
Computer Communication Networks

Application



UNIVERSITY
AT ALBANY
State University of New York

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Prof. Dola Saha

Problem

- Applications need their own protocols.
- These applications are part network protocol (in the sense that they exchange messages with their peers on other machines) and part traditional application program (in the sense that they interact with the windowing system, the file system, and ultimately, the user).
- We will explore some of the most popular network applications available today.

Outline

- Traditional Applications
- Multimedia Applications
- Infrastructure Services
- Overlay Networks

Some network apps

- e-mail
- web
- text messaging
- remote login
- P2P file sharing
- multi-user network games
- streaming stored video
(YouTube, Hulu, Netflix)
- voice over IP (e.g.,
Skype)
- real-time video
conferencing
- social networking
- search
- ...
- ...

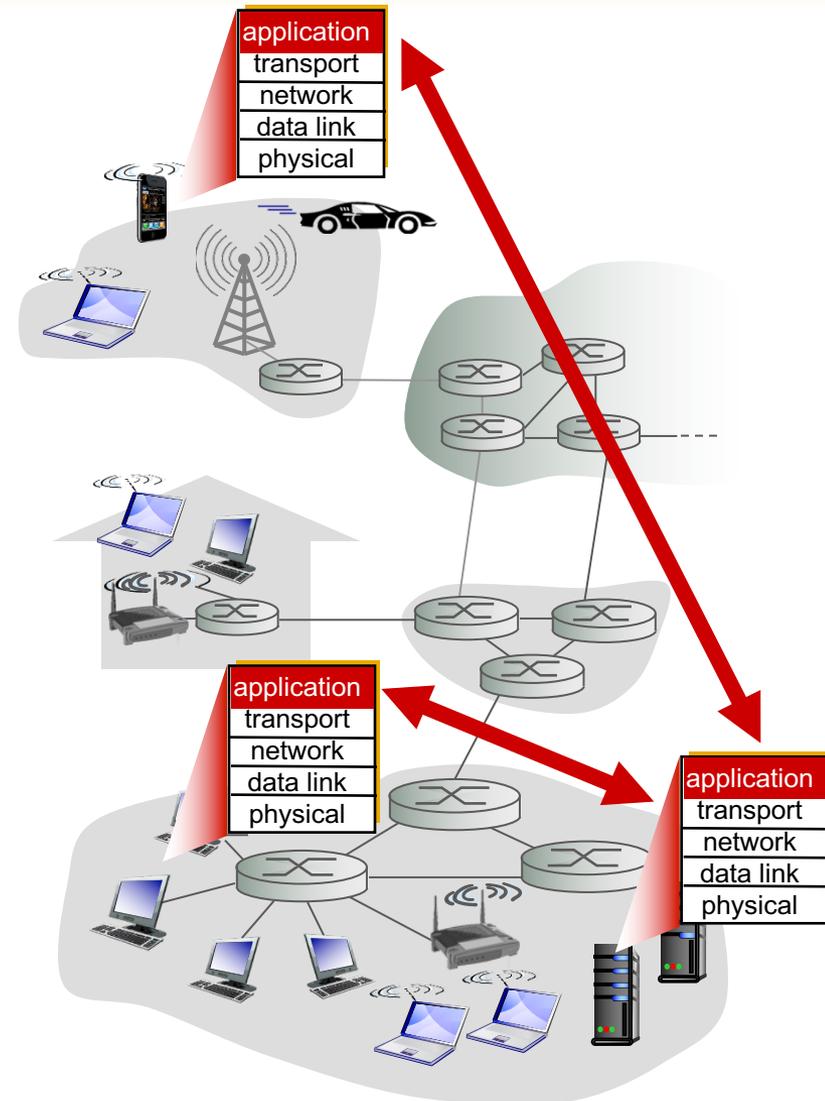
Creating a network app

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



Traditional Applications

- Two of the most popular—
 - The World Wide Web and
 - Email.
- Broadly speaking, both of these applications use the **request/reply paradigm**—users send requests to servers, which then respond accordingly.

Traditional Applications

- It is important to distinguish between *application programs* and *application protocols*.
- For example, *the HyperText Transport Protocol* (HTTP) is an application protocol that is used to retrieve Web pages from remote servers.
- There can be many different application programs—that is, Web clients like Internet Explorer, Chrome, Firefox, and Safari—that provide users with a different look and feel, but all of them use the same HTTP protocol to communicate with Web servers over the Internet.

Application

application	application program	application layer protocol	underlying transport protocol
e-mail	Outlook	SMTP [RFC 2821]	TCP
remote terminal access	Telnet	Telnet [RFC 854]	TCP
Web	Firefox	HTTP [RFC 2616]	TCP
file transfer	FileZilla	FTP [RFC 959]	TCP
streaming multimedia	YouTube	HTTP (e.g., YouTube),	TCP or UDP
	Cisco WebEx	RTP [RFC 1889]	
Internet telephony	Google voice	SIP, RTP, proprietary	
	Vonage	(e.g., Skype)	TCP or UDP

Traditional Applications

- Two very widely-used, standardized application protocols:
 - **HTTP**: HyperText Transport Protocol is used to communicate between Web browsers and Web servers.
 - RFC 2616 - <https://www.ietf.org/rfc/rfc2616.txt>
 - **SMTP**: Simple Mail Transfer Protocol is used to exchange electronic mail.
 - RFC 821 - <https://tools.ietf.org/rfc/rfc821.txt>

Traditional Applications

➤ World Wide Web

- The World Wide Web has been so successful and has made the Internet accessible to so many people that sometimes it seems to be synonymous with the Internet.
- In fact, the design of the system that became the Web started around 1989, long after the Internet had become a widely deployed system.
- The original goal of the Web was to find a way to organize and retrieve information, drawing on ideas about hypertext—interlinked documents—that had been around since at least the 1960s.

Traditional Applications

➤ World Wide Web

- The core idea of hypertext is that one document can link to another document, and the protocol (HTTP) and document language (HTML) were designed to meet that goal.
- One helpful way to think of the Web is as a set of cooperating clients and servers, all of whom speak the same language: HTTP.
- Most people are exposed to the Web through a graphical client program, or Web browser, like Safari, Chrome, Firefox or Internet Explorer.

Traditional Applications

➤ World Wide Web

- Clearly, if you want to organize information into a system of linked documents or objects, you need to be able to retrieve one document to get started.
- Hence, any Web browser has a function that allows the user to obtain an object by “opening a URL.”
- URLs (Uniform Resource Locators) are so familiar to most of us by now that it’s easy to forget that they haven’t been around forever.
- They provide information that allows objects on the Web to be located, and they look like the following:

- `http://www.cs.princeton.edu/index.html`

`www.someschool.edu/someDept/pic.gif`

host name

path name

Traditional Applications

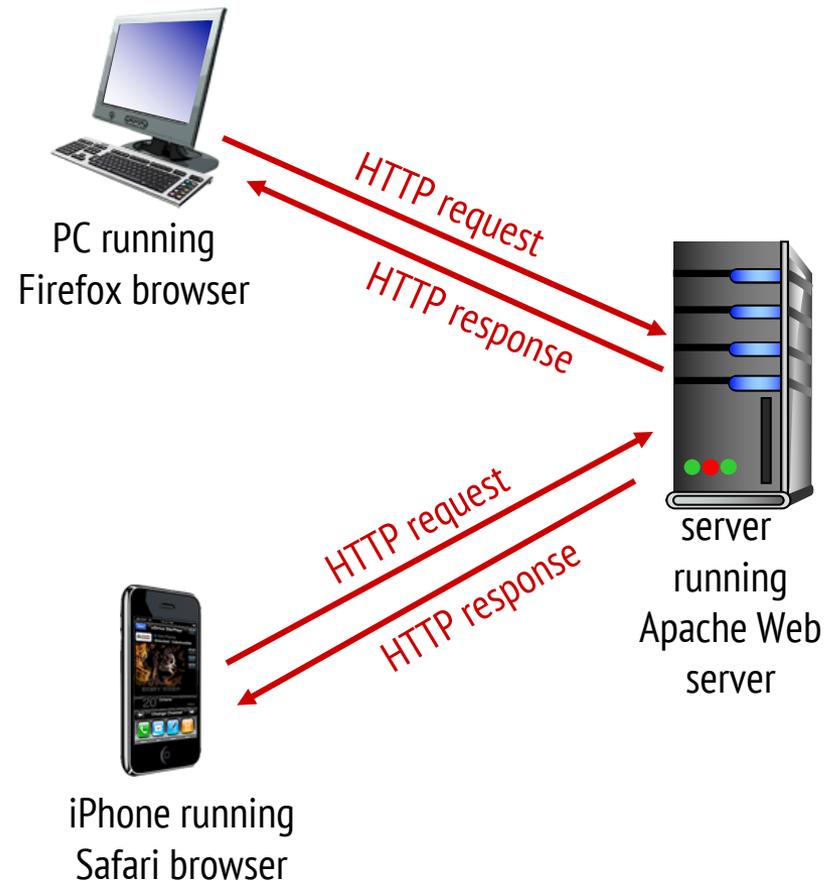
➤ World Wide Web

- If you opened that particular URL, your Web browser would open a TCP connection to the Web server at a machine called `www.cs.princeton.edu` and immediately retrieve and display the file called `index.html`.
- Most files on the Web contain images and text and many have other objects such as audio and video clips, pieces of code, etc.
- They also frequently include URLs that point to other files that may be located on other machines, which is the core of the “hypertext” part of HTTP and HTML.

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



Traditional Applications

➤ World Wide Web

- When you ask your browser to view a page, your browser (the client) fetches the page from the server using HTTP running over TCP.
- HTTP is a text oriented protocol.
- At its core, HTTP is a request/response protocol, where every message has the general form

START_LINE <CRLF>

MESSAGE_HEADER <CRLF>

<CRLF>

MESSAGE_BODY <CRLF>

- <CRLF> stands for carriage-return-line-feed.
- The first line (START LINE) indicates whether this is a request message or a response message.

HTTP overview (continued)

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

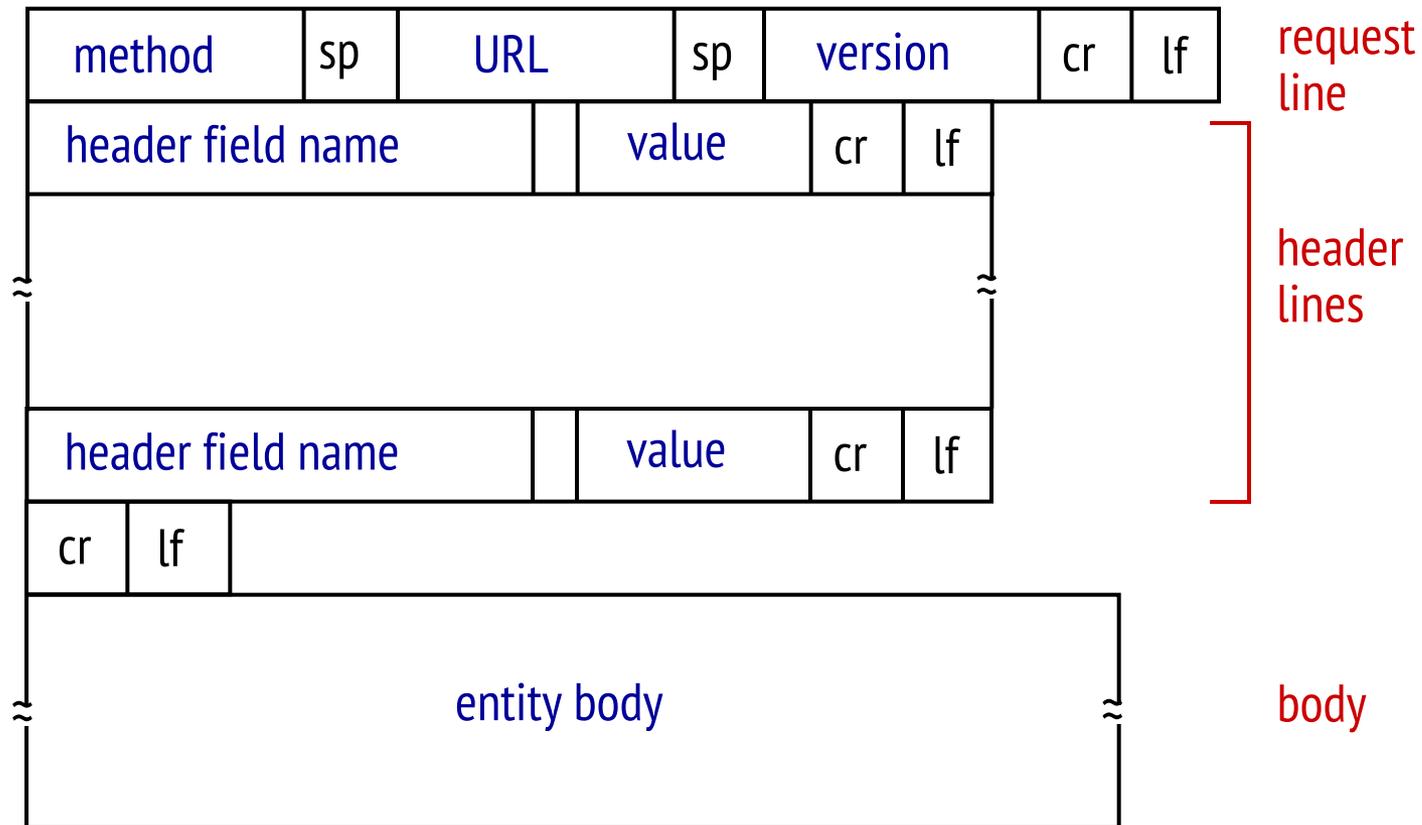
Traditional Applications

➤ World Wide Web

■ Request Messages

- The first line of an HTTP request message specifies three things:
 - ✓ the operation to be performed,
 - ✓ the Web page the operation should be performed on, and
 - ✓ the version of HTTP being used.
- Although HTTP defines a wide assortment of possible request operations—including “write” operations that allow a Web page to be posted on a server—the two most common operations are GET (fetch the specified Web page) and HEAD (fetch status information about the specified Web page).

HTTP Request Message



Traditional Applications

- World Wide Web
 - Request Messages

Operation	Description
OPTIONS	Request information about available options
GET	Retrieve document identified in URL
HEAD	Retrieve metainformation about document identified in URL
POST	Give information (e.g., annotation) to server
PUT	Store document under specified URL
DELETE	Delete specified URL
TRACE	Loopback request message
CONNECT	For use by proxies

HTTP request operations

Traditional Applications

➤ World Wide Web

■ Response Messages

- Like request messages, response messages begin with a single START LINE.
- In this case, the line specifies the version of HTTP being used, a three-digit code indicating whether or not the request was successful, and a text string giving the reason for the response.

HTTP Result Codes

- Response Messages

Code	Type	Example Reasons
1xx	Informational	request received, continuing process
2xx	Success	action successfully received, understood, and accepted
3xx	Redirection	further action must be taken to complete the request
4xx	Client Error	request contains bad syntax or cannot be fulfilled
5xx	Server Error	server failed to fulfill an apparently valid request

Five types of HTTP result codes

Uniform Resource Identifiers

- The URLs that HTTP uses as addresses are one type of *Uniform Resource Identifier (URI)*.
- A URI is a character string that identifies a resource, where a resource can be anything that has identity, such as a document, an image, or a service.
- The format of URIs allows various more-specialized kinds of resource identifiers to be incorporated into the URI space of identifiers.
- The first part of a URI is a *scheme* that names a particular way of identifying a certain kind of resource, such as mailto for email addresses or file for file names.
- The second part of a URI, separated from the first part by a colon, is the *scheme-specific part*.

Traditional Applications

➤ World Wide Web

■ TCP Connections

- The original version of HTTP (1.0) established a separate TCP connection for each data item retrieved from the server.
- It's not too hard to see how this was a very inefficient mechanism: connection setup and teardown messages had to be exchanged between the client and server even if all the client wanted to do was verify that it had the most recent copy of a page.
- Thus, retrieving a page that included some text and a dozen icons or other small graphics would result in 13 separate TCP connections being established and closed.

Traditional Applications

➤ World Wide Web

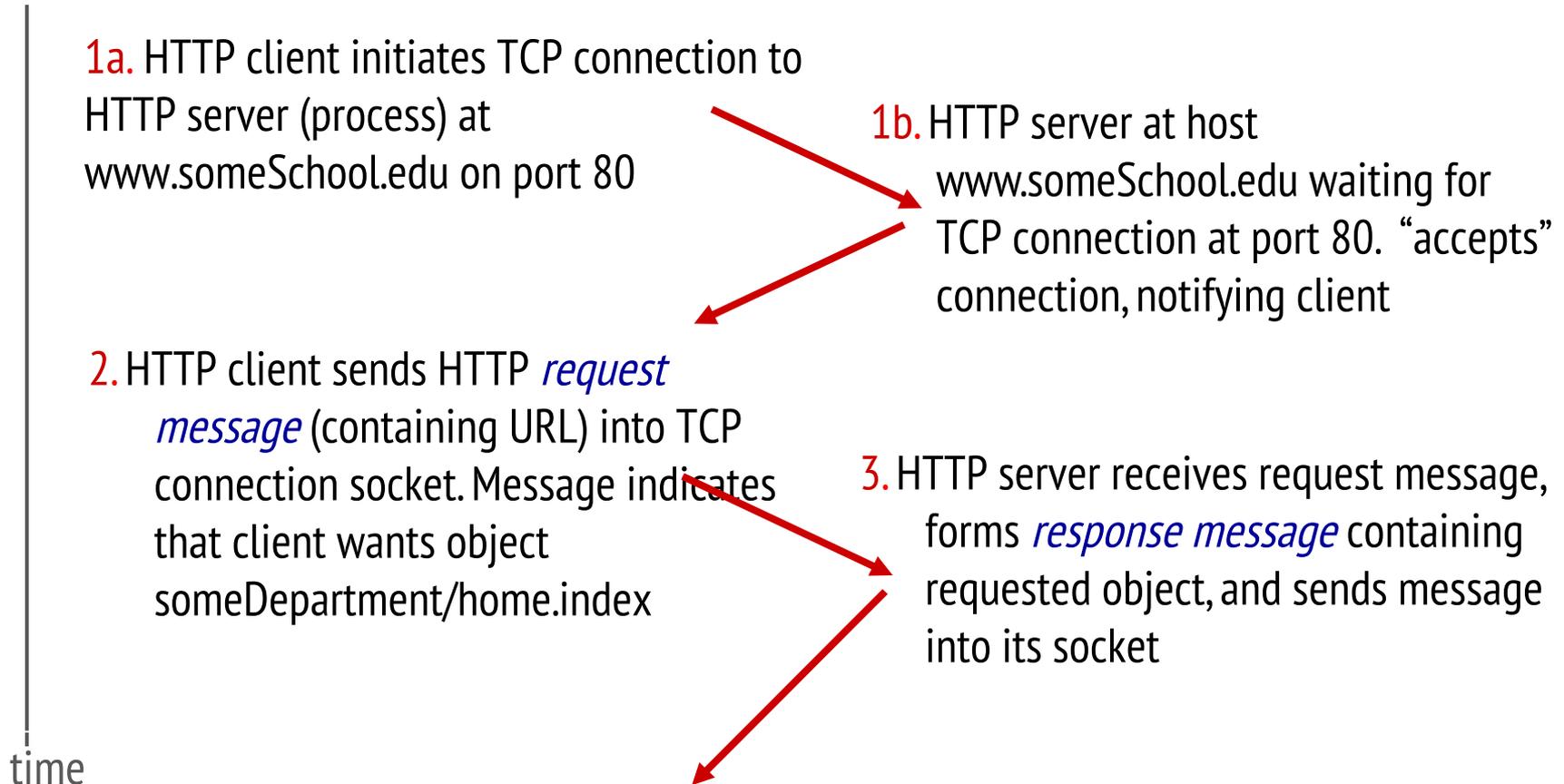
■ TCP Connections

- To overcome this situation, HTTP version 1.1 introduced *persistent connections*—the client and server can exchange multiple request/response messages over the same TCP connection.
- Persistent connections have many advantages.
 - ✓ First, they obviously eliminate the connection setup overhead, thereby reducing the load on the server, the load on the network caused by the additional TCP packets, and the delay perceived by the user.
 - ✓ Second, because a client can send multiple request messages down a single TCP connection, TCP's congestion window mechanism is able to operate more efficiently.
 - This is because it's not necessary to go through the slow start phase for each page.

