



Computer Networks

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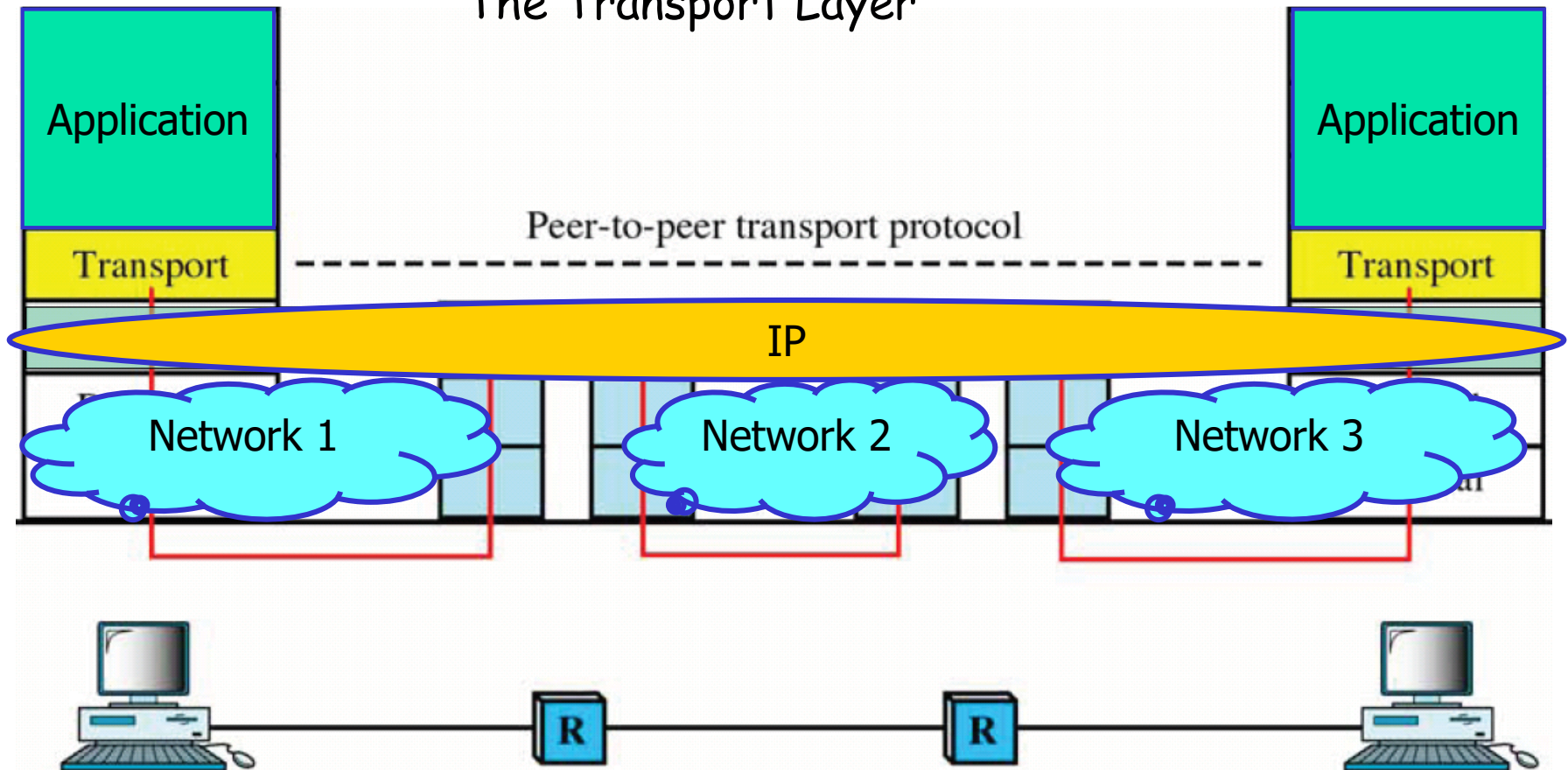


Chapter 5. End-to-End Protocols

- **Transport Services and Mechanisms**
- User Datagram Protocol (UDP)
- Transmission Control Protocol (TCP)
- TCP Congestion Control
- Real-time Transport Protocol (RTP)
- Session Initiation Protocol (SIP)
- Real Time Streaming Protocol (RTSP)

Transport Services and Mechanisms

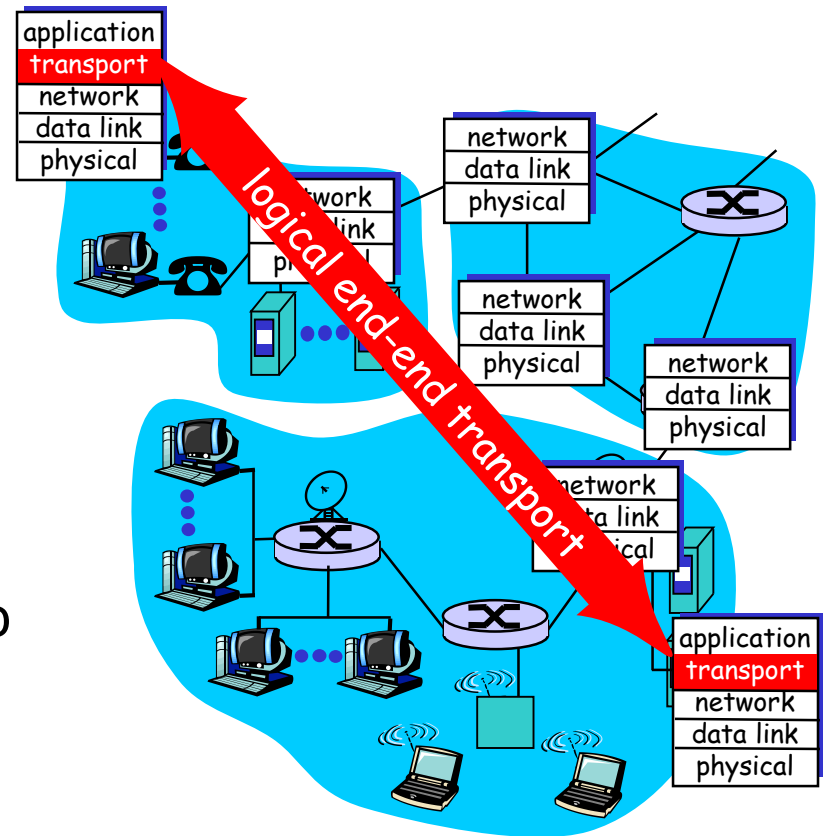
The Transport Layer





Internet Transport Services

- Provide **logical communication** between app processes running on different hosts
- Transport protocols run in end systems
 - **Send side**: breaks app messages into **segments**, passes to network layer
 - **Receive side**: reassembles segments into messages, passes to app layer
- More than one transport protocol available to apps
 - Internet: TCP and UDP





Transport vs. network layer

- Network layer:
 - logical communication between hosts
- Transport layer:
 - logical communication between processes
 - Relies on, enhances, network layer services
 - Multiplexing (复用) & Demultiplexing (分用)



Internet Transport-Layer Protocols

- Reliable, in-order delivery (TCP)

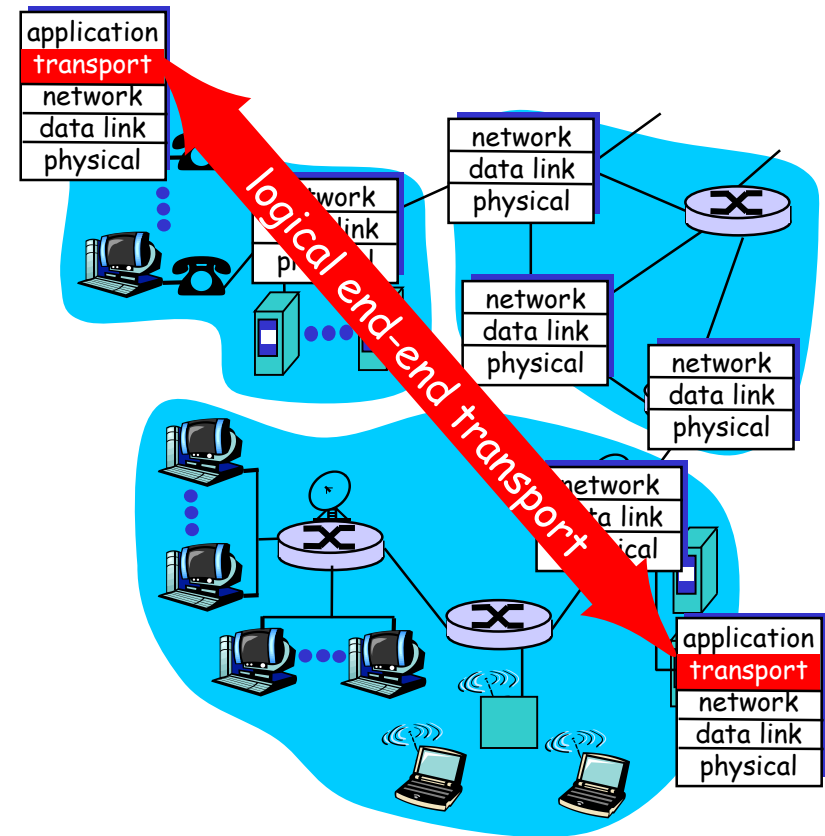
- Connection oriented
- Congestion control
- Flow control

- Unreliable, unordered delivery (UDP)

- No-frills extension of "best-effort" IP

- Services not available:

- delay guarantees
- bandwidth guarantees





Internet Transport Mechanisms

- Addressing and multiplexing
- Connection-oriented mechanisms
 - Flow control
 - Connection establishment and termination
 - Reliable sequencing communication



Addressing

3-level address for application processes on hosts

■ Process identification

- SAP on transport entity, represents a particular transport service (TS) user
- Port number on TCP/UDP

■ Transport entity identification

- Generally only one of each type per host
- Transport protocol identity (TCP, UDP)

