# CSCI-1680 Transport Layer I

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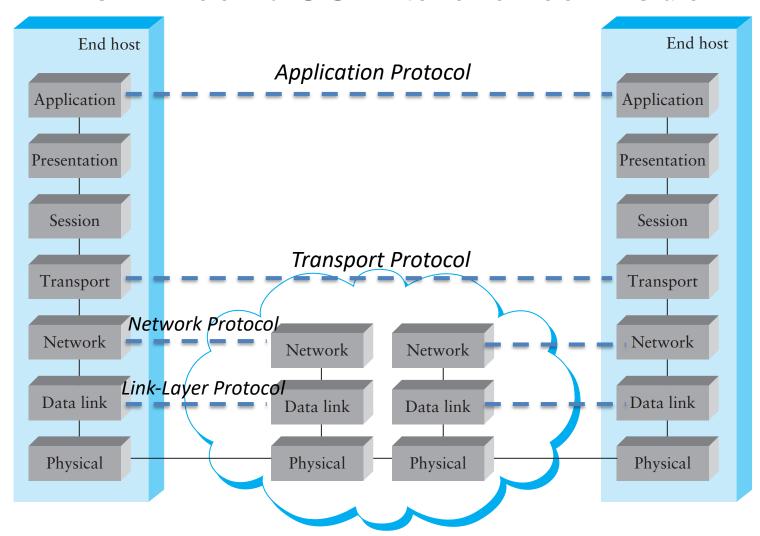


# **Today**

- Transport Layer
  - UDP
  - TCP Intro
    - Connection Establishment



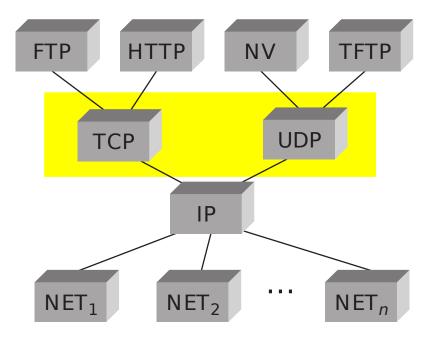
### From Lec 2: OSI Reference Model





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.

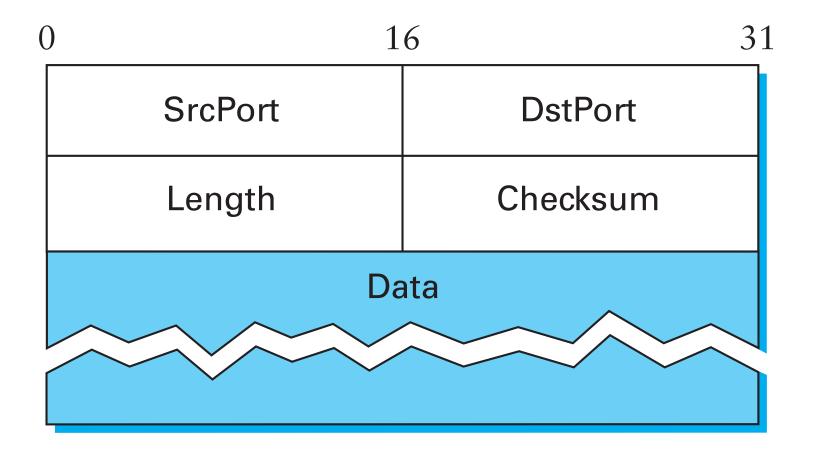


## **UDP – User Datagram Protocol**

- 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

```
0 7 8 15 16 23 24 31
+-----+----+-----+
| source address |
+-----+-----+
| destination address |
+-----+-----+
| zero |protocol| UDP length |
+-----+
```

- UDP Checksum is computer 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

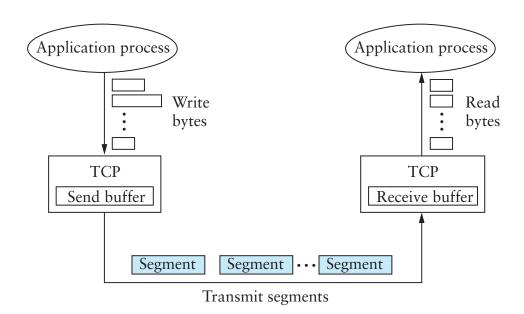
Must be able to discover receiver buffer: flow control

#### 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	ТСР	SCTP	DCCP
Multiplexing				
Connection				
Reliablity				
In-order			optional	
Message				
Stream				
Flow Control				
Congestion Control				
Multiple Streams		*		
Multiple Paths		*		



<sup>\*</sup>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



# 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**

0 3 Source Port Destination Port Sequence Number Acknowledgment Number Data U|A|P|R|S|F| Offset| Reserved |R|C|S|S|Y|I| Window G|K|H|T|N|N| Checksum Urgent Pointer Options Padding data



