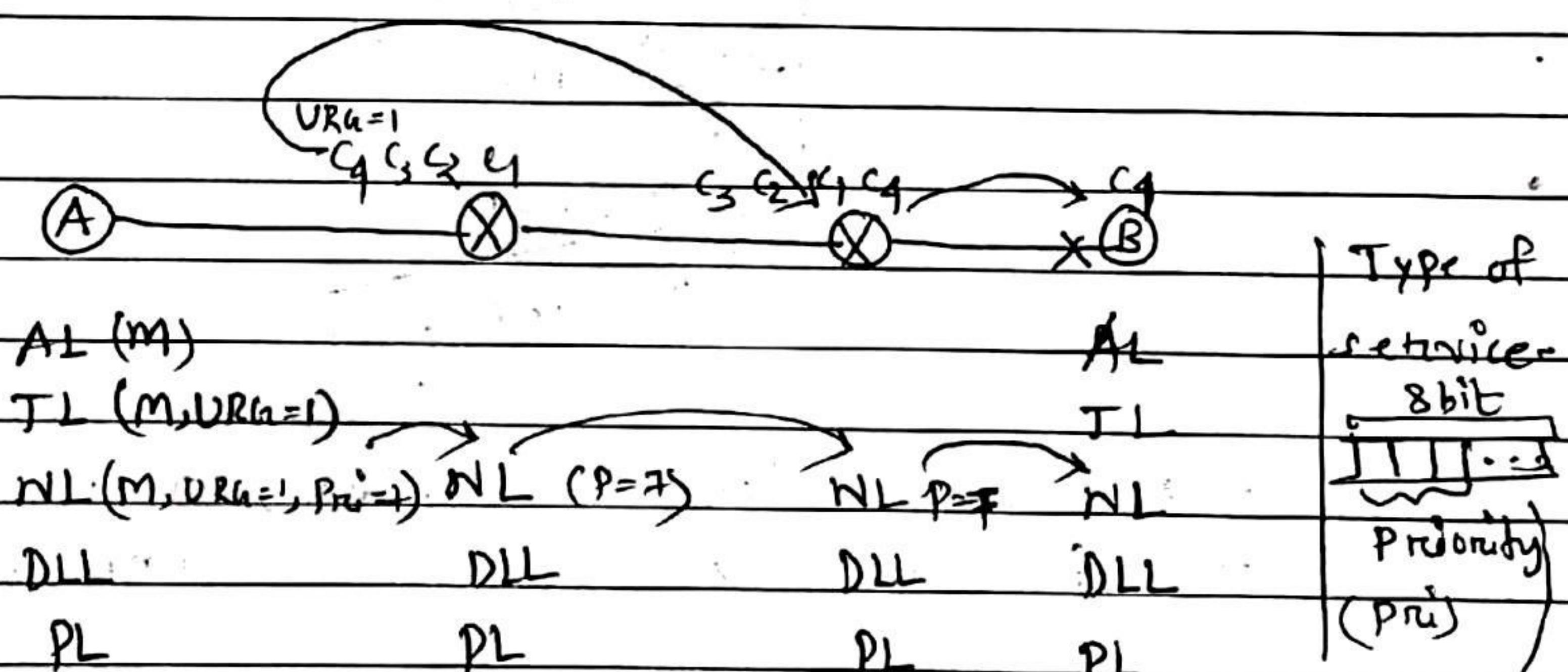
Sender



Receiver



• URG flag and Urgent pointers -



When Delay = 1 means - send data with least delay.

Cost = 1 " - " " " cost.

Reliability = 1 " - " " high reliability

Urgent pointer -

Urgent pointer indicates till what part Data is urgent.