■ Host address

- A global Internet address for attached hosts

■ Internet TCP/UDP addressing

- $\langle \text{HostIP}, \text{Port} \rangle$, called a **socket**

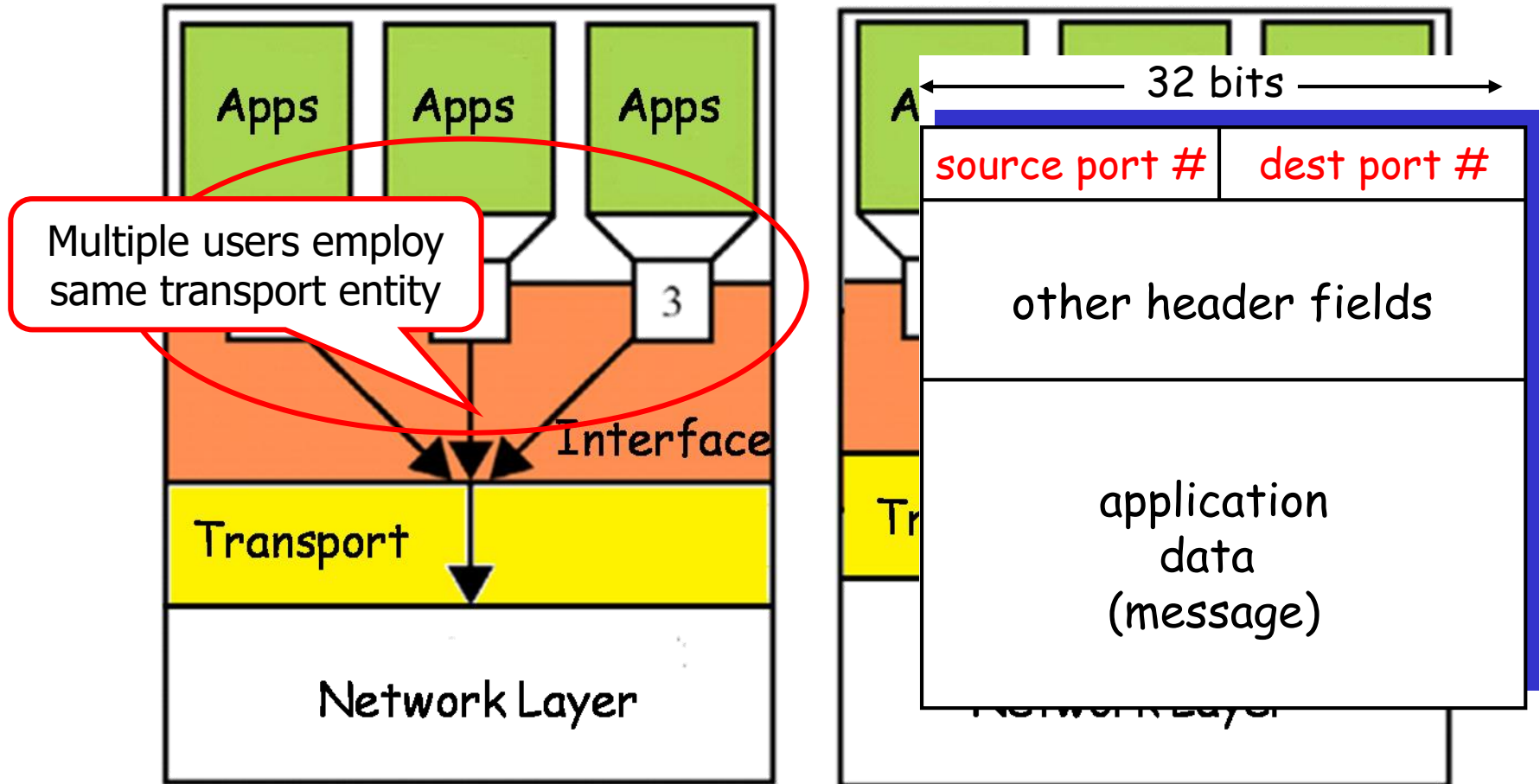


Finding Addresses

- How TS user get to know each other
 - Target port: use conventions
- 4 methods can be used for target host
 - Know address ahead of time
 - Well known addresses
 - Name server, directory service
 - Sending process request to known address
 - Create a TS user on target host



Multiplexing





Multiplexing / Demultiplexing

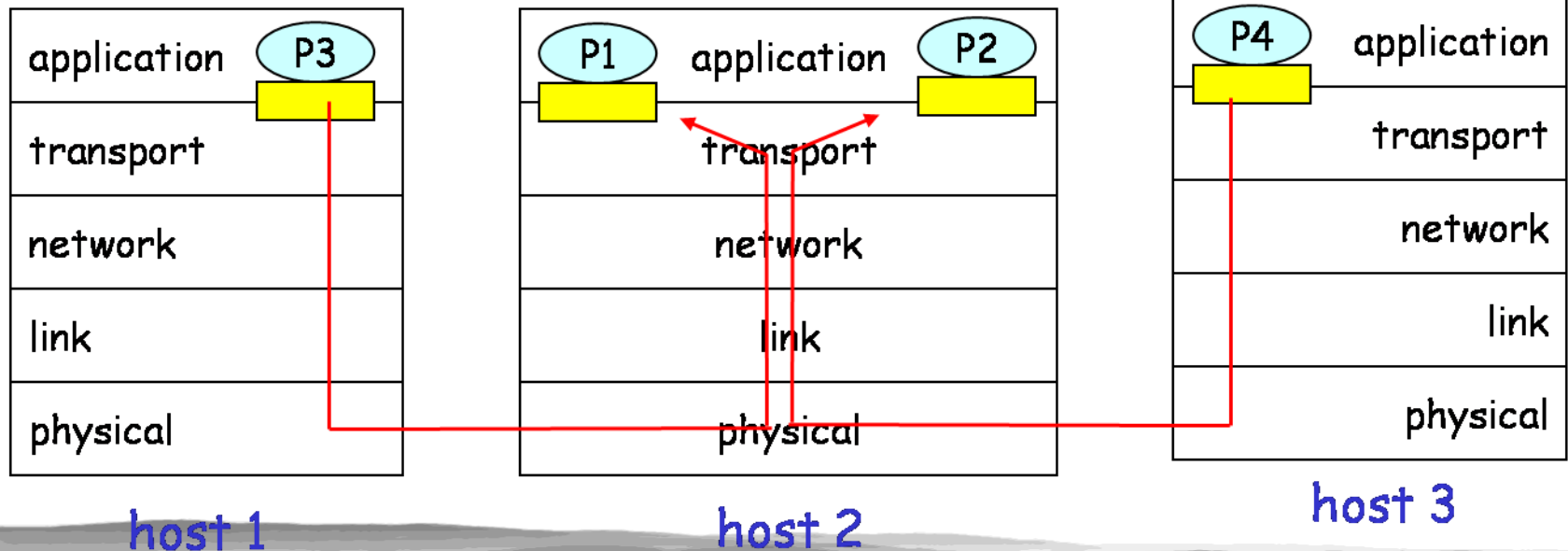
Multiplexing at send host:

Gathering data from multiple sockets, enveloping data with header (for demultiplexing)

Demultiplexing at rcv host:

Delivering received segments to correct socket

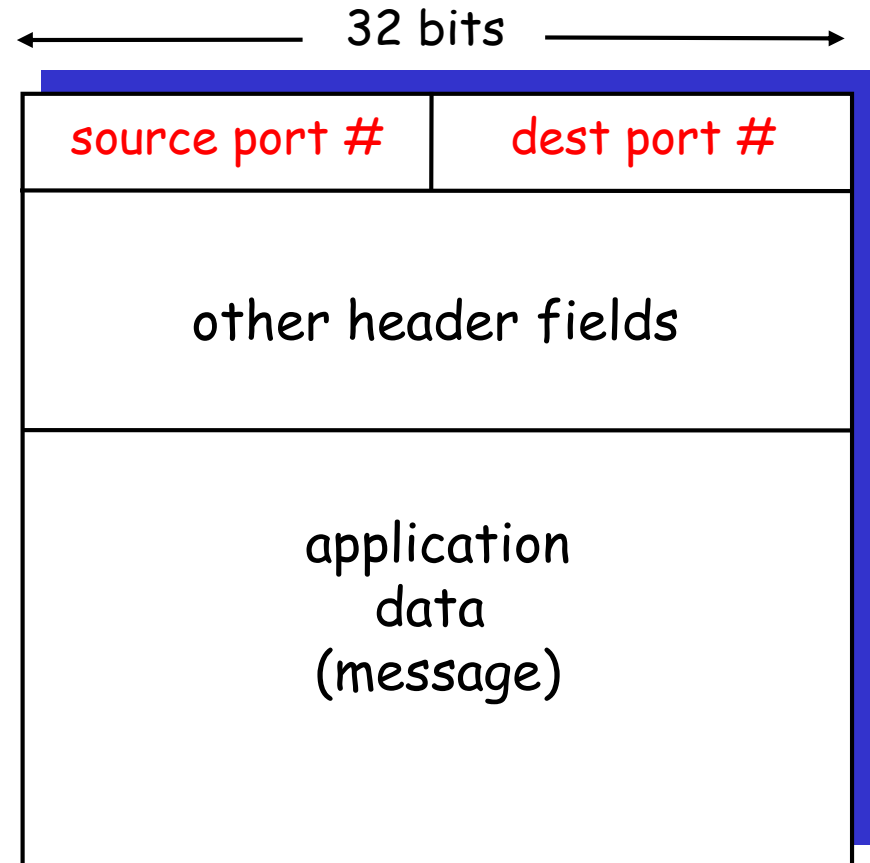
■ = socket ○ = process





Demultiplexing

- TE receives IP datagrams
- Each datagram carries 1 transport-layer segment
- Each datagram has source IP address, destination IP address
- Each segment has source, destination port number
- TE uses IP addresses & port numbers to direct segment to appropriate socket

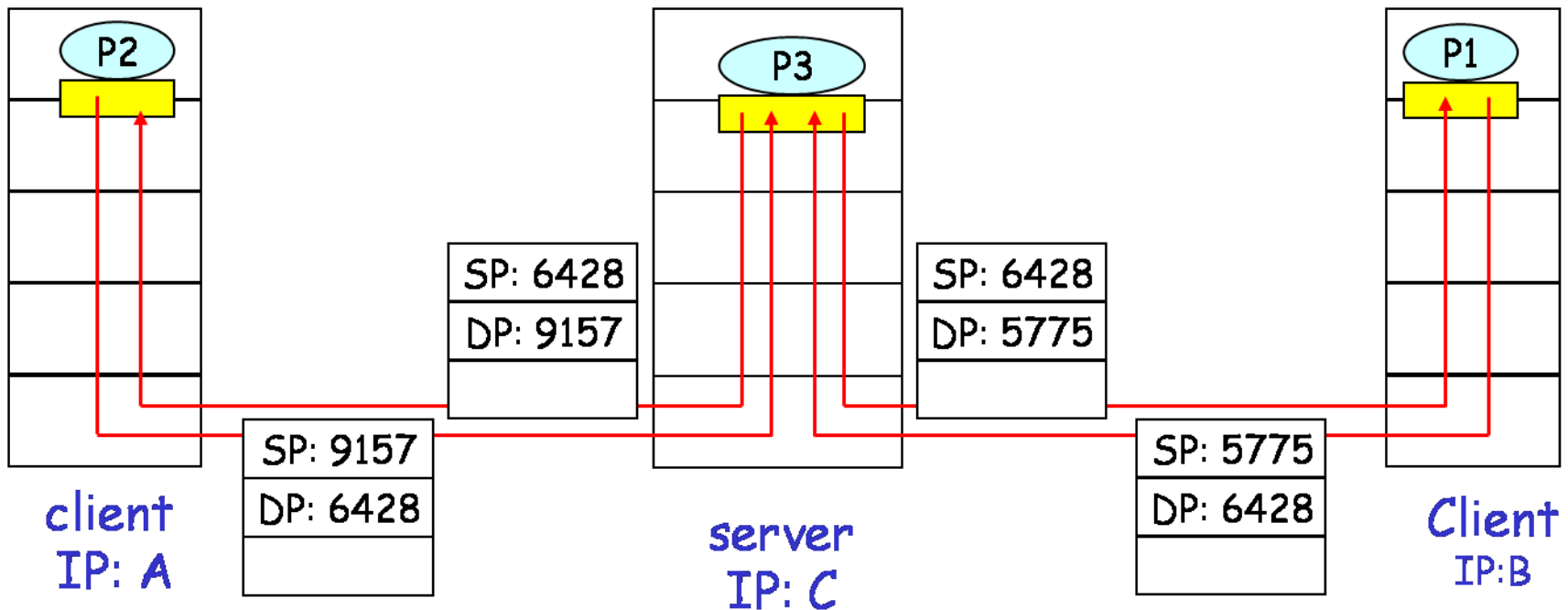


TCP/UDP segment format



Demultiplexing in UDP

```
DatagramSocket serverSocket = new DatagramSocket(6428);
```

