Chapter 2 - Application layer

Our goals:

- conceptual and implementation aspects of application-layer protocols
 - transport-layer service models
 - client-server paradigm
 - peer-to-peer paradigm

- learn about protocols by examining popular application-layer protocols and infrastructure
 - HTTP
 - SMTP, IMAP
 - DNS
 - video streaming systems, CDNs
- programming network applications
 - socket API

Some network apps

- social networking
- Web
- text messaging
- e-mail
- multi-user network games
- streaming stored video (YouTube, Hulu, Netflix)
- P2P file sharing

- voice over IP (e.g., Skype)
- real-time video conferencing (e.g., Zoom)
- Internet search
- remote login

• • •

Q: your favorites?

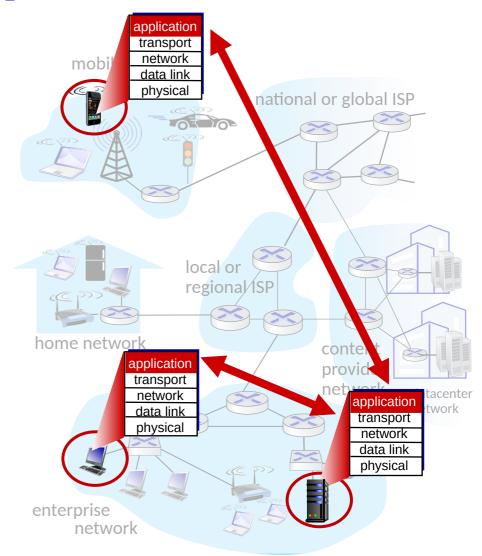
Creating a network app

write programs that:

- run on (different) end systems
- communicate over network
- e.g., web server software communicates with browser software

no need to write software for network-core devices

- network-core devices do not run user applications
- applications on end systems allows for rapid app development, propagation



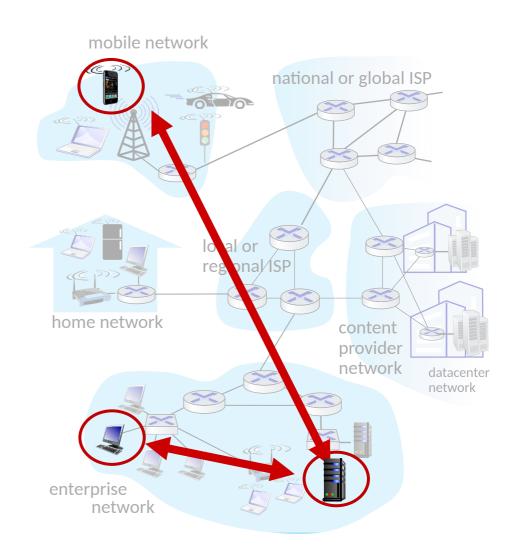
Client-server paradigm

server:

- always-on host
- permanent IP address
- often in data centers, for scaling

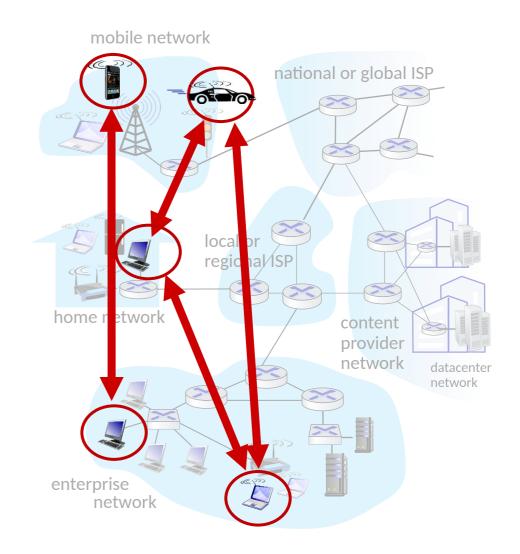
clients:

- contact, communicate with server
- may be intermittently connected
- may have dynamic IP addresses
- do not communicate directly with each other
- examples: HTTP, IMAP, FTP



Peer-peer architecture

- no always-on server
- arbitrary end systems directly communicate
- peers request service from other peers, provide service in return to other peers
 - self scalability new peers bring new service capacity, as well as new service demands
- peers are intermittently connected and change IP addresses
 - complex management
- example: P2P file sharing [BitTorrent]



Processes communicating

process: program running
 within a host

- within same host, two processes communicate using inter-process communication (defined by OS)
- processes in different hosts communicate by exchanging messages

clients, servers

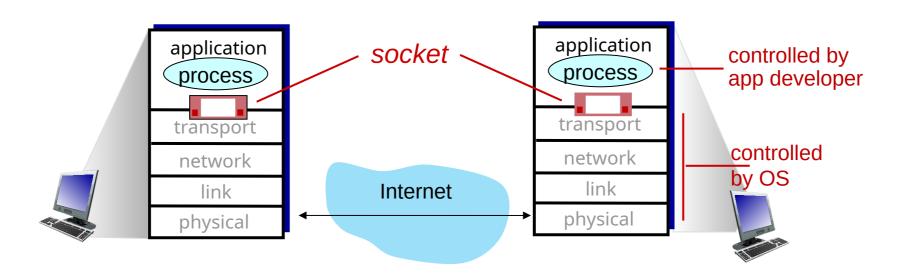
client process: process that initiates communication

server process: process that waits to be contacted

 note: applications with P2P architectures have client processes & server processes

Sockets

- process sends/receives messages to/from its socket
- socket analogous to door
 - sending process shoves message out door
 - sending process relies on transport infrastructure on other side of door to deliver message to socket at receiving process
 - two sockets involved: one on each side



Addressing processes

- to receive messages, process must have identifier
- host device has unique 32-bit IP address
- Q: does IP address of host on which process runs suffice for identifying the process?
 - A: no, many processes can be running on same host

- identifier includes both IP address and port numbers associated with process on host.
- example port numbers:
 - HTTP server: 80
 - mail server: 25
- to send HTTP message to gaia.cs.umass.edu web server:
 - IP address: 128.119.245.12
 - port number: 80
- more shortly...

An application-layer protocol defines:

- types of messages exchanged,
 - e.g., request, response
- message syntax:
 - what fields in messages & how fields are delineated
- message semantics
 - meaning of information in fields
- rules for when and how processes send & respond to messages

open protocols:

- defined in RFCs, everyone has access to protocol definition
- allows for interoperability
- e.g., HTTP, SMTP

proprietary protocols:

e.g., Skype, Zoom

What transport service does an app need?

data integrity

- some apps (e.g., file transfer, web transactions) require
 100% reliable data transfer
- other apps (e.g., audio) can tolerate some loss

timing

some apps (e.g., Internet telephony, interactive games) require low delay to be "effective"

throughput

- some apps (e.g., multimedia) require minimum amount of throughput to be "effective"
- other apps ("elastic apps")
 make use of whatever
 throughput they get

security

encryption, data integrity,

Transport service requirements: common apps

app	olication	data loss	throughput	time sensitive?
file transfer/do	ownload	no loss	elastic	no
	e-mail	no loss	elastic	no
Web documents		no loss	elastic	no
real-time aud	io/video	loss-tolerant	audio: 5Kbps-1Mbps	yes, 10's msec
			video:10Kbps-5Mbps	
streaming aud	io/video	loss-tolerant	same as above	yes, few secs
interactiv	e games	loss-tolerant	Kbps+	yes, 10's msec
text me	essaging	no loss	elastic	yes and no

Internet transport protocols services

TCP service:

- reliable transport between sending and receiving process
- flow control: sender won't overwhelm receiver
- congestion control: throttle sender when network overloaded
- connection-oriented: setup required between client and server processes
- does not provide: timing, minimum throughput guarantee, security

UDP service:

- unreliable data transfer between sending and receiving process
- does not provide: reliability, flow control, congestion control, timing, throughput guarantee, security, or connection setup.

