

CSCI-1680

Transport Layer I

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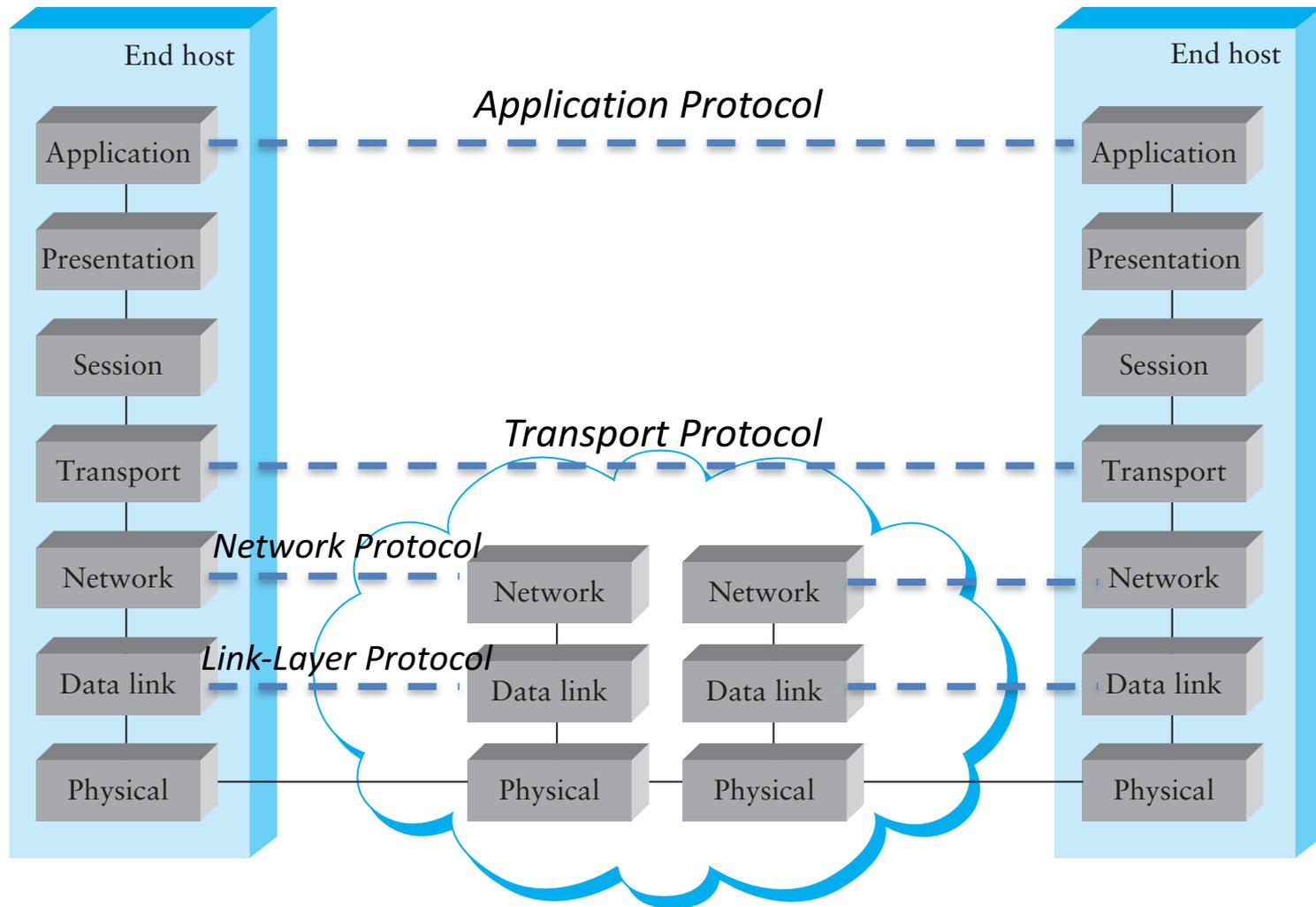


