Transport layer and UDP







Computer Networking: A Top Down Approach

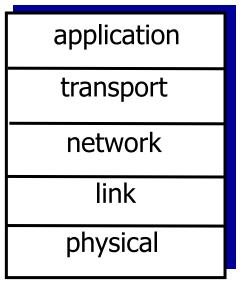
6th edition Jim Kurose, Keith Ross Addison-Wesley





Overview

- Principles underlying transport layer
 - Multiplexing/demultiplexing
 - Detecting errors
 - Reliable delivery
 - Flow control
 - Congestion control
- Major transport layer protocols:
 - User Datagram Protocol (UDP)
 - Simple unreliable message delivery
 - Transmission Control Protocol (TCP)
 - Reliable bidirectional stream of bytes



Chapter 3: Transport Layer

Goals:

- Understand
 principles behind
 transport layer
 services:
 - Multiplexing,demultiplexing
 - Reliable data transfer
 - Flow control
 - Congestion control

- Learn about Internet transport layer protocols:
 - UDP: connectionless transport
 - TCP: connection-oriented reliable transport
 - TCP congestion control

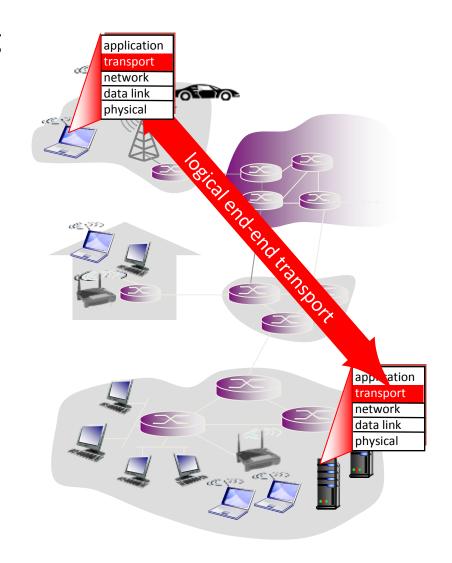
Chapter 3 outline

- 3.1 Transport-layer services
- 3.2 Multiplexing and demultiplexing
- 3.3 Connectionless transport: UDP
- 3.4 Principles of reliable data transfer

- 3.5 Connection-oriented transport: TCP
 - Segment structure
 - Reliable data transfer
 - Flow control
 - Connection management
- 3.6 Principles of congestion control
- 3.7 TCP congestion control

Transport services and protocols

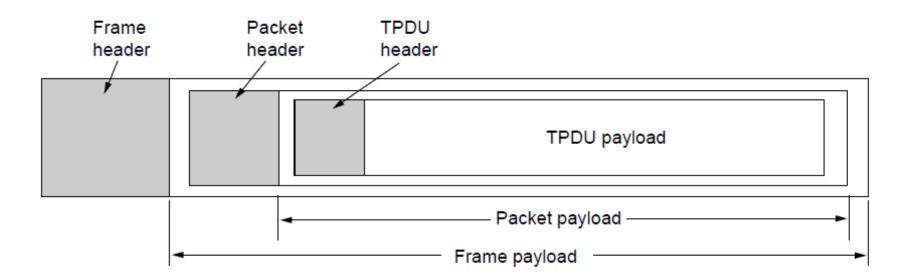
- Provide *logical communication* between app processes running
 on different hosts
- Transport protocols run in end systems
 - Send side: breaks app messages into segments, passes to network layer
 - Recv side: reassembles segments into messages, passes to app layer
- More than one transport protocol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