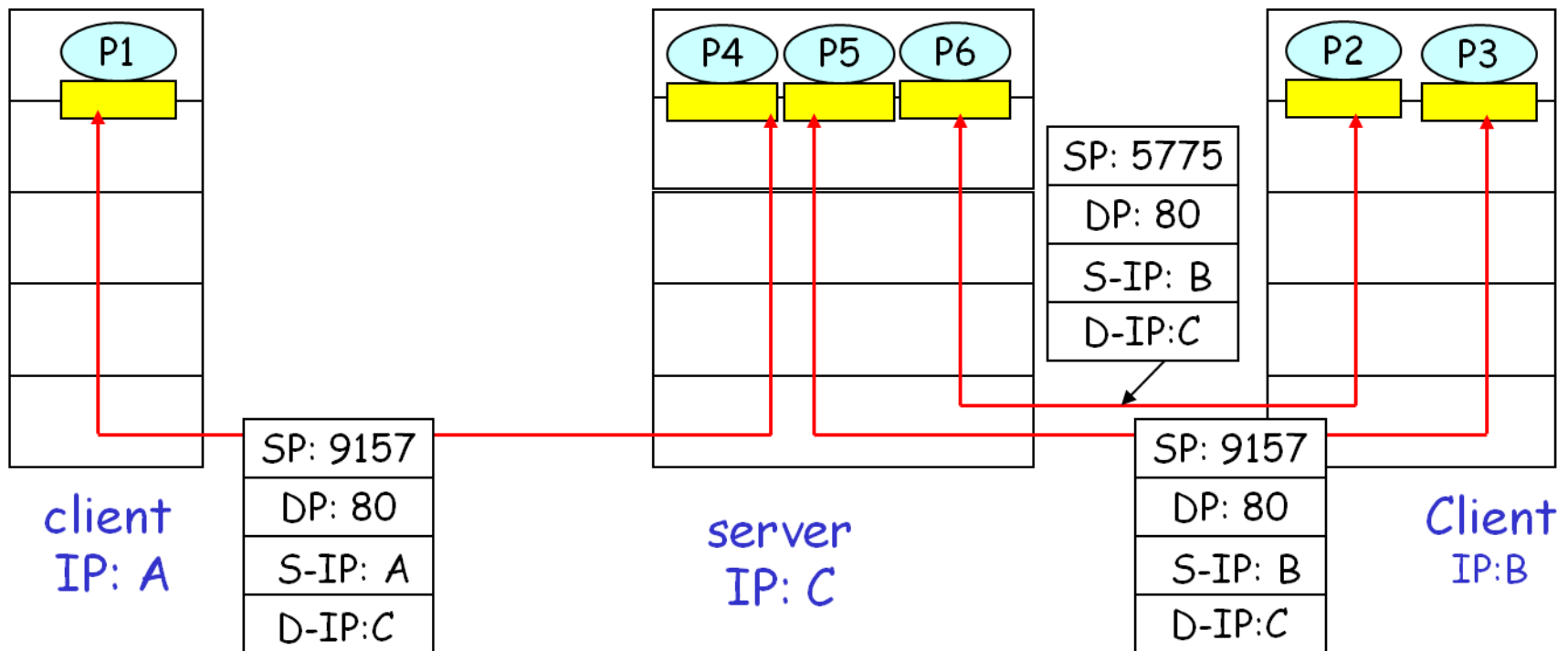


SP provides "return address"

Demultiplexing in TCP

TCP connection identified by 4-tuple:

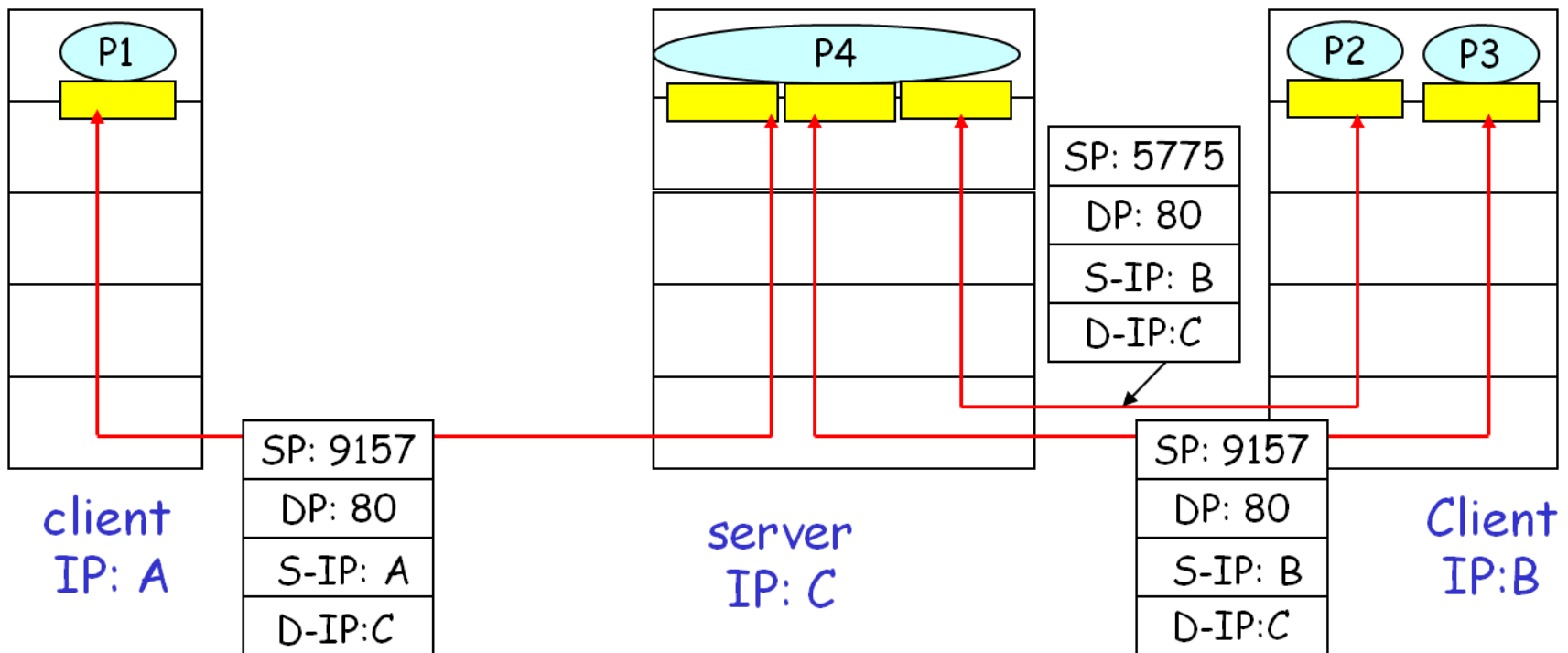
<Source IP, Source Port, Destination IP, Destination Port>





Threaded Web Server

Application with multiple threads, can be seen as
a **downward multiplexing**



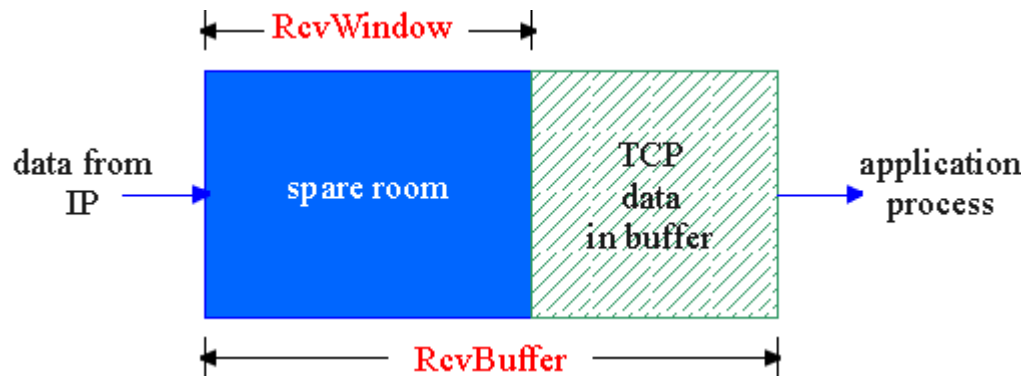


Flow Control

- Flow control in **multi-layers**
 - Sender won't overflow receiver's buffer by transmitting **too much too fast**
 - The receiving TS user can not keep up
 - Transport buffer may overflow
 - The receiving transport entity can not keep up
 - Network buffer may overflow
- **Speed-matching** service
 - Matching the send rate to the receiving app's receive rate

Receive Buffer

- The receive side of TS connection has a **receive buffer**



- App process may be slow at reading from buffer

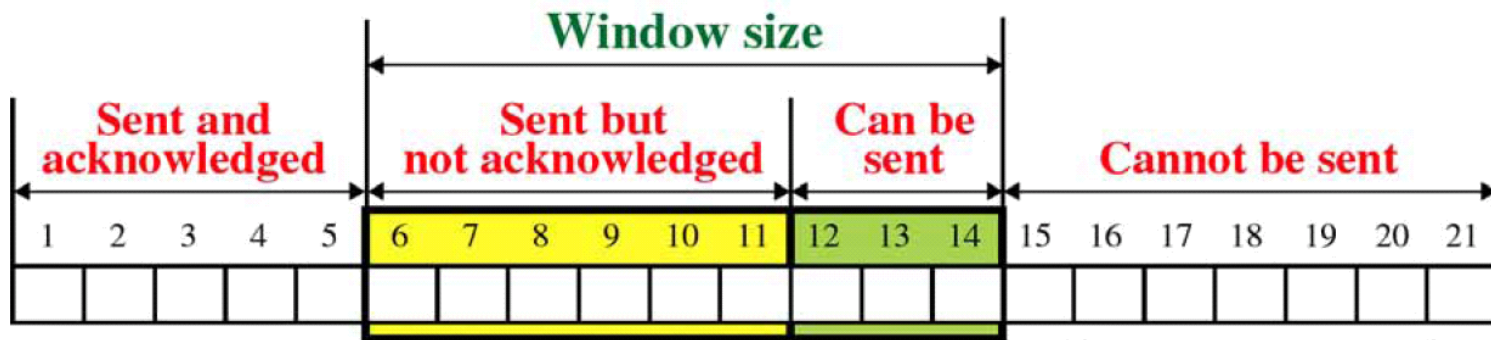
Handle Flow Control

■ Fixed sliding window

- Works well on reliable direct links

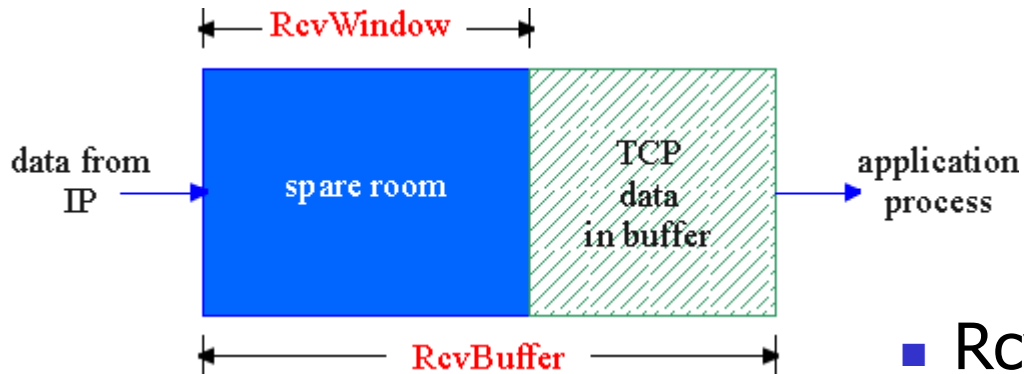
■ Problem:

- Failure to receive *ACK* is taken as flow control indication
- The receiver can achieve flow control by stop sending *ACK*, but the sender can not distinguish between lost segment and flow control
- Not flexible for congestion control mandated in Internet





Credit Scheme (1)



- **Spare room** in receive buffer

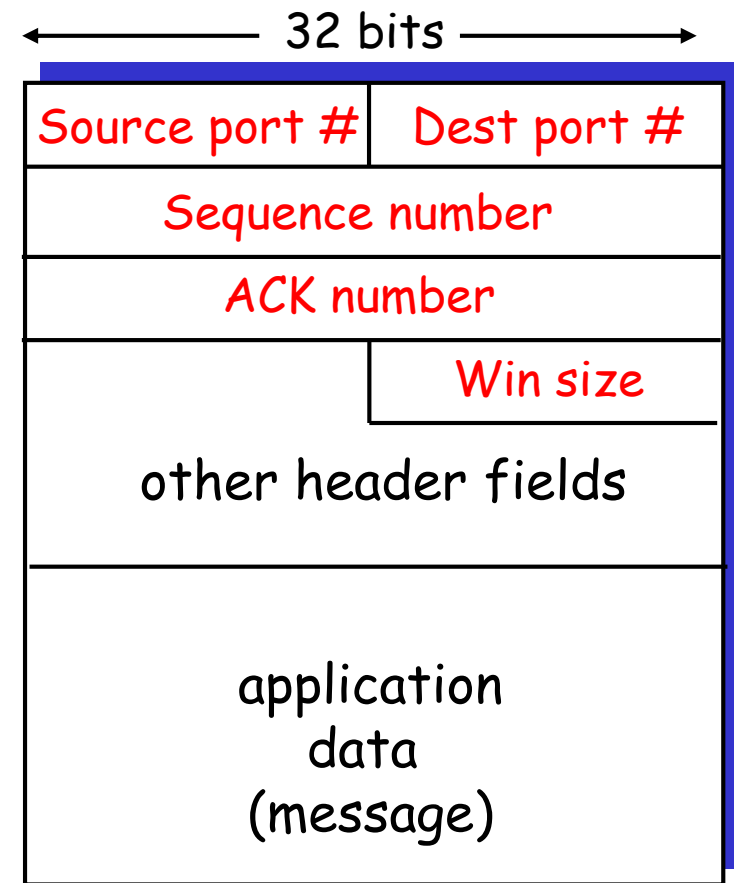
$$\text{RcvWindow} = \text{RcvBuffer} - [\text{LastByteRcvd} - \text{LastByteRead}]$$

- Rcvr advertises spare room (**credits**) by including value of RcvWindow in segments
- Sender limits unACKed data to RcvWindow



Credit Scheme (2)

- Greater control on Internet
- Decouples flow control from ACK
 - May ACK without granting credit
- Each octet has a sequence number
- Each transport segment has seq number, ack number and window size in header



TCP segment format

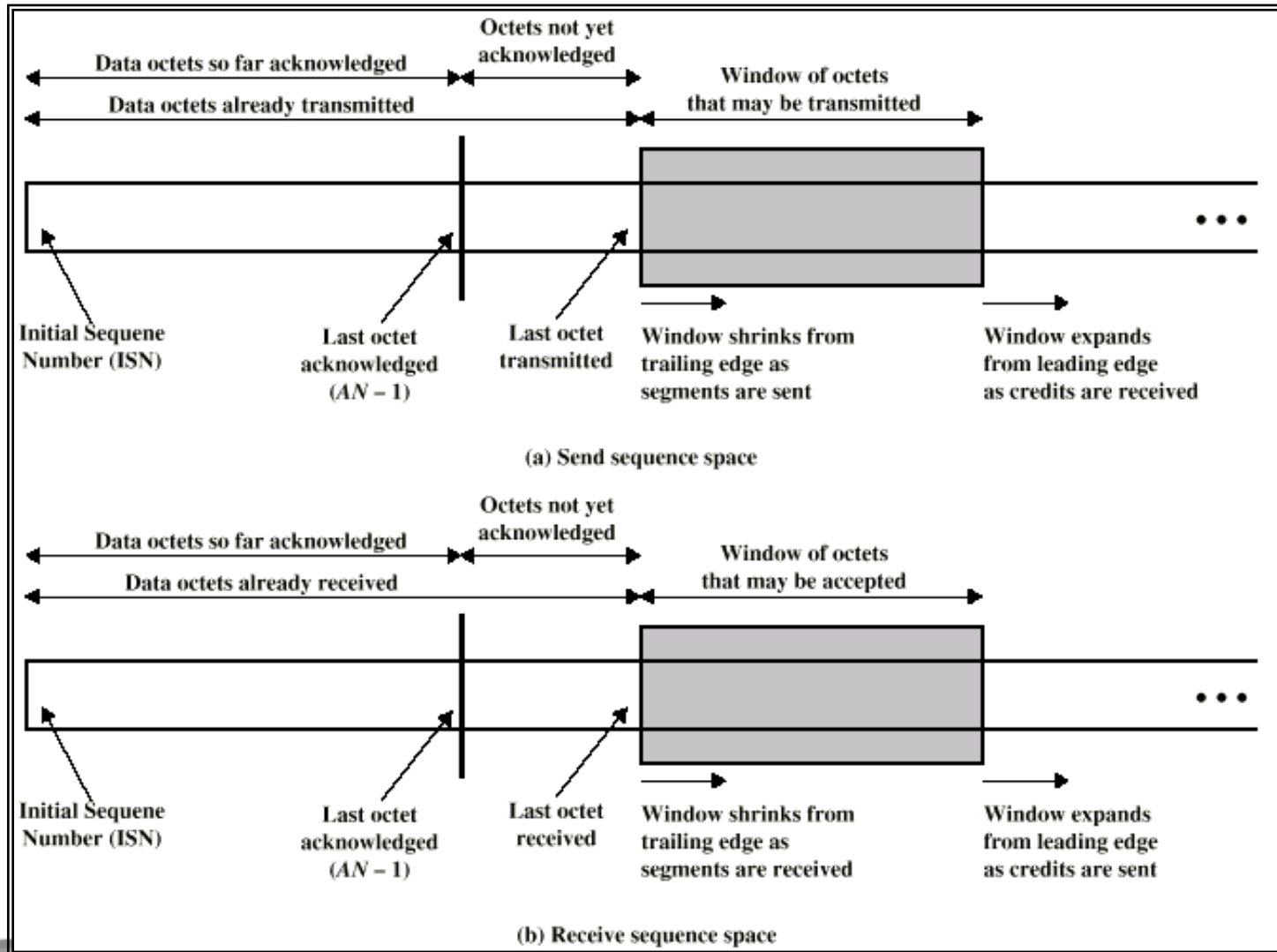


Use of Header Fields

- When **sending a segment**
 - seq number (SN) is that of first octet in segment
 - ACK includes $AN=i$, $W=j$
- All octets through $SN=i-1$ **acknowledged**
 - Next expected octet is i
- Permission to send **additional window** of $W=j$ octets
 - i.e. octets from i to $i+j-1$

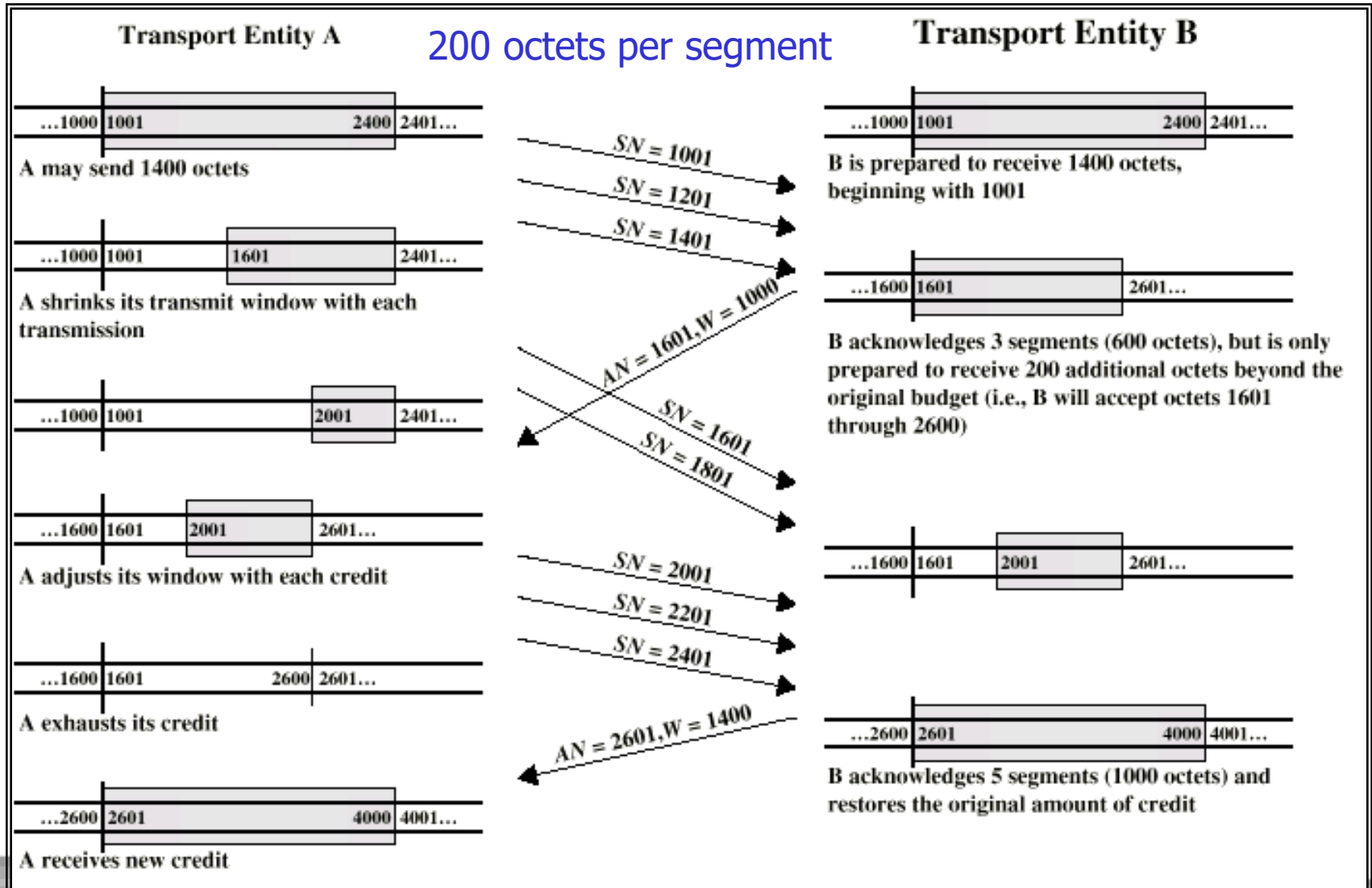


Sending and Receiving Windows





Credit Allocation Procedure





Connection Establishment and Termination



- 2 ends establish **connection** before exchanging data segments
 - Allow each end to know the other exists
 - Negotiation of **optional parameters**
 - e.g. initial *seq* numbers, max segment size, max window size, IP QOS
 - Allocation of transport entity resources
 - e.g. buffers
- Gets **mutual agreement**
 - On reliable sequencing networks
 - On unreliable IP internets