$$Seq = 1000$$

$$URG = 1$$

$$Urg\ pointer = 100$$

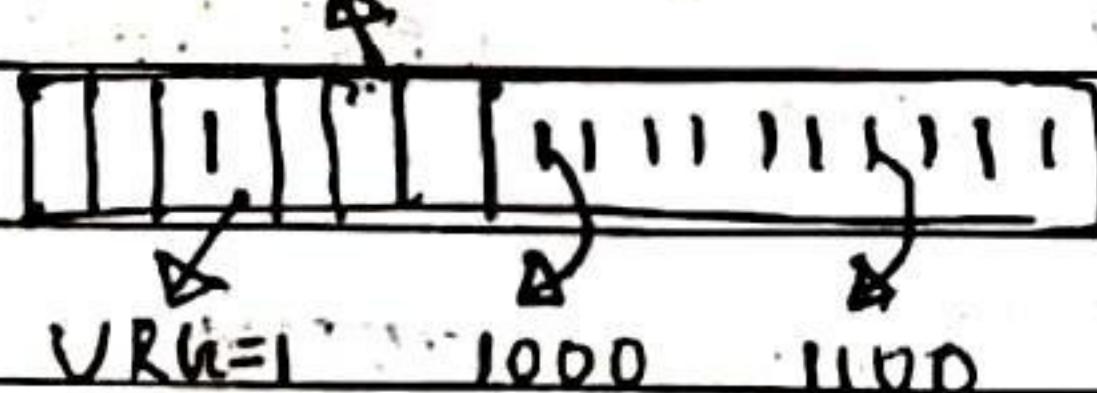
$$\Rightarrow Seq + Urg\ pointer$$

$$\Rightarrow 1000 + 100$$

$$\Rightarrow 1100 \text{ — Urgent Byte.}$$

$$URG\ pointer = 100$$

TCP segment



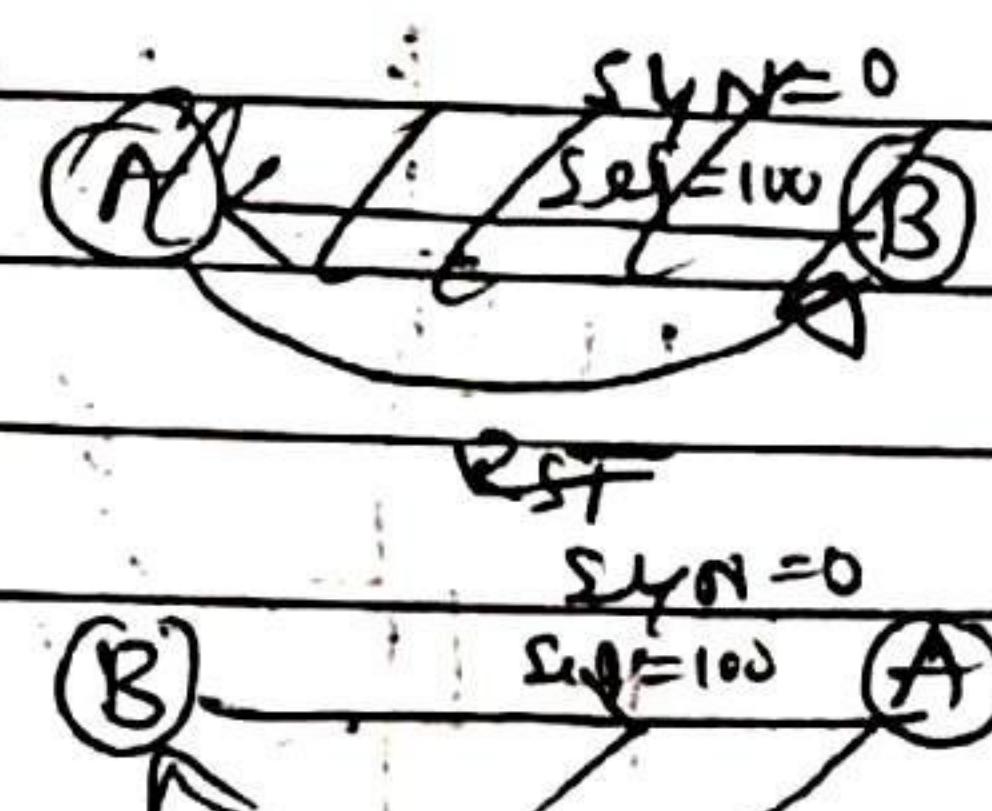
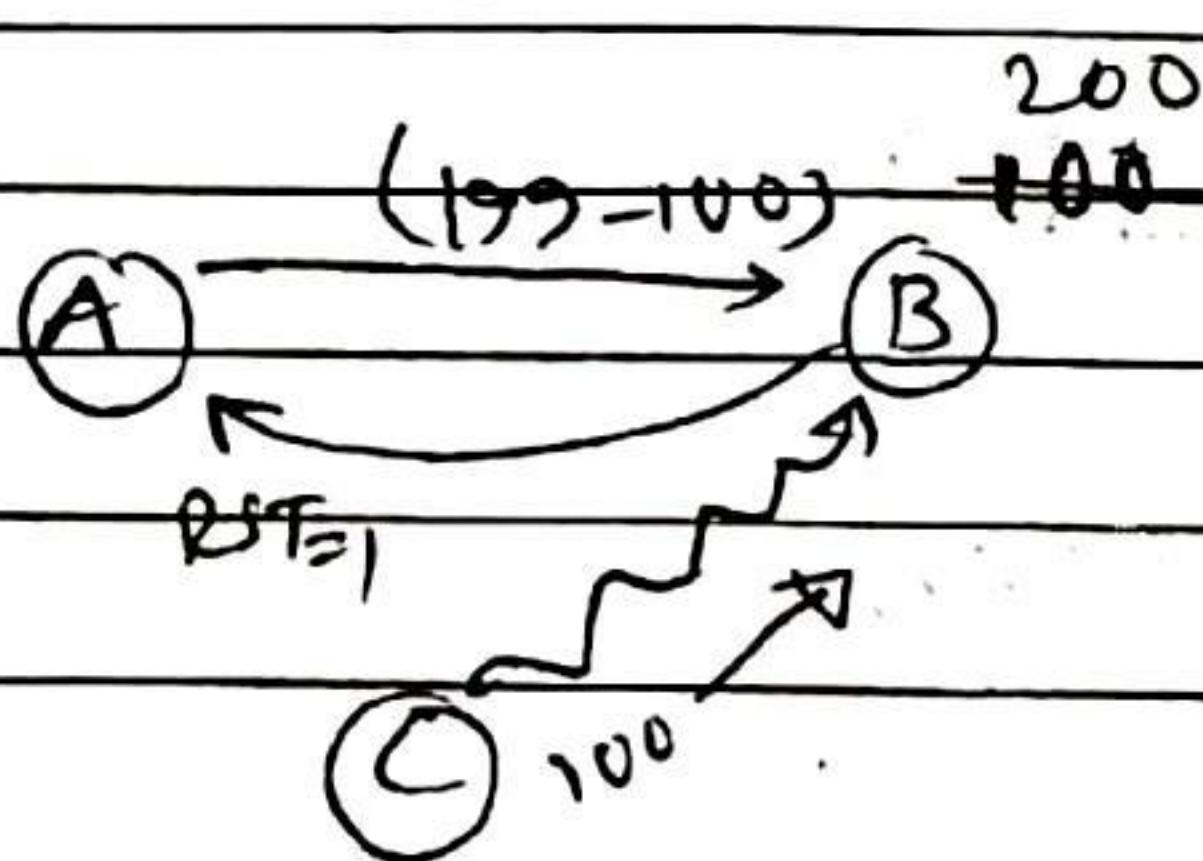
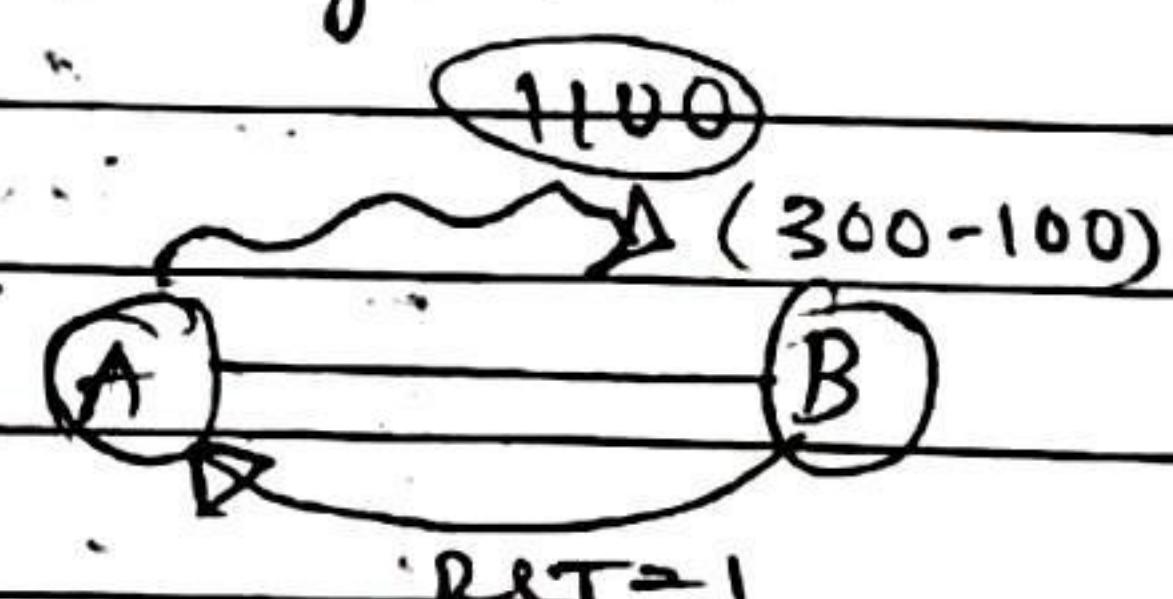
$$1000 \text{ — } 1100$$

$$\Rightarrow 101 \text{ — Urgent.}$$

RST Flag (Reset flag) -

to terminate connection.

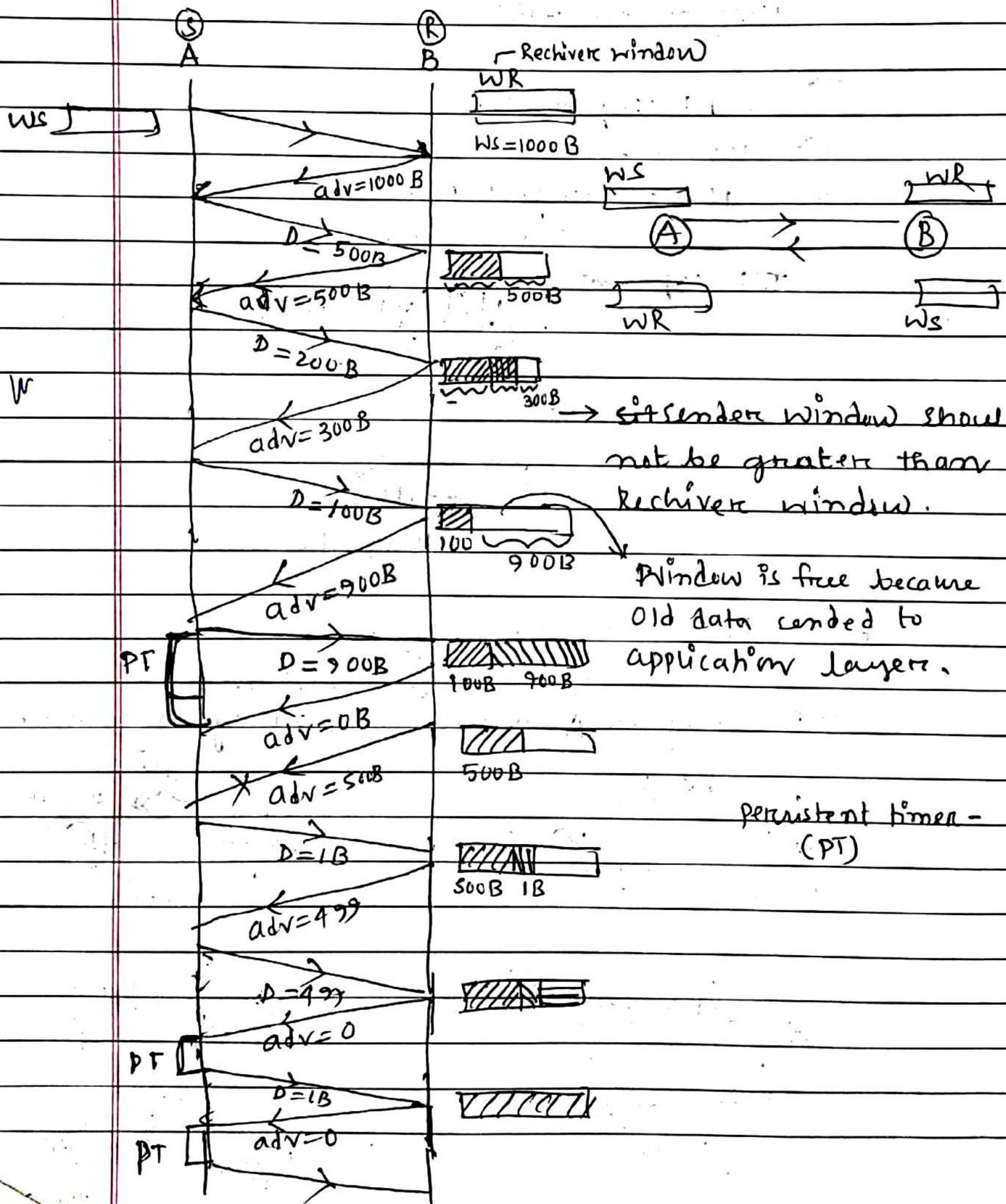
→ RST Flag used whenever any wrong. in connection.