Q: why bother? Why is there a UDP?

Internet applications, and transport protocols

application layer protocol	transport protocol
FTP [RFC 959]	TCP
SMTP [RFC 5321]	TCP
HTTP [RFC 7230, 9110]	TCP
SIP [RFC 3261], RTP [RFC	TCP or UDP
3550], or proprietary	
HTTP [RFC 7230], DASH	TCP
WOW, FPS (proprietary)	UDP or TCP
	Iayer protocol FTP [RFC 959] SMTP [RFC 5321] HTTP [RFC 7230, 9110] SIP [RFC 3261], RTP [RFC 3550], or proprietary HTTP [RFC 7230], DASH

Securing TCP

Vanilla TCP & UDP sockets:

- no encryption
- cleartext passwords sent into socket traverse Internet in cleartext (!)

Transport Layer Security (TLS)

- provides encrypted TCP connections
- data integrity
- end-point authentication

TLS implemented in application layer

- apps use TLS libraries, that use TCP in turn
- cleartext sent into "socket" traverse Internet encrypted
- more: Chapter 8

Application layer: overview

- Principles of network applications
- Web and HTTP
- E-mail, SMTP, IMAP
- The Domain Name System DNS

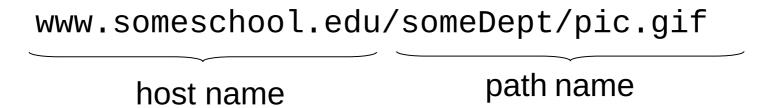
- P2P applications
- video streaming and content distribution networks
- socket programming with UDP and TCP



Web and HTTP

First, a quick review...

- web page consists of objects, each of which can be stored on different Web servers
- object can be HTML file, JPEG image, Java applet, audio file,...
- web page consists of base HTML-file which includes several referenced objects, each addressable by a URL, e.g.,



HTTP overview

HTTP: hypertext transfer protocol

- Web's application-layer protocol
- client/server model:
 - client: browser that requests, receives, (using HTTP protocol) and "displays" Web objects
 - server: Web server sends (using HTTP protocol) objects in response to requests



HTTP overview (continued)

HTTP uses TCP:

- client initiates TCP connection (creates socket) to server, port 80
- server accepts TCP connection from client
- HTTP messages (application-layer protocol messages) exchanged between browser (HTTP client) and Web server (HTTP server)
- TCP connection closed

HTTP is "stateless"

 server maintains no information about past client requests

aside

protocols that maintain "state" are complex!

- past history (state) must be maintained
- if server/client crashes, their views of "state" may be inconsistent, must be reconciled

HTTP connections: two types

Non-persistent HTTP

- 1. TCP connection opened
- 2. at most one object sent over TCP connection
- 3. TCP connection closed

downloading multiple objects required multiple connections

Persistent HTTP

- *TCP connection opened to a server
- multiple objects can be sent over *single* TCP connection between client, and that server
- TCP connection closed

Non-persistent HTTP: example

User enters URL: www.someSchool.edu/someDepartment/home.index (containing text, references to 10 jpeg images)

- 1a. HTTP client initiates TCP connection to HTTP server (process) at www.someSchool.edu on port 80
- 2. HTTP client sends HTTP request message (containing URL) into TCP connection socket. Message indicates that client wants object someDepartment/home.index

- 1b. HTTP server at host www.someSchool.edu waiting for TCP connection at port 80 "accepts" connection, notifying client
- 3. HTTP server receives request message, forms *response message* containing requested object, and sends message into its socket

time

Non-persistent HTTP: example (cont.)

User enters URL: www.someSchool.edu/someDepartment/home.index (containing text, references to 10 jpeg images)



5. HTTP client receives response message containing html file, displays html. Parsing html file, finds 10 referenced jpeg objects



4. HTTP server closes TCP connection.

6. Steps 1-5 repeated for each of 10 jpeg objects

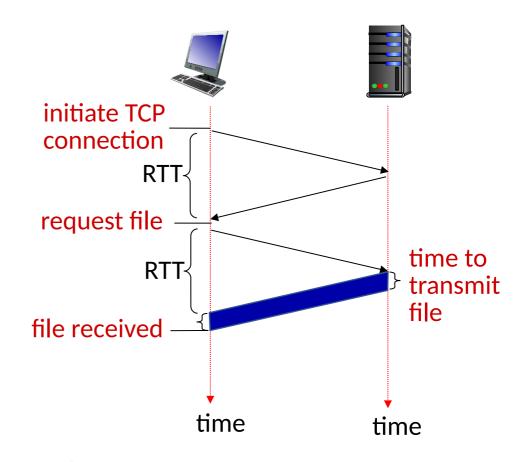


Non-persistent HTTP: response time

RTT (definition): time for a small packet to travel from client to server and back

HTTP response time (per object):

- one RTT to initiate TCP connection
- one RTT for HTTP request and first few bytes of HTTP response to return
- object/file transmission time



Non-persistent HTTP response time = 2RTT+ file transmission time

Persistent HTTP (HTTP 1.1)

Non-persistent HTTP issues:

- requires 2 RTTs per object
- OS overhead for each TCP connection
- browsers often open multiple parallel TCP connections to fetch referenced objects in parallel

Persistent HTTP (HTTP1.1):

- server leaves connection open after sending response
- subsequent HTTP messages between same client/server sent over open connection
- client sends requests as soon as it encounters a referenced object
- as little as one RTT for all the referenced objects (cutting response time in half)

HTTP request message

- two types of HTTP messages: request, response
- HTTP request message:
 - ASCII (human-readable format)

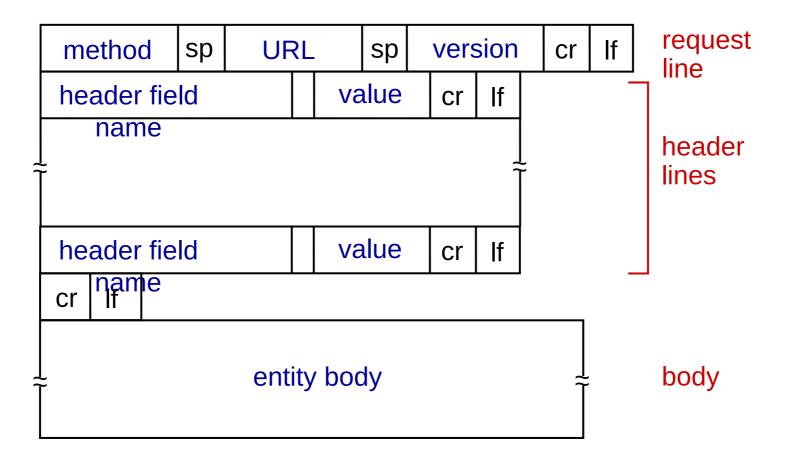
```
request line (GET, POST, HEAD commands)
```

carriage return character line-feed character

carriage return, line feed at start of line indicates end of header lines

^{*} Check out the online interactive exercises for more examples: http://gaia.cs.umass.edu/kurose_ross/interactive/

HTTP request message: general format



Other HTTP request messages

POST method:

- web page often includes form input
- user input sent from client to server in entity body of HTTP POST request message

GET method (for sending data to server):

 include user data in URL field of HTTP GET request message (following a '?'):

www.somesite.com/animalsearch?monkeys&banana

HEAD method:

requests headers (only) that would be returned if specified URL were requested with an HTTP GET method.