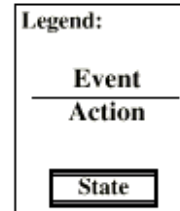
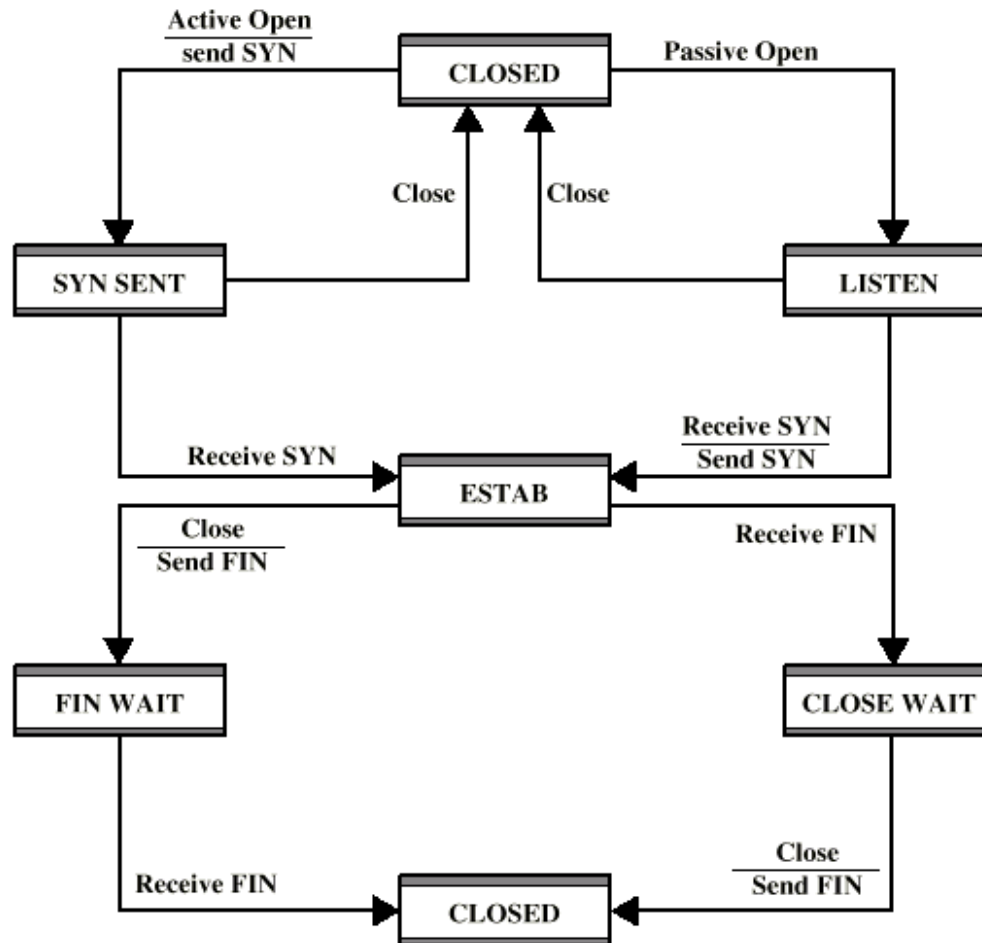


Reliable Sequencing Network Service

- Assume arbitrary length message delivered in sequence
- Assume virtually 100% reliable delivery by network service
- Examples
 - Reliable packet switched network using X.25
 - Frame relay using LAPF control protocol
 - IEEE 802.3 using connection oriented LLC service

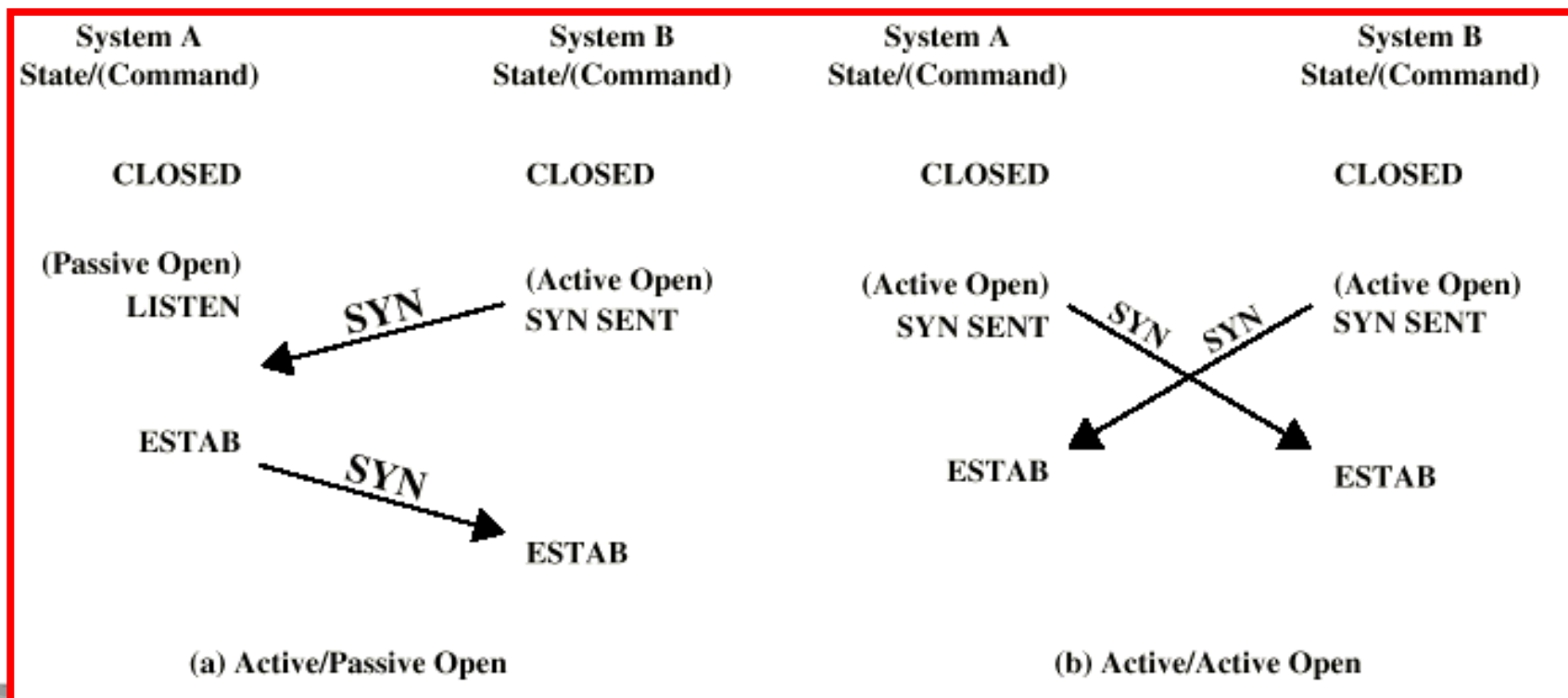
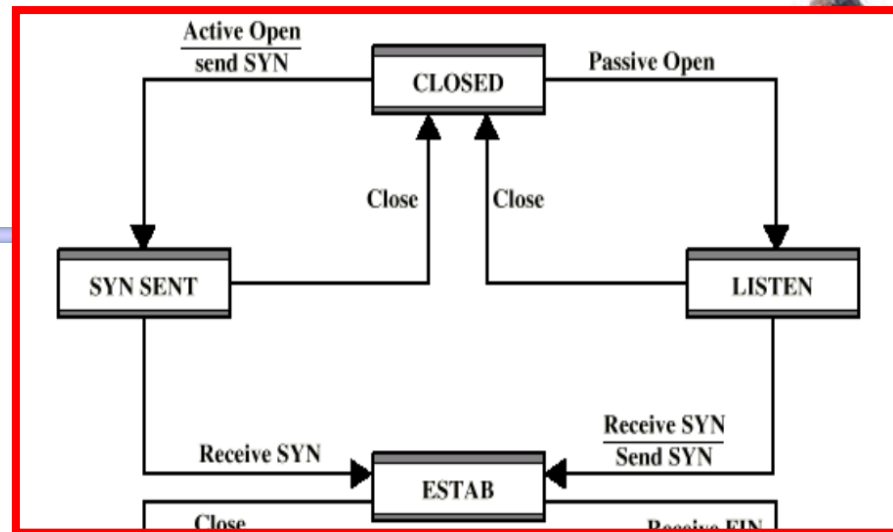


Simple Connection State Diagram





Connection Establishment





Handle Pending Request

- SYN comes while requested TS user is **not listening**
 - Reject with RST (Reset)
 - Queue request until matching open issued
 - Signal TS user to notify of pending request
 - Just accept without passive open



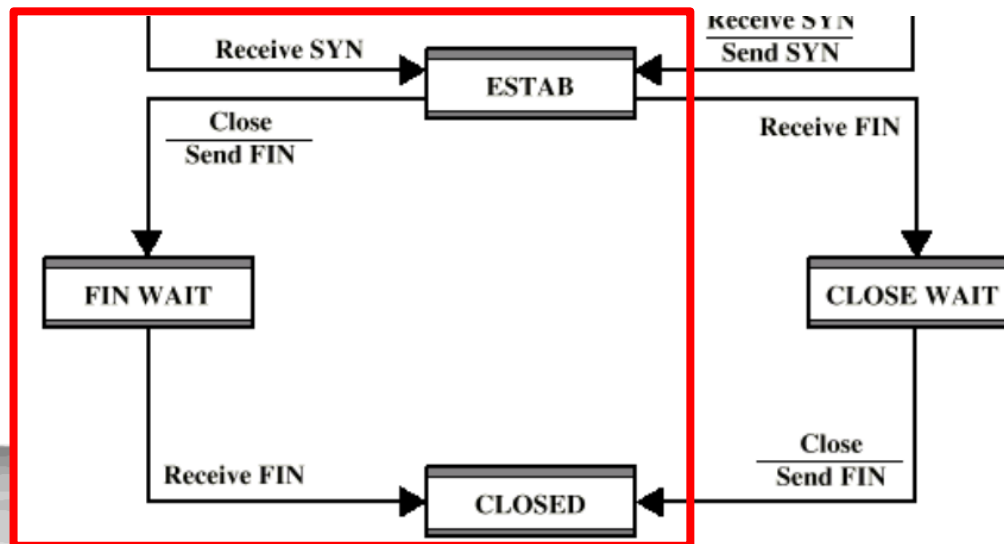
Termination

- Either or both sides issue **terminate**
- Reach **mutual agreement**
- **Abrupt termination**
 - Pending segments from other end may lost
- **Graceful termination**
 - All outstanding data is transmitted from both sides
 - Both sides agree to terminate