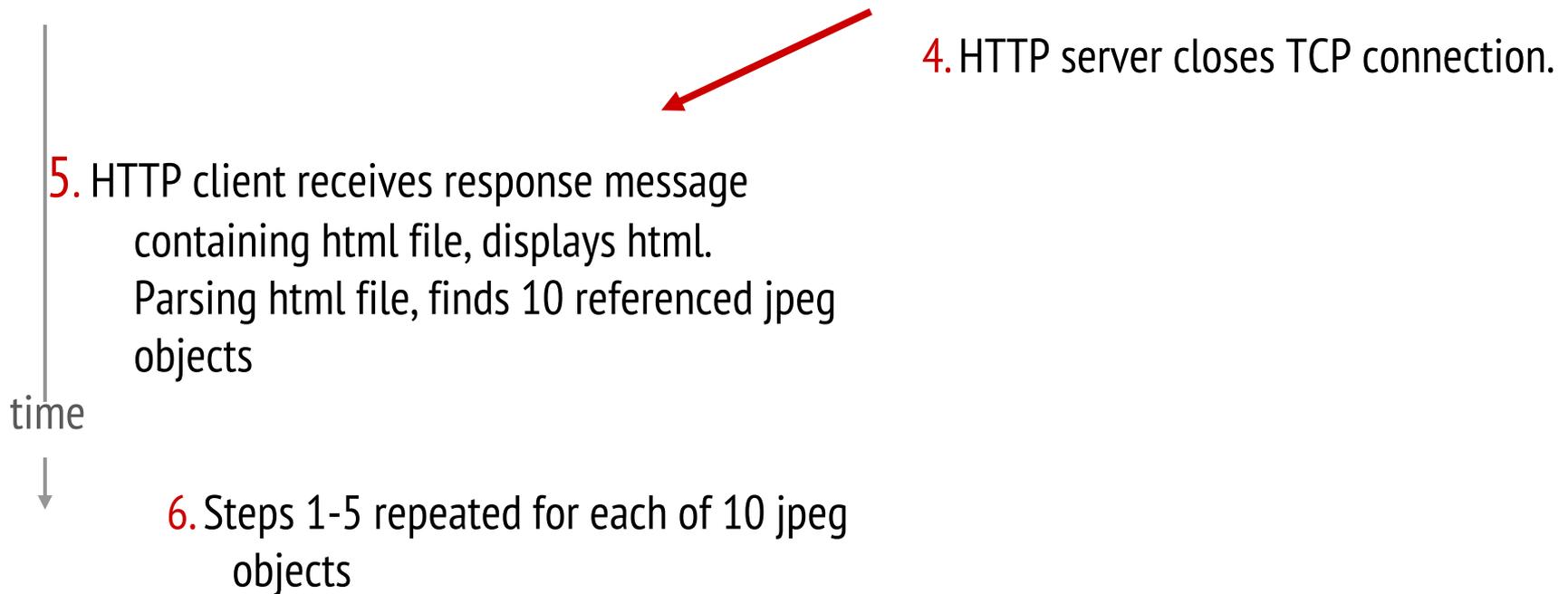
Non-persistent HTTP

suppose user enters URL:
`www.someSchool.edu/someDepartment/home.index`

(contains text,
references to 10
jpeg images)



Non-persistent HTTP (cont.)

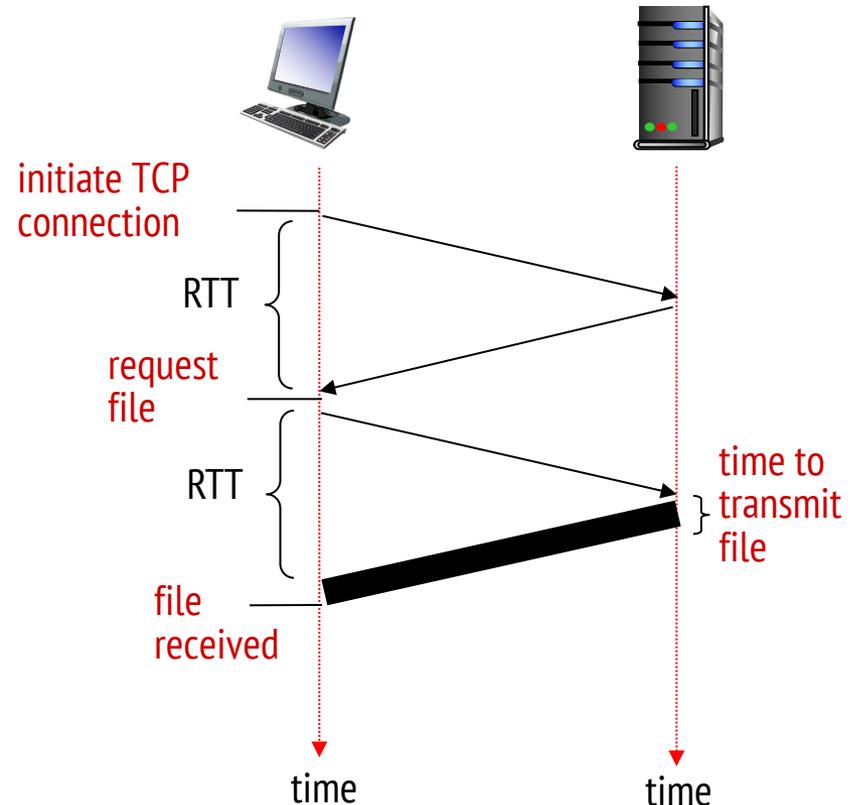


Non-persistent HTTP: response time

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

HTTP response time:

- one RTT to initiate TCP connection
- one RTT for HTTP request and first few bytes of HTTP response to return
- file transmission time
- non-persistent HTTP response time =
2RTT+ file transmission time



Persistent and non-persistent HTTP

non-persistent HTTP issues:

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

persistent HTTP:

- 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

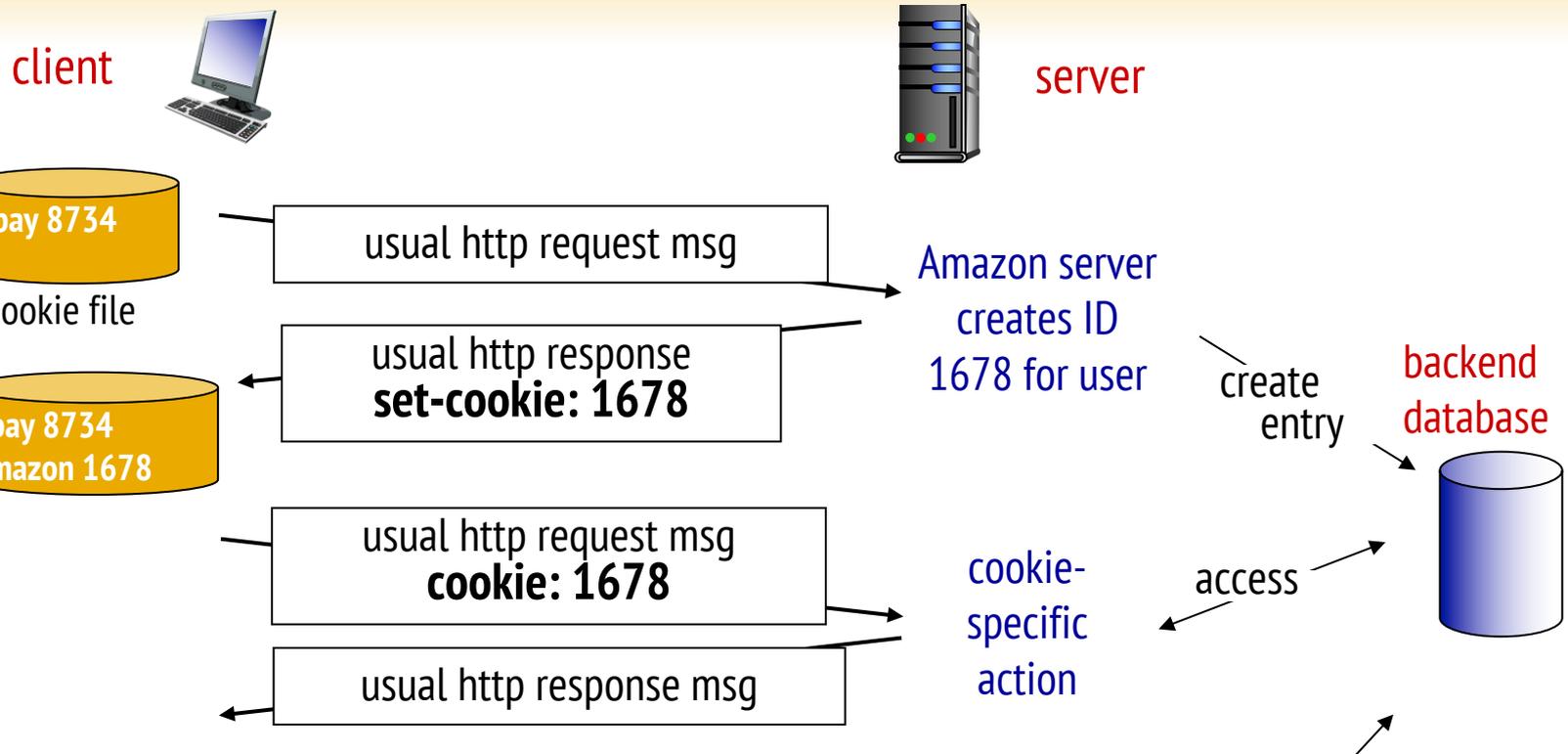
Caching

- One of the most active areas of research (and entrepreneurship) in the Internet today is how to effectively cache Web pages.
- Caching has benefits.
 - From the client's perspective, a page that can be retrieved from a nearby cache can be displayed much more quickly than if it has to be fetched from across the world.
 - From the server's perspective, having a cache intercept and satisfy a request reduces the load on the server.

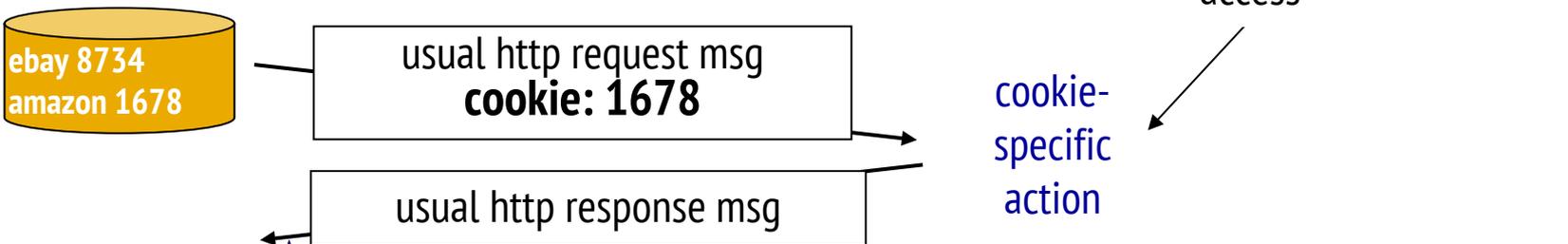
Caching

- Caching can be implemented in many different places.
 - a user's browser can cache recently accessed pages, and simply display the cached copy if the user visits the same page again.
 - a site can support a single site-wide cache.
- This allows users to take advantage of pages previously downloaded by other users.
- Closer to the middle of the Internet, ISPs can cache pages.
- Note that in the second case, the users within the site most likely know what machine is caching pages on behalf of the site, and they configure their browsers to connect directly to the caching host. This node is sometimes called a *proxy*

Cookies: keeping "state" (cont.)



one week later:



Cookies (continued)

what cookies can be used for:

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

aside

cookies and privacy:

- cookies permit sites to learn a lot about you
- you may supply name and e-mail to sites

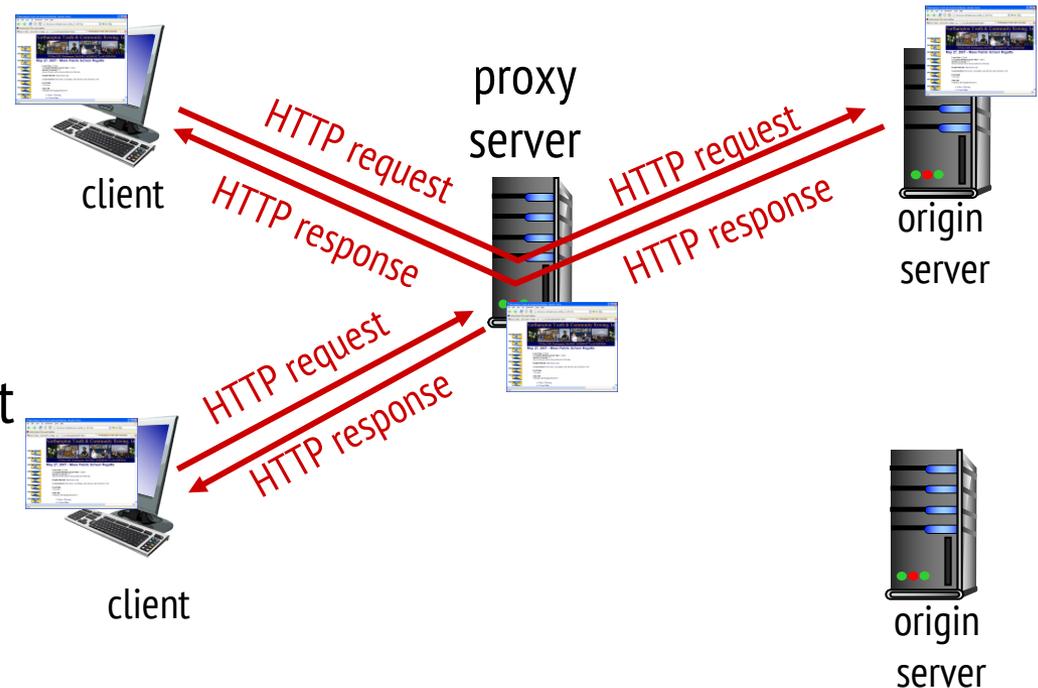
how to keep “state”:

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

Web caches (proxy server)

goal: satisfy client request without involving origin server

- user sets browser: Web accesses via cache
- browser sends all HTTP requests to cache
 - object in cache: cache returns object
 - else cache requests object from origin server, then returns object to client



More about Web caching

- cache acts as both client and server
 - server for original requesting client
 - client to origin server
- typically cache is installed by ISP (university, company, residential ISP)

why Web caching?

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

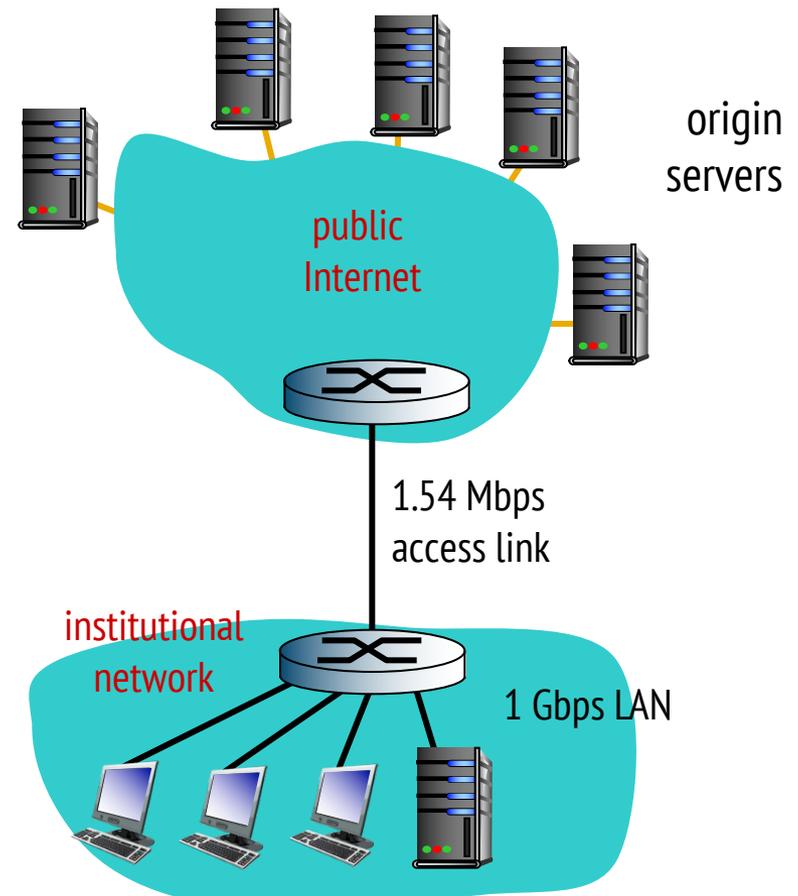
Caching example:

assumptions:

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

consequences:

- LAN utilization: 15%
- access link utilization = 99% *problem!*
- total delay = Internet delay + access delay + LAN delay
= 2 sec + minutes + usecs



Caching example: fatter access link

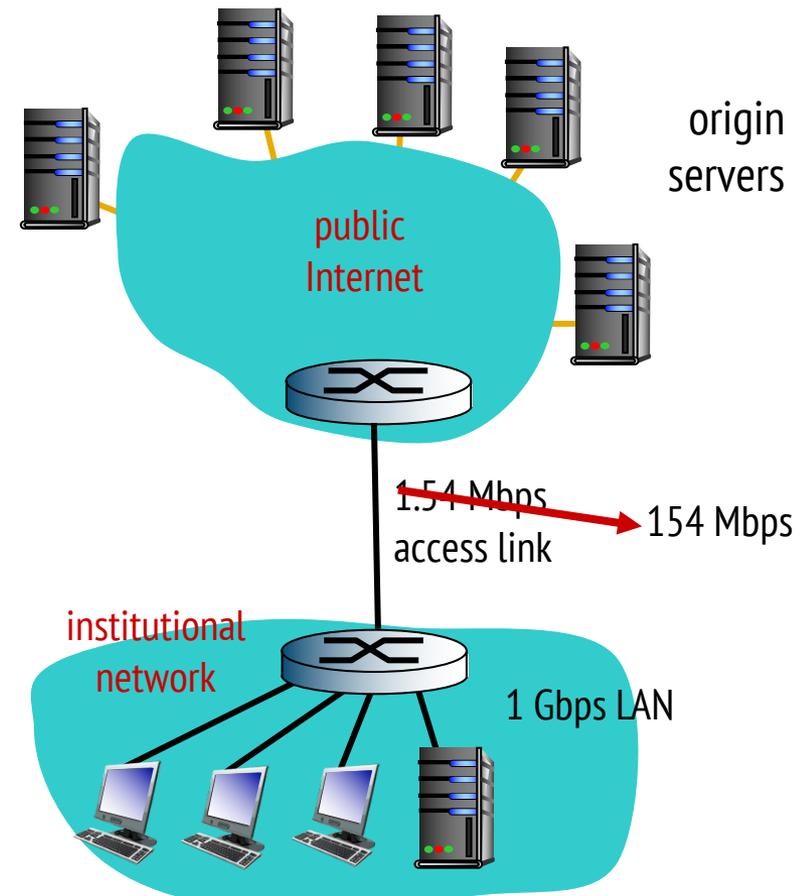
assumptions:

- avg object size: 100K bits
- avg request rate from browsers to origin servers: 15/sec
- avg data rate to browsers: 1.50 Mbps
- RTT from institutional router to any origin server: 2 sec
- access link rate: ~~1.54 Mbps~~ → 154 Mbps

consequences:

- LAN utilization: 15%
- access link utilization = ~~99%~~ → 9.9%
- total delay = Internet delay + access delay + LAN delay
- = ~~2 sec + minutes + usecs~~ → msec

Cost: increased access link speed (not cheap!)