PUT method:

- uploads new file (object) to server
- completely replaces file that exists at specified URL with content in entity body of POST HTTP request message

HTTP response message

HTTP response status codes

- status code appears in 1st line in server-to-client response message.
- some sample codes:

200 OK

request succeeded, requested object later in this message

301 Moved Permanently

 requested object moved, new location specified later in this message (in Location: field)

400 Bad Request

request msg not understood by server

404 Not Found

requested document not found on this server

505 HTTP Version Not Supported

Trying out HTTP (client side) for yourself

1. netcat to your favorite Web server:

```
% nc -c -v gaia.cs.umass.edu 80 (for Mac)
```

>ncat -C gaia.cs.umass.edu 80 (for Windows)

Or in CMD do telnet gaia.cs.umass.edu 80

2. type in a GET HTTP request:

- opens TCP connection to port 80 (default HTTP server port) at gaia.cs.umass.edu.
- anything typed in will be sent to port 80 at gaia.cs.umass.edu

```
GET /kurose_ross/interactive/index.php HTTP/1.1
```

By typing this in (hit carriage return twice), you send this minimal (but complete) GET request to HTTP server

3. look at response message sent by HTTP server!

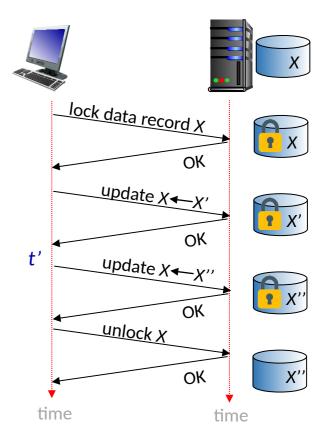
(or use Wireshark to look at captured HTTP request/response)

Maintaining user/server state: cookies

Recall: HTTP GET/response interaction is *stateless*

- no notion of multi-step exchanges of HTTP messages to complete a Web "transaction"
 - no need for client/server to track "state" of multi-step exchange
 - all HTTP requests are independent of each other
 - no need for client/server to "recover" from a partially-completed-but-nevercompletely-completed transaction

a stateful protocol: client makes two changes to X, or none at all



Q: what happens if network connection or client crashes at t?

Maintaining user/server state: cookies

Web sites and client browser use cookies to maintain some state between transactions

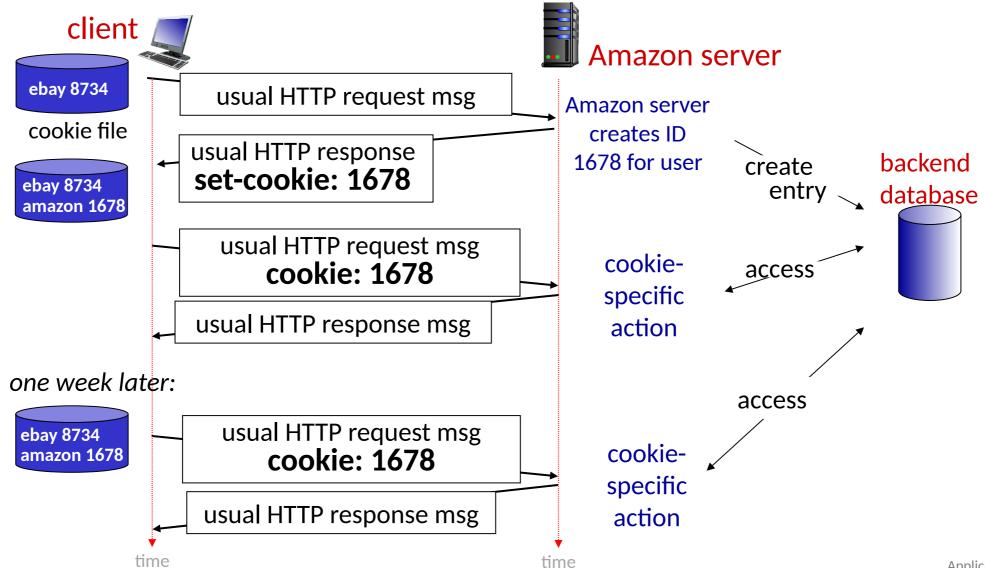
four components:

- 1) cookie header line of HTTP *response* message
- 2) cookie header line in next HTTP request message
- 3) cookie file kept on user's host, managed by user's browser
- 4) back-end database at Web site

Example:

- Susan uses browser on laptop, visits specific e-commerce site for first time
- when initial HTTP requests arrives at site, site creates:
 - unique ID (aka "cookie")
 - entry in backend database for ID
- subsequent HTTP requests from Susan to this site will contain cookie ID value, allowing site to "identify" Susan

Maintaining user/server state: cookies



HTTP cookies: comments

What cookies can be used for:

- authorization
- shopping carts
- recommendations
- user session state (Web e-mail)

Challenge: How to keep state?

- at protocol endpoints: maintain state at sender/receiver over multiple transactions
- in messages: cookies in HTTP messages carry state

cookies and privacy:

- cookies permit sites to learn a lot about you on their site.
- third party persistent cookies (tracking cookies) allow common identity (cookie value) to be tracked across multiple web sites

Cookies: tracking a user's browsing behavior

Cookies can be used to:

- track user behavior on a given website (first party cookies)
- track user behavior across multiple websites (third party cookies) without user ever choosing to visit tracker site (!)
- tracking may be invisible to user:
 - rather than displayed ad triggering HTTP GET to tracker, could be an invisible link

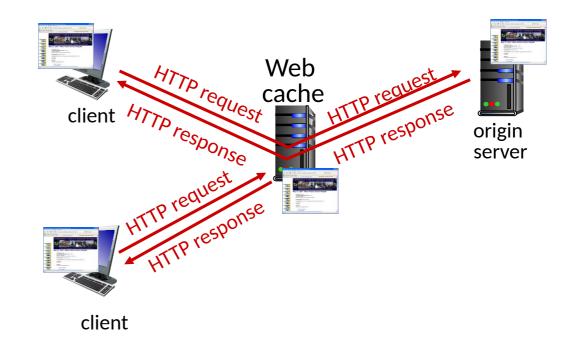
third party tracking via cookies:

- disabled by default in Firefox, Safari browsers
- to be disabled in Chrome browser in 2023 ????