- Flow control using advertisement window - (16 Bit)

- window size & Adv window -

↓
(flow control).



~~Advantage of window size~~

→ adv window used for flow control.

$$2^{16} - 1$$

$$65,535$$

1GB

(A)

(B)

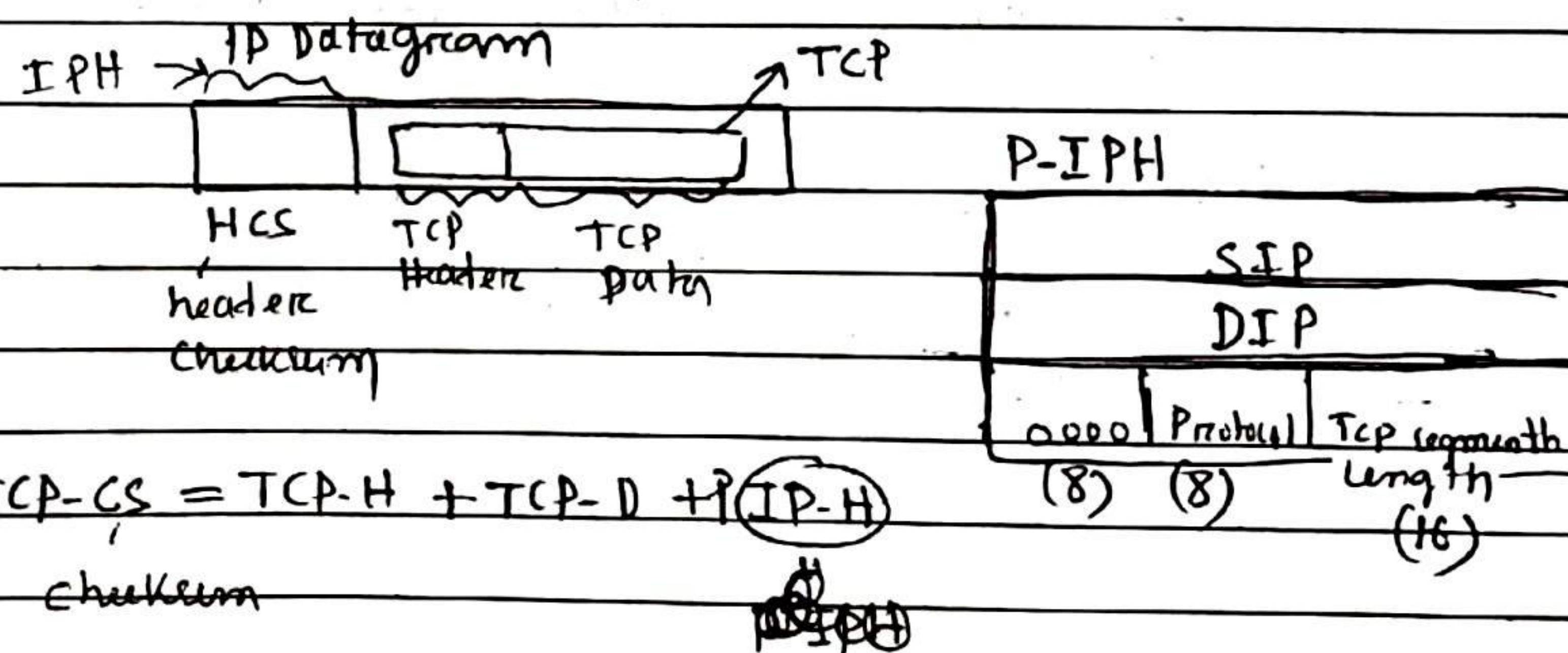
$$16 + 14 \text{ (option)}$$

$$= 30 \text{ bit}$$

$$= 2^{30} - 1$$

$$= 1GB$$

• checksum - (16)



B. Why checksum used two time in IPH & TCPH.

→ for double checking.

- Options :

- W_{AT}
 amount
 time
- life
 time
- (I) Time stamp (WAT < LT) $\rightarrow (00)'100$
 - (II) window size extension $\rightarrow (01)'100$
 - (III) Parameters negotiation.
 - (IV) padding.

- Retransmissions in TCP :

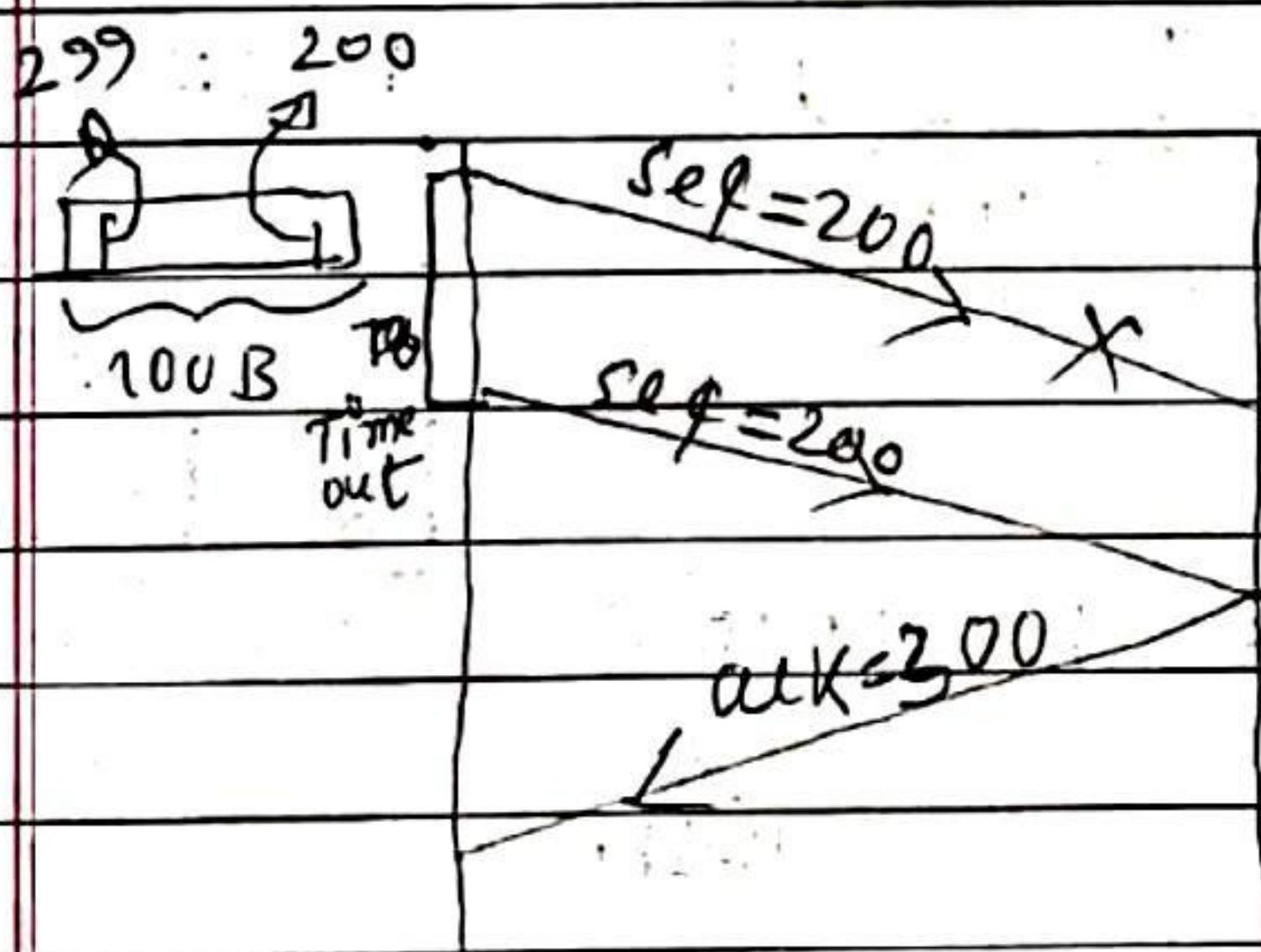
\rightarrow TCP uses as SR + GBN (Go Back N)