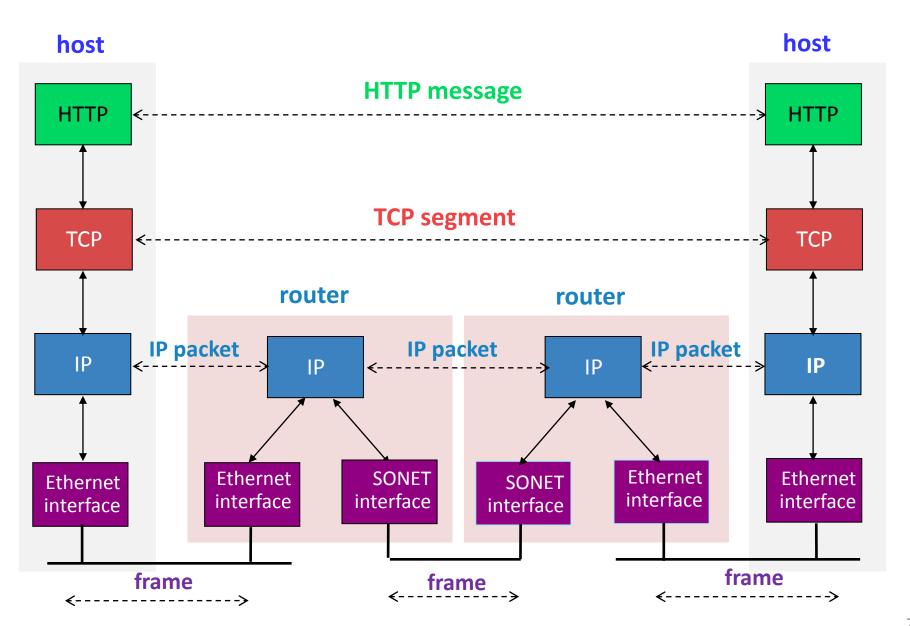
Segments

Segment

- Message sent from one transport entity to another transport entity
- Term used by TCP, UDP, other Internet protocols
- aka TPDU (Transport Protocol Data Unit)



Internet layering model

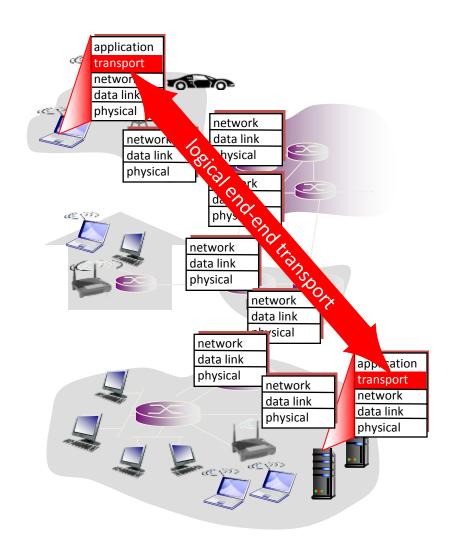


Transport layer challenges

- Running on best-effort network:
 - Messages may be dropped
 - Messages may be reordered
 - Duplicate messages may be delivered
 - Messages have some finite size
 - Messages may arrive after long delay
- Sender must not overrun receiver
- Network may be congested
- Hosts must support multiple applications

Internet transport-layer protocols

- Reliable, in-order delivery: TCP
 - Congestion control
 - Flow control
 - Connection setup
- Unreliable, unordered delivery: UDP
 - No-frills extension of "best-effort" IP
- Services not available:
 - Delay guarantees
 - Bandwidth guarantees



Transport layer

- Goal: End-to-end data transfer
 - Just getting to host machine isn't enough
 - Deliver data from process on sending host to correct process on receiving host
- Solution: OS demultiplexes to correct process
 - Port number, an abstract locater
 - OS demuxes combining with other info

```
UDP <port, host>
TCP <source port, source IP, dest port, dest IP>
```

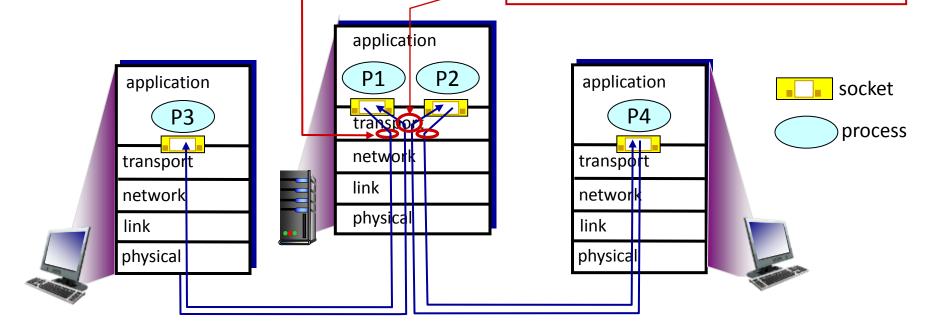
Multiplexing/demultiplexing

Multiplexing at sender:

Handle data from multiple sockets, add transport header (later used for demultiplexing)

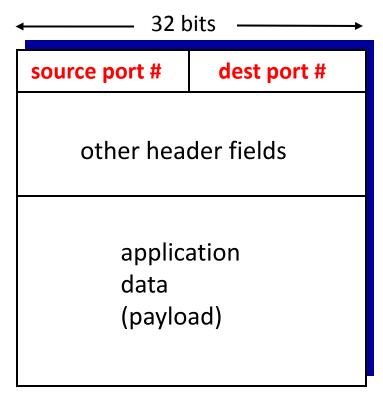
Demultiplexing at receiver:

Use header info to deliver received segments to correct socket



How demultiplexing works

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



TCP/UDP segment format

Connectionless demultiplexing

 Recall: created socket can specify host-local port #:

DatagramSocket mySocket1
= new DatagramSocket(12534);

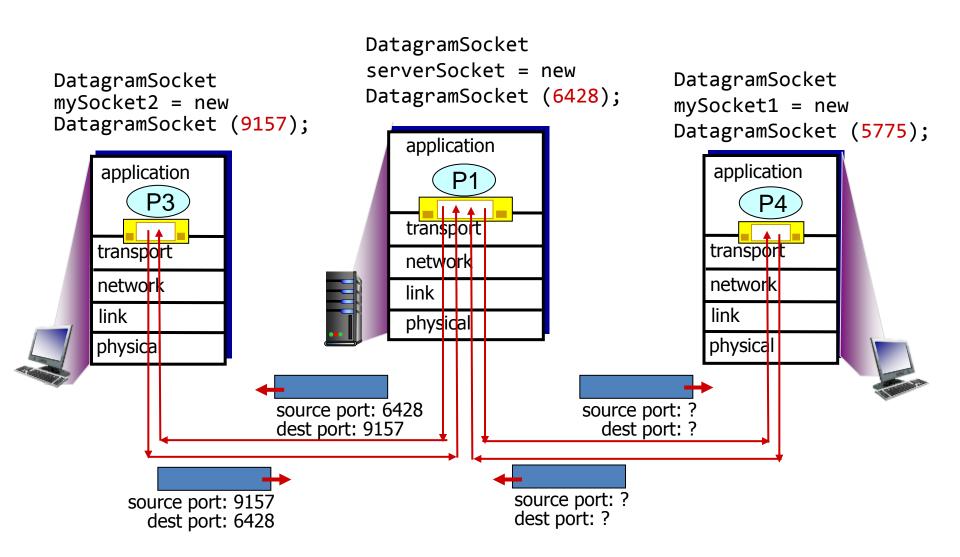
- Recall: when creating datagram to sent into UDP socket, must specify:
 - Destination IP
 - Destination port #

- When host receives UDP segment:
 - Checks destination port # in segment
 - Directs UDP segment to socket with that port #



IP datagrams with same destination port #, but different source IP addresses and/or source port numbers will be directed to same socket at destination