Today

- **Transport Layer**
 - UDP
 - TCP Intro
 - Connection Establishment



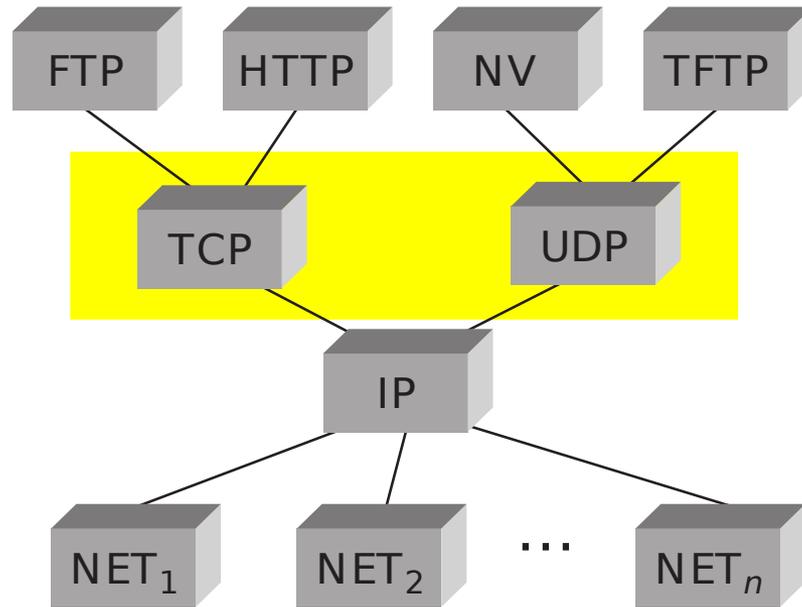
From Lec 2: OSI Reference Model



One or more nodes
within the network



Transport Layer



- **Transport protocols sit on top of network layer**
- **Problem solved: communication among processes**
 - Application-level multiplexing (“ports”)
 - Error detection, reliability, etc.