$\rightarrow W_s = W_R \rightarrow$ Acknowledge after retransmission - time

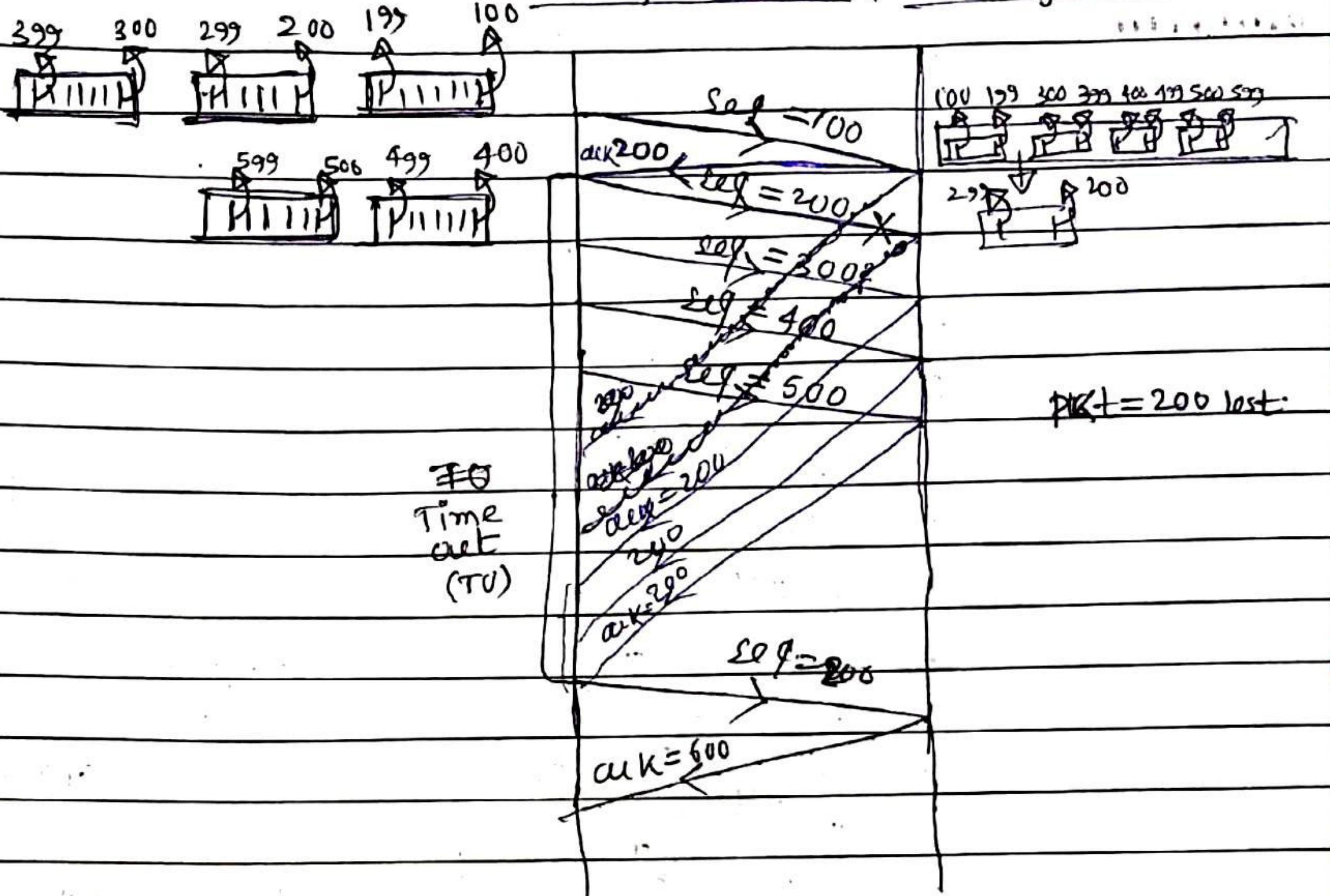
\rightarrow out of order

75% 25%

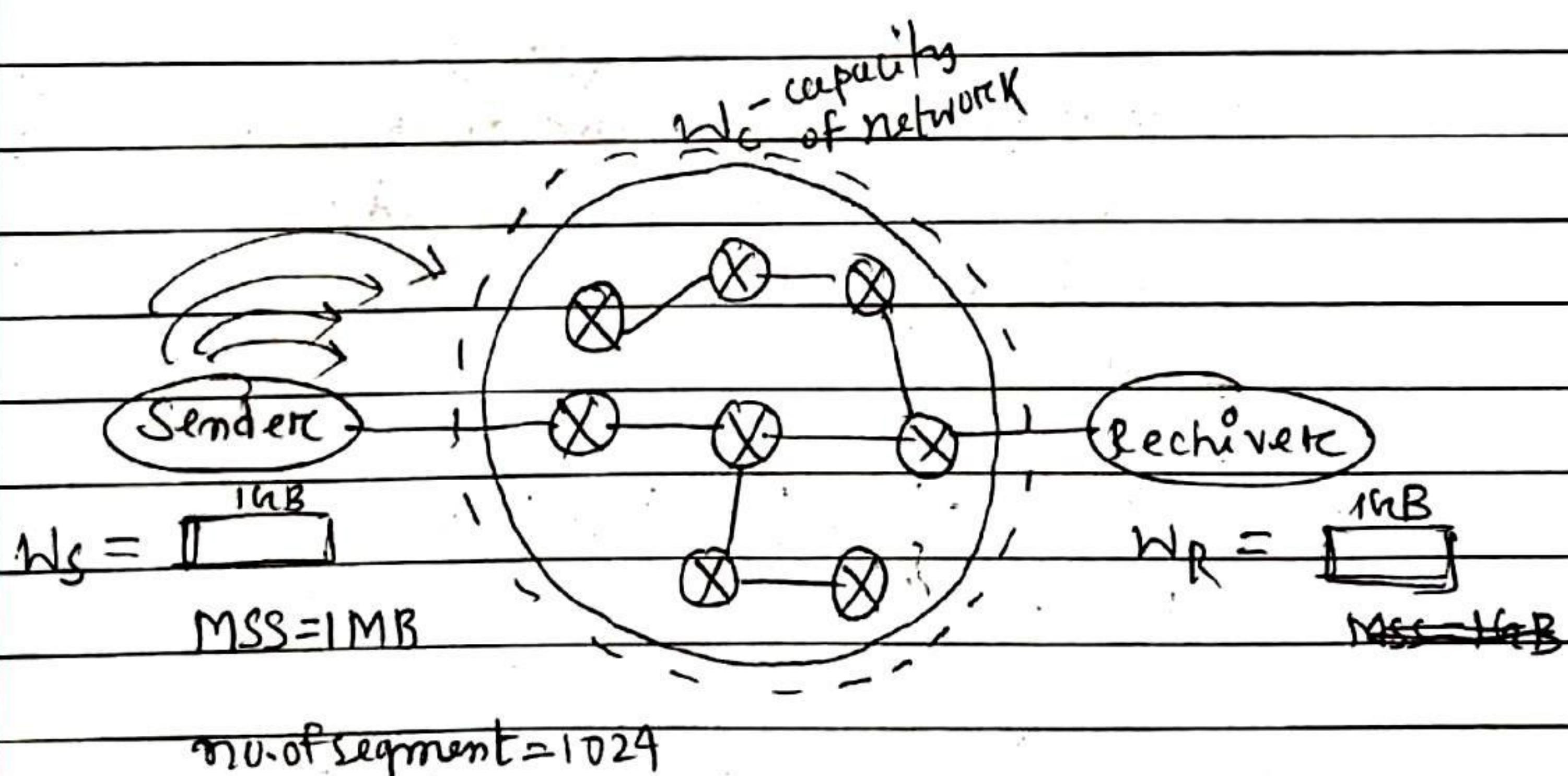
- Retransmission after time out, —



Retransmission after 3-duplicate ack: Early retransmission



• TCP Congestion control —



before send the data we should know capacity of = \square
 sending data = $\min(W_S, W_R)$.