Web caches

Goal: satisfy client requests without involving origin server

- user configures browser to point to a (local) Web cache
- browser sends all HTTP requests to cache
 - *if* object in cache: cache returns object to client
 - else cache requests object from origin server, caches received object, then returns object to client



Web caches (aka proxy servers)

- Web cache acts as both client and server
 - server for original requesting client
 - client to origin server
- server tells cache about object's allowable caching in response header:

```
Cache-Control: max-age=<seconds>
```

Cache-Control: no-cache

Why Web caching?

- reduce response time for client request
 - cache is closer to client
- reduce traffic on an institution's access link
- Internet is dense with caches
 - enables "poor" content providers to more effectively deliver content

Caching example

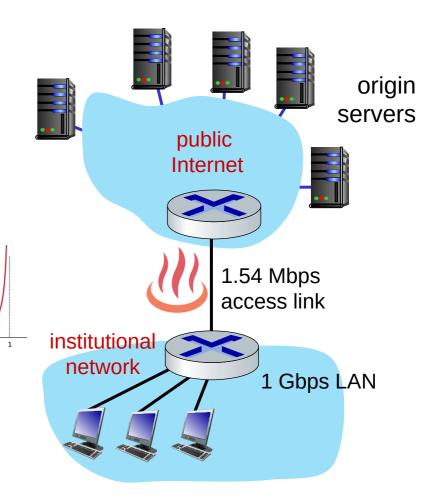
Scenario:

- access link rate: 1.54 Mbps
- RTT from institutional router to server: 2 sec
- web object size: 100K bits
- average request rate from browsers to origin servers: 15/sec
 - avg data rate to browsers: 1.50 Mbps

Performance:

- access link utilization € .97
- LAN utilization: .0015

- problem: large full queueing delays at
- high utilization!
- end-end delay = Internet delay + access link delay + LAN delay
 - = 2 sec +(minutes) + usecs



Option 1: buy a faster access link

Scenario: __154 Mbps

- access link rate: 1.54 Mbps
- RTT from institutional router to server: 2 sec
- web object size: 100K bits
- average request rate from browsers to origin servers: 15/sec
 - avg data rate to browsers: 1.50 Mbps

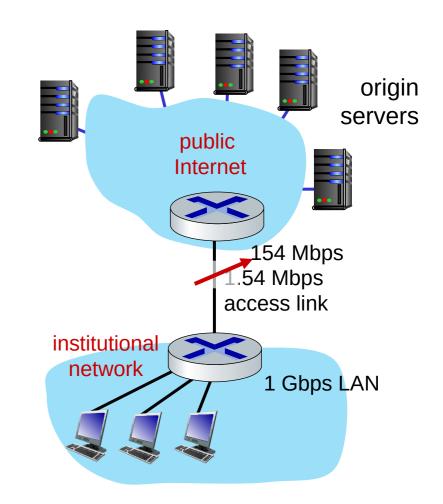
Performance:

- access link utilization = .97 → .0097
- LAN utilization: .0015
- end-end delay = Internet delay + access link delay + LAN delay

= 2 sec + minutes + usecs

msecs

Cost: faster access link (expensive!)



Option 2: install a web cache

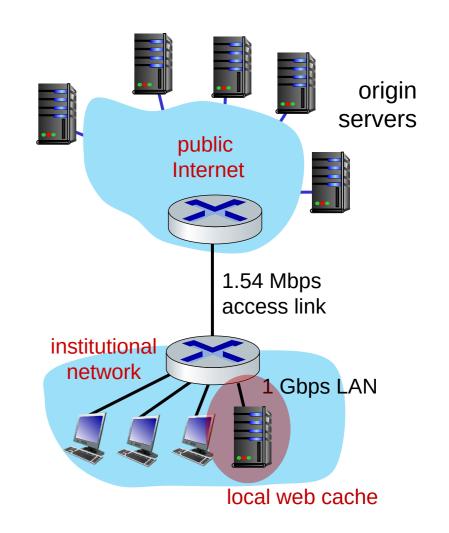
Scenario:

- access link rate: 1.54 Mbps
- RTT from institutional router to server: 2 sec
- web object size: 100K bits
- average request rate from browsers to origin servers: 15/sec
 - avg data rate to browsers: 1.50 Mbps

Cost: web cache (cheap!)

Performance:

- LAN utilization: .? How to compute link
 access link utilization = ? utilization, delay?
- average end-end delay = ?



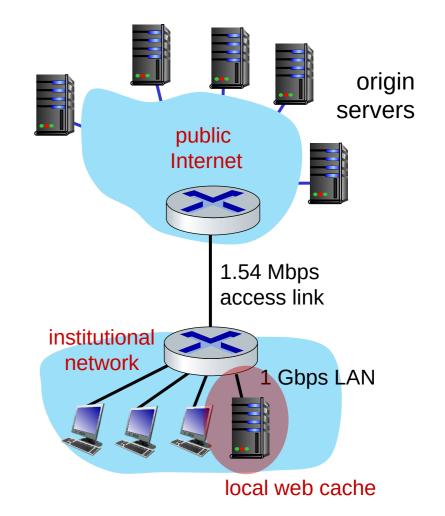
Calculating access link utilization, end-end delay with cache:

suppose cache hit rate is 0.4:

- 40% requests served by cache, with low (msec) delay
- 60% requests satisfied at origin
 - rate to browsers over access link

$$= 0.6 * 1.50 \text{ Mbps} = .9 \text{ Mbps}$$

- access link utilization = 0.9/1.54 = .58 means low (msec) queueing delay at access link
- average end-end delay:
 - = 0.6 * (delay from origin servers)+ 0.4 * (delay when satisfied at cache)
 - $= 0.6 (2.01) + 0.4 (\sim msecs) = \sim 1.2 secs$



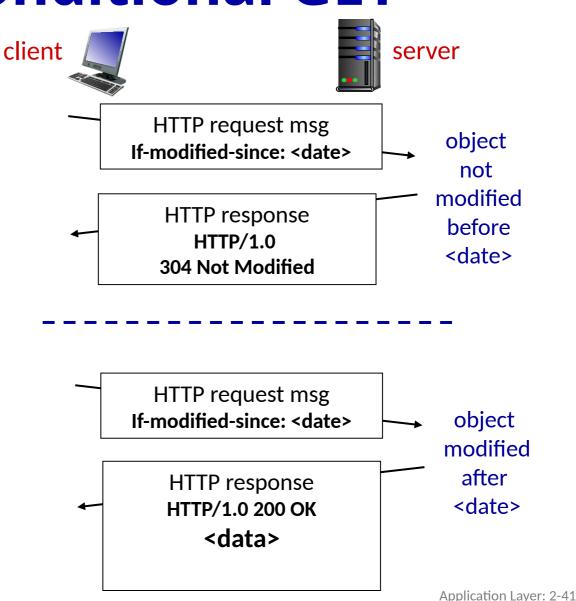
lower average end-end delay than with 154 Mbps link (and cheaper too!)