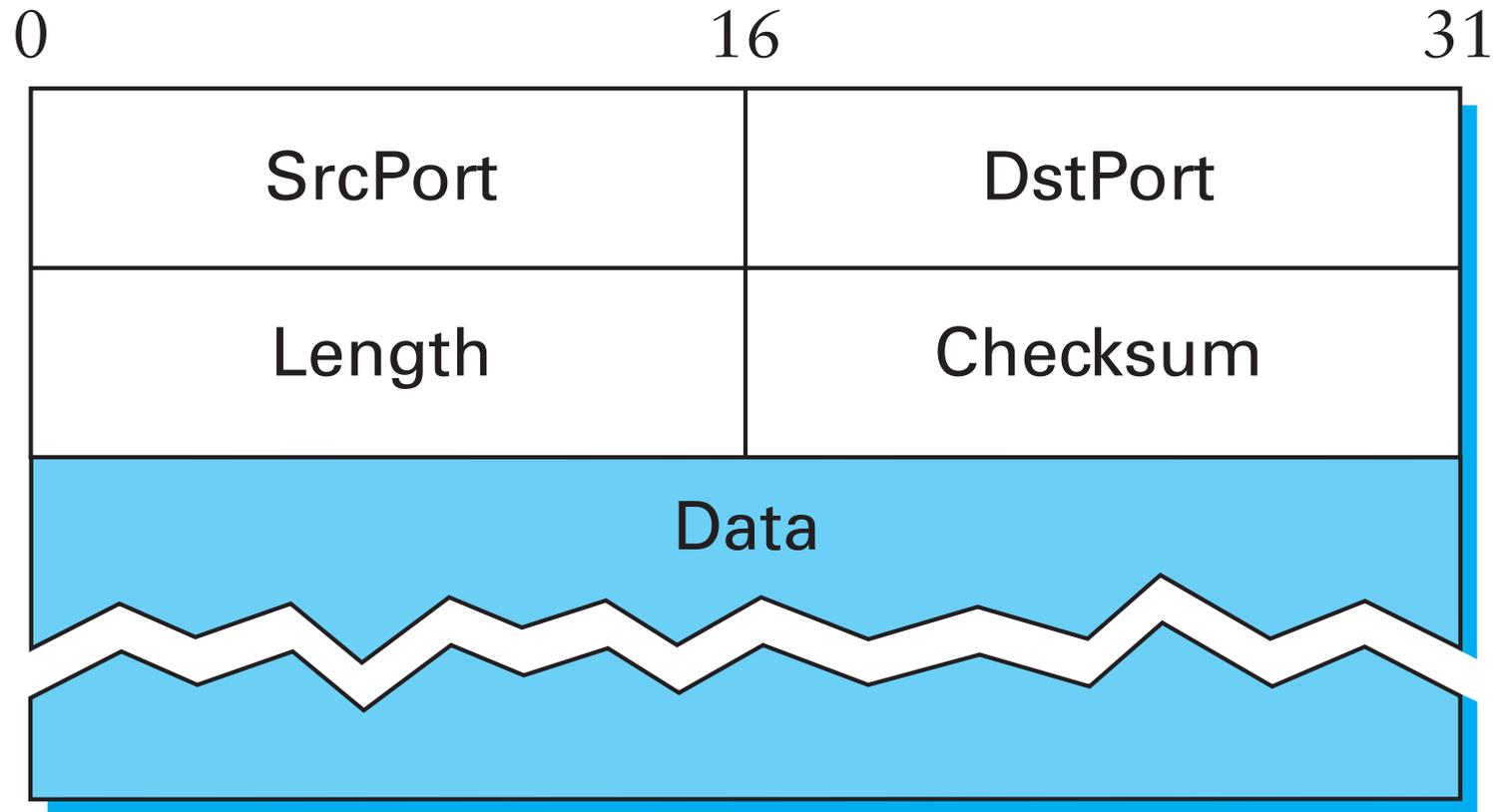


UDP – User Datagram Protocol

- **Unreliable, unordered datagram service**
- **Adds multiplexing, checksum**
- **End points identified by *ports***
 - Scope is an IP address (interface)
- **Checksum aids in error detection**



UDP Header

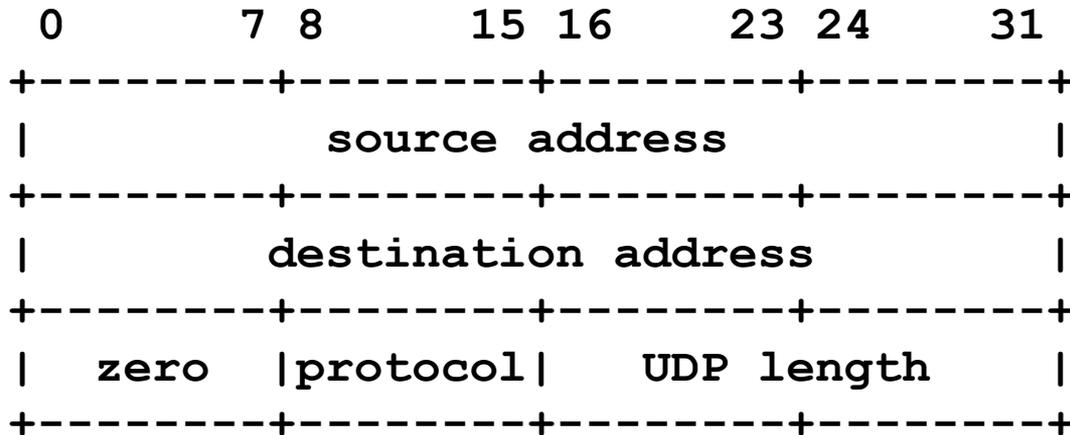


UDP Checksum

- **Uses the same algorithm as the IP checksum**
 - Set Checksum field to 0
 - Sum all 16-bit words, adding any carry bits to the LSB
 - Flip bits to get checksum (except 0xffff->0xffff)
 - To check: sum whole packet, including sum, should get 0xffff
- **How many errors?**
 - Catches any 1-bit error
 - Not all 2-bit errors
- **Optional in IPv4: not checked if value is 0**



Pseudo Header



- **UDP Checksum is computed over *pseudo-header* prepended to the UDP header**
 - For IPv4: IP Source, IP Dest, Protocol (=17), plus UDP length
- **What does this give us?**
- **What is a problem with this?**
 - Is UDP a layer on top of IP?



Next Problem: Reliability

- **Review: reliability on the link layer**

Problem	Mechanism
Dropped Packets	Acknowledgments + Timeout
Duplicate Packets	Sequence Numbers
Packets out of order	Receiver Window
Keeping the pipe full	Sliding Window (Pipelining)

- **Single link: things were easy... ☺**



Transport Layer Reliability

- **Extra difficulties**
 - Multiple hosts
 - Multiple hops
 - Multiple potential paths
- **Need for connection establishment, tear down**
 - Analogy: dialing a number versus a direct line
- **Varying RTTs**
 - Both across connections and *during* a connection
 - Why do they vary? What do they influence?