Connectionless demux: example

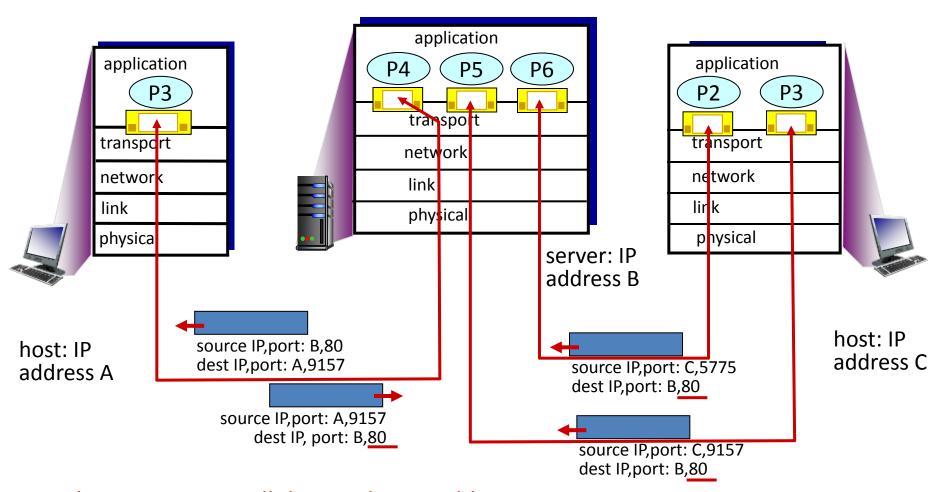


Connection-oriented demux

- TCP socket identified by 4-tuple:
 - Source IP address
 - Source port number
 - Dest IP address
 - Dest port number
- Demux: receiver uses all four values to direct segment to appropriate socket

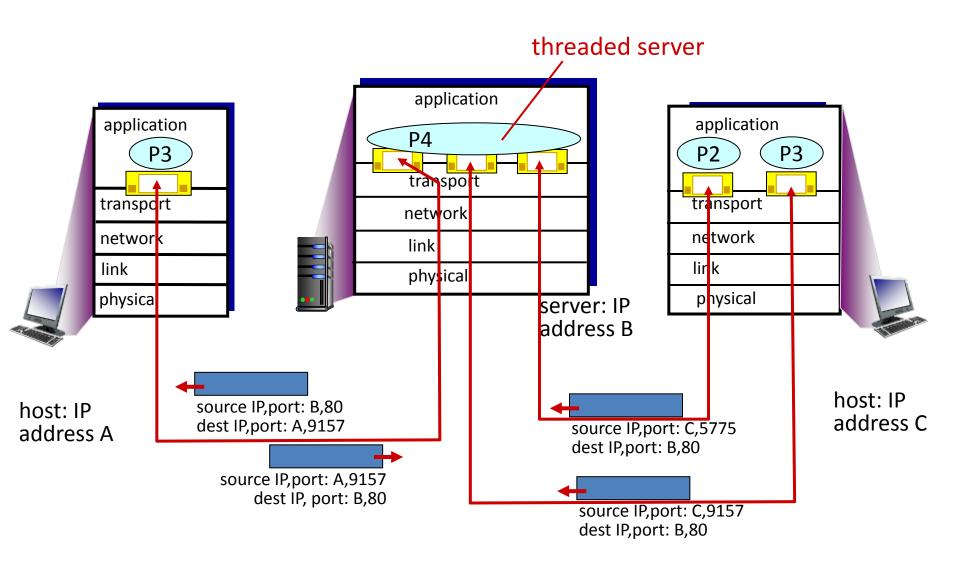
- Server host may support many simultaneous TCP sockets:
 - Each socket identified by its own 4-tuple
- Web servers have different sockets for each connecting client
 - Non-persistent HTTP will have different socket for each request

Connection-oriented demux: example



Three segments, all destined to IP address: B, dest port: 80 are demultiplexed to *different* sockets

Connection-oriented demux: example



Why use UDP?

• Provides:

- Lightweight communication between processes
- Avoid overhead and delays of ordered, reliable delivery
- Precise control of when data is sent
 - As soon as app writes to socket, UDP packages and sends
- No delay establishing a connection
- No connection state, scales to more clients
- Small packet overhead, header only 8 bytes long

Does not provide:

- Flow control
- Congestion control
- Retransmission on error

UDP checksum

Goal: detect errors (e.g. flipped bits) in transmitted segment

Sender:

- Treat segment contents, including header fields, as sequence of 16-bit integers
- Checksum: addition (one's complement sum) of segment contents
- Sender puts checksum value into UDP checksum field