Browser caching: Conditional GET

Goal: don't send object if browser has up-to-date cached version

- no object transmission delay (or use of network resources)
- client: specify date of browsercached copy in HTTP request
 If-modified-since: <date>
- server: response contains no object if browser-cached copy is up-to-date:

HTTP/1.0 304 Not Modified



HTTP/2

Key goal: decreased delay in multi-object HTTP requests

<u>HTTP1.1:</u> introduced multiple, pipelined GETs over single TCP connection

- server responds in-order (FCFS: first-come-first-served scheduling) to GET requests
- with FCFS, small object may have to wait for transmission (head-of-line (HOL) blocking) behind large object(s)
- loss recovery (retransmitting lost TCP segments) stalls object transmission

HTTP/2

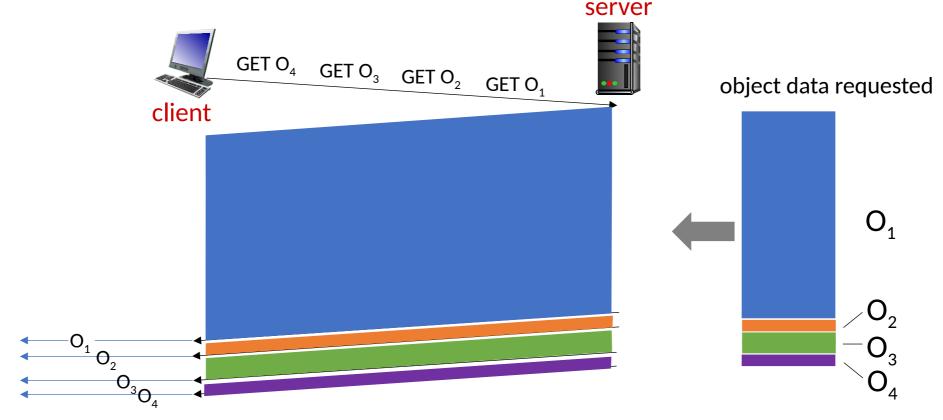
Key goal: decreased delay in multi-object HTTP requests

<u>HTTP/2:</u> [RFC 7540, 2015] increased flexibility at server in sending objects to client:

- methods, status codes, most header fields unchanged from HTTP 1.1
- transmission order of requested objects based on client-specified object priority (not necessarily FCFS)
- push unrequested objects to client
- divide objects into frames, schedule frames to mitigate HOL blocking

HTTP/2: mitigating HOL blocking

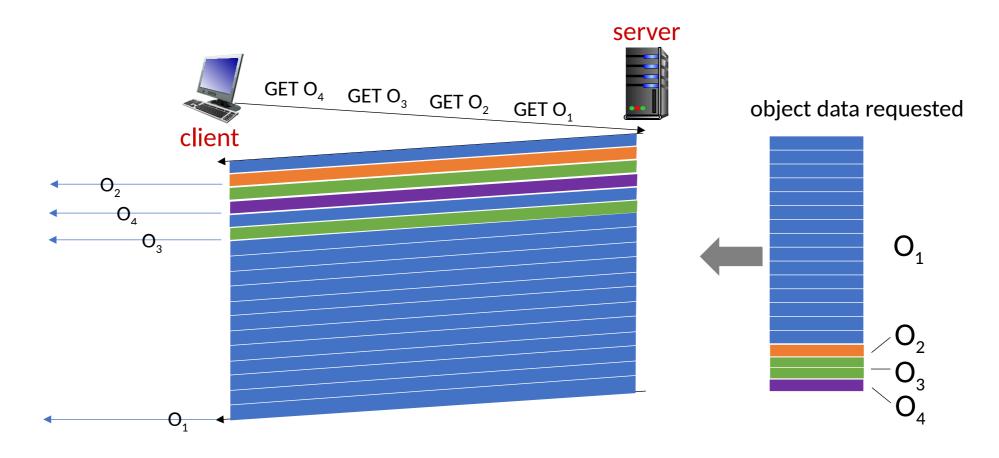
HTTP 1.1: client requests 1 large object (e.g., video file) and 3 smaller objects



objects delivered in order requested: O_2 , O_3 , O_4 wait behind O_1

HTTP/2: mitigating HOL blocking

HTTP/2: objects divided into frames, frame transmission interleaved



O₂, O₃, O₄ delivered quickly, O₁ slightly delayed

HTTP/2 to HTTP/3

HTTP/2 over single TCP connection means:

- recovery from packet loss still stalls all object transmissions
 - as in HTTP 1.1, browsers have incentive to open multiple parallel TCP connections to reduce stalling, increase overall throughput
- no security over vanilla TCP connection
- HTTP/3: adds security, per object error- and congestioncontrol (more pipelining) over UDP
 - more on HTTP/3 in transport layer

Application layer: overview

- Principles of network applications
- Web and HTTP
- E-mail, SMTP, IMAP
- The Domain Name System DNS

- P2P applications
- video streaming and content distribution networks
- socket programming with UDP and TCP



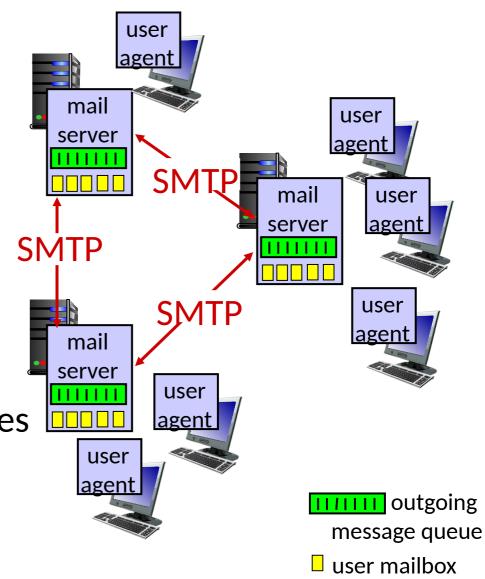
E-mail

Three major components:

- user agents
- mail servers
- simple mail transfer protocol: SMTP

User Agent

- a.k.a. "mail reader"
- composing, editing, reading mail messages
- e.g., Outlook, iPhone mail client
- outgoing, incoming messages stored on server



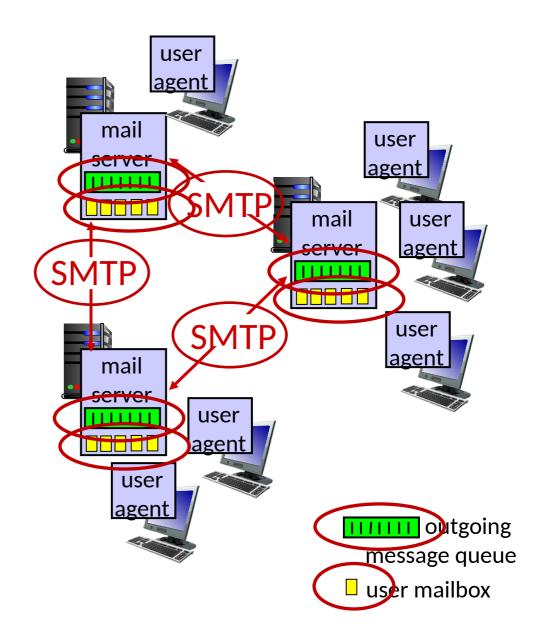
E-mail: mail servers

mail servers:

- mailbox contains incoming messages for user
- message queue of outgoing (to be sent) mail messages