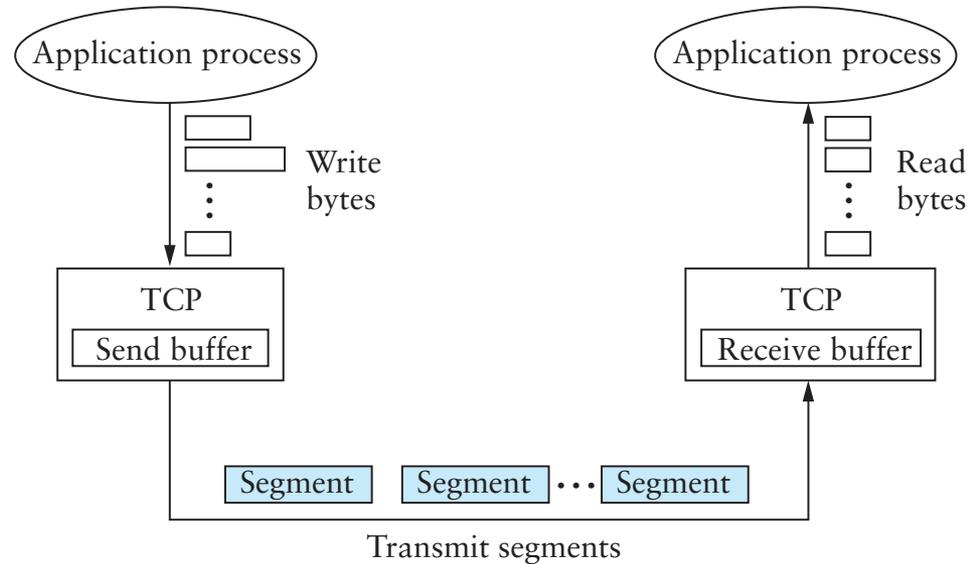


Extra Difficulties (cont.)

- **Out of order packets**
 - Not only because of drops/retransmissions
 - Can get very old packets (up to 120s), must not get confused
- **Unknown resources at other end**
 - Must be able to discover receiver buffer: flow control
- **Unknown resources in the network**
 - Should not overload the network
 - But should use as much as safely possible
 - Congestion Control (next class)



TCP – Transmission Control Protocol



- **Service model: “reliable, connection oriented, full duplex ordered byte stream”**
 - Endpoints: <IP Address, Port>
- **Flow control**
 - If one end stops reading, writes at other eventually stop/fail
- **Congestion control**
 - Keeps sender from overloading the network



TCP

- **Specification**
 - RFC 793 (1981), RFC 1222 (1989, some corrections), RFC 5681 (2009, congestion control), ...
- **Was born coupled with IP, later factored out**
 - We talked about this, don't always need everything!
- **End-to-end protocol**
 - Minimal assumptions on the network
 - All mechanisms run on the end points
- **Alternative idea:**
 - Provide reliability, flow control, etc, link-by-link
 - Does it work?



Not the only options...

	UDP	TCP	SCTP	DCCP
Multiplexing	✓	✓	✓	✓
Connection		✓	✓	✓
Reliability		✓	✓	
In-order		✓	optional	
Message	✓		✓	✓
Stream		✓		
Flow Control		✓	✓	
Congestion Control		✓	✓	✓
Multiple Streams		*	✓	
Multiple Paths		*	✓	

*MPTCP adds multiple streams and multiple paths

This table is not exhaustive!



Why not provide (*) on the network layer?

- **Cost**
 - These functionalities are not free: don't burden those who don't need them
- **Conflicting**
 - Timeliness and in-order delivery, for example
- **Insufficient**
 - Example: reliability

* may be security, reliability, ordering guarantees, ...