Caching example: install local cache

assumptions:

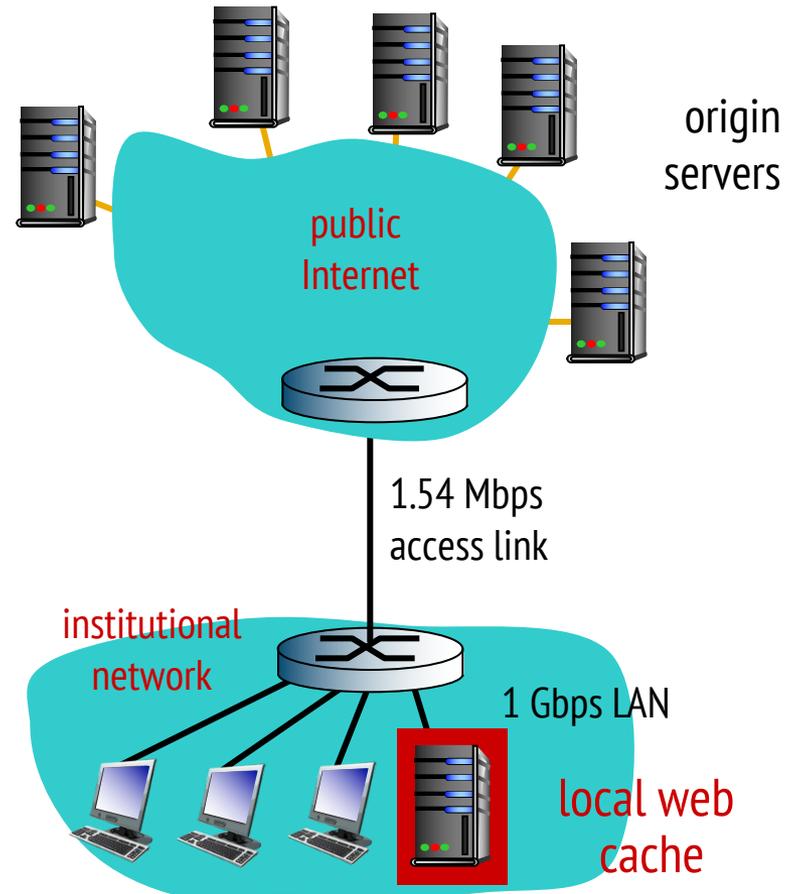
- avg object size: 100K bits
- avg request rate from browsers to origin servers: 15/sec
- avg data rate to browsers: 1.50 Mbps
- RTT from institutional router to any origin server: 2 sec
- access link rate: 1.54 Mbps

consequences:

- LAN utilization: 15%
- access link utilization = ?
- total delay = Internet delay + access delay + LAN delay = ?

How to compute link utilization, delay?

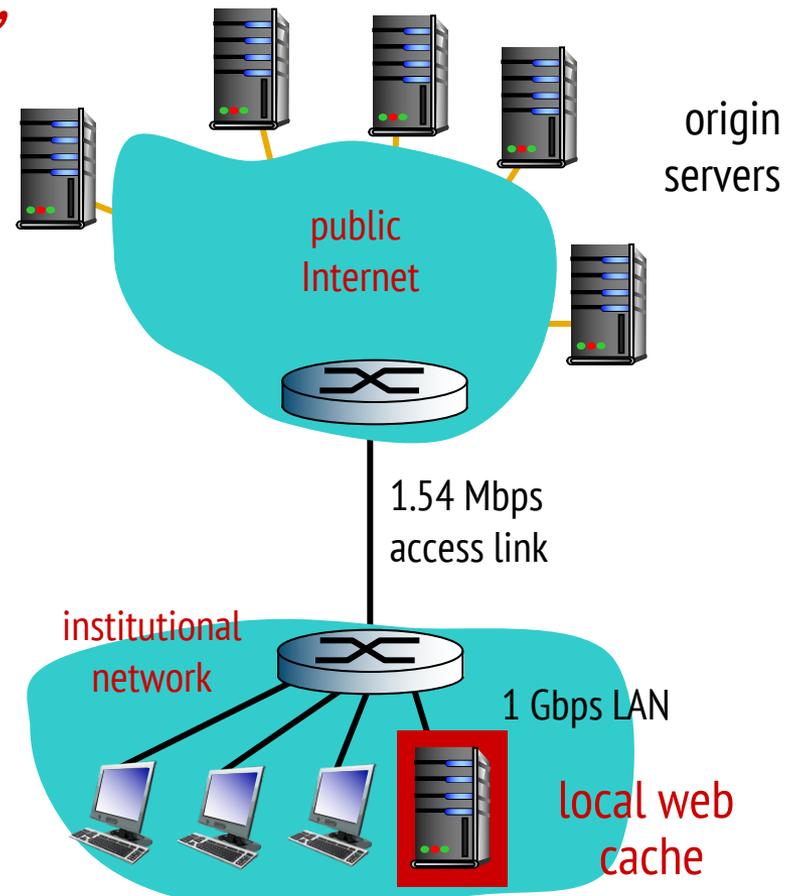
Cost: web cache (cheap!)



Caching example: install local cache

Calculating access link utilization, delay with cache:

- **suppose cache hit rate is 0.4**
 - 40% requests satisfied at cache, 60% requests satisfied at origin
- **access link utilization:**
 - 60% of requests use access link
- **data rate to browsers over access link**
 $= 0.6 * 1.50 \text{ Mbps} = .9 \text{ Mbps}$
 - utilization = $0.9 / 1.54 = .58$
- **total delay**
 - = $0.6 * (\text{delay from origin servers}) + 0.4 * (\text{delay when satisfied at cache})$
 - = $0.6 (2.01) + 0.4 (\sim \text{msecs}) = \sim 1.2 \text{ secs}$
 - less than with 154 Mbps link (and cheaper too!)

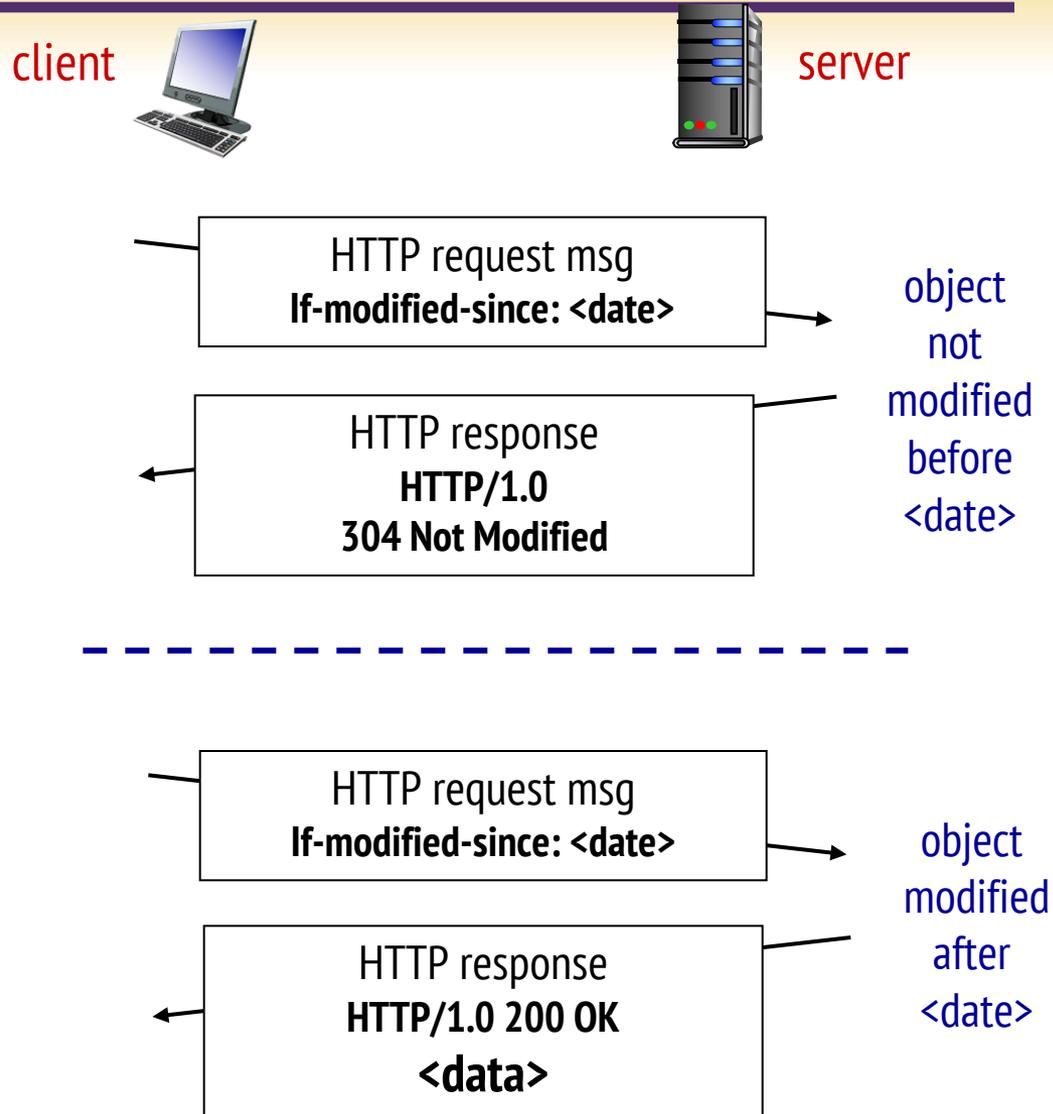


Conditional GET

- *Goal:* don't send object if cache has up-to-date cached version
 - no object transmission delay
 - lower link utilization
- *cache:* specify date of cached copy in HTTP request

If-modified-since: <date>

- *server:* response contains no object if cached copy is up-to-date:
HTTP/1.0 304 Not Modified



Electronic Mail

- SMTP, MIME, IMAP
 - Email is one of the oldest network applications
 - It is important
 - (1) to distinguish the user interface (i.e., your mail reader) from the underlying message transfer protocols (such as SMTP or IMAP), and
 - (2) to distinguish between this transfer protocol and a companion protocol (RFC 822 and MIME) that defines the format of the messages being exchanged

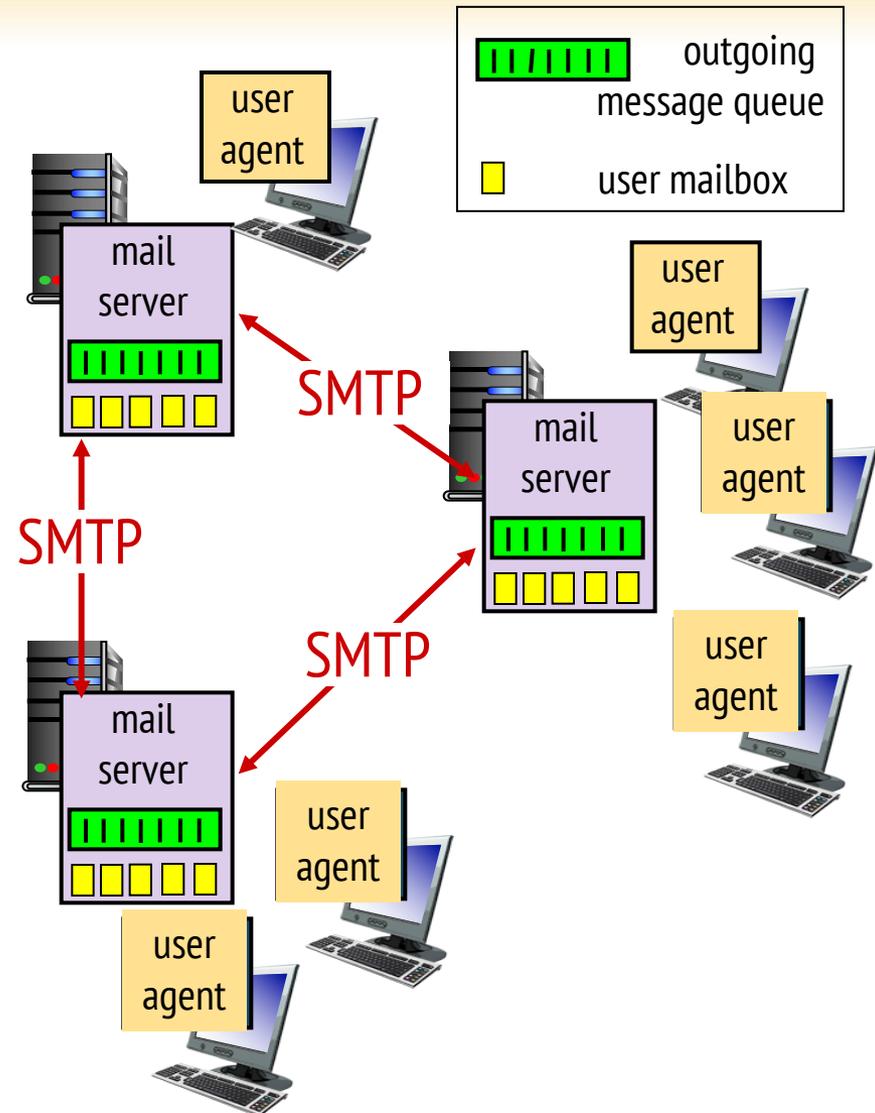
Electronic Mail

Three major components:

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

User Agent

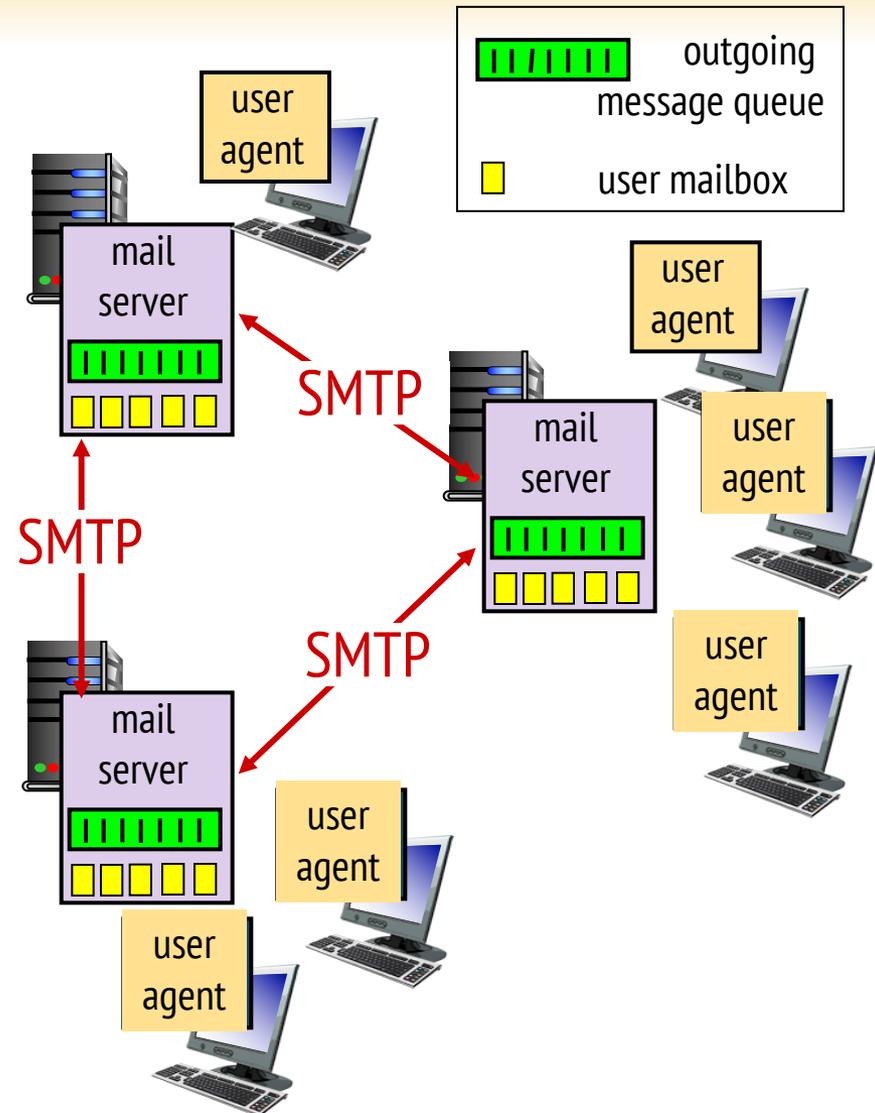
- a.k.a. “mail reader”
- composing, editing, reading mail messages
- e.g., Outlook, Thunderbird, iPhone mail client
- outgoing, incoming messages stored on server



Electronic 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

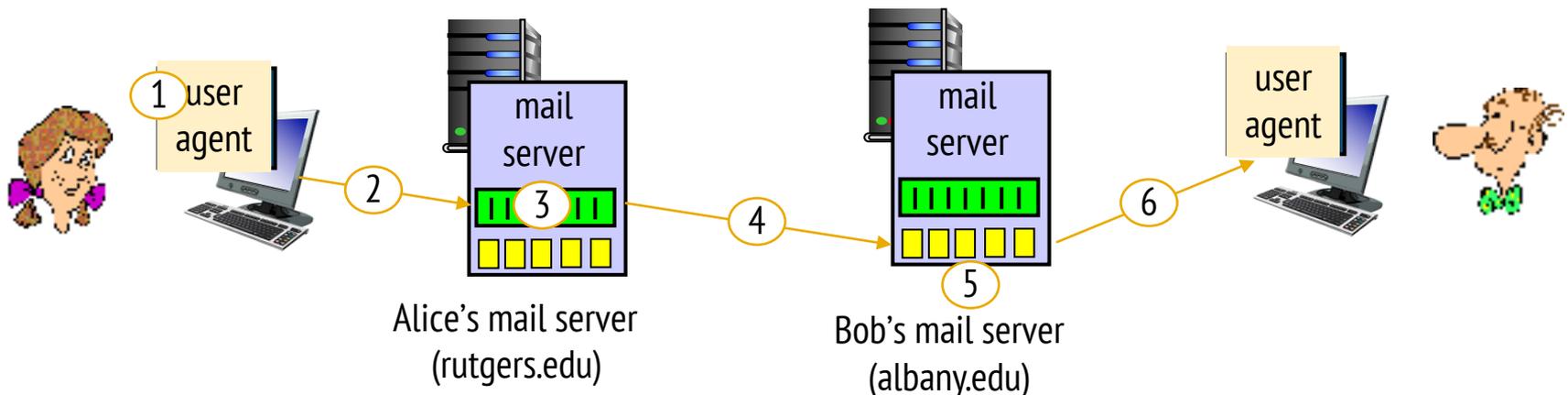


Electronic Mail: SMTP [RFC 2821]

- uses TCP to reliably transfer email message from client to server, port 25
- direct transfer: sending server to receiving server
- three phases of transfer
 - handshaking (greeting)
 - transfer of messages
 - closure
- command/response interaction (like HTTP)
 - **commands:** ASCII text
 - **response:** status code and phrase
- messages must be in 7-bit ASCII

Scenario: Alice sends message to Bob

- 1) Alice uses UA to compose message “to” bob@some.edu
- 2) Alice’s UA sends message to her mail server; message placed in message queue
- 3) client side of SMTP 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



SMTP: final words

- 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

comparison with HTTP:

- HTTP: pull
- SMTP: 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

Mail message format

SMTP: protocol for exchanging email messages

RFC 822: standard for text message format:

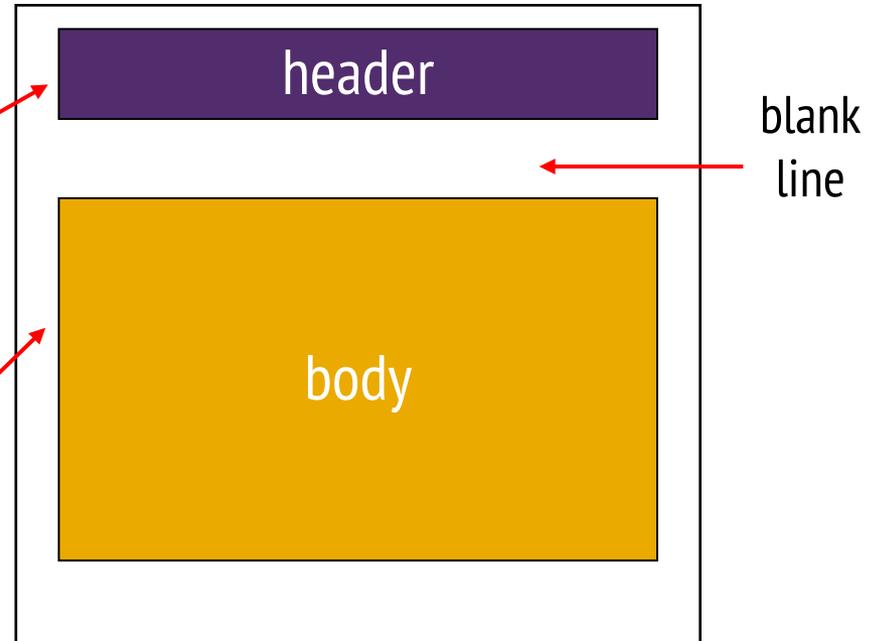
➤ header lines, e.g.,

- To:
- From:
- Subject:

different from SMTP MAIL FROM, RCPT TO: commands!