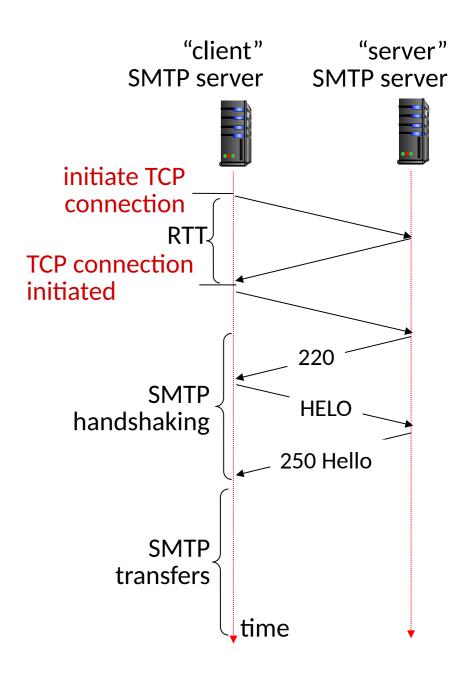
SMTP protocol between mail servers to send email messages

- client: sending mail server
- "server": receiving mail server



SMTP RFC (5321)

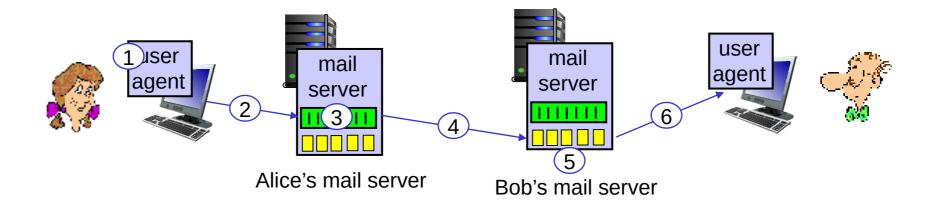
- uses TCP to reliably transfer email message from client (mail server initiating connection) to server, port 25
 - direct transfer: sending server (acting like client) to receiving server
- three phases of transfer
 - SMTP handshaking (greeting)
 - SMTP transfer of messages
 - SMTP closure
- command/response interaction (like HTTP)
 - commands: ASCII text
 - response: status code and phrase



Scenario: Alice sends e-mail to Bob

- 1) Alice uses UA to compose e-mail message "to" bob@someschool.edu
- 2) Alice's UA sends message to her mail server using SMTP; message placed in message queue
- 3) client side of SMTP at mail server opens TCP connection with Bob's mail server

- 4) SMTP client sends Alice's message over the TCP connection
- 5) Bob's mail server places the message in Bob's mailbox
- 6) Bob invokes his user agent to read message



Sample SMTP interaction

S: 220 hamburger.edu

SMTP: observations

comparison with HTTP:

- HTTP: client pull
- SMTP: client push
- both have ASCII command/response interaction, status codes
- HTTP: each object encapsulated in its own response message
- SMTP: multiple objects sent in multipart message

- SMTP uses persistent connections
- SMTP requires message (header & body) to be in 7-bit ASCII
- SMTP server uses CRLF.CRLF to determine end of message

Mail message format

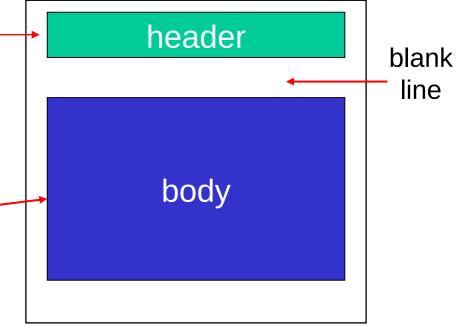
SMTP: protocol for exchanging e-mail messages, defined in RFC 5321 (like RFC 7231 defines HTTP)

RFC 2822 defines *syntax* for e-mail message itself (like HTML defines syntax for web documents)

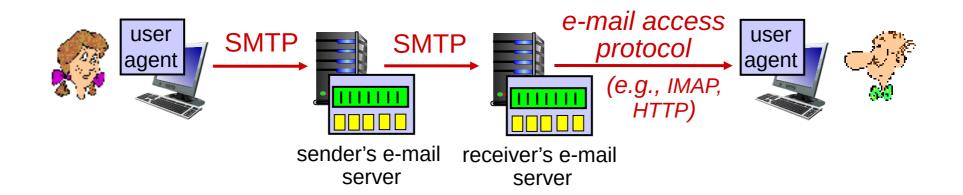
- header lines, e.g.,
 - To:
 - From:
 - Subject:

these lines, within the body of the email message area different from SMTP MAIL FROM:, RCPT TO: commands!

Body: the "message", ASCII characters only



Retrieving email: mail access protocols



- SMTP: delivery/storage of e-mail messages to receiver's server
- mail access protocol: retrieval from server
 - IMAP: Internet Mail Access Protocol [RFC 3501]: messages stored on server, IMAP provides retrieval, deletion, folders of stored messages on server
- HTTP: gmail, Hotmail, Yahoo!Mail, etc. provides web-based interface on top of STMP (to send), IMAP (or POP) to retrieve e-mail messages

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- P2P applications
- video streaming and content distribution networks
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DNS: Domain Name System

people: many identifiers:

SSN, name, passport #

Internet hosts, routers:

- IP address (32 bit) used for addressing datagrams
- "name", e.g., cs.umass.edu used by humans

Q: how to map between IP address and name, and vice versa?

Domain Name System (DNS):

- distributed database implemented in hierarchy of many name servers
- application-layer protocol: hosts, DNS servers communicate to resolve names (address/name translation)
 - note: core Internet function, implemented as application-layer protocol
 - complexity at network's "edge"

DNS: services, structure

DNS services:

- hostname-to-IP-address translation
- host aliasing
 - canonical, alias names
- mail server aliasing
- load distribution
 - replicated Web servers: many IP addresses correspond to one name

Q: Why not centralize DNS?

- single point of failure
- traffic volume
- distant centralized database
- maintenance

A: doesn't scale!

- Comcast DNS servers alone: 600B DNS queries/day
- Akamai DNS servers alone:2.2T DNS queries/day

Thinking about the DNS

humongous distributed database:

~ billion records, each simple

handles many trillions of queries/day:

- many more reads than writes
- performance matters: almost every Internet transaction interacts with DNS - msecs count!

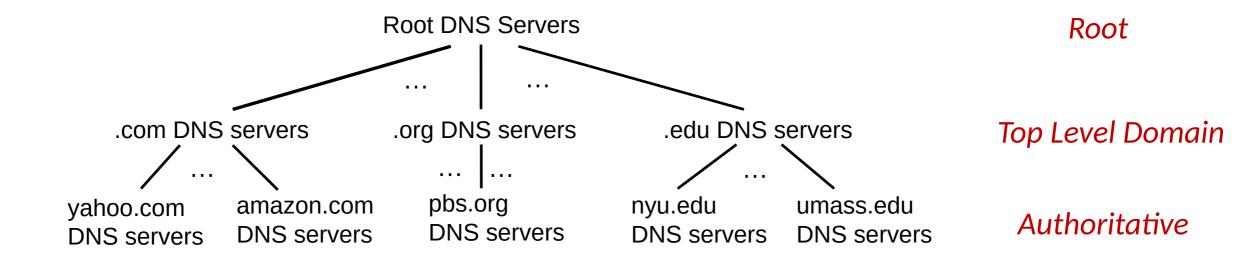
organizationally, physically decentralized:

 millions of different organizations responsible for their records

"bulletproof": reliability, security



DNS: a distributed, hierarchical database

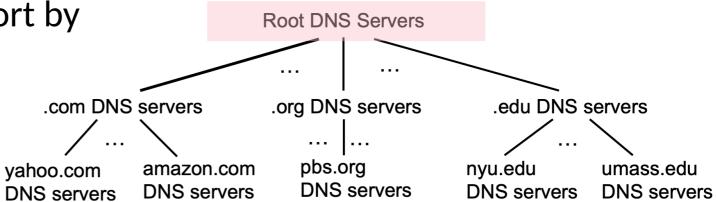


Client wants IP address for www.amazon.com; 1st approximation:

- client queries root server to find .com DNS server
- client queries .com DNS server to get amazon.com DNS server
- client queries amazon.com DNS server to get IP address for www.amazon.com

DNS: root name servers

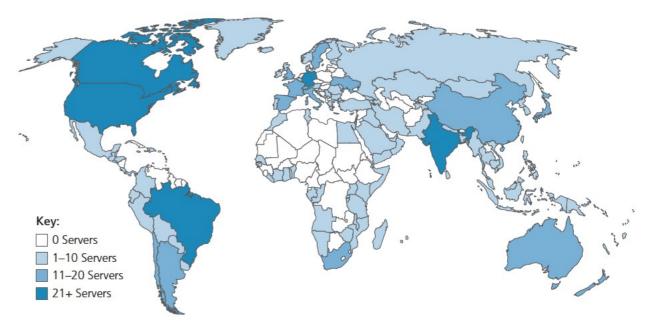
 official, contact-of-last-resort by name servers that can not resolve name



DNS: root name servers

- official, contact-of-last-resort by name servers that can not resolve name
- incredibly important Internet function
 - Internet couldn't function without it!
 - DNSSEC provides security (authentication, message integrity)
- ICANN (Internet Corporation for Assigned Names and Numbers) manages root DNS domain

13 logical root name "servers" worldwide each "server" replicated many times (~200 servers in US)

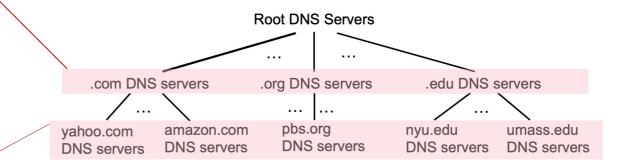


Top-Level Domain, and authoritative servers

Top-Level Domain (TLD) servers:

- responsible for .com, .org, .net, .edu, .aero, .jobs, .museums, and all top-level country domains, e.g.: .cn, .uk, .fr, .ca, .jp
- Network Solutions: authoritative registry for .com, .net TLD

Educause: .edu TLD



authoritative DNS servers:

- organization's own DNS server(s), providing authoritative hostname to IP mappings for organization's named hosts
- can be maintained by organization or service provider

Local DNS name servers

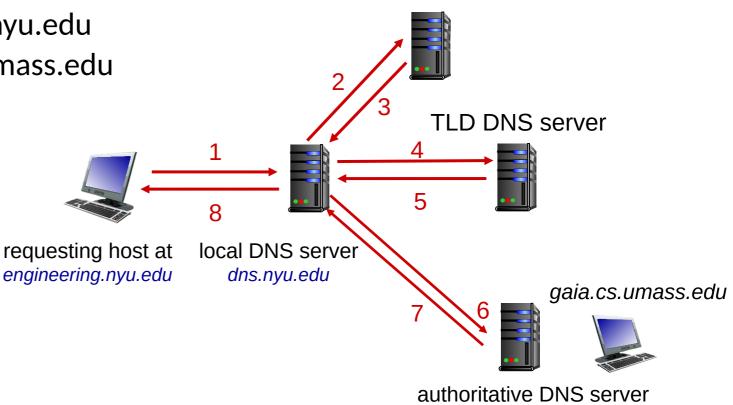
- when host makes DNS query, it is sent to its local DNS server
 - Local DNS server returns reply, answering:
 - from its local cache of recent name-to-address translation pairs (possibly out of date!)
 - forwarding request into DNS hierarchy for resolution
 - each ISP has local DNS name server; to find yours:
 - MacOS: % scutil --dns
 - Windows: >ipconfig /all
- local DNS server doesn't strictly belong to hierarchy

DNS name resolution: iterated query

Example: host at engineering.nyu.edu wants IP address for gaia.cs.umass.edu

Iterated query:

- contacted server replies with name of server to contact
- "I don't know this name, but ask this server"



root DNS server

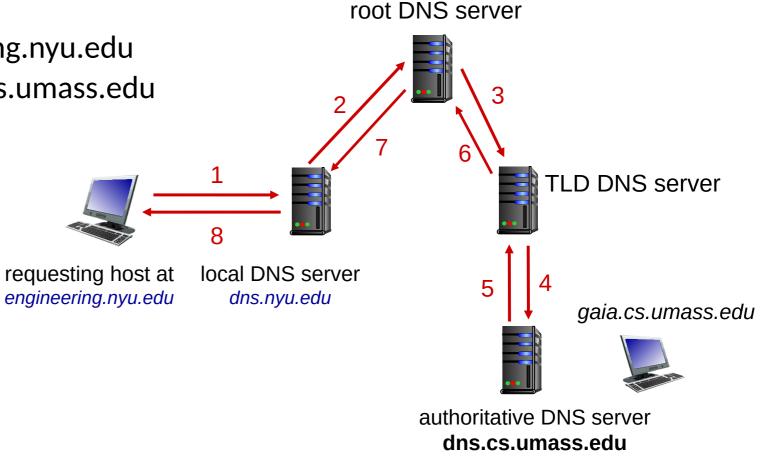
authoritative DNS server dns.cs.umass.edu

DNS name resolution: recursive query

Example: host at engineering.nyu.edu wants IP address for gaia.cs.umass.edu

Recursive query:

- puts burden of name resolution on contacted name server
- heavy load at upper levels of hierarchy?



Caching DNS Information

- once (any) name server learns mapping, it caches mapping, and immediately returns a cached mapping in response to a query
 - caching improves response time
 - cache entries timeout (disappear) after some time (TTL)
 - TLD servers typically cached in local name servers
- cached entries may be out-of-date
 - if named host changes IP address, may not be known Internetwide until all TTLs expire!
 - best-effort name-to-address translation!