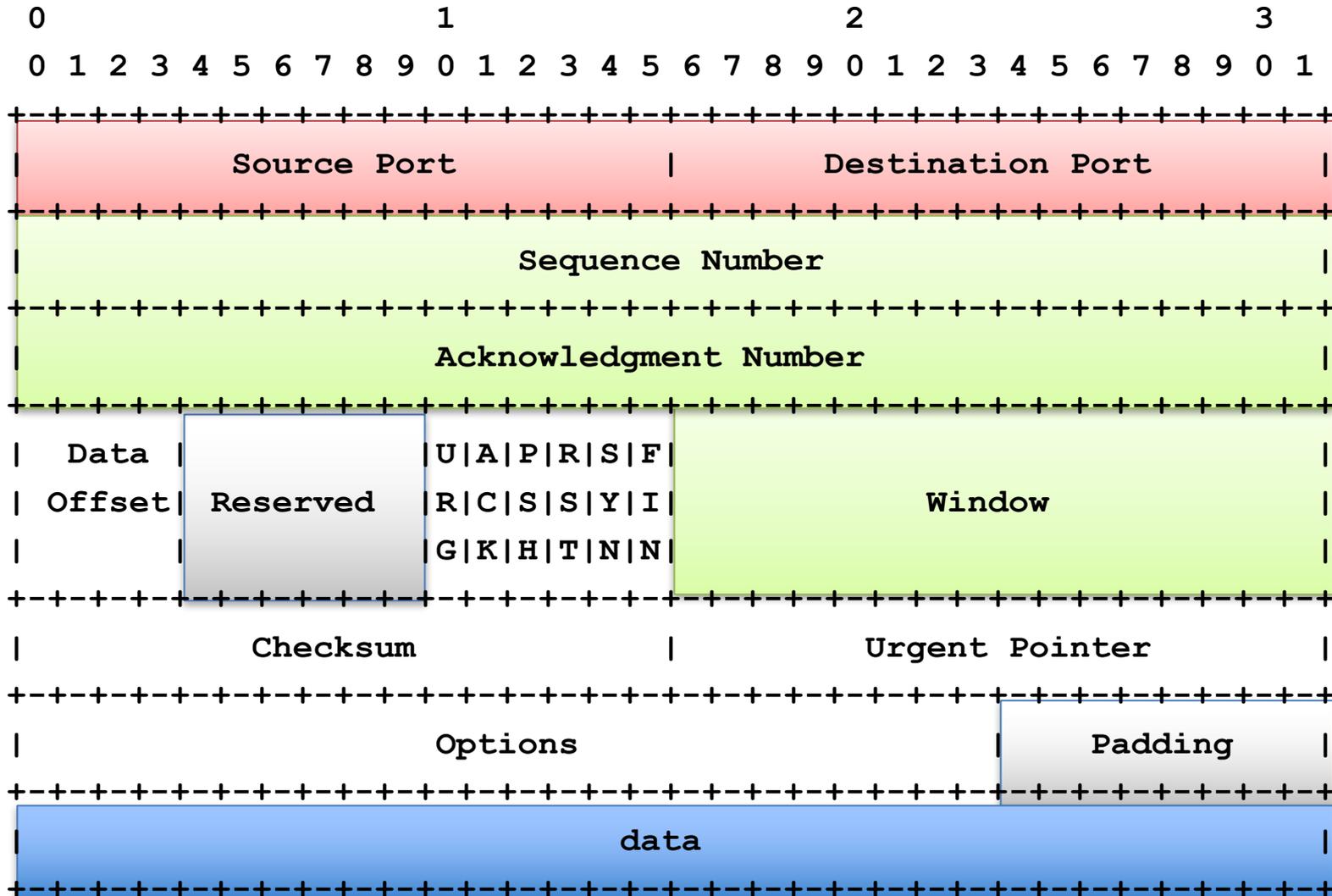


End-to-end argument

- **Functions placed at lower levels of a system may be redundant or of little value**
 - They may **need** to be performed at a higher layer anyway
- **But they may be justified for performance reasons**
 - Or just because they provide *most* of what is needed
 - Example: retransmissions
- **Lesson: weigh the costs and benefits at each layer**
 - Also: the *end* also varies from case to case



TCP Header



Header Fields

- **Ports: multiplexing**
- **Sequence number**
 - Correspond to *bytes*, not packets!
- **Acknowledgment Number**
 - Next expected sequence number
- **Window: willing to receive**
 - Lets receiver limit SWS (even to 0) for flow control
- **Data Offset: # of 4 byte (header + option bytes)**
- **Flags, Checksum, Urgent Pointer**