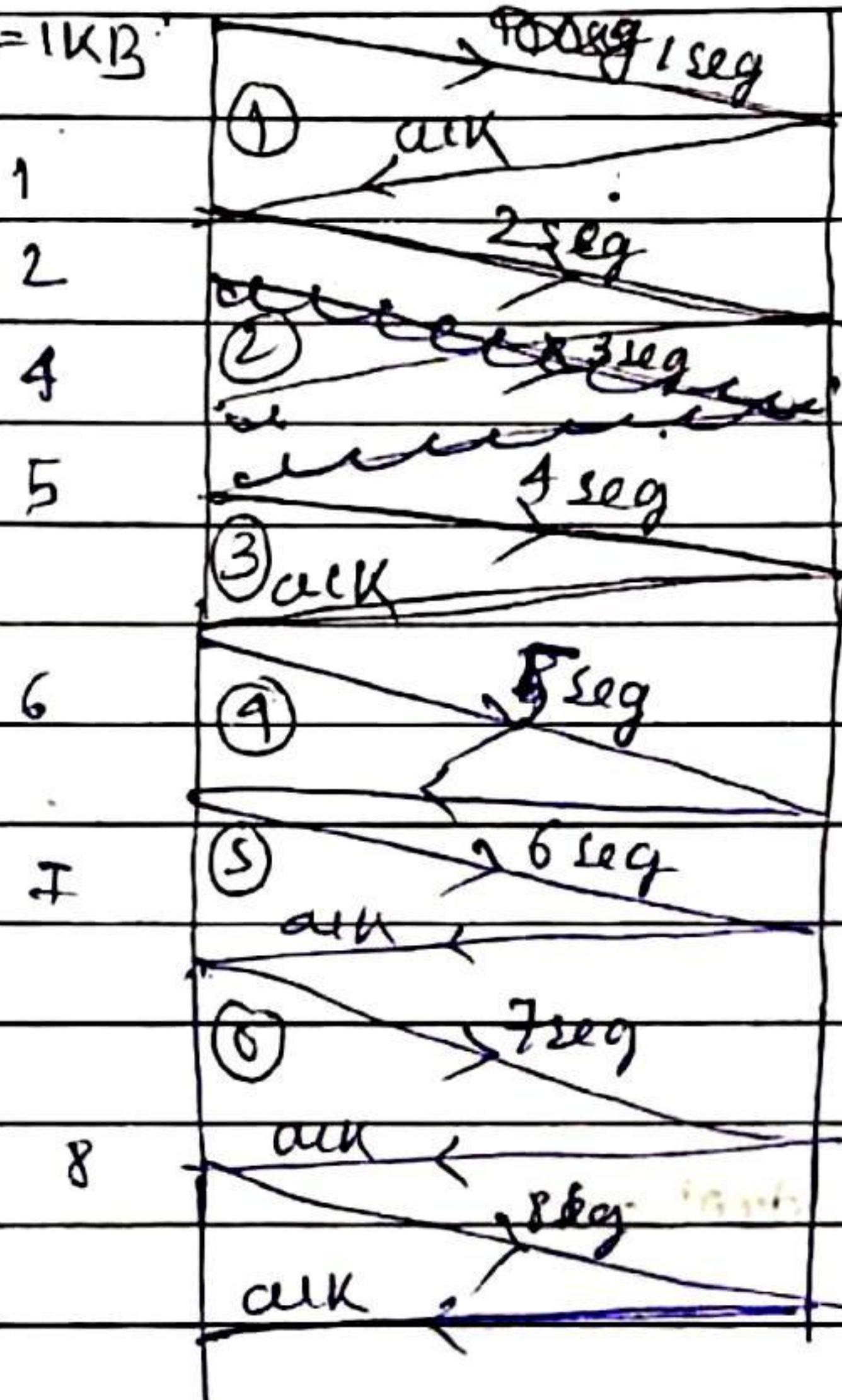
$$\begin{matrix} 1MB & 1MB \\ 1GB & 1MB \end{matrix}$$

MMMM

→ TCP (Congestion Control) is going to be protective layers for network.

adv window = 8 KB

MSS = 1 KB



adv win = 8 KB

MSS = 1 KB

WP = 8 segment

$W_C = 1 \text{ segment}$

congestion window
stalling with 1 segment

RTO = 1

$$\text{Threshold} = \frac{WP}{\alpha} = 4$$

WS = 1 segment

2 seg

4 seg

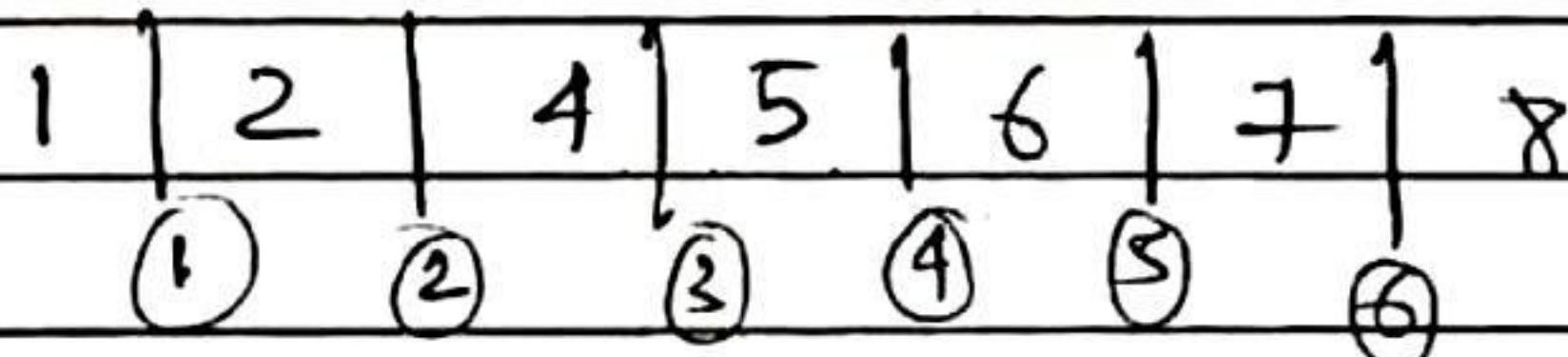
5 "

6 "

7 "

8 "

Q. after how many round trip time will be reach the max sender capacity?
⇒ Answer is 6

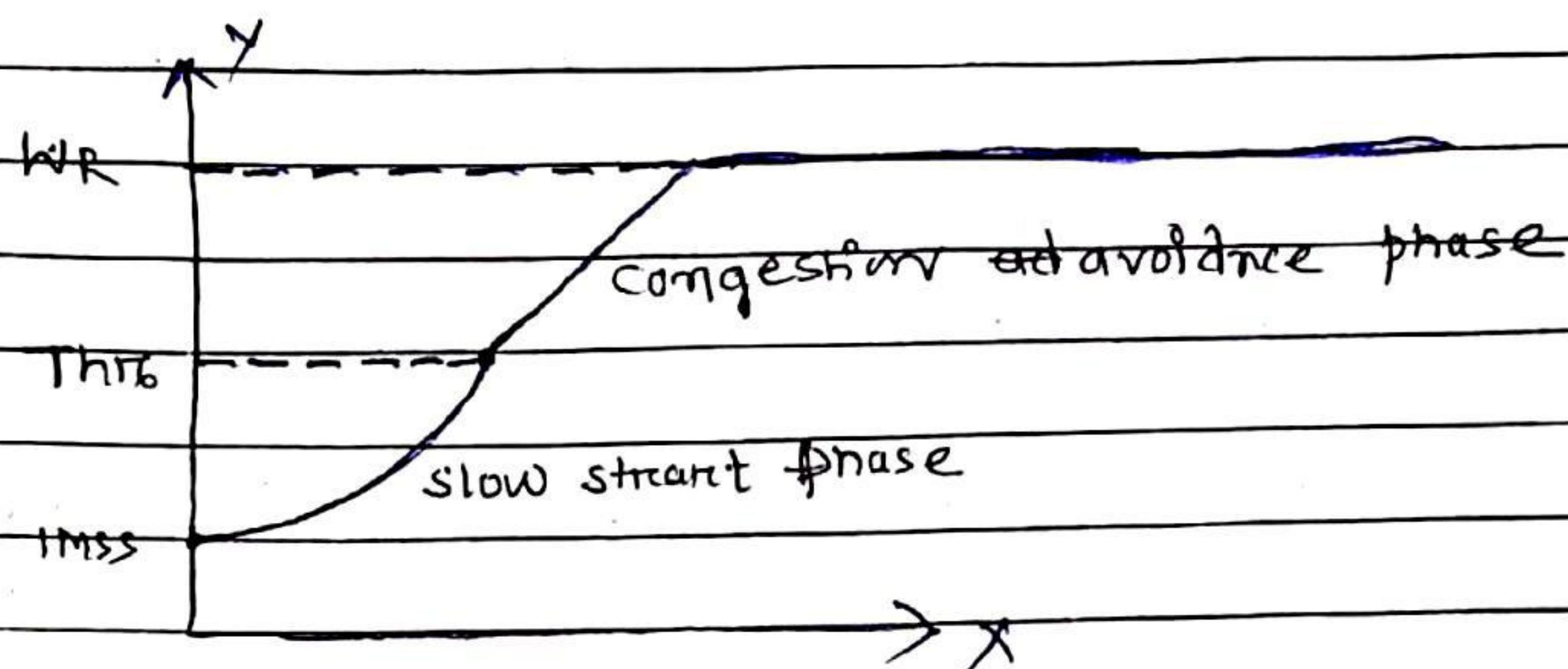


Q. when what will be RTT = ? (to reach max sender capacity)
when, Threshold = 8 / WP = 16

⇒ 1 | 2 | 4 | 8 | 9 | 10 | 11 | 12 | 13 | 14 | 15 | 16

Answer = (11) RTT

↓
Fast full window



Congestion control algorithm

(i). slow start phase .