➤ Body: the “message”

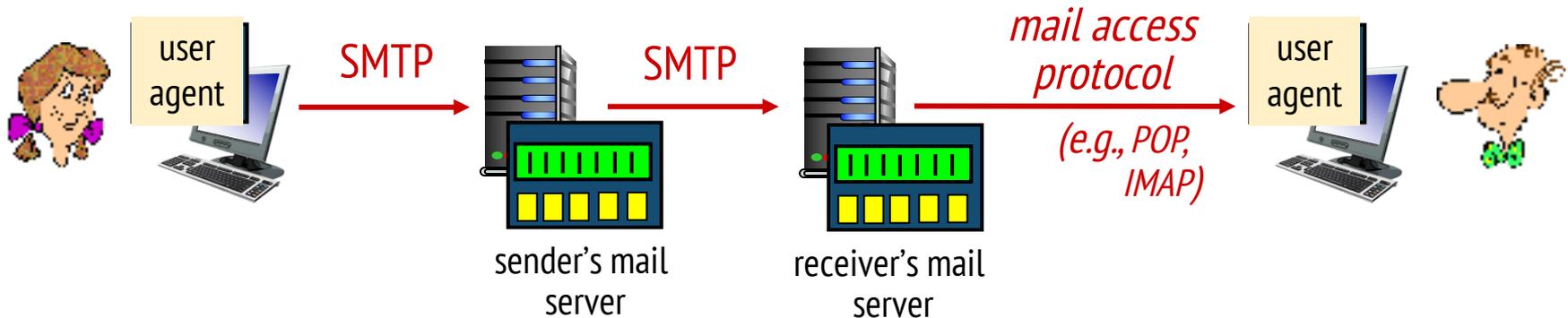
- ASCII characters only



➤ **MIME – Multipurpose Internet Mail Extensions**

- Supports non-text attachments

Mail access protocols



- **SMTP:** delivery/storage to receiver's server
- mail access protocol: retrieval from server
 - **POP:** Post Office Protocol [RFC 1939]: authorization, download
 - **IMAP:** Internet Mail Access Protocol [RFC 1730]: more features, including manipulation of stored messages on server
 - **HTTP:** gmail, Hotmail, Yahoo! Mail, etc.

POP3 protocol

authorization phase

- client commands:
 - **user**: declare username
 - **pass**: password
- server responses
 - **+OK**
 - **-ERR**

```
S: +OK POP3 server ready
C: user bob
S: +OK
C: pass hungry
S: +OK user successfully logged on
```

transaction phase, client:

- **list**: list message numbers
- **retr**: retrieve message by number
- **dele**: delete
- **quit**

```
C: list
S: 1 498
S: 2 912
S: .
C: retr 1
S: <message 1 contents>
S: .
C: dele 1
C: retr 2
S: <message 1 contents>
S: .
C: dele 2
C: quit
S: +OK POP3 server signing off
```

POP3 (more) and IMAP

more about POP3

- previous example uses POP3 “download and delete” mode
 - Bob cannot re-read e-mail if he changes client
- POP3 “download-and-keep”:
copies of messages on different clients
- POP3 is stateless across sessions

IMAP

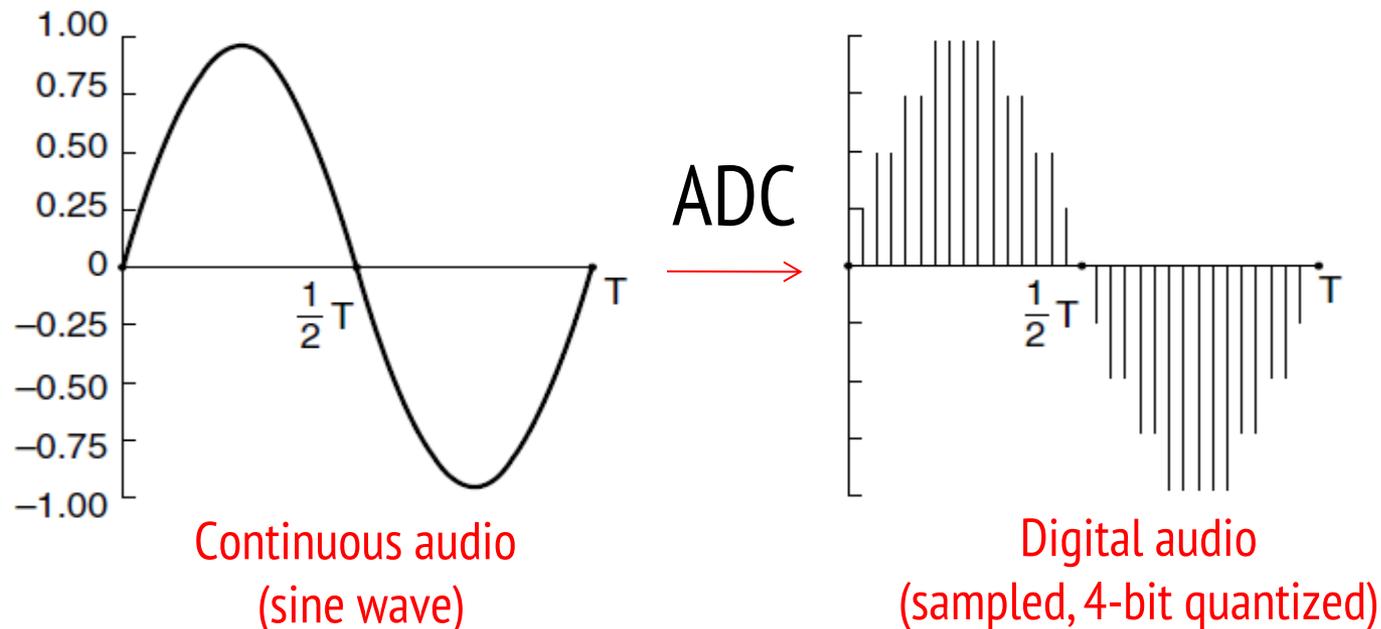
- keeps all messages in one place: at server
- allows user to organize messages in folders
- keeps user state across sessions:
- names of folders and mappings between message IDs and folder name

Multimedia Applications

- Audio and video have become key types of traffic, e.g., voice over IP, and video streaming.
 - Digital audio
 - Digital video
 - Streaming stored media
 - Streaming live media
 - Real-time conferencing

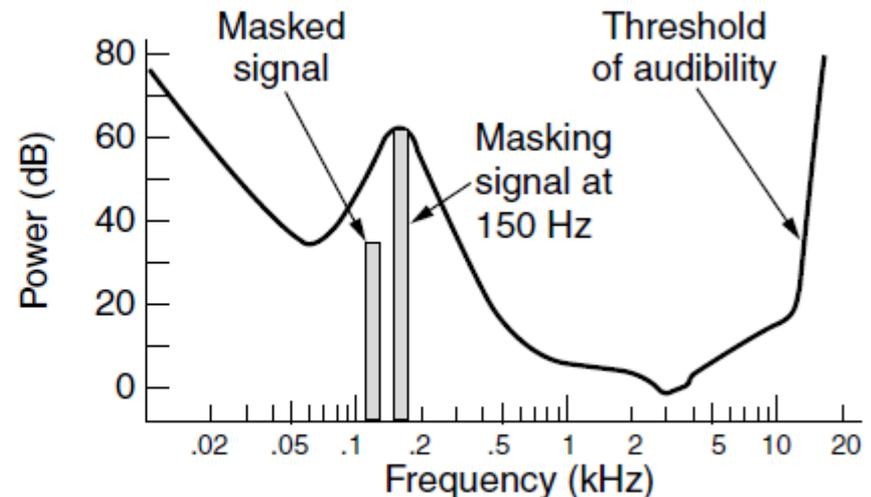
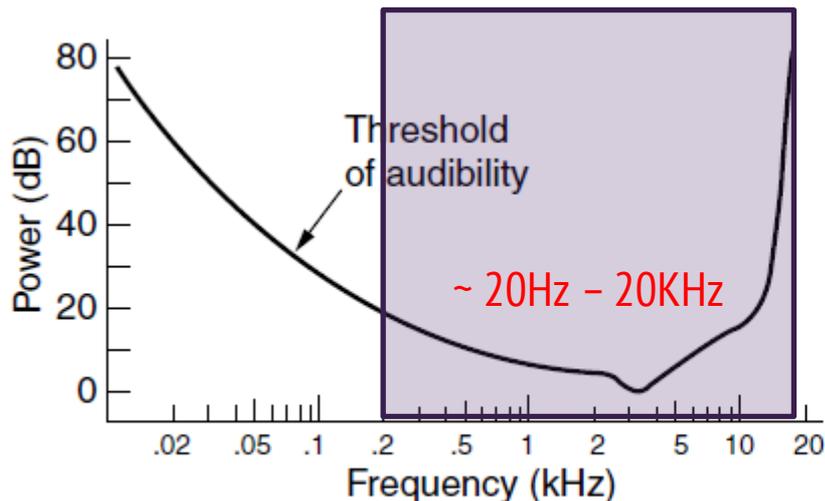
Digital Audio (1)

- ADC (Analog-to-Digital Converter) produces digital audio from a microphone
 - Telephone: 8000 8-bit samples/second (64 Kbps); computer audio is usually better quality (e.g., 16 bit)



Digital Audio (2)

- Digital audio is typically compressed before it is sent
 - Lossy encoders (like Advanced Audio Coding - AAC) exploit human perception
 - It is not necessary to encode signals whose power falls below Threshold of audibility.
 - Large compression ratios (can be >10X)



Sensitivity of the ear varies with frequency



A loud tone can mask nearby tones

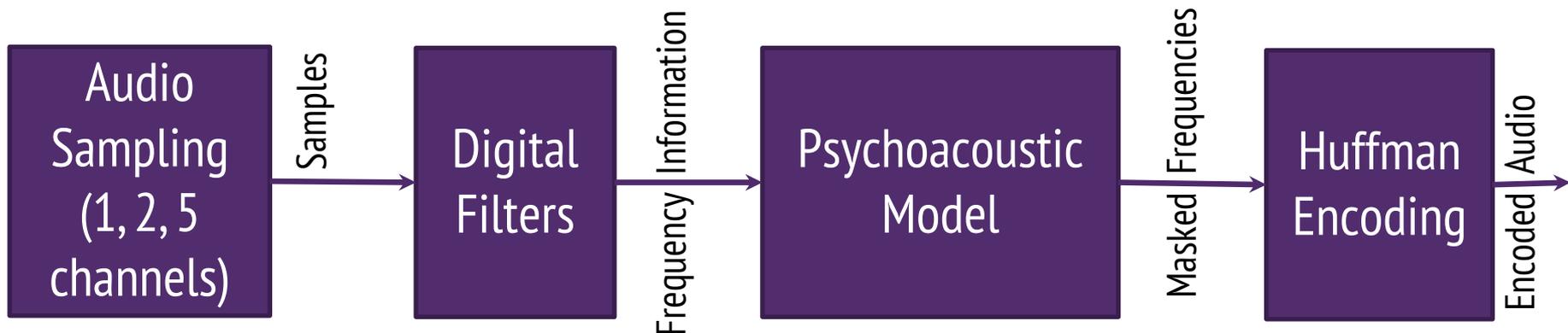
Digital Audio (3)

➤ Perceptual Coding

- Based on science of psychoacoustics (way humans perceive sound)
- Frequency Masking
- Temporal Masking
- Examples: MP3, AAC

➤ Huffman Encoding

- Assigns short codes to numbers that appear frequently

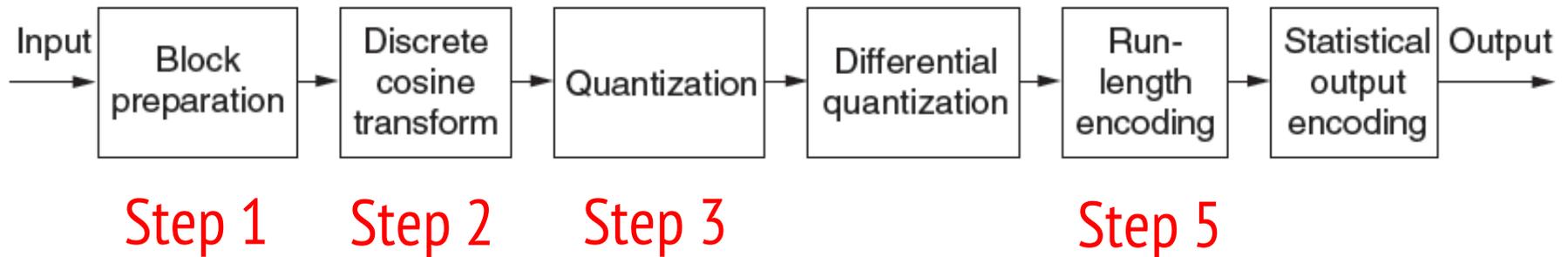


Digital Video (1)

- Video is digitized as pixels (sampled, quantized)
 - SD TV quality: 640x480 pixels, 24-bit color, 30 times/sec
 - HD TV quality: 1280x720 pixels
 - Player refreshes 50-100 times to minimize flicker
- Video is sent compressed due to its large bandwidth
 - Lossy compression exploits human perception
 - E.g., JPEG for still images, MPEG, H.264 for video
 - Large compression ratios (often 50X for video)
 - Video is normally > 1 Mbps, versus >10 kbps for speech and >100 kbps for music

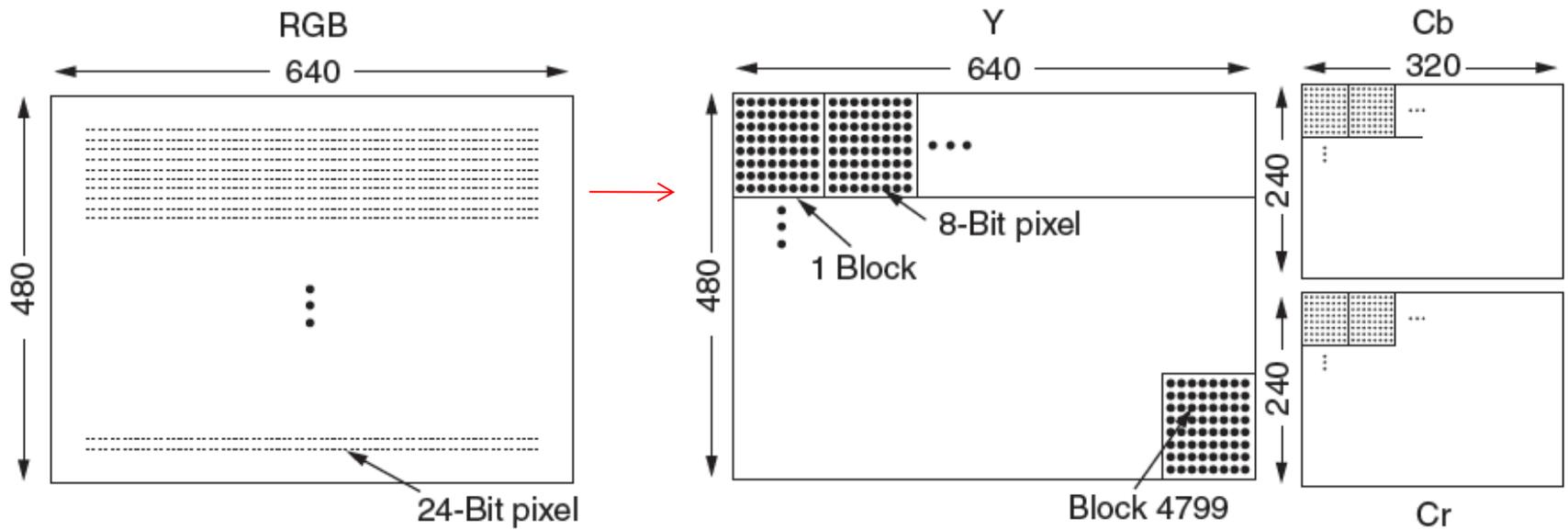
Digital Video (2)

- JPEG lossy compression sequence for one image:



Digital Video (3)

- Step 1: Pixels are mapped to luminance (brightness)/chrominance (YCbCr) color space
 - Luma signal (Y), Chroma signal: 2 components (Cb and Cr)
 - Chrominance is sub-sampled, the eye is less sensitive to chrominance



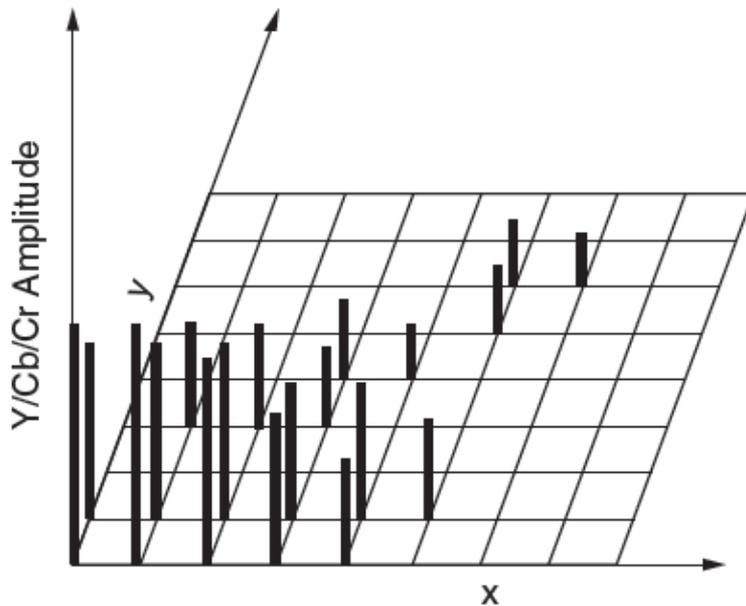
Input 24-bit RGB pixels

8-bit luminance
pixels

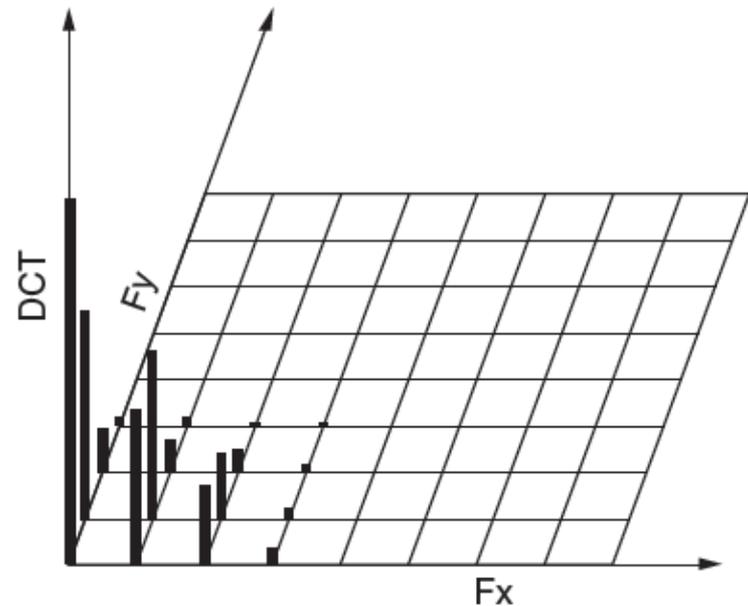
8-bit chrominances
for every 4 pixels

Digital Video (4)

- Step 2: Each component block is transformed to spatial frequencies with DCT (Discrete Cosine Transformation)
 - Captures the key image features



One component block



Transformed block



Digital Video (5)

- Step 3: DCT coefficients are quantized by dividing by thresholds; reduces bits in higher spatial frequencies
 - Top left element is differenced over blocks (Step 4)

DCT coefficients

150	80	40	14	4	2	1	0
92	75	36	10	6	1	0	0
52	38	26	8	7	4	0	0
12	8	6	4	2	1	0	0
4	3	2	0	0	0	0	0
2	2	1	1	0	0	0	0
1	1	0	0	0	0	0	0
0	0	0	0	0	0	0	0