### **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

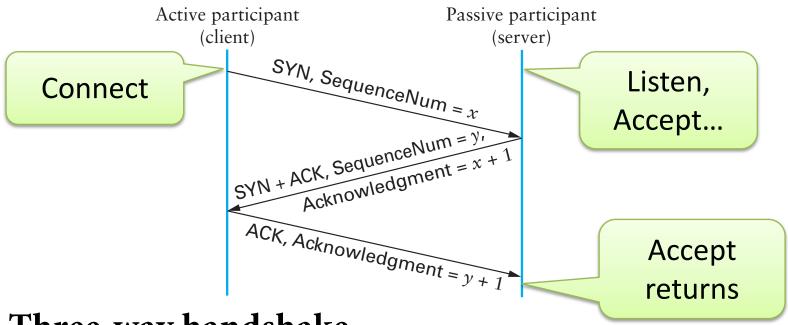


## Header Flags

- 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**



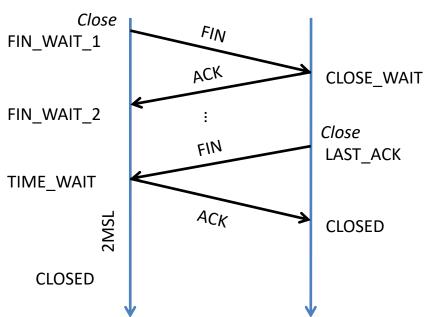
- Three-way handshake
  - Two sides agree on respective initial sequence nums
- If no one is listening on port: server sends RST
- If server is overloaded: ignore SYN
- If no SYN-ACK: retry, timeout



### **Connection Termination**

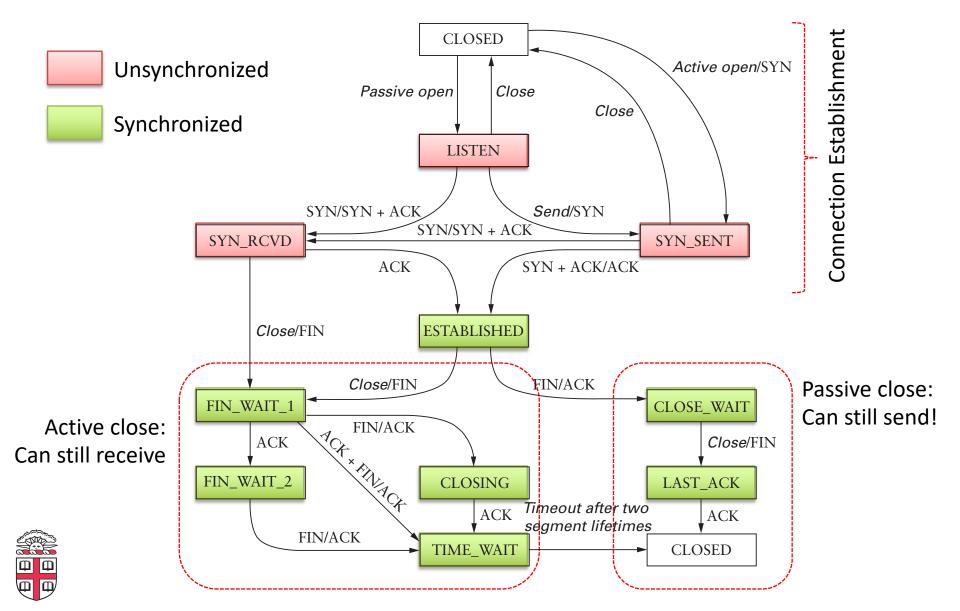
- FIN bit says no more data to send
  - Caused by close or shutdown
  - Both sides must send FIN to close a connection

### Typical close





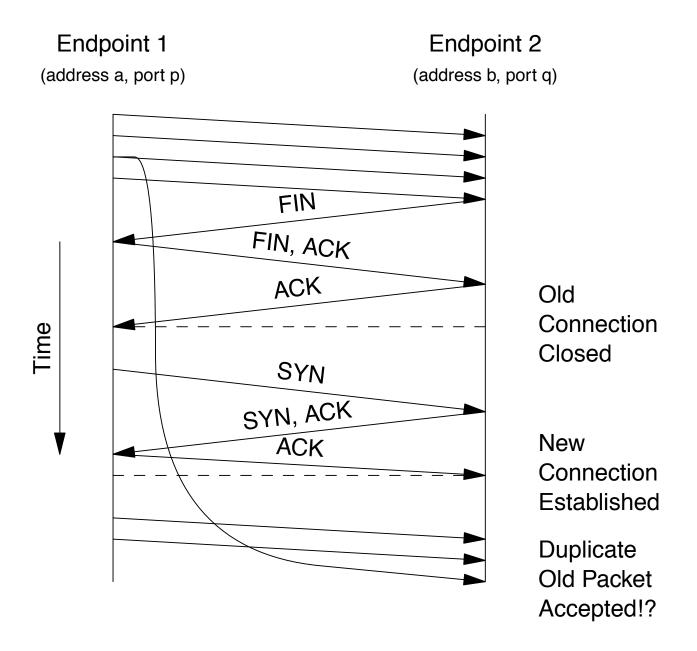
# **Summary of TCP States**



### 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