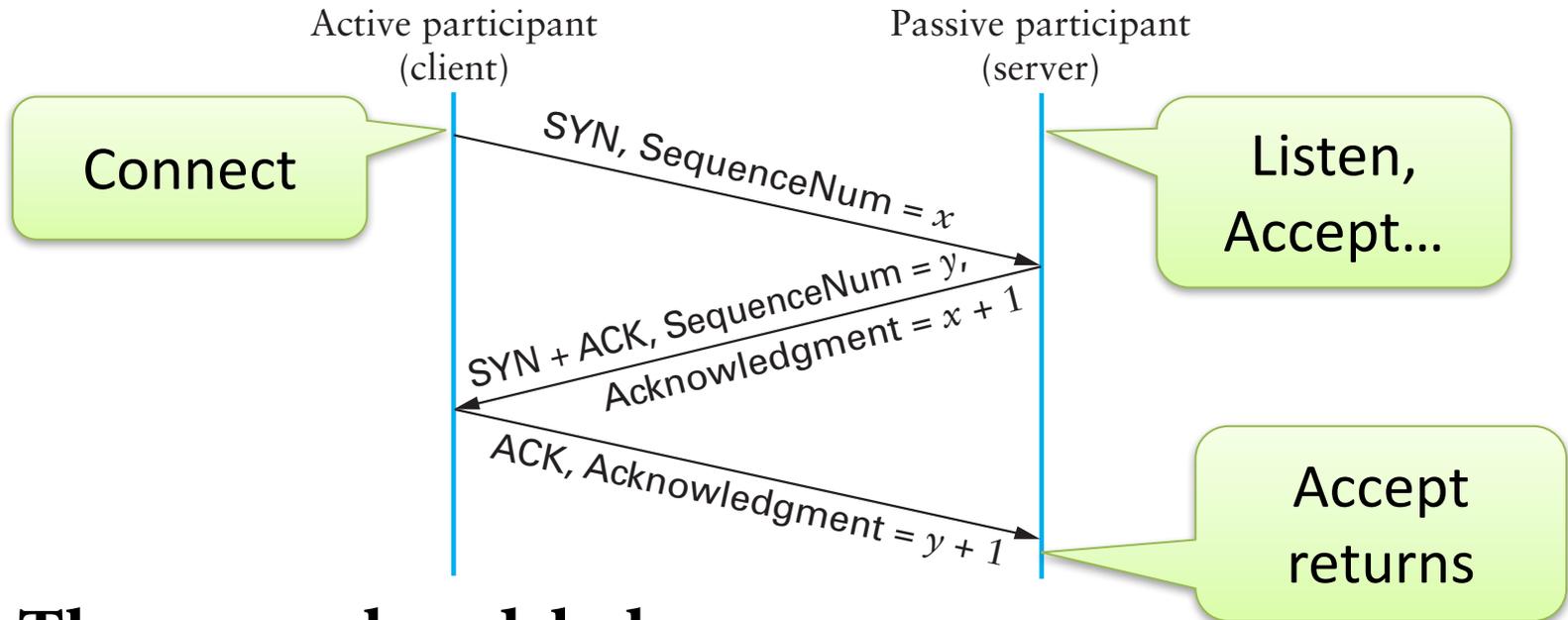


Header Flags

- **URG: whether there is urgent data**
- **ACK: ack no. valid (all but first segment)**
- **PSH: push data to the application immediately**
- **RST: reset connection**
- **SYN: synchronize, establishes connection**
- **FIN: close connection**



Establishing a Connection

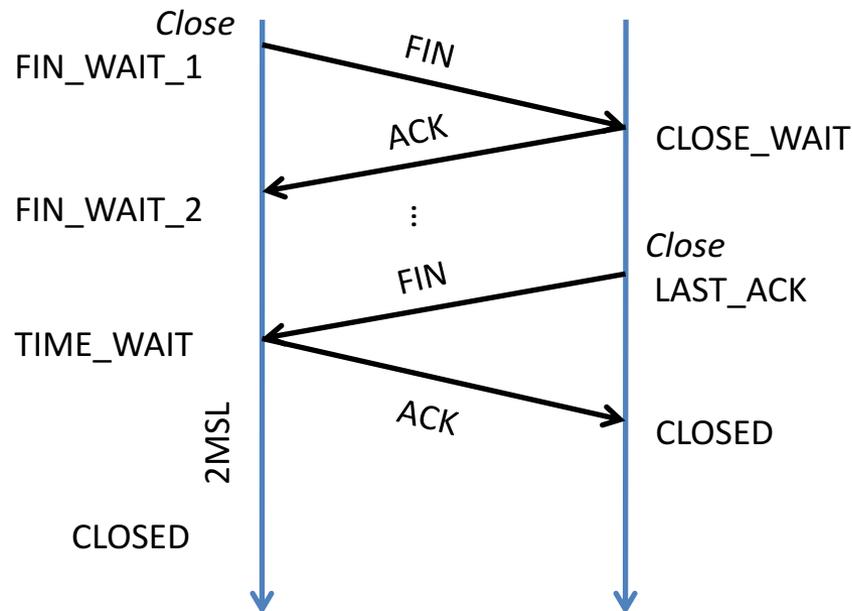


- **Three-way handshake**
 - Two sides agree on respective initial sequence nums
- **If no one is listening on port: server sends RST**
- **If server is overloaded: ignore SYN**
- **If no SYN-ACK: retry, timeout**