Quantization table

1	1	2	4	8	16	32	64
1	1	2	4	8	16	32	64
2	2	2	4	8	16	32	64
4	4	4	4	8	16	32	64
8	8	8	8	8	16	32	64
16	16	16	16	16	16	32	64
32	32	32	32	32	32	32	64
64	64	64	64	64	64	64	64

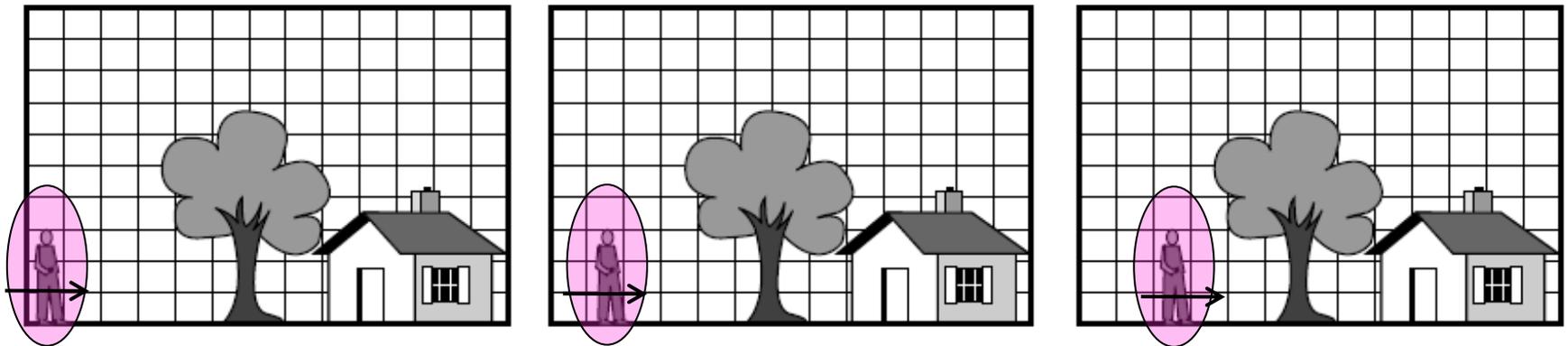
Quantized coefficients

150	80	20	4	1	0	0	0
92	75	18	3	1	0	0	0
26	19	13	2	1	0	0	0
3	2	2	1	0	0	0	0
1	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0
0	0	0	0	0	0	0	0

Input / Thresholds = Output

Digital Video (7)

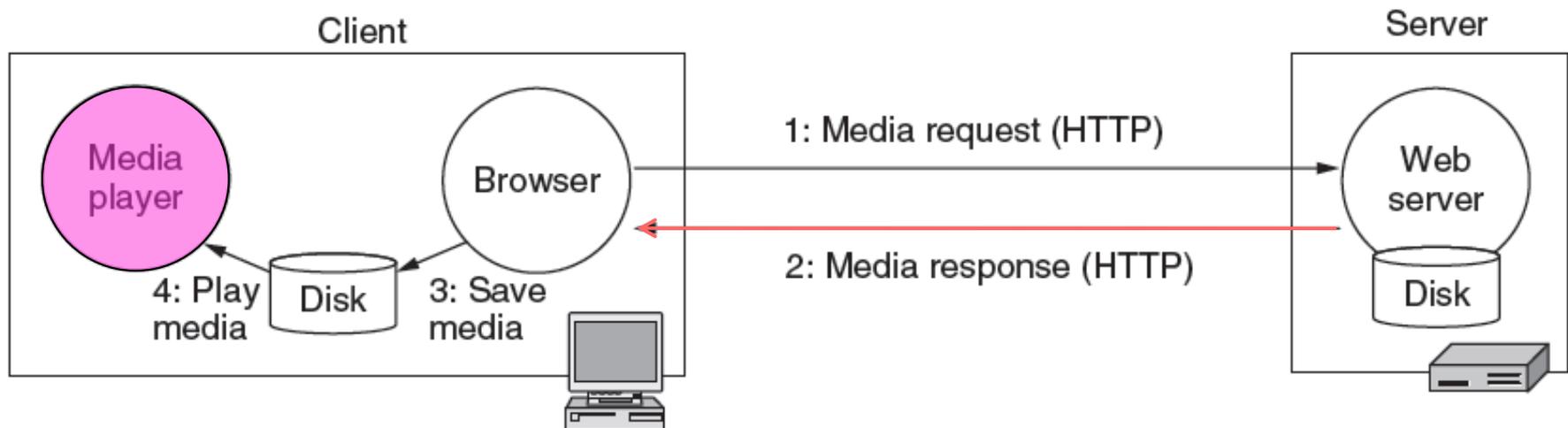
- MPEG compresses over a sequence of frames, further using motion tracking to remove temporal redundancy
 - MPEG Output
 - I (Intra-coded) frames are self-contained
 - P (Predictive) frames use block motion predictions
 - B (Bidirectional) frames may base prediction on future frame



Three consecutive frames with stationary and moving components

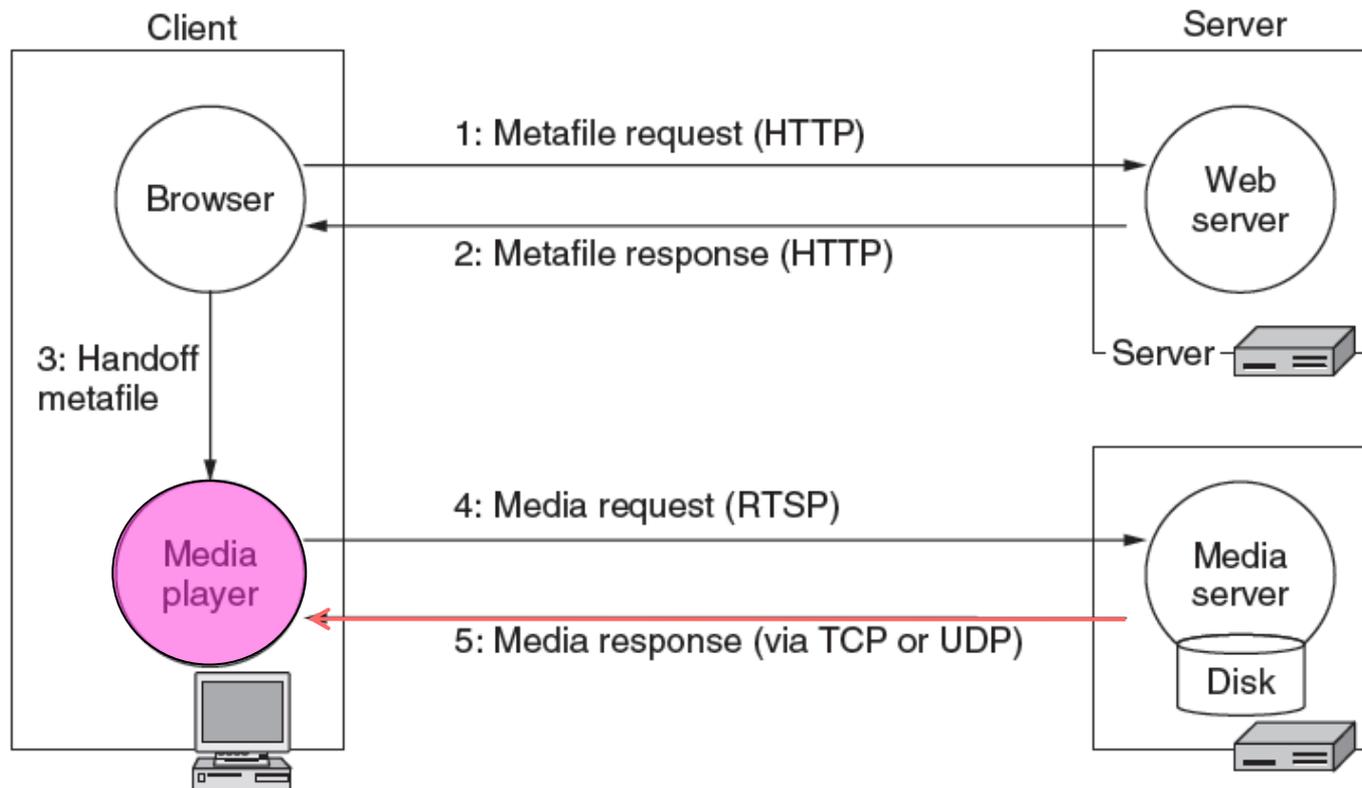
Streaming Stored Media (1)

- A simple method to stream stored media, e.g., for video on demand, is to fetch the video as a file download
 - But has large startup delay, except for short files



Streaming Stored Media (2)

- Effective streaming starts the playout during transport
 - With RTSP (Real-Time Streaming Protocol)



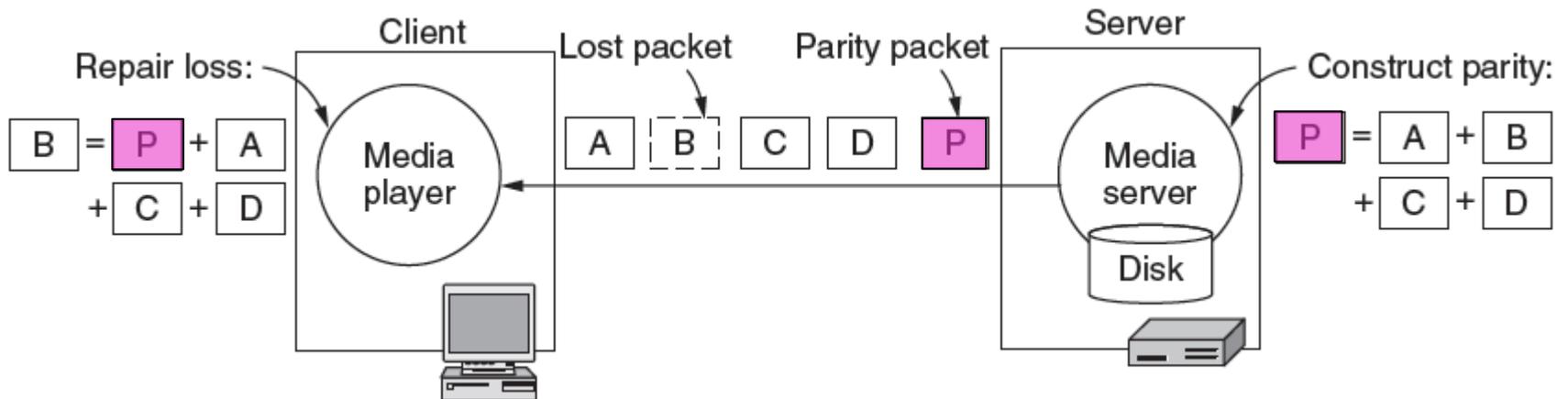
Streaming Stored Media (3)

- Key problem: how to handle transmission errors

Strategy	Advantage	Disadvantage
Use reliable transport (TCP)	Repairs all errors	Increases jitter significantly
Add FEC (e.g., parity)	Repairs most errors	Increases overhead, decoding complexity and jitter
Interleave media	Masks most errors	Slightly increases decoding complexity and jitter

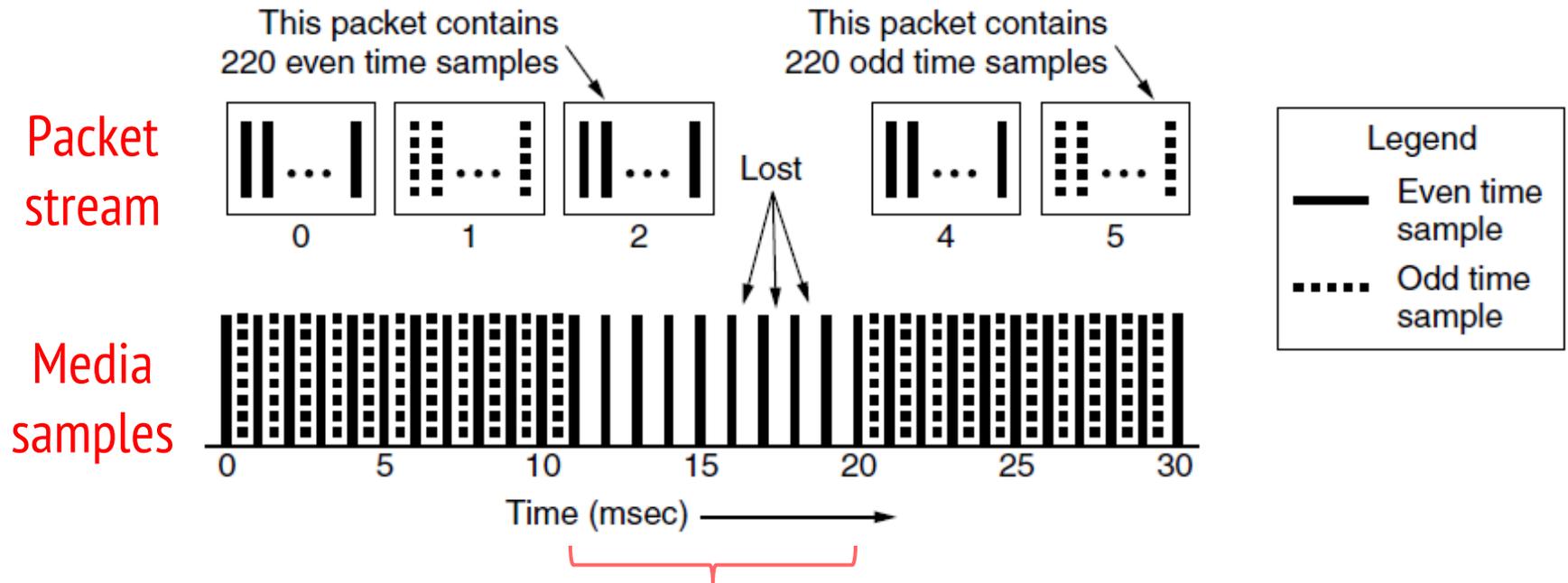
Streaming Stored Media (4)

- Parity packet can repair one lost packet in a group of N
 - Decoding is delayed for N packets



Streaming Stored Media (5)

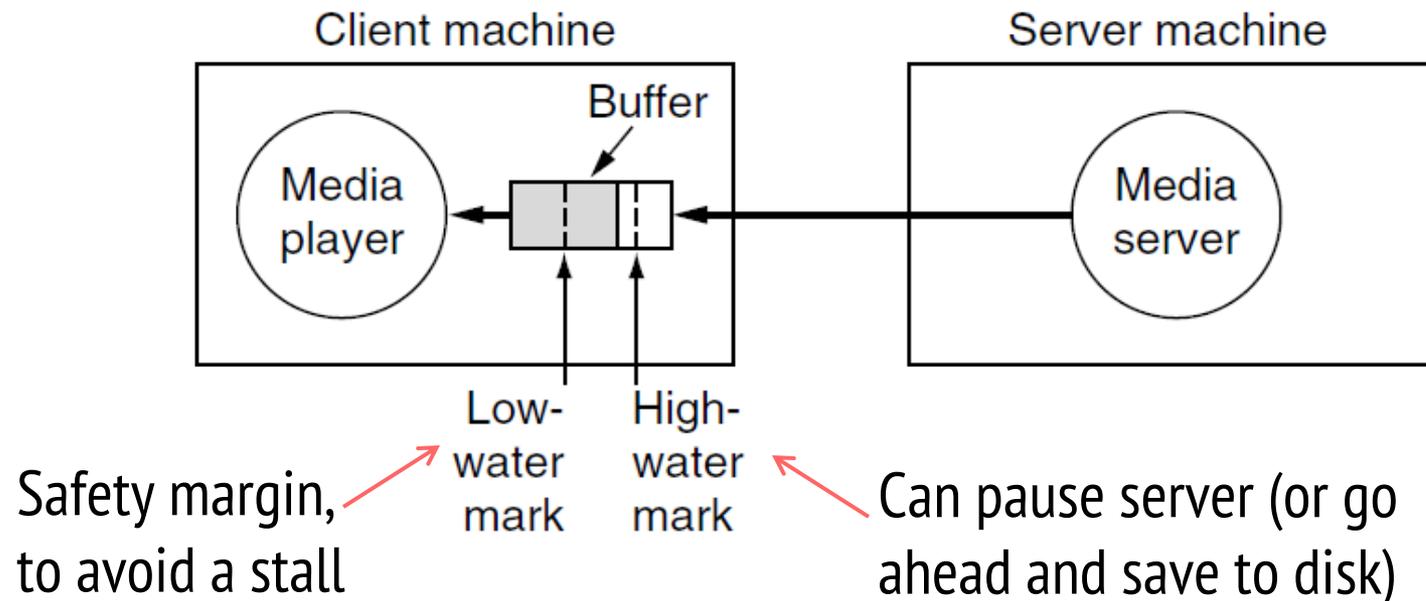
- Interleaving spreads nearby media samples over different transmissions to reduce the impact of loss



Loss reduces temporal resolution; doesn't leave a gap

Streaming Stored Media (6)

- Key problem: media may not arrive in time for playout due to variable bandwidth and loss/retransmissions
 - Client buffers media to absorb jitter; we still need to pick an achievable media rate



Streaming Stored Media (7)

- RTSP commands
 - Sent from player to server to adjust streaming

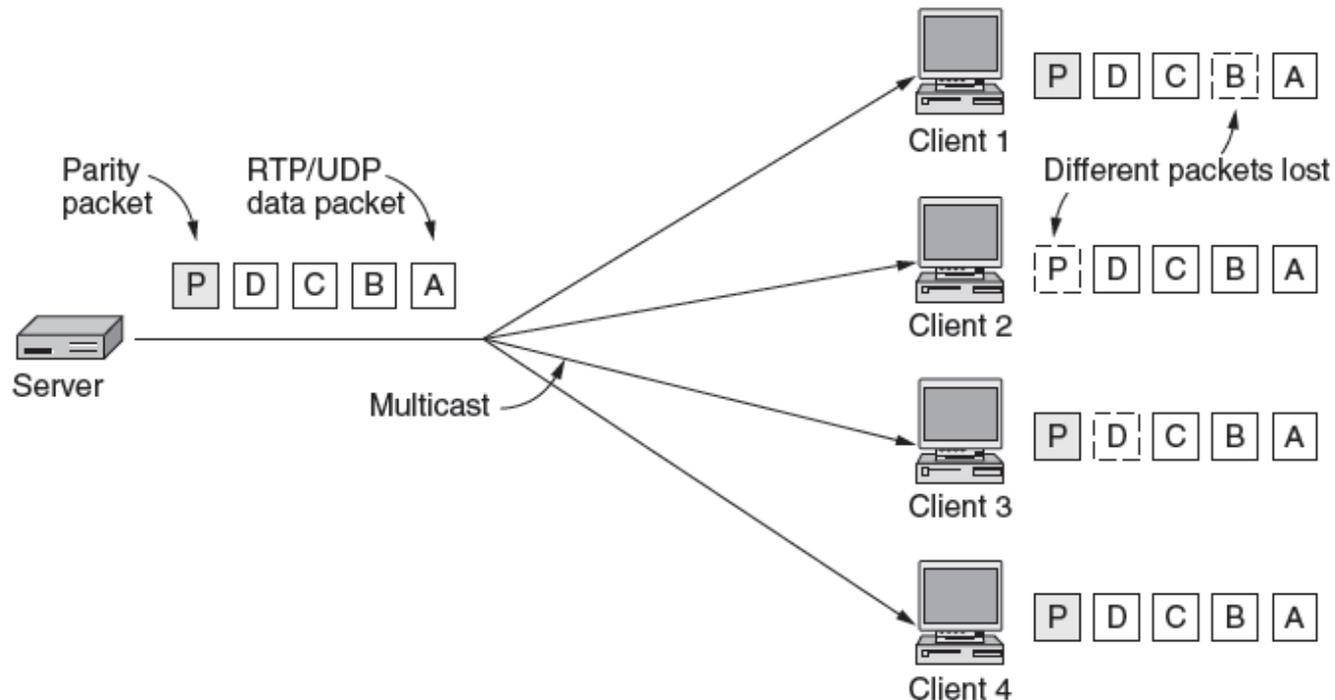
Command	Server action
DESCRIBE	List media parameters
SETUP	Establish a logical channel between the player and the server
PLAY	Start sending data to the client
RECORD	Start accepting data from the client
PAUSE	Temporarily stop sending data
TEARDOWN	Release the logical channel

Streaming Live Media (1)

- Streaming live media is similar to the stored case plus:
 - Can't stream faster than “live rate” to get ahead
 - Usually need larger buffer to absorb jitter
 - Often have many users viewing at the same time
 - UDP with multicast greatly improves efficiency. It is rarely available, so many TCP connections are used.
 - For very many users, content distribution is used [later]

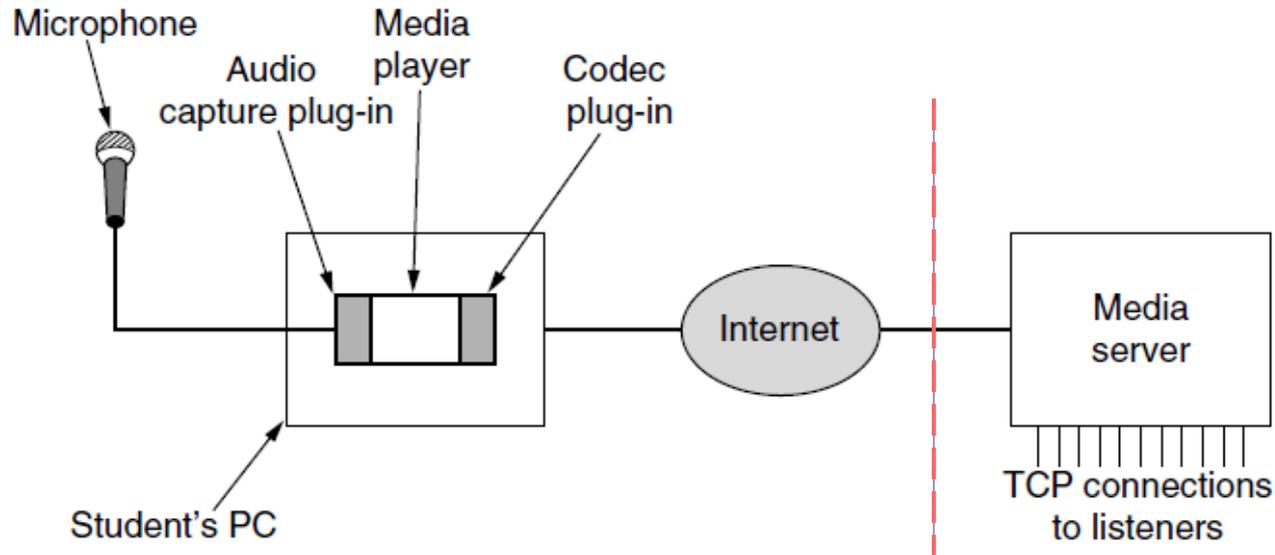
Streaming Live Media (2)

- With multicast streaming media, parity is effective
 - Clients can each lose a different packet and recover



Streaming Live Media (2)

- Production side of a student radio station.