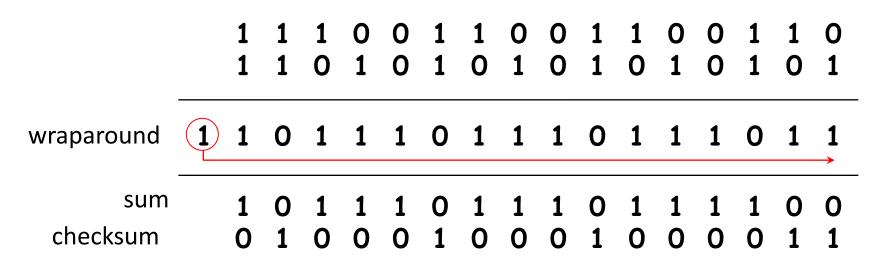
Receiver:

- Compute checksum of received segment
- Check if computed checksum equals checksum field value:
 - NO error detected
 - YES no error detected.
 But maybe errors
 nonetheless? More later

• • • •

Internet checksum: example

example: add two 16-bit integers

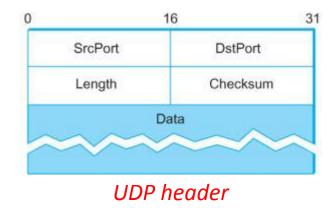


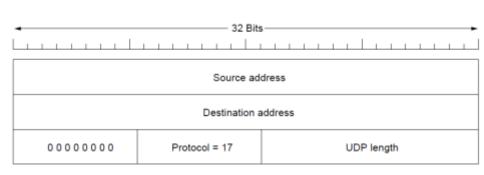
Note: when adding numbers, a carryout from the most significant bit needs to be added to the result

UDP checksums

UDP checksum

- Add up 16-bit words in one's complement
- Take one's complement of the sum
- Done on UDP header, data, IP pseudoheader
 - Helps detect misdelivered packets
 - Violates layers, looking into network layer





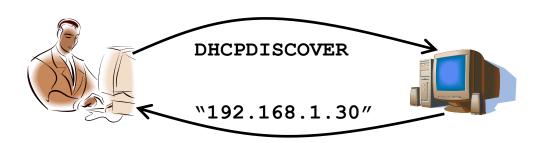
IP pseudoheader

Type of UDP apps, part 1/3

- Simple query protocols
 - Overhead of connection establishment is overkill
 - Easier to have application retransmit if needed
 - e.g. DNS, UDP port 53

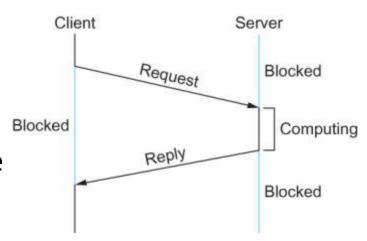


– e.g. DHCP, UDP port 67/68



Type of UDP apps, part 2/3

- Request/reply style interaction
 - Client sends request to server
 - Blocks while waiting for reply
 - Server responds with reply
 - Must deal with:
 - Identify process that can handle request
 - Possible loss of request or reply
 - Correlate request with reply



Request/reply example

- Remote Procedure Call (RPC)
 - Request/reply paradigm over UDP
 - Allow programs to call procedures located on a remote host
 - Invisible to the application programmer
 - Client code blocks while request made and response waited for from remote host
 - Object-oriented languages:
 - Remote Method Invocation(RMI), e.g. Java RMI

Type of UDP apps, part 3/3

- Multimedia streaming
 - e.g. Voice over IP, video conferencing
 - Time is of the essence
 - By time packet is retransmitted, it's too late!
 - Interactive applications:
 - Human-to-human interaction
 - e.g. conference, first-person shooters
 - Streaming applications:
 - Computer-to-human interaction
 - e.g. Netflix, Spotify





Summary

- Transport layer
 - Providing end-to-end process communication
 - Port numbers allow multiple processes per host
 - Provide reliable transport on best-effort network
- User Datagram Protocol (UDP)
 - Lightweight protocol running on top of IP
 - Three typical classes of applications:
 - Simple queries (DNS, DHCP)
 - Request/reply semantics (RPC)
 - Real-time data (Skype)