

Chapter 3 Transport Layer

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Chapter 3: Transport Layer

our 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 of Lecture 6

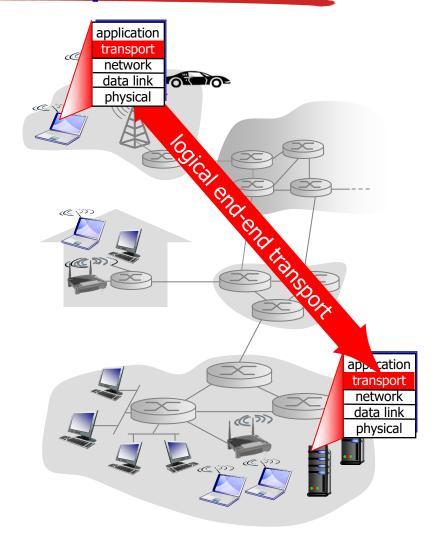


- 3.1 transport-layer services
- 3.2 multiplexing and demultiplexing
- 3.3 connectionless transport: UDP



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
 - rcv 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

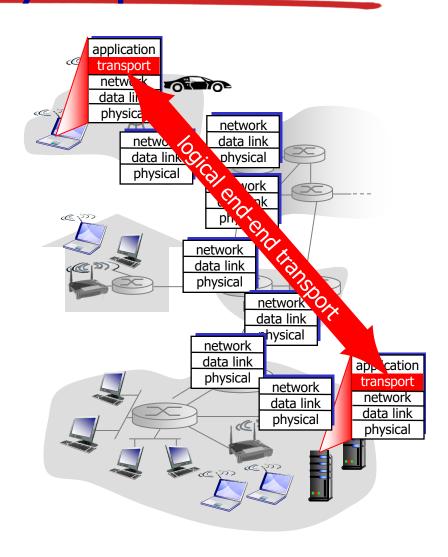
household analogy:

- 12 kids in Ann's house sending letters to 12 kids in Bill's house:
- hosts = houses
- processes = kids
- app messages = letters in envelopes
- transport protocol = Ann and Bill who demux to inhouse siblings
- network-layer protocol = postal service



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





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



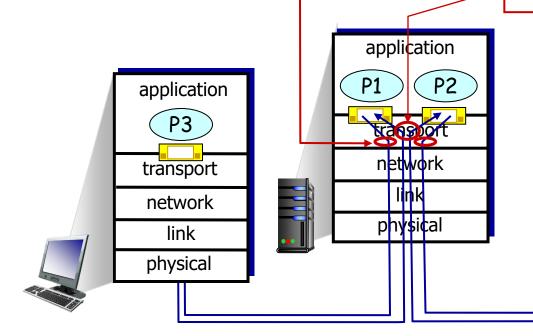
Multiplexing/demultiplexing

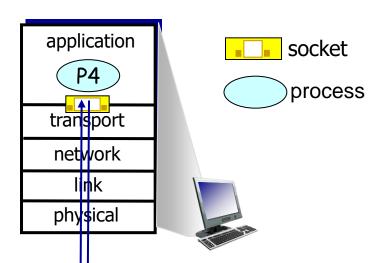
multiplexing at sender:

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

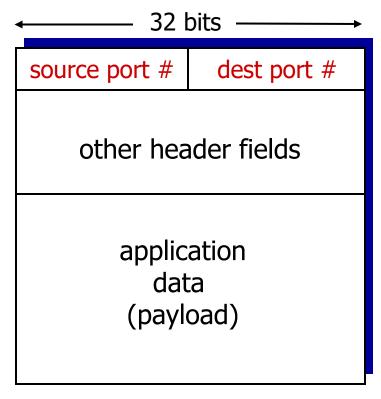






How demultiplexing works

- 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

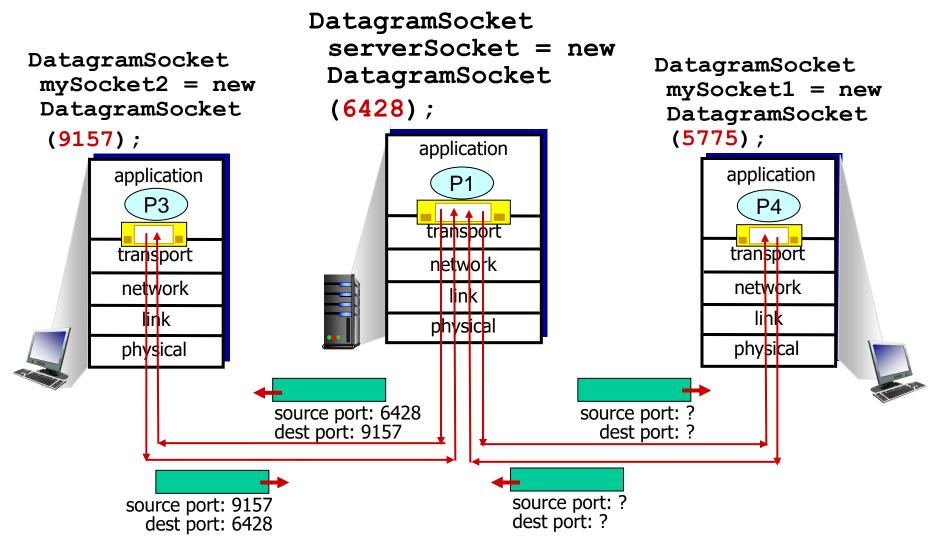
- recall: created socket has host-local port #:
 - DatagramSocket mySocket1
 - = new DatagramSocket(12534);
- recall: when creating datagram to send into UDP socket, must specify
 - destination IP address
 - destination port #

- when host receives UDP segment:
 - checks destination port # in segment
 - directs UDP segment to socket with that port #