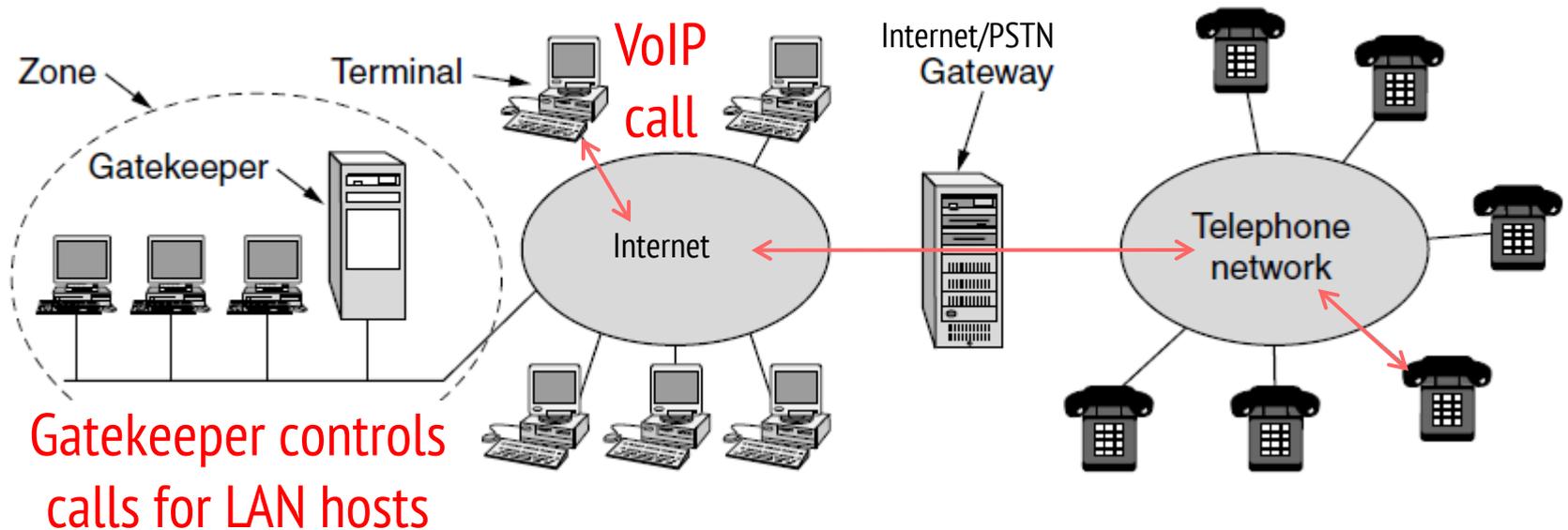


Real-Time Conferencing (1)

- Real-time conferencing has two or more connected live media streams, e.g., voice over IP, Skype video call
- Key issue over live streaming is low (interactive) latency
 - Want to reduce delay to near propagation
 - Benefits from network support, e.g., QoS
 - Or, benefits from ample bandwidth (no congestion)

Real-Time Conferencing (2)

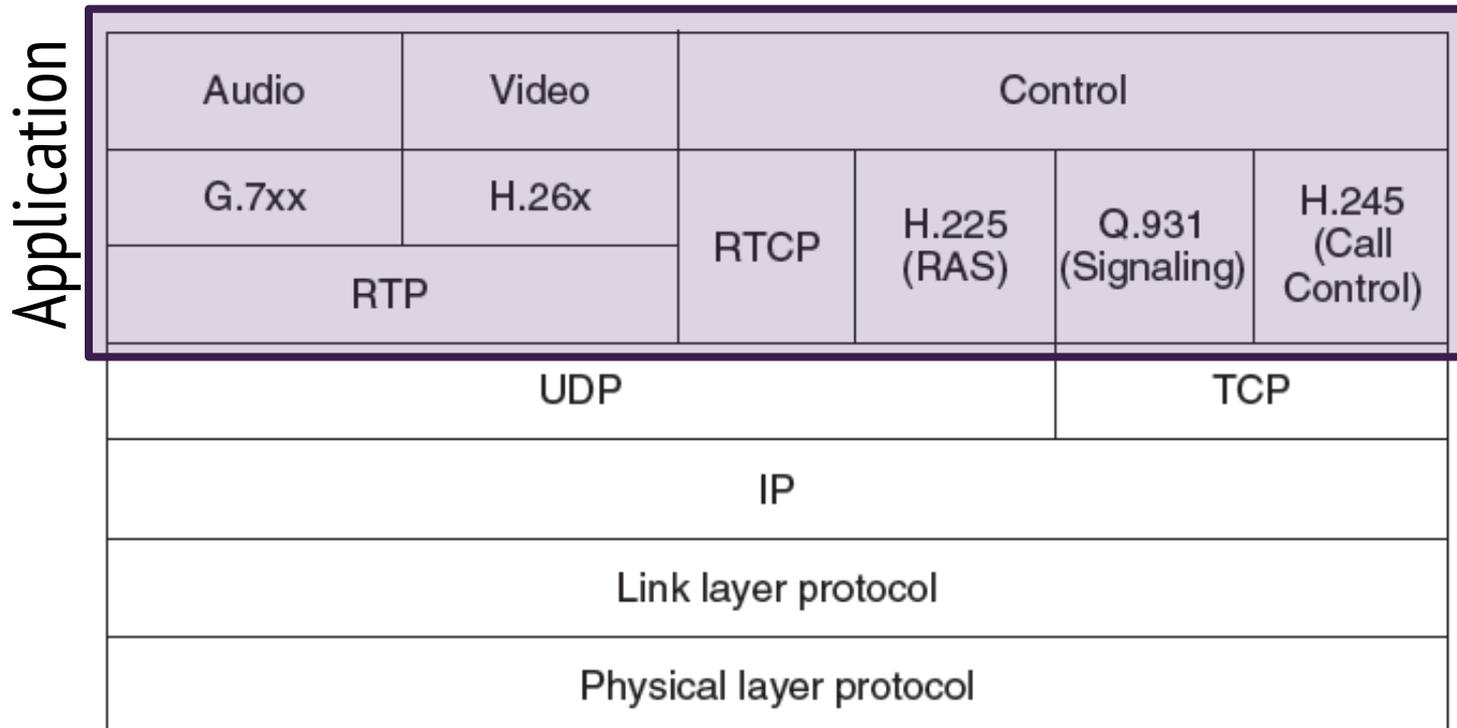
- H.323 architecture for Internet telephony supports calls between Internet computers and PSTN phones.



Real-Time Conferencing (3)

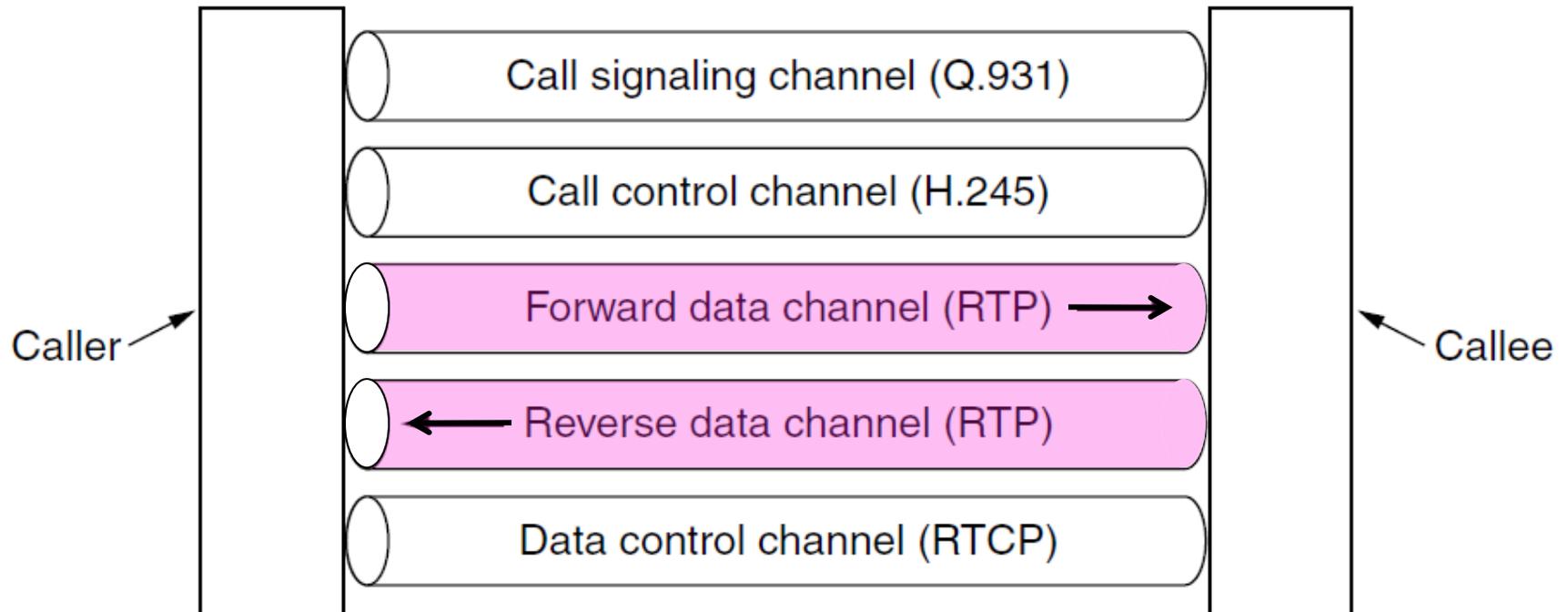
➤ H.323 protocol stack:

- Call is digital audio/video over RTP/UDP/IP
- Call setup is handled by other protocols (Q.931 etc.)



Real-Time Conferencing (4)

- Logical channels that make up an H.323 call



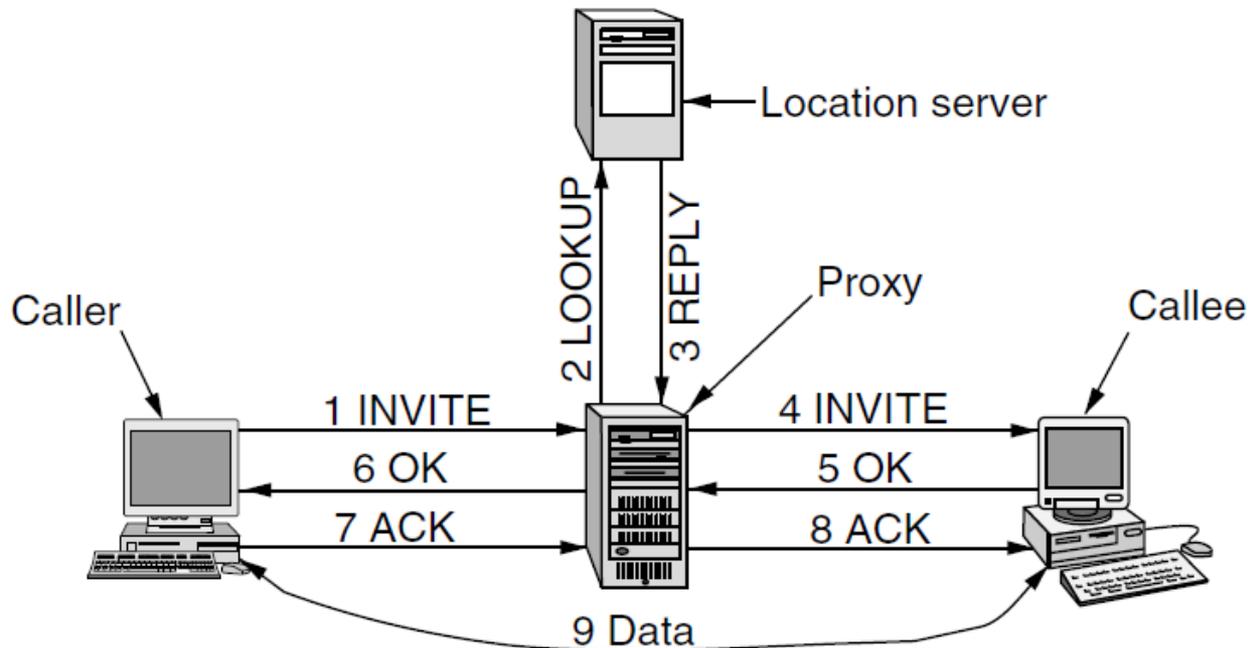
Real-Time Conferencing (5)

- SIP (Session Initiation Protocol) is an alternative to H.323 to set up real-time calls
 - Simple, text-based protocol with URLs for addresses
 - Data is carried with RTP / RTCP as before

Method	Description
INVITE	Request initiation of a session
ACK	Confirm that a session has been initiated
BYE	Request termination of a session
OPTIONS	Query a host about its capabilities
CANCEL	Cancel a pending request
REGISTER	Inform a redirection server about the user's current location

Real-Time Conferencing (6)

- Setting up a call with the SIP three-way handshake
 - Proxy server lets a remote callee be connected
 - Call data flows directly between caller/callee



Real-Time Conferencing (7)

Item	H.323	SIP
Designed by	ITU	IETF
Compatibility with PSTN	Yes	Largely
Compatibility with Internet	Yes, over time	Yes
Architecture	Monolithic	Modular
Completeness	Full protocol stack	SIP just handles setup
Parameter negotiation	Yes	Yes
Call signaling	Q.931 over TCP	SIP over TCP or UDP
Message format	Binary	ASCII
Media transport	RTP/RTCP	RTP/RTCP
Multiparty calls	Yes	Yes
Multimedia conferences	Yes	No
Addressing	URL or phone number	URL
Call termination	Explicit or TCP release	Explicit or timeout
Instant messaging	No	Yes
Encryption	Yes	Yes
Size of standards	1400 pages	250 pages
Implementation	Large and complex	Moderate, but issues
Status	Widespread, esp. video	Alternative, esp. voice

Comparison of H.323 and SIP.

DNS: domain name system

people: many identifiers:

- SSN, name, passport #

Internet hosts, routers:

- IP address (32 bit) - used for addressing datagrams
- “name”, e.g., www.yahoo.com - used by humans

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

Domain Name System:

- *distributed database*
implemented in hierarchy of many name servers
- *application-layer protocol*: hosts, name 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

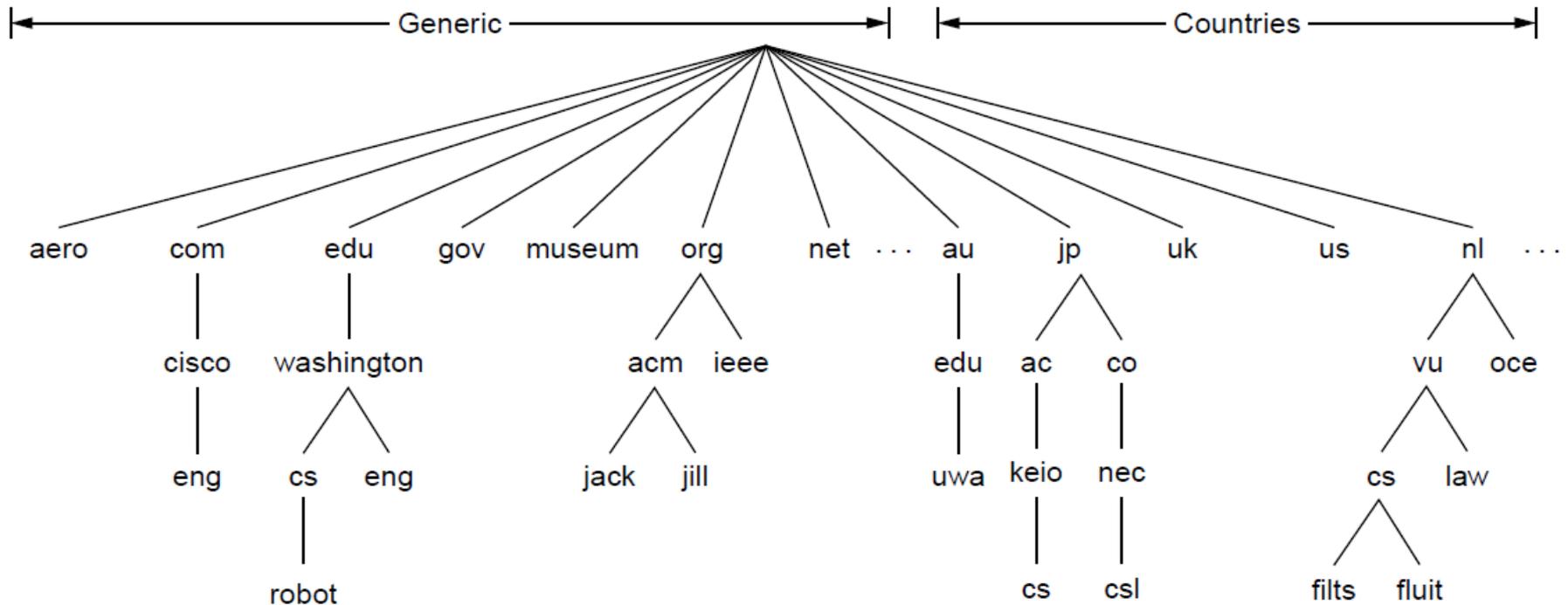
why not centralize DNS?