DNS records

DNS: distributed database storing resource records (RR)

RR format: (name, value, type, ttl)

type=A

- name is hostname
- value is IP address

type=NS

- name is domain (e.g., foo.com)
- value is hostname of authoritative name server for this domain

type=CNAME

- name is alias name for some "canonical" (the real) name
- www.ibm.com is really servereast.backup2.ibm.com
- value is canonical name

type=MX

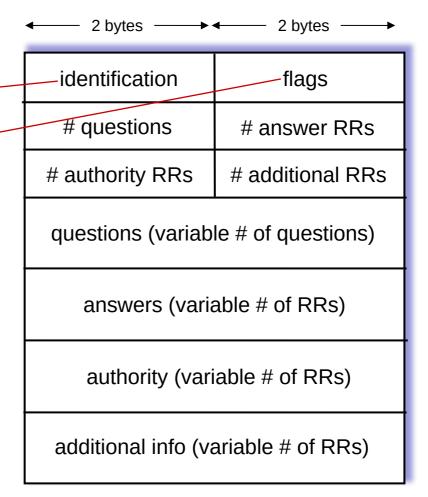
value is name of SMTP mail server associated with name

DNS protocol messages

DNS query and reply messages, both have same format:

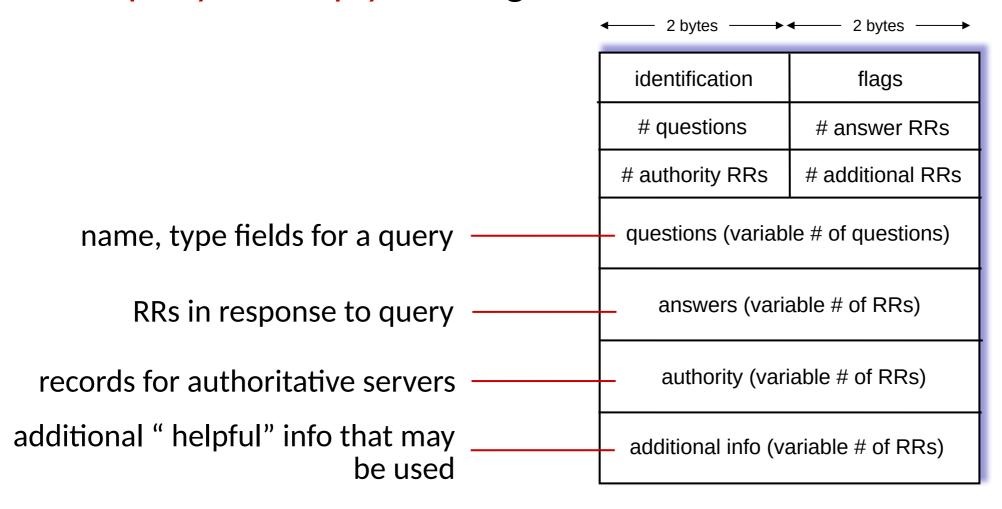
message header:

- identification: 16 bit # for query, reply to query uses same #
- flags:
 - query or reply
 - recursion desired
 - recursion available
 - reply is authoritative



DNS protocol messages

DNS query and reply messages, both have same format:



Getting your info into the DNS

example: new startup "Network Utopia"

- register name networkuptopia.com at DNS registrar (e.g., Network Solutions)
 - provide names, IP addresses of authoritative name server (primary and secondary)
 - registrar inserts NS, A RRs into .com TLD server: (networkutopia.com, dns1.networkutopia.com, NS)

```
(dns1.networkutopia.com, 212.212.212.1, A)
```

- create authoritative server locally with IP address 212.212.212.1
 - type A record for www.networkuptopia.com
 - type MX record for networkutopia.com

DNS security

DDoS attacks

- bombard root servers with traffic
 - not successful to date
 - traffic filtering
 - local DNS servers cache IPs of TLD servers, allowing root server bypass
- bombard TLD servers
 - potentially more dangerous

Spoofing attacks

- intercept DNS queries, returning bogus replies
 - DNS cache poisoning
 - RFC 4033: DNSSEC authentication services

Application Layer: Overview

- Principles of network applications
- Web and HTTP
- E-mail, SMTP, IMAP
- The Domain Name System DNS

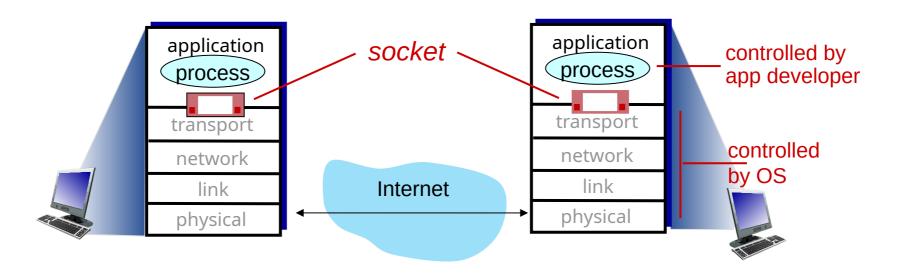
- P2P applications
- video streaming and content distribution networks
- socket programming with UDP and TCP



Socket programming

goal: learn how to build client/server applications that communicate using sockets

socket: door between application process and end-end-transport protocol



Socket programming

Two socket types for two transport services:

- UDP: unreliable datagram
- TCP: reliable, byte stream-oriented

Application Example:

- 1. client reads a line of characters (data) from its keyboard and sends data to server
- 2. server receives the data and converts characters to uppercase
- 3. server sends modified data to client
- 4. client receives modified data and displays line on its screen

Socket programming with UDP

UDP: no "connection" between client and server:

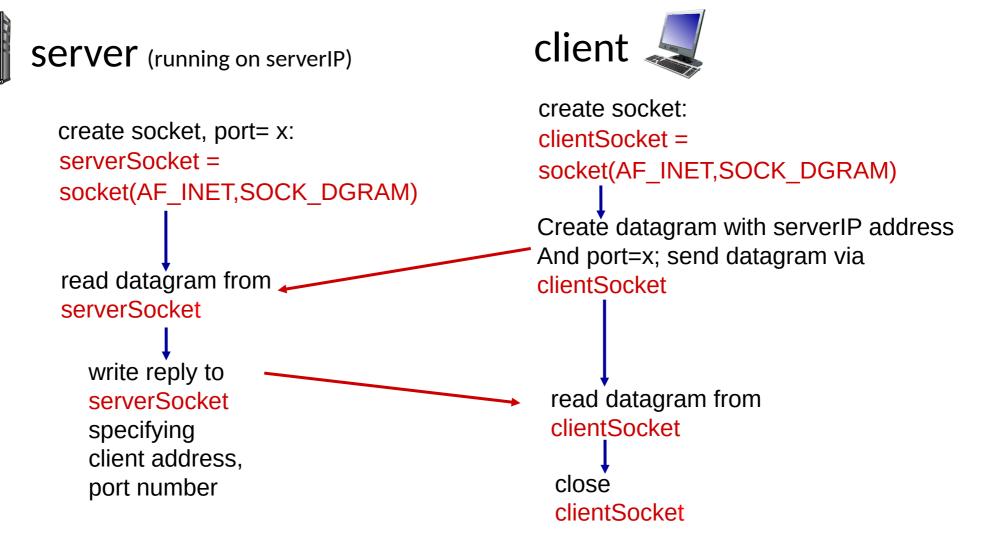
- no handshaking before sending data
- sender explicitly attaches IP destination address and port # to each packet
- receiver extracts sender IP address and port# from received packet

UDP: transmitted data may be lost or received out-of-order

Application viewpoint:

 UDP provides unreliable transfer of groups of bytes ("datagrams") between client and server processes

Client/server socket interaction: UDP



Socket programming with TCP

Client must contact server

- server process must first be running
- server must have created socket (door) that welcomes client's contact

Client contacts server by:

- Creating TCP socket, specifying IP address, port number of server process
- when client creates socket: client TCP establishes connection to server TCP

- when contacted by client, server TCP creates new socket for server process to communicate with that particular client
 - allows server to talk with multiple clients
 - client source port # and IP address used to distinguish clients (more in Chap 3)

Application viewpoint

TCP provides reliable, in-order byte-stream transfer ("pipe") between client and server processes

Client/server socket interaction: TCP