Connection Termination

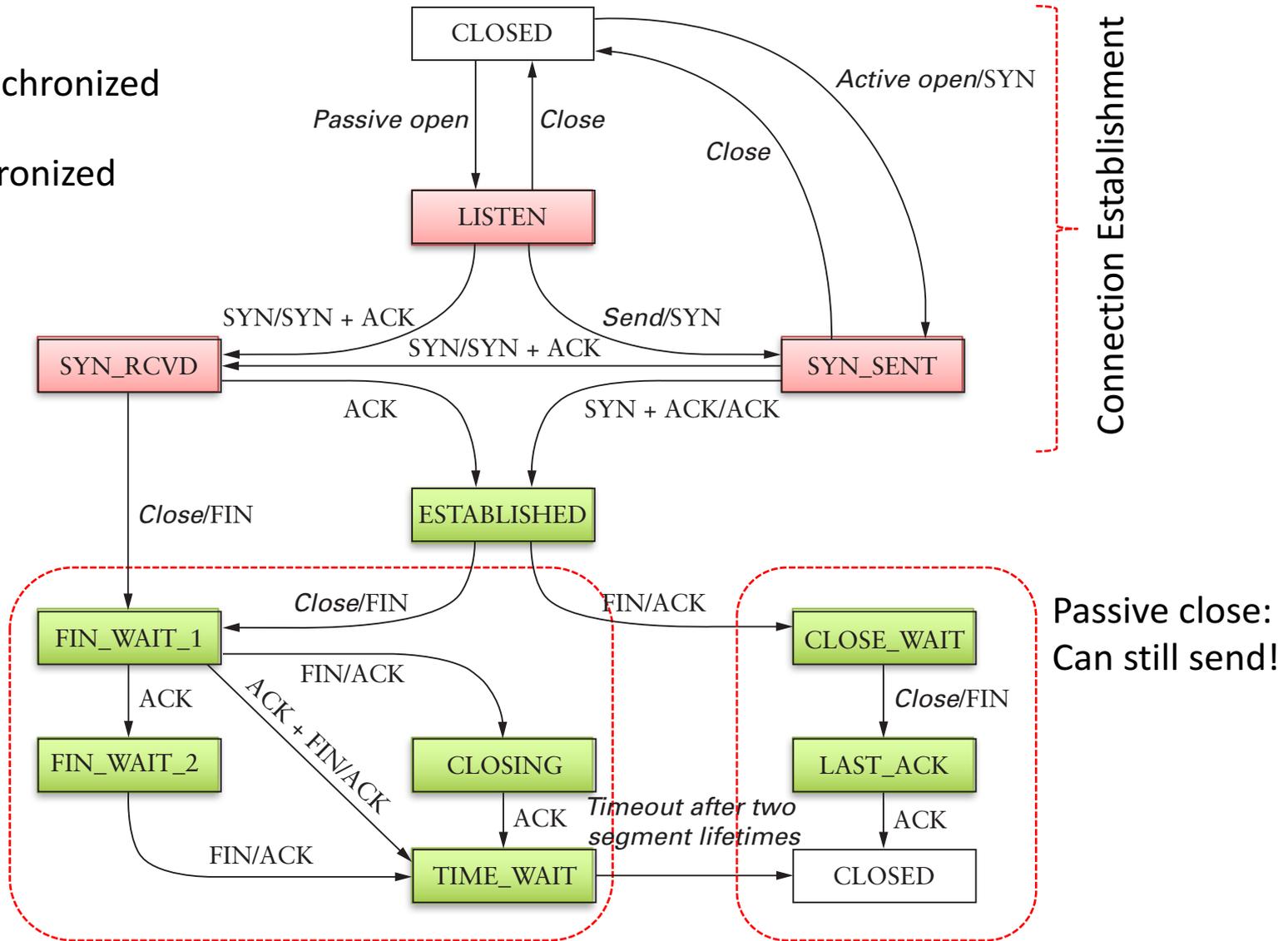
- **FIN bit says no more data to send**
 - Caused by close or shutdown
 - Both sides must send FIN to close a connection
- **Typical close**