(ii). congestion avoidance phase .

(iii). congestion Detection phase .

- way { (a) Time out . (congation severe) } $\text{newTh} = \frac{1}{2} W_L$
- to detect (b) & 3 duplicate ack . (congation mild)
- congection (c) ICMP source quench . }

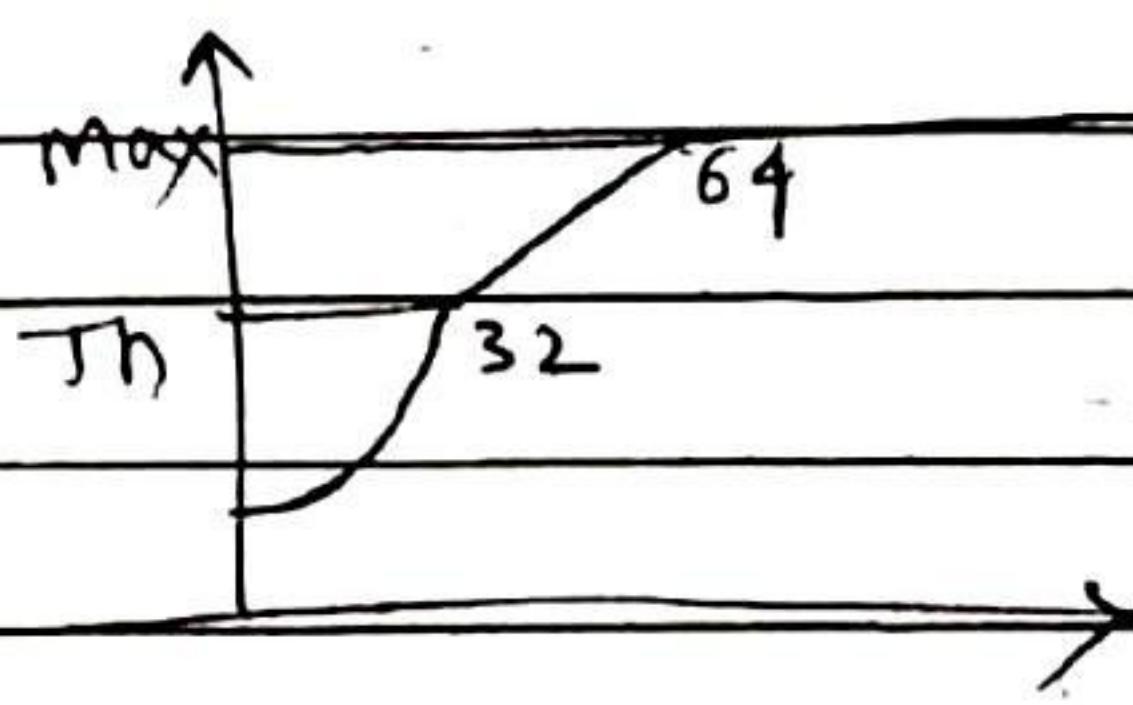
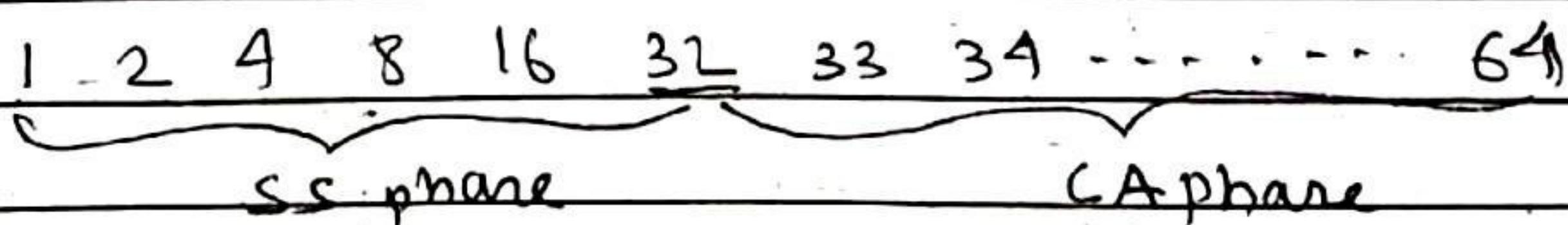
ex: (with example)

$$W_p = 64 \text{ KB}$$

$$WSS = \cancel{1KB} + 1 \text{ KB}$$

$$WR = 64 \text{ MSS}$$

$$Th = \frac{64}{2} = 32 \text{ MSS}$$



congestion detect

(a) T0 growth $\rightarrow \frac{1}{2} W_c \rightarrow$ (SS phase)

(b) 3 dup acks \rightarrow new Th $\rightarrow \frac{1}{2} W_c \rightarrow$ (ACK phase)

$W_R = 64KB$ (congestion detection)