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

Connectionless demux: example



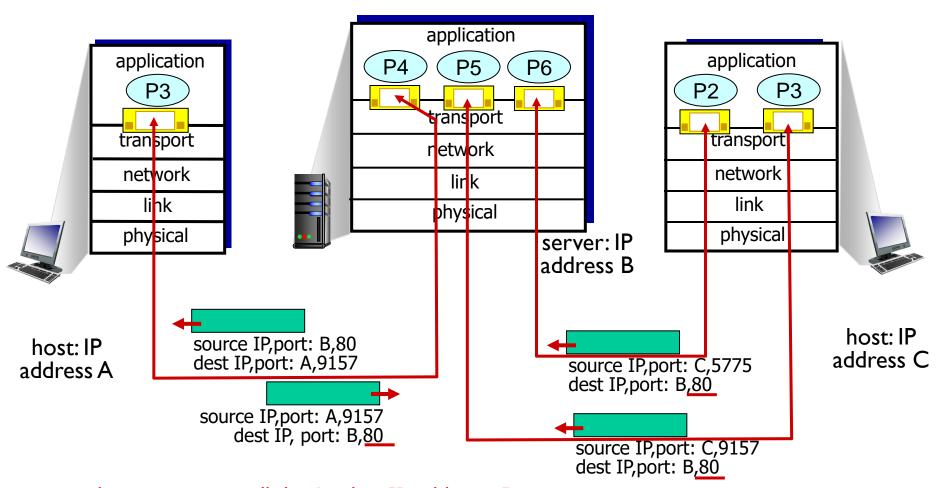


Connection-oriented demux

- 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

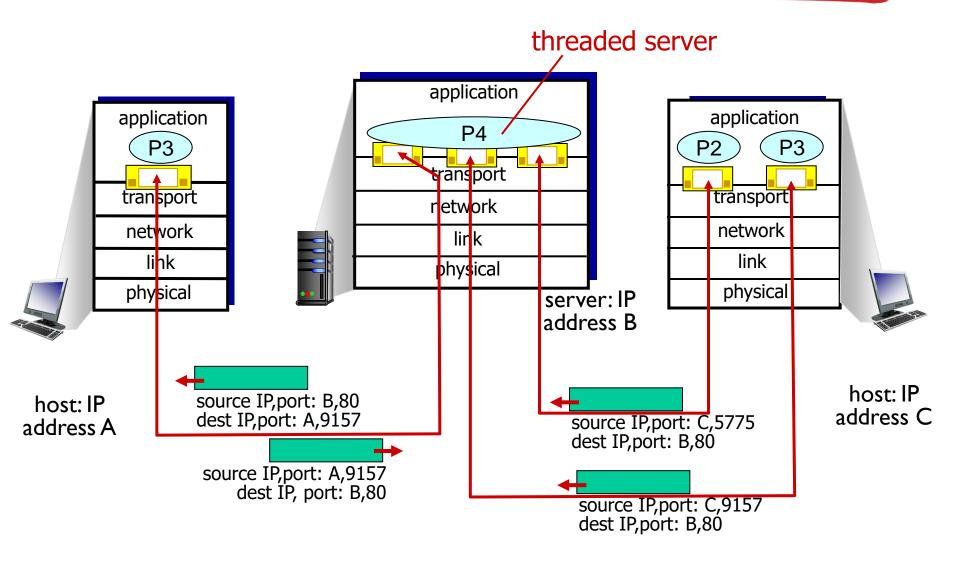
- server host may support many simultaneous TCP sockets:
 - each socket identified by its own 4-tuple
- web servers have different sockets for each connecting client
 - non-persistent HTTP will have different socket for each request

Connection-oriented demux: example



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

Connection-oriented demux: example





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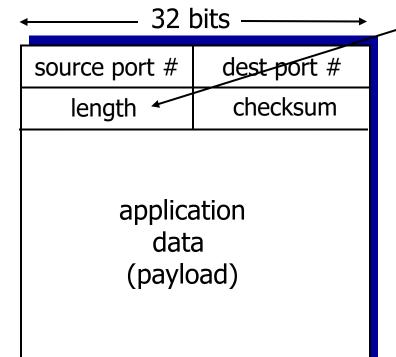
UDP: User Datagram Protocol [RFC 768]

- "no frills," "bare bones" Internet transport protocol
- "best effort" service, UDP segments may be:
 - lost
 - delivered out-of-order to app
- connectionless:
 - no handshaking between UDP sender, receiver
 - each UDP segment handled independently of others

- UDP use:
 - streaming multimedia apps (loss tolerant, rate sensitive)
 - DNS
 - SNMP
- reliable transfer over UDP:
 - add reliability at application layer
 - application-specific error recovery!



UDP: segment header



UDP segment format

length, in bytes of UDP segment, including header

why is there a UDP?

- no connection establishment (which can add delay)
- simple: no connection state at sender, receiver
- small header size
- no congestion control:
 UDP can blast away as fast as desired



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

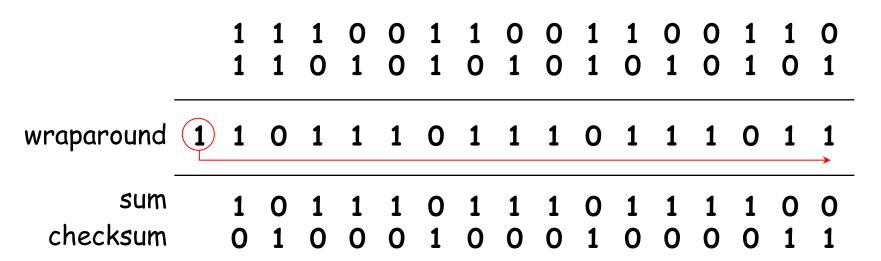
receiver:

- 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

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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

Class Test Time