Side Initiating Termination

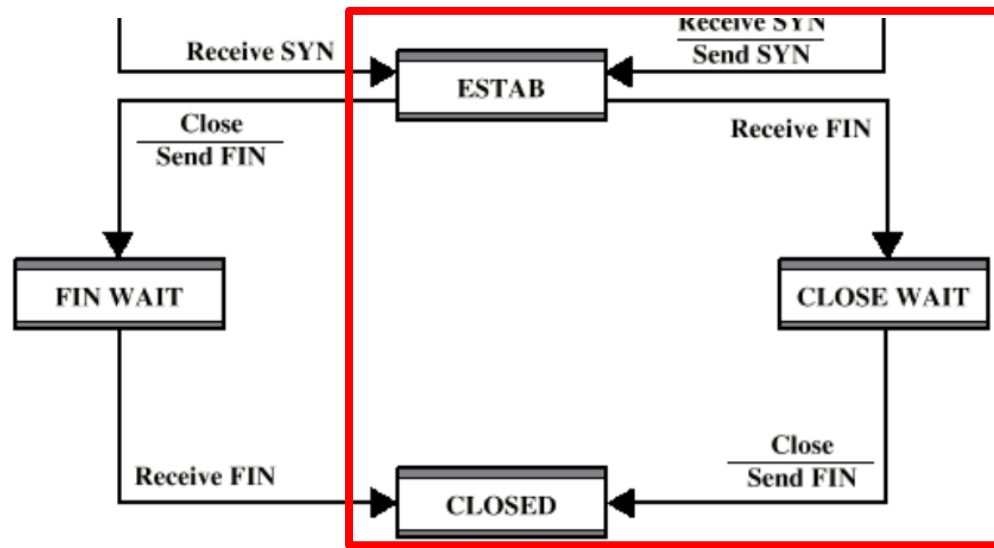
- TS user issues *Close* request
- Transport entity sends FIN, *requesting termination*
- Connection placed in FIN Wait state
 - Continues to accept data and deliver data to user
 - Not sends any more data
- When FIN received, inform user and close connection





Side Not Initiating Termination

- FIN received, Inform TS user
 - No more data come from the other end
- Place connection in *CLOSE Wait* state
 - Continue to accept data from TS user and transmit it
- TS user issues *CLOSE* primitive
 - Transport entity sends FIN
 - Connection closed





Unreliable Network Service

- Segments may **get lost**
- Segments may arrive **out of order**
- Examples
 - Internet using IP
 - Frame relay using LAPF core
 - IEEE 802.3 using unacknowledged connectionless LLC



Seven Issues

- Ordered Delivery
- Retransmission Strategy
- Duplication Detection
- Traffic Control
- Connection Establishment
- Connection Termination
- Crash Recovery

- 按序交付
- 重传策略
- 副本检测
- 流量控制
- 连接建立
- 连接终止
- 崩溃恢复



(1) Ordered Delivery

- Problem
 - Segments may arrive out of order
- Handle
 - Number segments sequentially
 - TCP numbers each octet sequentially
 - Segments are numbered by the first octet number in the segment



(2) Retransmission Strategy

■ Problem

- Segment damaged in transit, or
- Segment dropped due to buffer overflow at router
- Sender may not know of failure

■ Handle

- Receiver: **acknowledge successful receipt**
- Can use cumulative acknowledgement
- Sender: waiting **Timer for ACK** timeout triggers re-transmission



Setting Re-transmission Timer

- Should adapt to **changing network conditions**
 - Fixed timer is not suitable
 - Too small leads to unnecessary re-transmissions
 - Too large and response to lost segments is slow
- Can be set a **bit longer than round-trip time**
 - Receiver may not ACK immediately
 - Sender can not distinguish between ACK of original segment and re-transmitted segment
 - Should adapt to **network congestion**



(3) Duplication Detection

- If re-transmission Timer timeout, sender re-transmits segment
 - If segment just delayed, receiver must recognize duplicates
- Duplicate received within a connection
 - Receiver assumes ACK lost and re-transmits ACKs
 - Sender must not get confused with multiple ACKs
 - Space of seq number should be large enough to not cycle within maximum life of segment
- Duplicate received after connection closed





(4) Traffic Control

Credit allocation flow control mechanism

- Suppose that the last octet of data received by B was octet number $i-1$, and that the last segment issued by B was $(AN=i, W=j)$. Then
 - To increase credit to an amount k ($k>j$) when no additional data have arrived, B issues $(AN=i, W=k)$
 - To acknowledge an incoming segment containing m octets of data ($m<j$) without granting additional credit, B issues $(AN=i+m, W=j-m)$
- If an ACK/CREDIT segment is lost, little harm is done. Future acknowledgments will resynchronize the protocol.
- Further, if the sender times out and retransmits a data segment, it triggers a new acknowledgment.



■ Credit allocation deadlock

- B sends A: segment with $AN=i$, $W=0$ closing rcv-window
- B sends A: $AN=i$, $W=j$ to reopen, but this maybe lost
- Now A thinks window is closed, B thinks it is open and wait

■ Handle

- Use window timer
- If timer expires without any receiving, send something
- Could be re-transmission of previous segment



(5) Connection Establishment

■ 2-way handshake

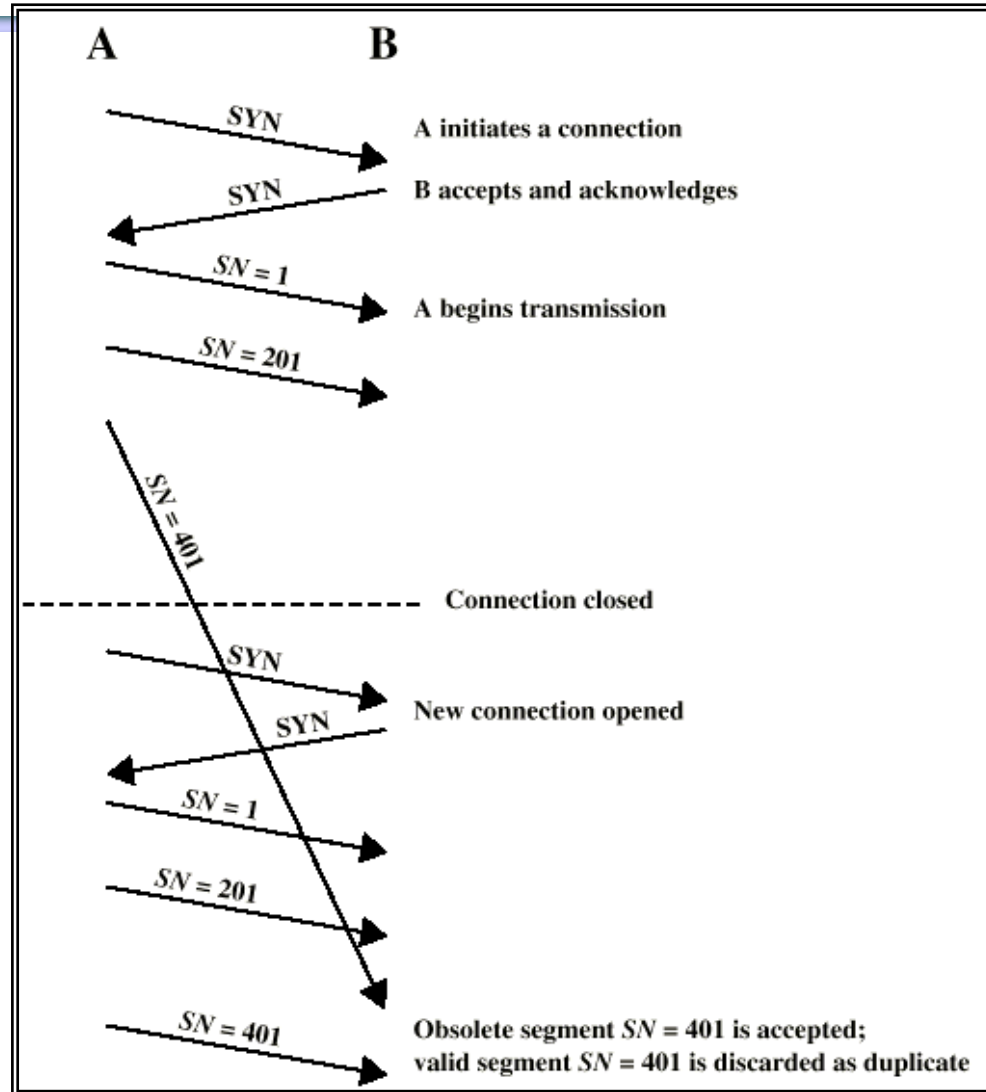
- A sends SYN, B replies with SYN
- Lost SYNs handled by re-transmission
- Ignore duplicate SYNs once connected

■ Problem

- How to recognize **slipped segments from old connection**
- How to recognize duplicated **obsolete SYN**