MSS = 1KB

$W_R = 64MSS$

Th = 32 MSS

Th=17

1 2 4 8 16 32 33 34 ↑ 1 2 4 8 16 17 18

Th=10

Th=6

Th=4

19 20 ↑ 10 11 12 ↑ 1 2 4 8 6 7 8 ↑ 4 5 6 7 - 64

3 dup ack

T0

3 dup ack

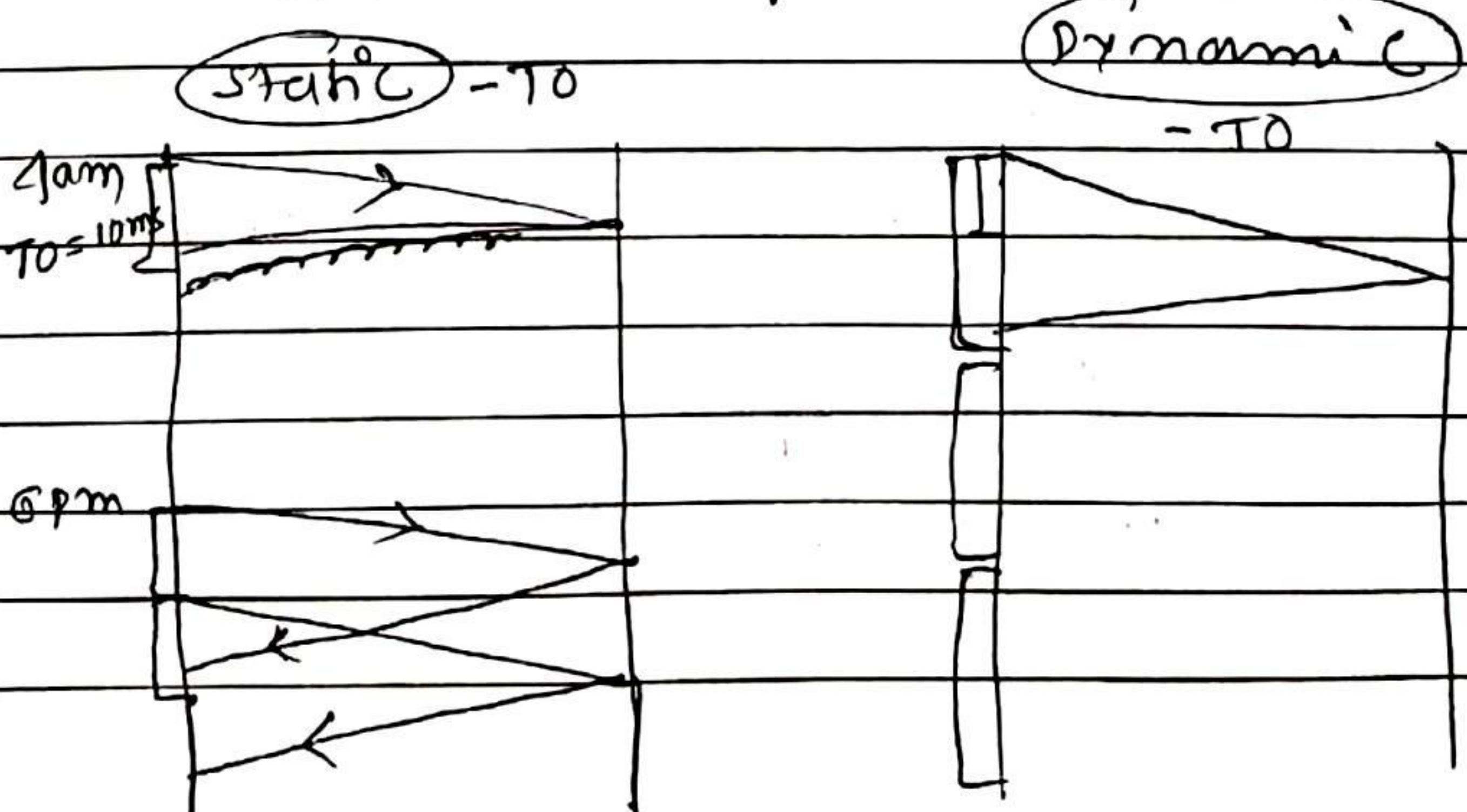
• TCP Timer Management —

- (1) Time-wait timer \rightarrow Adv (to take care of late pkt)
- (2) keep alive timer \rightarrow Adv (close idle connection)
- (3) persistent timer \rightarrow used (window size = 0 is advertised)
- (4) acknowledgement timer (ACK cum ARK, PB ACK)
- (5) time-out timer.

Time out timer —

(1) Basic algo.

(2) Jacobson's algo.



PKT segment

TO timer ↑
→ Time waste if packet lost

TO-timer ↓
→ unnecessary RTT Page

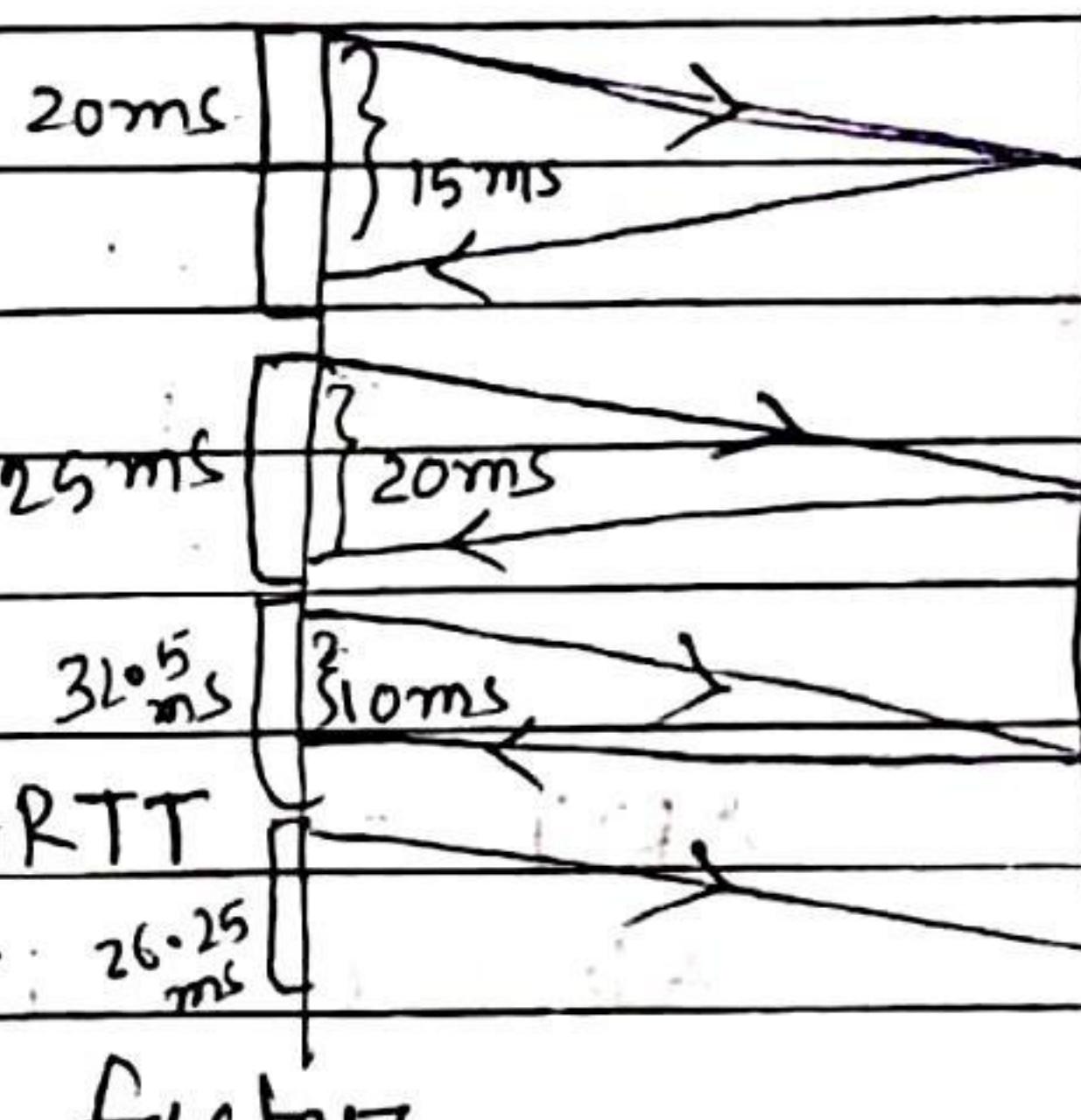
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• Basic algorithm for timeout timers -

RTT
initial round trip time

$$IRTT = 10 \text{ ms}$$

$$\begin{aligned} TO &= 2 * RTT \\ &= 20 \text{ ms} \end{aligned}$$



TO
→ time out

$$ARTT = 15 \text{ ms}$$

$$NRTT = \alpha * IRTT + (1-\alpha) ARTT$$

$\alpha \rightarrow$ Smoothing factor.

$$[0 \leq \alpha \leq 1] \quad \alpha = 0.5$$

ARTT
actual RTT

NRTT
→ next RTT

$$= 12.5 \text{ ms}$$

$$TO =$$

$$IRTT = 12.5 \text{ ms}$$

$$TO = 25 \text{ ms}$$

$$ARTT = 20 \text{ ms}$$

$$NRTT = \alpha IRTT + (1-\alpha) ARTT$$

$$\alpha = 0.5$$

$$= 16.25 \text{ ms}$$

$$IRTT = 16.25 \text{ ms}$$

$$TO = 2 * (2 * 16.25) = 32.5 \text{ ms.}$$

$$ARTT = 10 \text{ ms}$$

$$NRTT = 13.125 \text{ ms}$$

$$IRTT = 13.125 \text{ ms}$$

$$TO = (2 * 13.125) = 26.25 \text{ ms.}$$

✓ Disadvantage -

$$TO = 2 * RTT$$

Jacobson modify it.