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

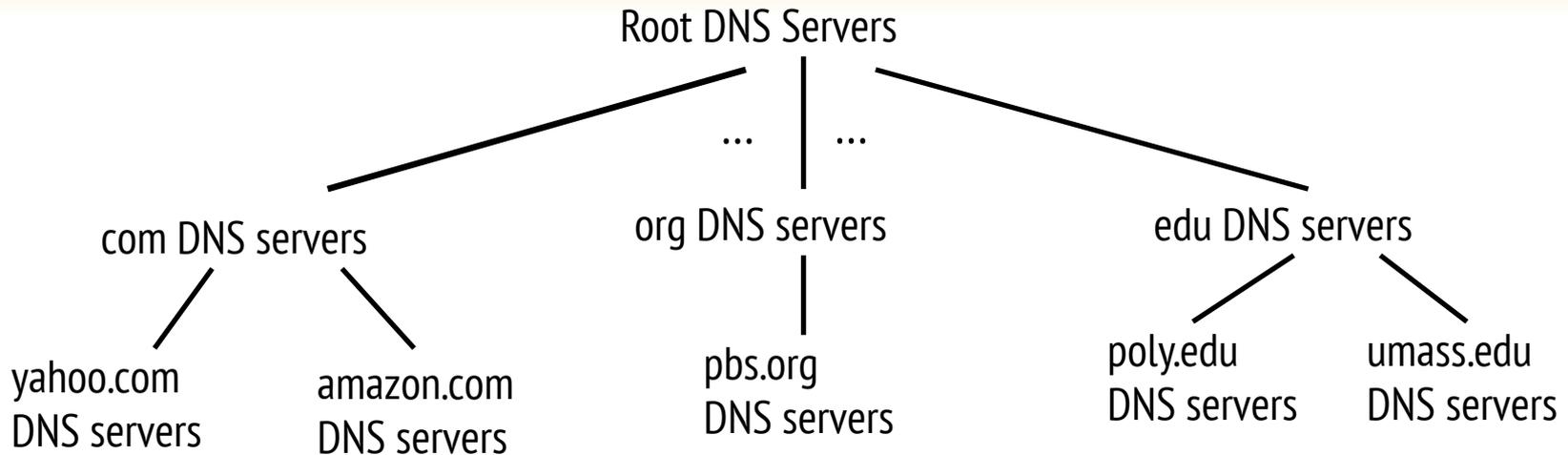
A: doesn't scale!

Domain Name Space

- DNS is hierarchical
- Assigned based on affiliation of institution



DNS: a distributed, hierarchical database

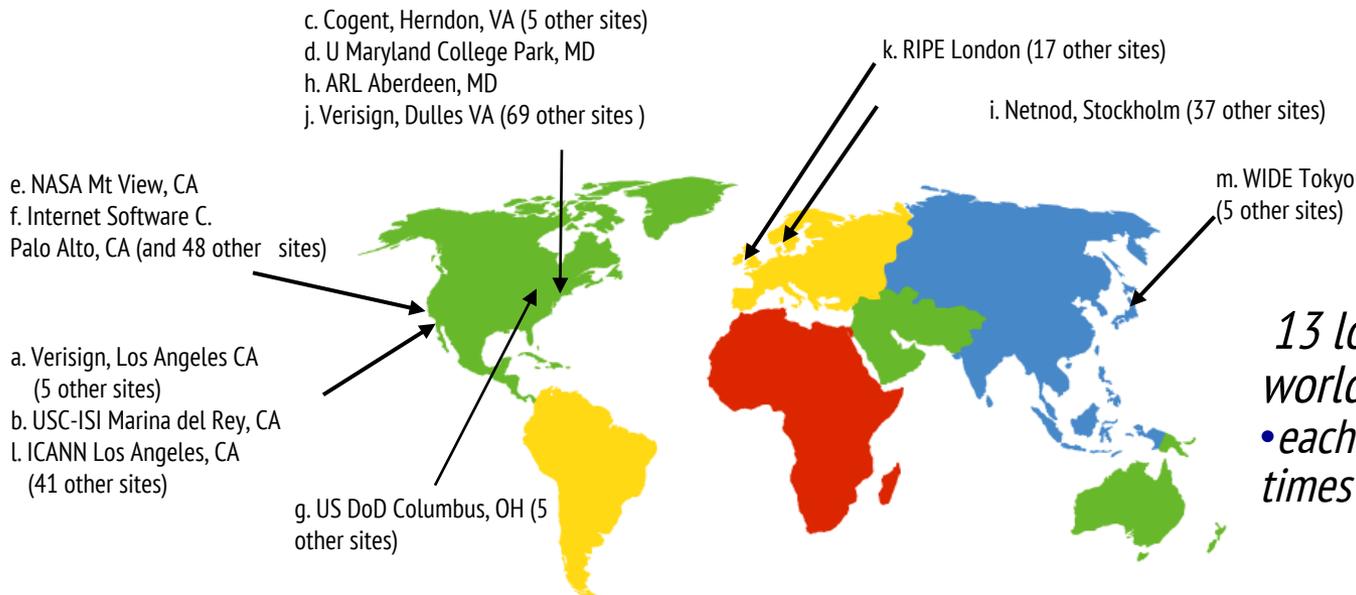


client wants IP for www.amazon.com; 1st approximation:

- 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

- contacted by local name server that can not resolve name
- root name server:
 - contacts authoritative name server if name mapping not known
 - gets mapping
 - returns mapping to local name server



*13 logical root name “servers”
worldwide
• each “server” replicated many
times*

TLD, authoritative servers

top-level domain (TLD) servers:

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

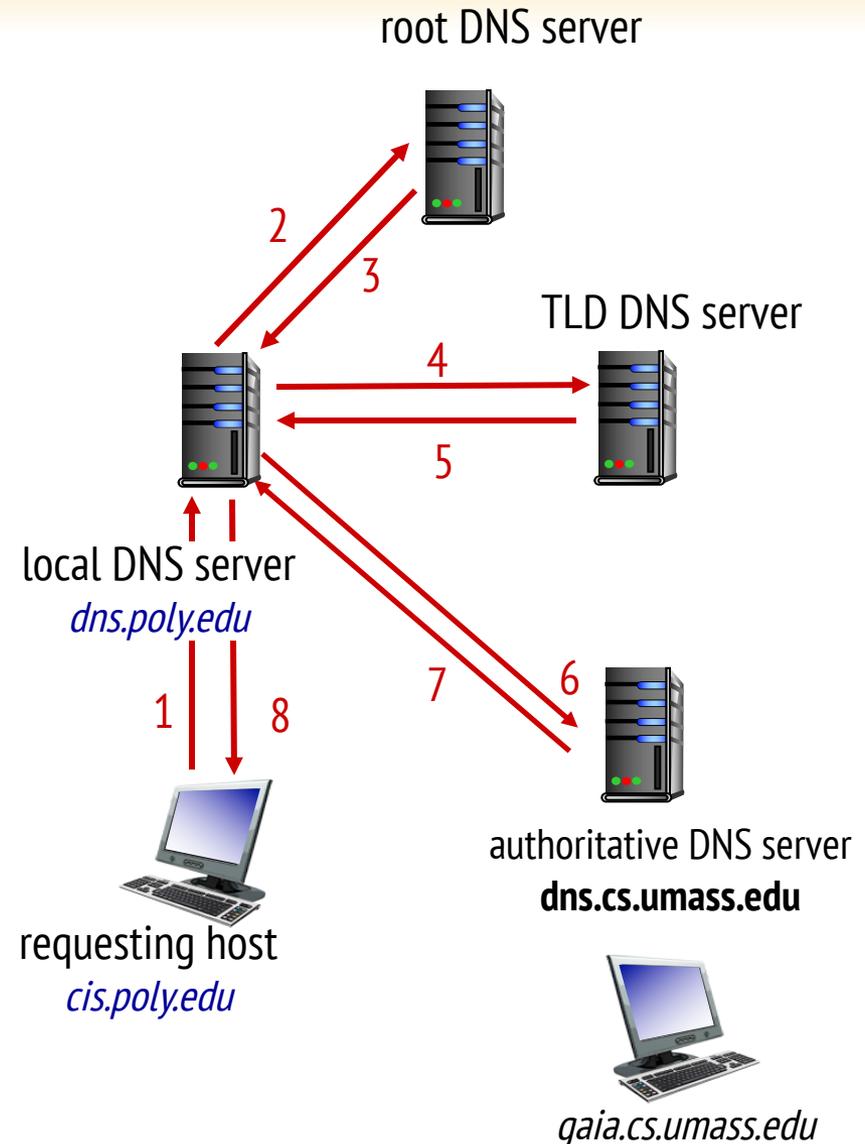
- does not strictly belong to hierarchy
- each ISP (residential ISP, company, university) has one
 - also called “default name server”
- when host makes DNS query, query is sent to its local DNS server
 - has local cache of recent name-to-address translation pairs (but may be out of date!)
 - acts as proxy, forwards query into hierarchy

DNS name resolution example

- host at cis.poly.edu wants IP address for gaia.cs.umass.edu

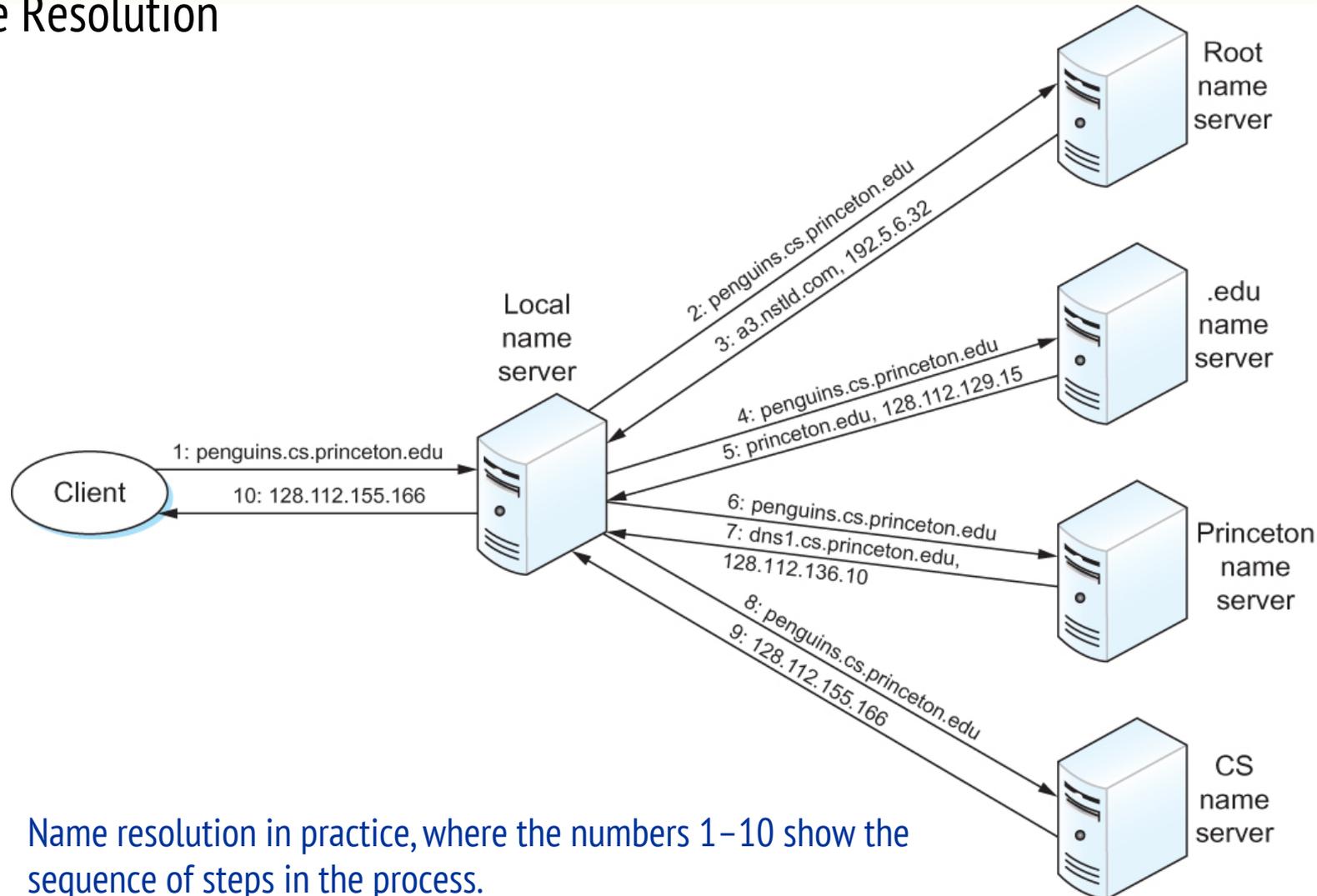
iterated query:

- contacted server replies with name of server to contact
- “I don’t know this name, but ask this server”



Infrastructure Services

➤ Name Resolution

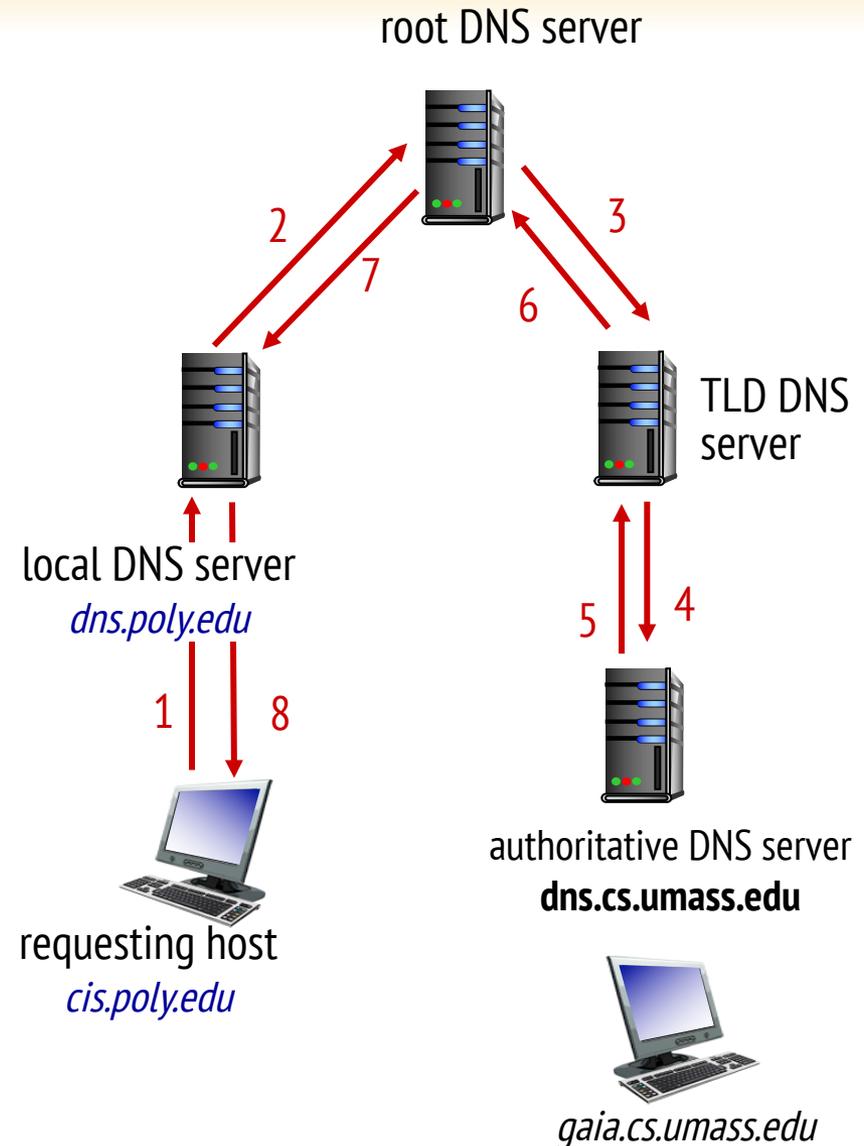


Name resolution in practice, where the numbers 1–10 show the sequence of steps in the process.

DNS name resolution example

recursive query:

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



DNS: caching, updating records

- once (any) name server learns mapping, it *caches* mapping
 - cache entries timeout (disappear) after some time (TTL)
 - TLD servers typically cached in local name servers
 - thus root name servers not often visited
- cached entries may be *out-of-date* (best effort name-to-address translation!)
 - if name host changes IP address, may not be known Internet-wide until all TTLs expire
- update/notify mechanisms proposed IETF standard
 - RFC 2136

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

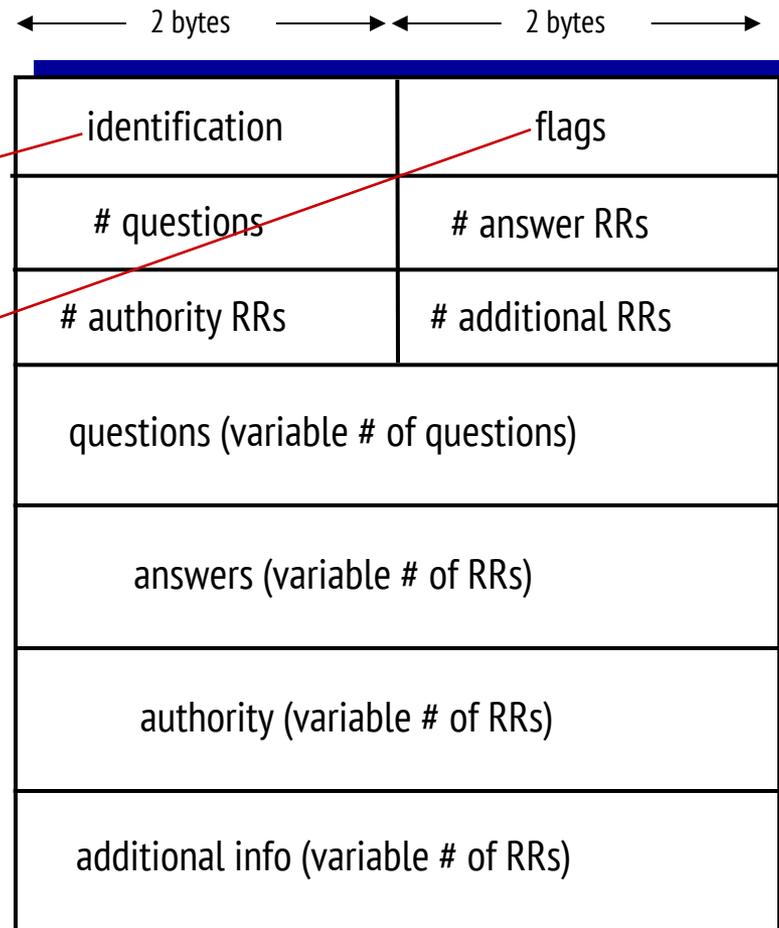
- **value** is name of mailserver associated with **name**