2-Way Handshake: Slipped Data Segment



两方握手：由滞后数据报文段带来问题



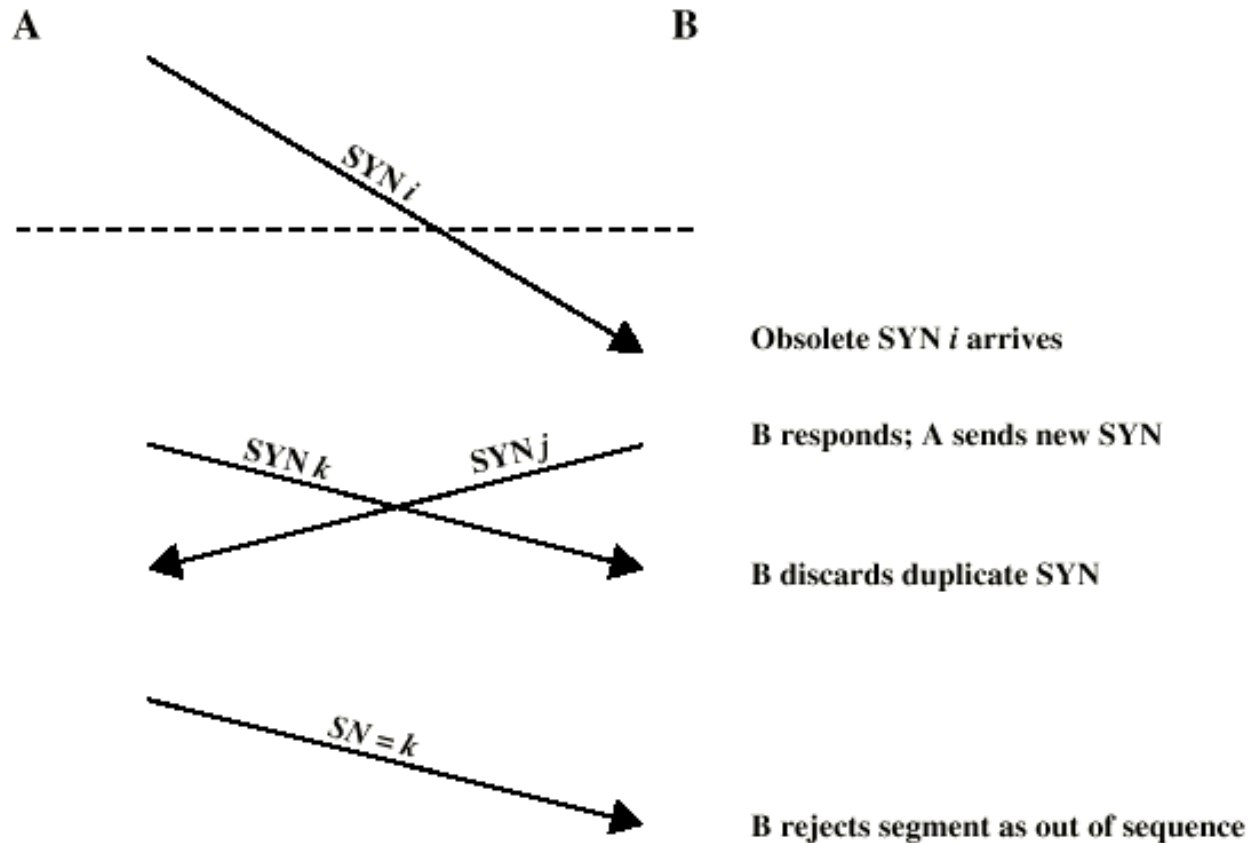
■ Handle

- Start each new connection with a different sequence number far from previous connection
- The connection request is of the form SYN $i+1$, where i is the sequence number of the first data segment that will be sent on this connection.

■ However:



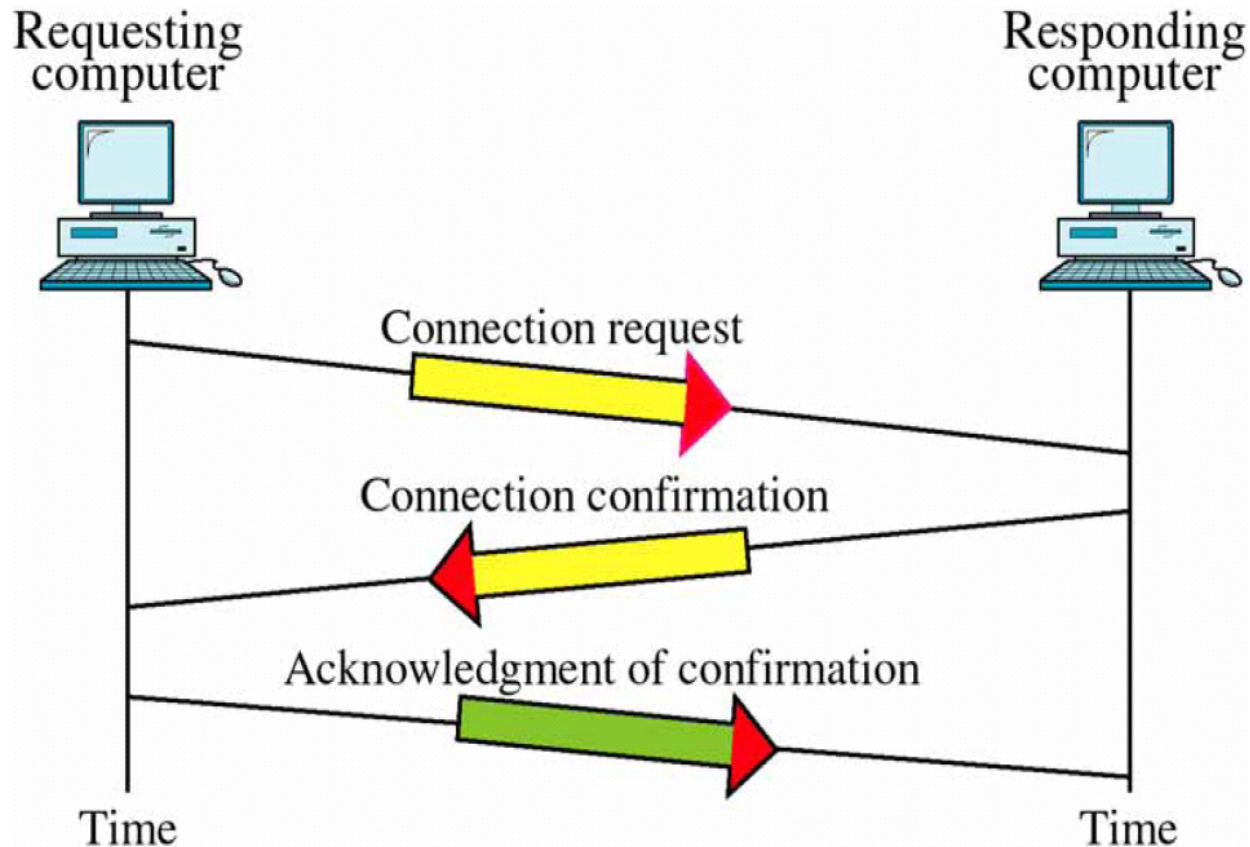
2-Way Handshake: Obsolete SYN



两方握手：由滞后SYN报文段带来问题



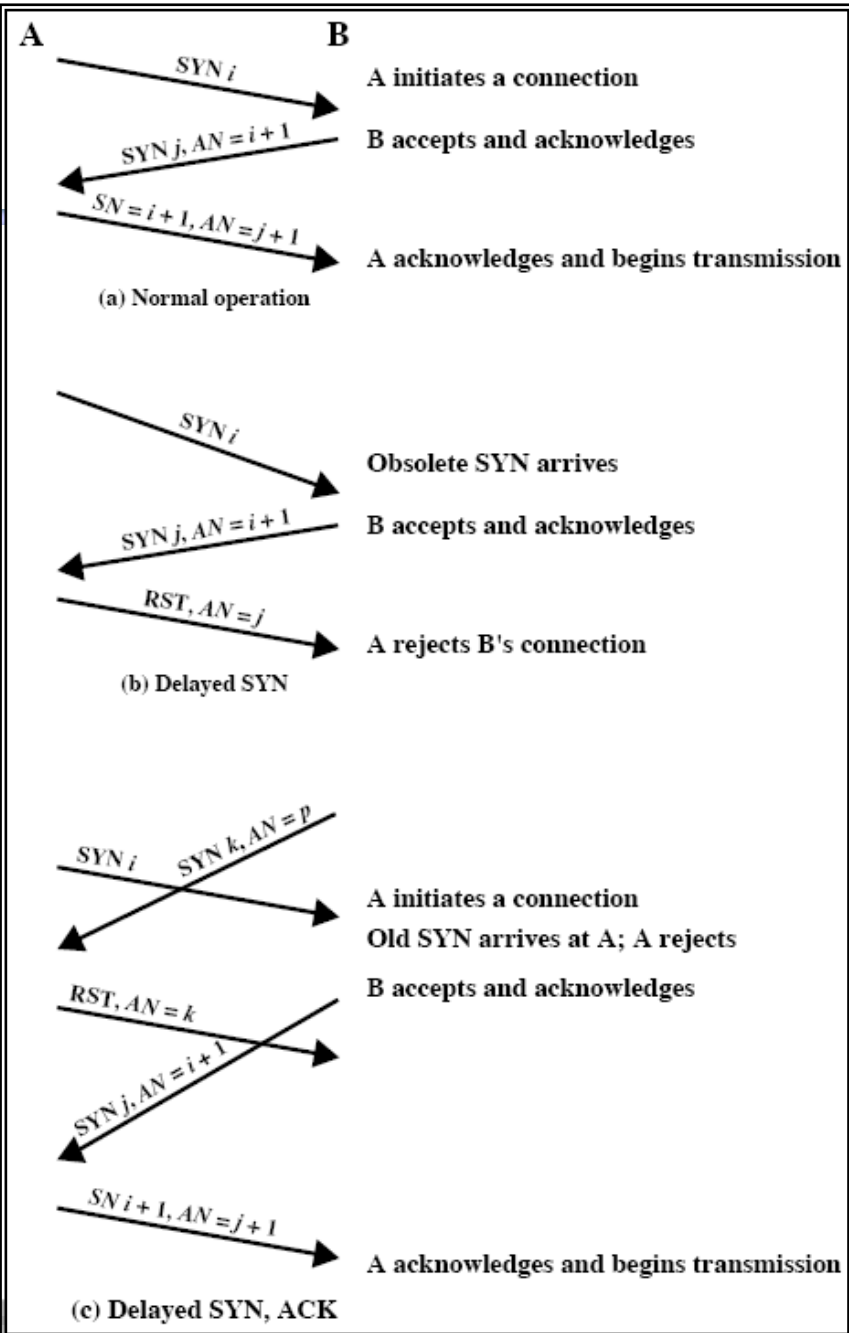
Solution: 3-Way Handshake



三方握手：确认对方的SYN和序号



3-Way Handshake: Examples





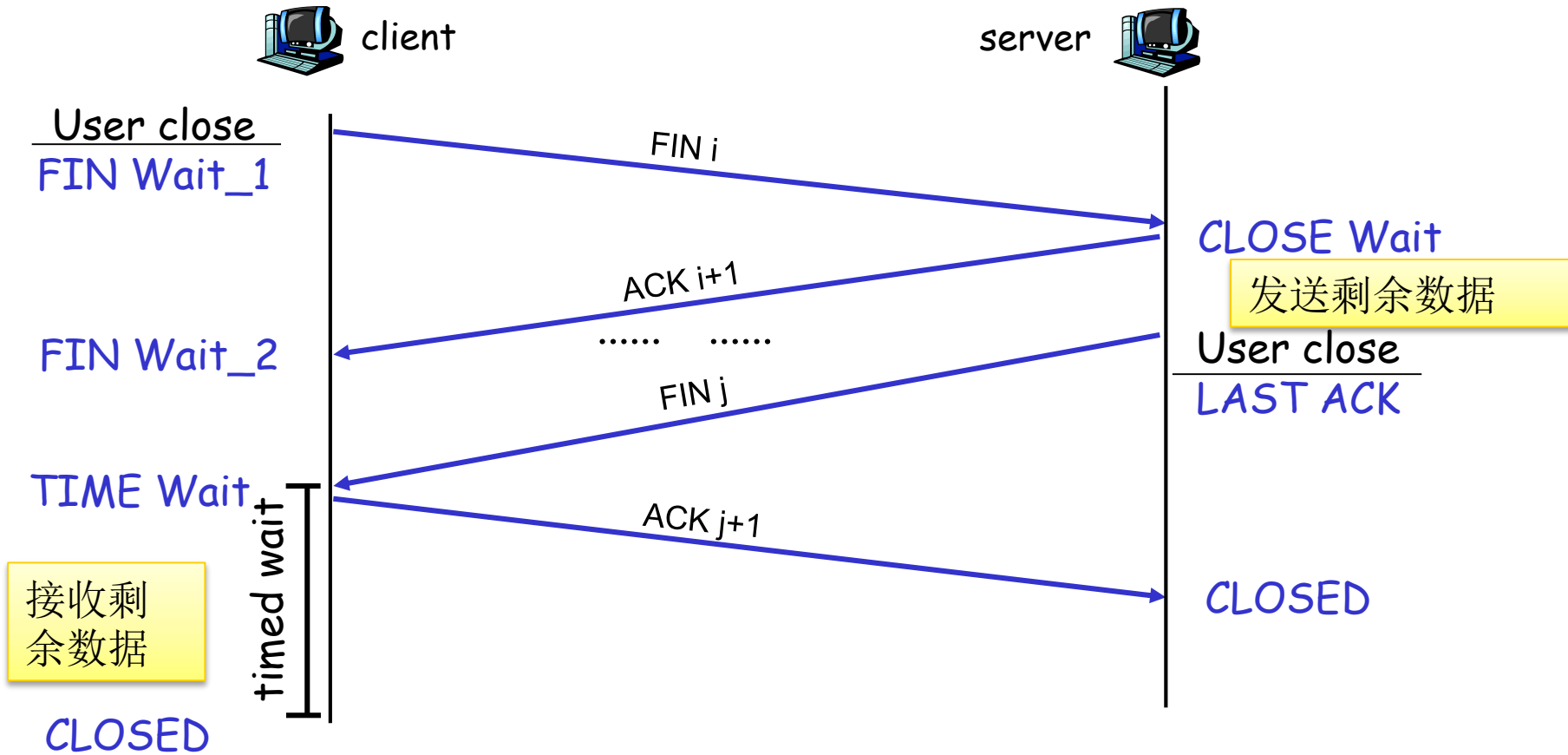
(6) Connection Termination

- A problem with 2-way termination
 - Entity in *CLOSE Wait* state sends last data segment, followed by *FIN*
 - *FIN* arrives before last data segment
 - Receiver accepts *FIN* and closes connection
 - Now **last data segment lost**
- Handle
 - Associate sequence number with *FIN*
 - Receiver waits for all segments before *FIN* seq number
 - *ACK FIN*, use **3-way termination**



Graceful Close

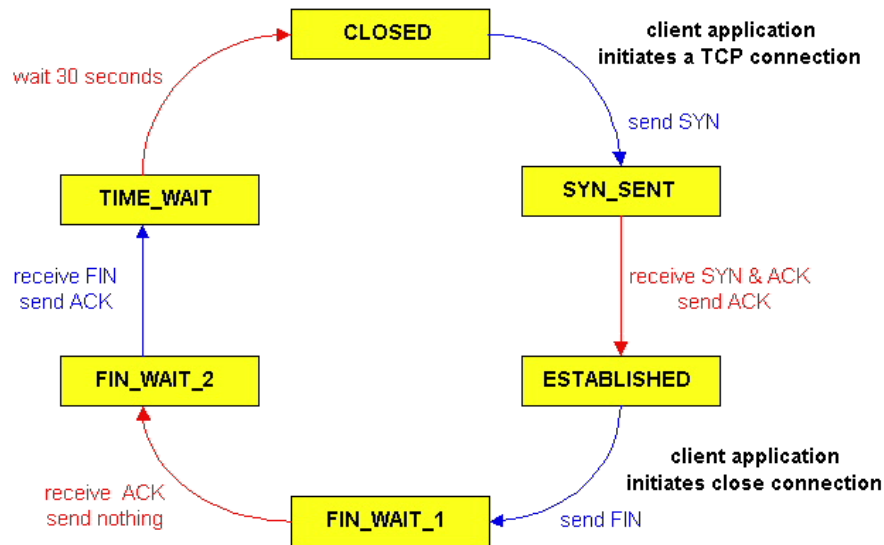
四次挥手



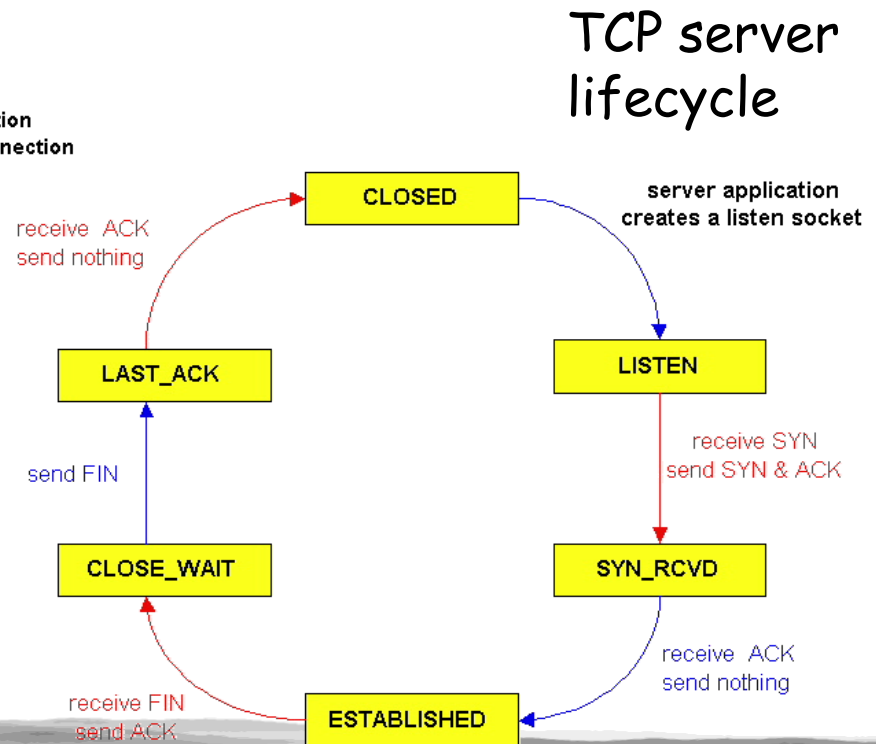
Q: What does TIME Wait stand for?



State Diagram: Client and Server

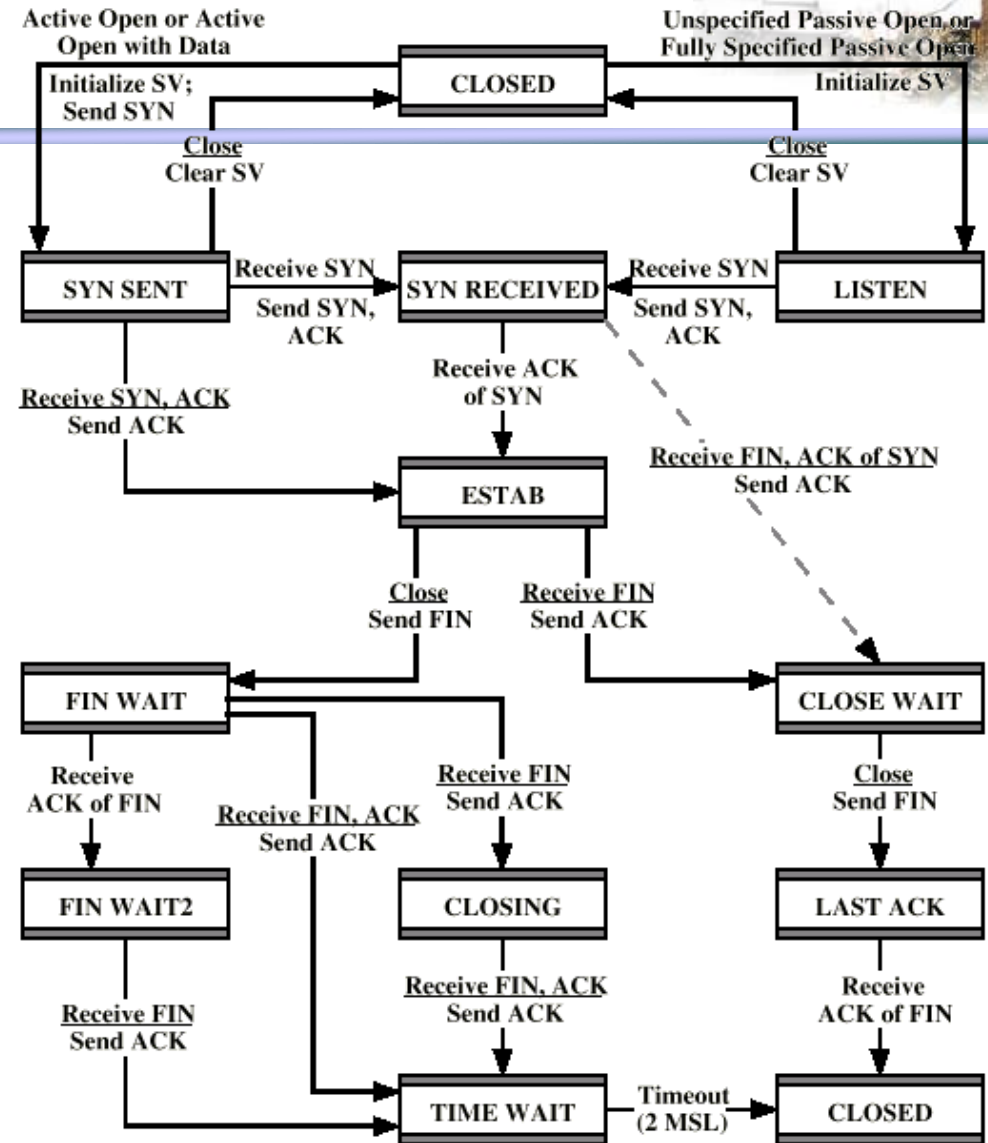


TCP client lifecycle





3-Way Handshake: State Diagram



SV = state vector
MSL = maximum segment lifetime



(7) Crash Recovery

Problem

- If a side restarts, all state info is lost
- Connection is **half open now**
 - Side that did not crash still thinks it is connected

Handle

- Close connection using **Persistence Timer**
 - Wait for *ACK* for $(\text{timeout}) \times (\text{number of retries})$
 - When expired, close connection and inform user
- Restarted side **sends RST i** in response to any i segment arriving
 - Other side verifies RST i , then closes connection
 - Restarted user can **reconnect immediately**



Summary

- Internet Transport Service
 - Addressing and multiplexing
 - Flow control
 - Connection-oriented mechanisms
 - Connection establishment and termination
 - Reliable sequencing communication
- Dealing with unreliable network service: seven issues
 - Ordered Delivery
 - Retransmission Strategy
 - Duplication Detection
 - Traffic Control
 - Connection Establishment: 3-way handshake
 - Connection Termination
 - Crash Recovery



Homework

- 第3章: R5, R8, P1, P19