Summary of TCP States

Unsynchronized

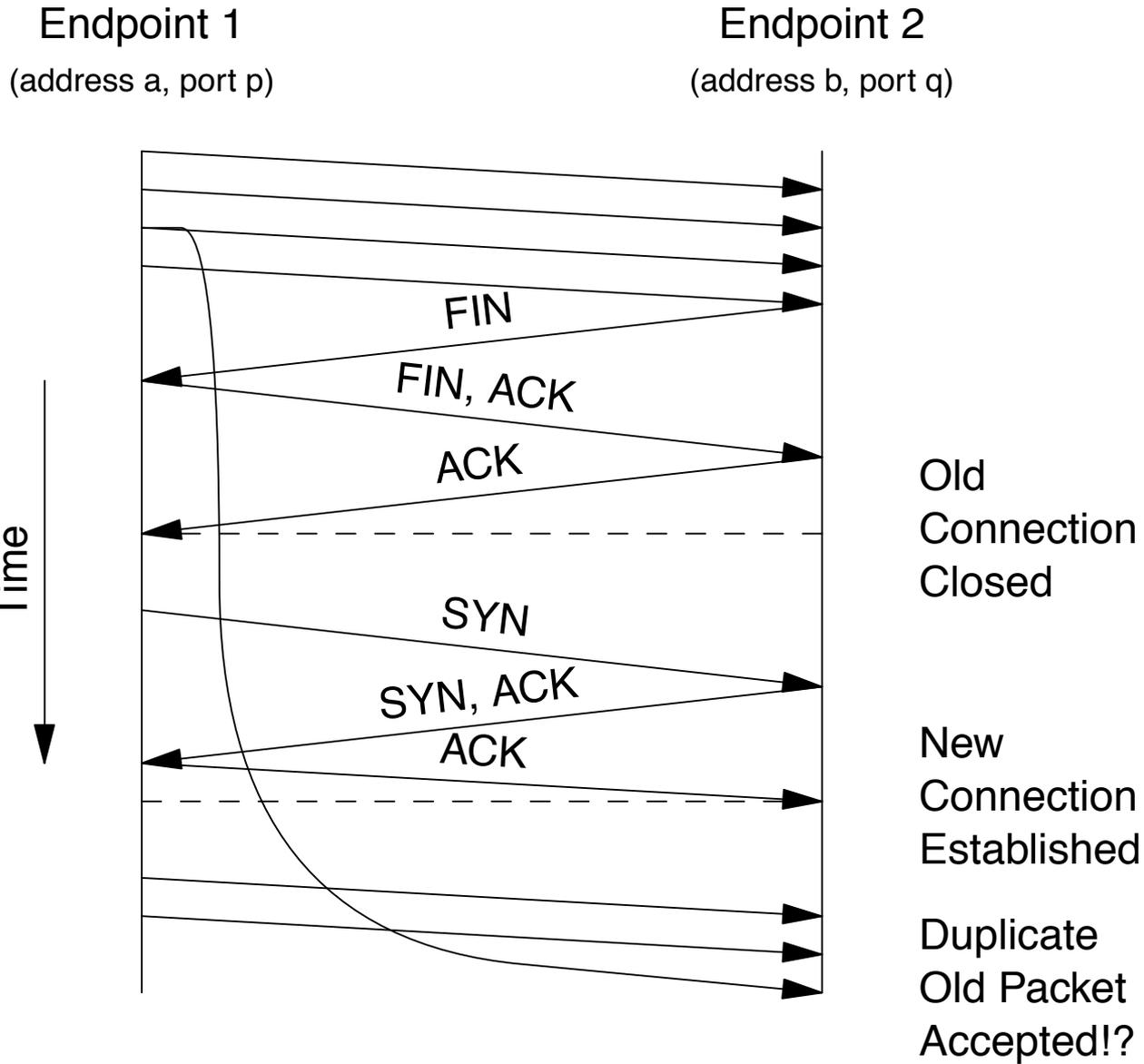
Synchronized



TIME_WAIT

- **Why do you have to wait for 2MSL in TIME_WAIT?**
 - What if last ack is severely delayed, AND
 - Same port pair is immediately reused for a new connection?
- **Solution: active closer goes into TIME_WAIT**
 - Waits for 2MSL (Maximum Segment Lifetime)
- **Can be problematic for active servers**
 - OS has too many sockets in TIME_WAIT, can accept less connections
 - Hack: send RST and delete socket, SO_LINGER = 0
 - OS won't let you re-start server because port in use
 - SO_REUSEADDR lets you rebind





From: The TIME-WAIT state in TCP and Its Effect on Busy Servers, Faber and Touch Infocom 1999



Next class

- **Sending data over TCP**