Jacobson's algorithm —

$$\textcircled{1} \quad IRTT = 10 \text{ ms}$$

ID
→ initial deviation

$$ID = 5 \text{ ms}$$

$$T_0 = 4 * D + RTT \\ = 4 * 5 + 10 \\ = 30 \text{ ms}$$

$$ARTT = 20 \text{ ms}$$

$$AD = 10 \text{ ms} (|IRT - ARTT|)$$

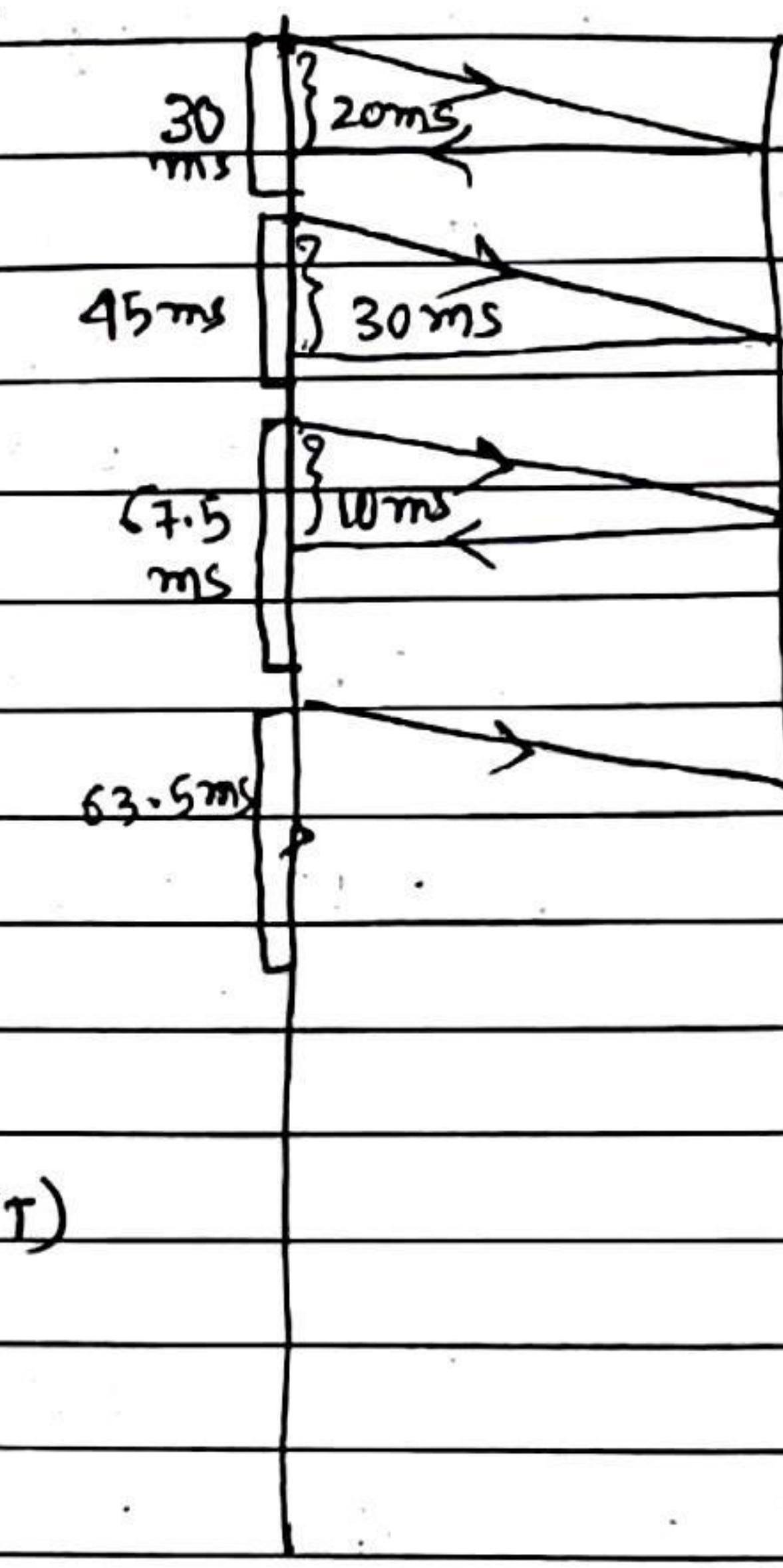
$$NRTT = \alpha (IRT) + (1-\alpha) (ARTT) \\ = 15 \text{ ms} \quad (\alpha = 0.5)$$

$$ND = \alpha (ID) + (1-\alpha) (AD) \\ = 7.5 \text{ ms}$$

$$\textcircled{2} \quad IRTT = 15 \text{ ms}$$

$$ID = 7.5 \text{ ms}$$

$$T_0 = 45 \text{ ms}$$



$$\textcircled{3} \quad IRTT = 22.5$$

$$ID = 7.5$$

$$T_0 = 67.5 \text{ ms}$$

$$ARTT = 30 \text{ ms}$$

$$AD = 15 \text{ ms}$$

$$ARTT = 10 \text{ ms}$$

$$AD = 12.5$$

$$NRTT = 22.5$$

$$NRTT = 16.25$$

$$ND = 11.25$$

$$ND = 11.25$$

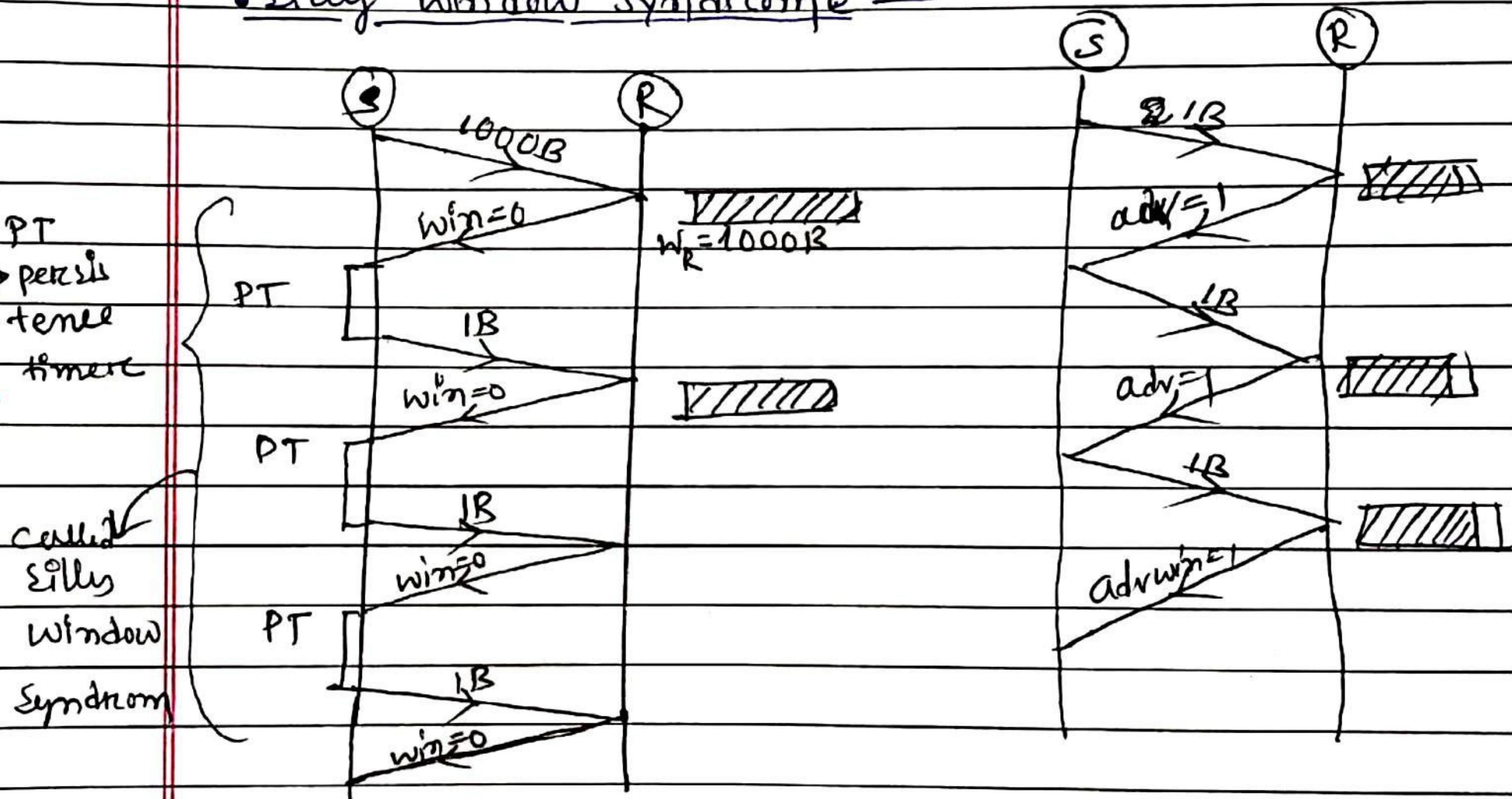
$$T_0 = 4 * D + RTT$$

$$= 63.75$$

Karn's modification on Basic & Jacobson's algo-

→ Karn's modification is said that, if the acknowledgement goes across that the range of Time out timers then next time Resend the segment with Double TO. ($2 \times TO$).

• Silly window syndrome —



✓ Nagles algorithm —

→ if the sender is very slow then Transport layer should not send 1B of Data, Transport layer should wait for 1 RTT.

✓ Clark —

→ Destination should not advertised 1B until it should wait until it can send at least 1 MSS or $\frac{1}{2}$ of Buffer.