DNS protocol, messages

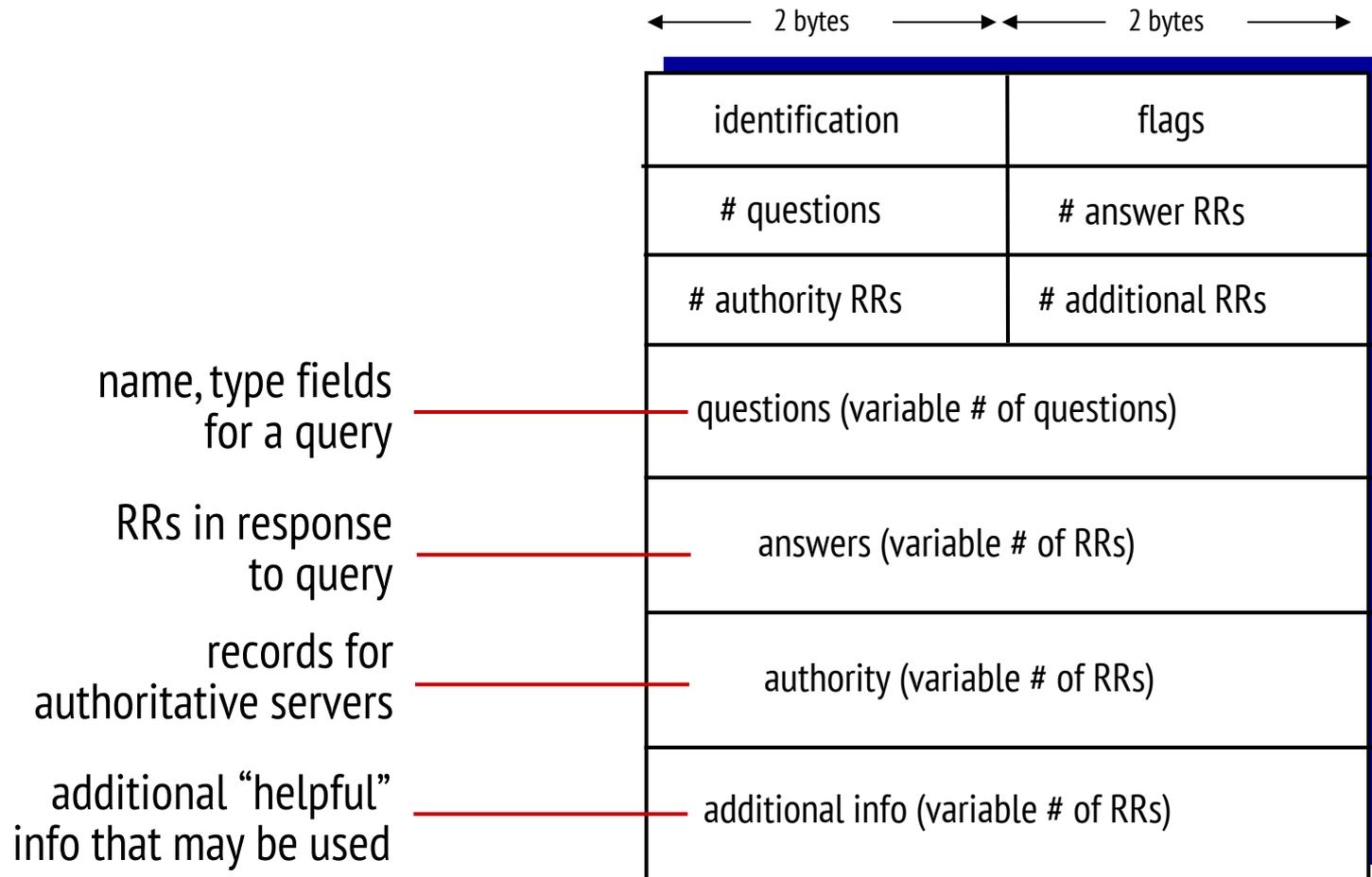
- *query* and *reply* messages, both with same *message format*

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



Inserting records into DNS

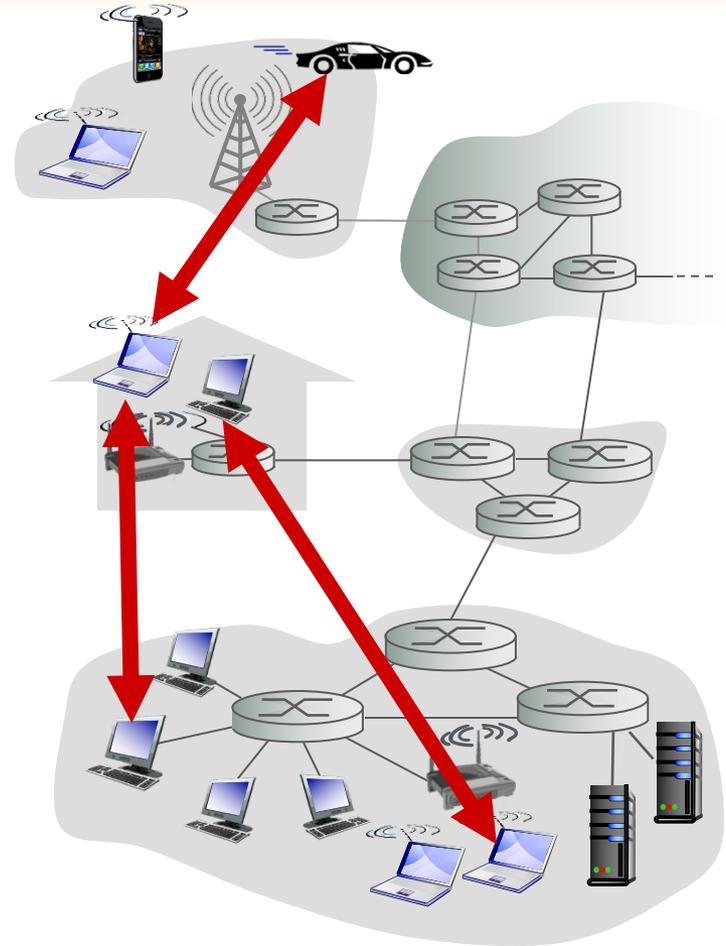
- example: new startup “Network Utopia”
- register name networkutopia.com at *DNS registrar* (e.g., Network Solutions)
 - provide names, IP addresses of authoritative name server (primary and secondary)
 - registrar inserts two RRs into .com TLD server:
 - (networkutopia.com, dns1.networkutopia.com, NS)
 - (dns1.networkutopia.com, 212.212.212.1, A)
- create authoritative server type A record for www.networkutopia.com; type MX record for networkutopia.com

Pure P2P architecture

- *no* always-on server
- arbitrary end systems directly communicate
- peers are intermittently connected and change IP addresses

examples:

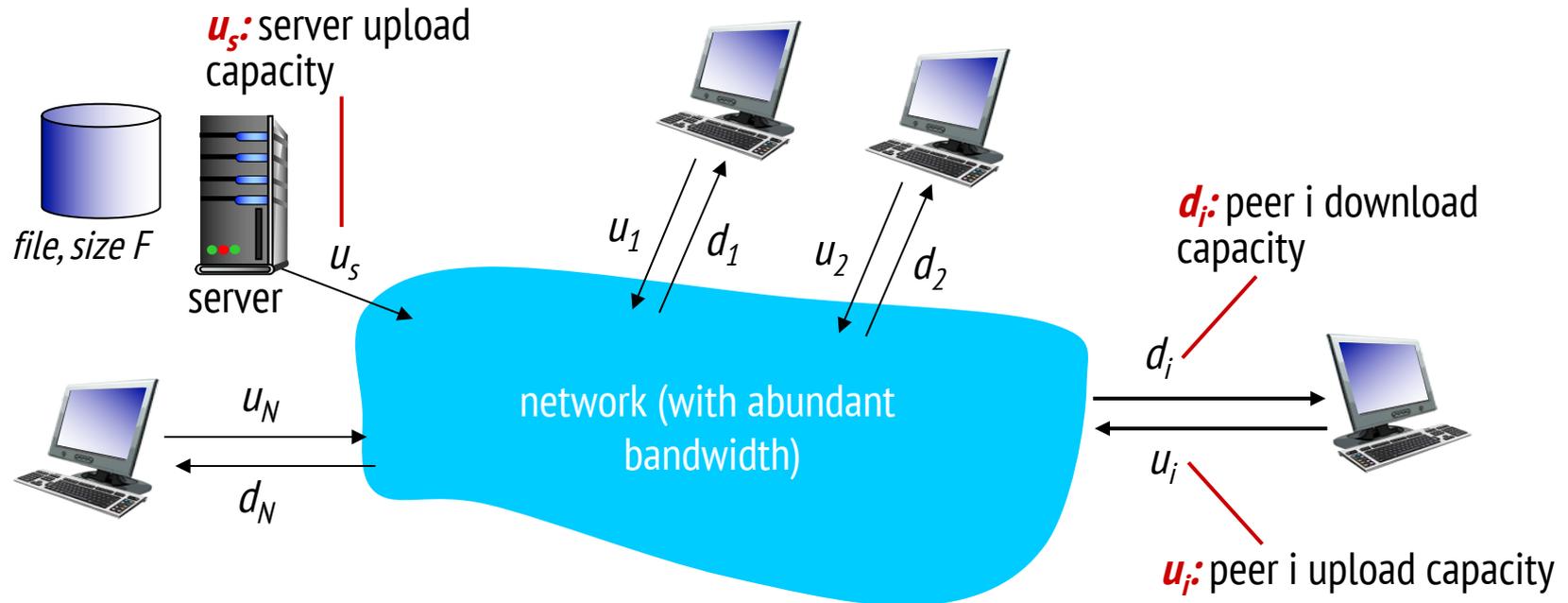
- file distribution (BitTorrent)
- Streaming (KanKan)
- VoIP (Skype)



File distribution: client-server vs P2P

Question: how much time to distribute file (size F) from one server to N peers?

- peer upload/download capacity is limited resource



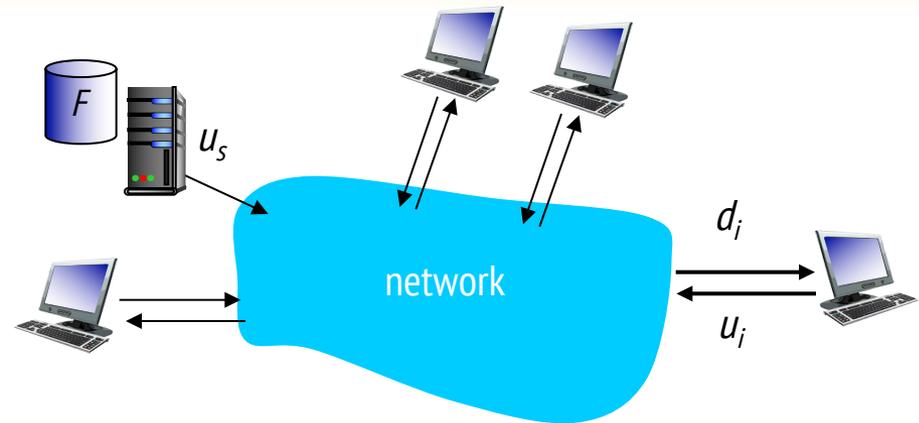
File distribution time: client-server

➤ *server transmission*: must sequentially send (upload) N file copies:

- time to send one copy: F/u_s
- time to send N copies: NF/u_s

➤ *client*: each client must download file copy

- d_{\min} = min client download rate
- min client download time: F/d_{\min}



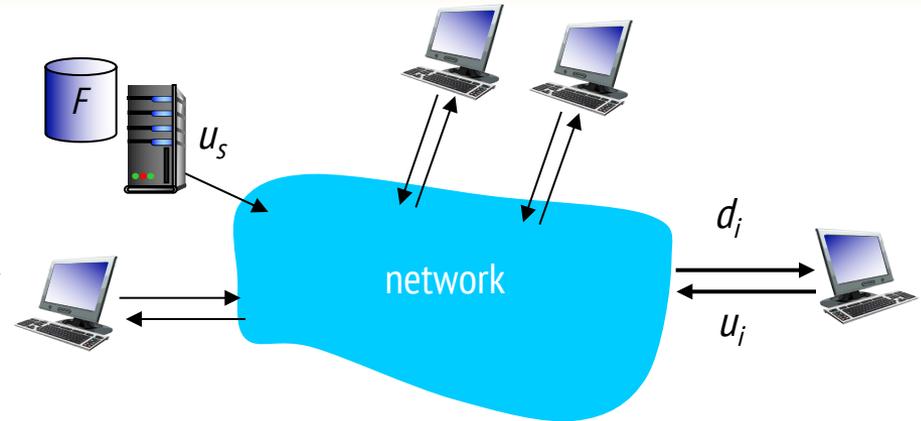
*time to distribute F
to N clients using
client-server approach*

$$D_{C-S} > \bar{m} \max\{NF/u_s, F/d_{\min}\}$$

increases linearly in N

File distribution time: P2P

- **server transmission**: must upload at least one copy
 - time to send one copy: F/u_s
- **client**: each client must download file copy
 - min client download time: F/d_{\min}
- **clients**: as aggregate must download NF bits
 - max upload rate (limiting max download rate) is $u_s + \sum u_i$



*time to distribute F
to N clients using
P2P approach*

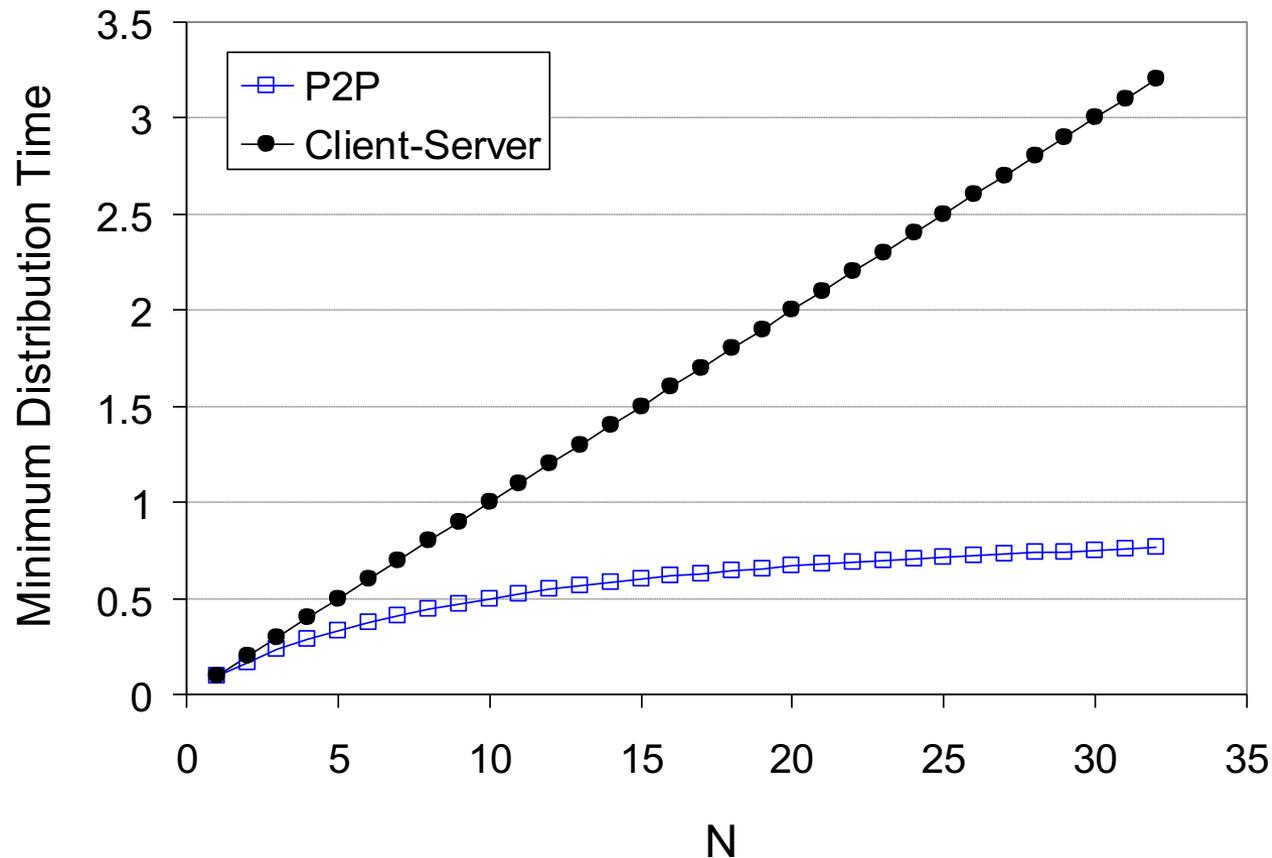
$$D_{P2P} \geq \max\{F/u_s, F/d_{\min}, NF/(u_s + \sum u_i)\}$$

increases linearly in N ...

... but so does this, as each peer brings service capacity

Client-server vs. P2P: example

client upload rate = u , $F/u = 1$ hour, $u_s = 10u$, $d_{min} \geq u_s$

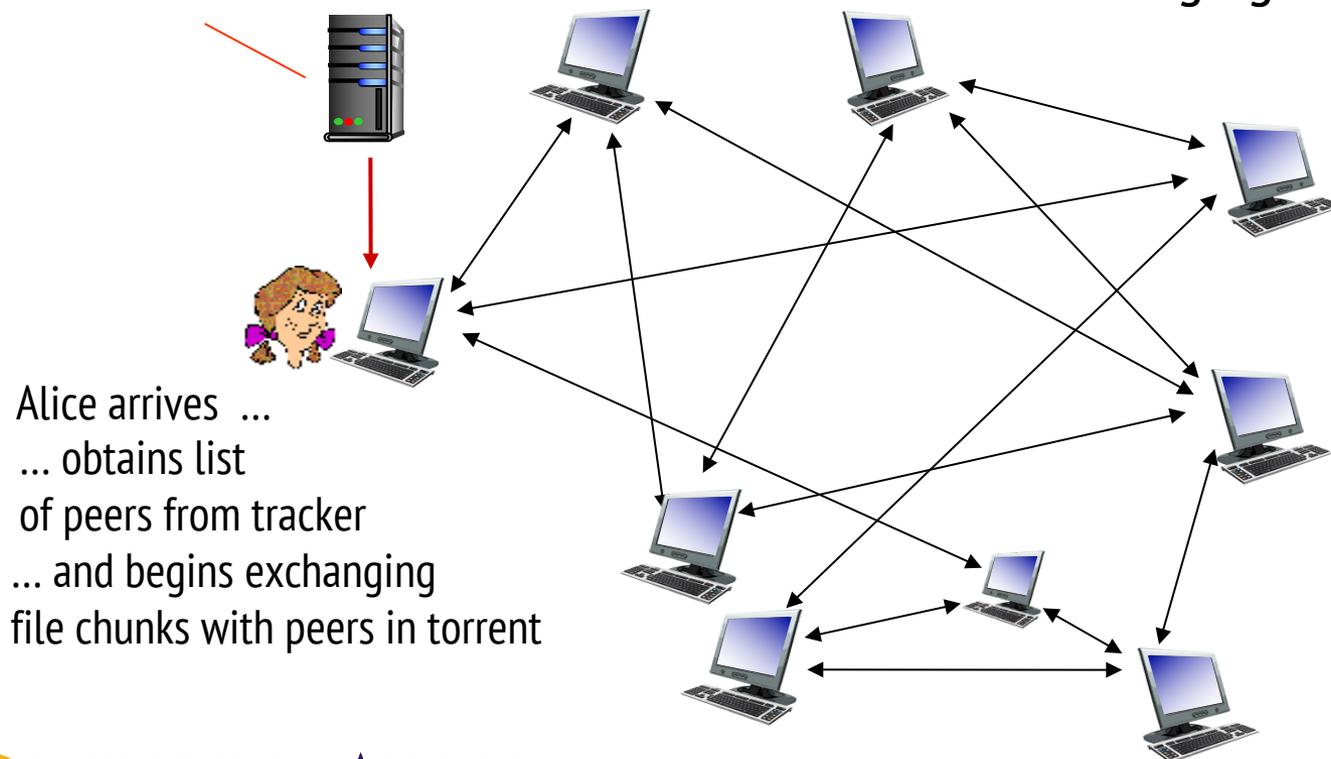


P2P file distribution: BitTorrent protocol

- file divided into 256Kb chunks
- peers in torrent send/receive file chunks

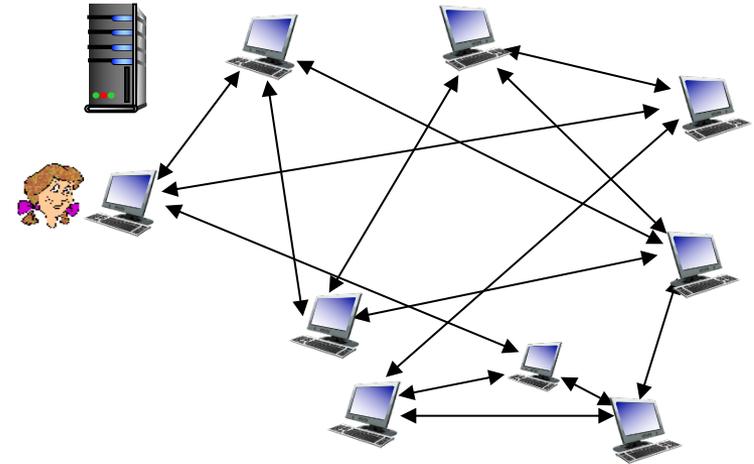
tracker: tracks peers participating in torrent

torrent: group of peers exchanging chunks of a file



P2P file distribution: BitTorrent

- peer joining torrent:
 - has no chunks, but will accumulate them over time from other peers
 - registers with tracker to get list of peers, connects to subset of peers (“neighbors”)
- while downloading, peer uploads chunks to other peers
- peer may change peers with whom it exchanges chunks
- churn: peers may come and go
- once peer has entire file, it may (selfishly) leave or (altruistically) remain in torrent



BitTorrent: requesting, sending file chunks

requesting chunks:

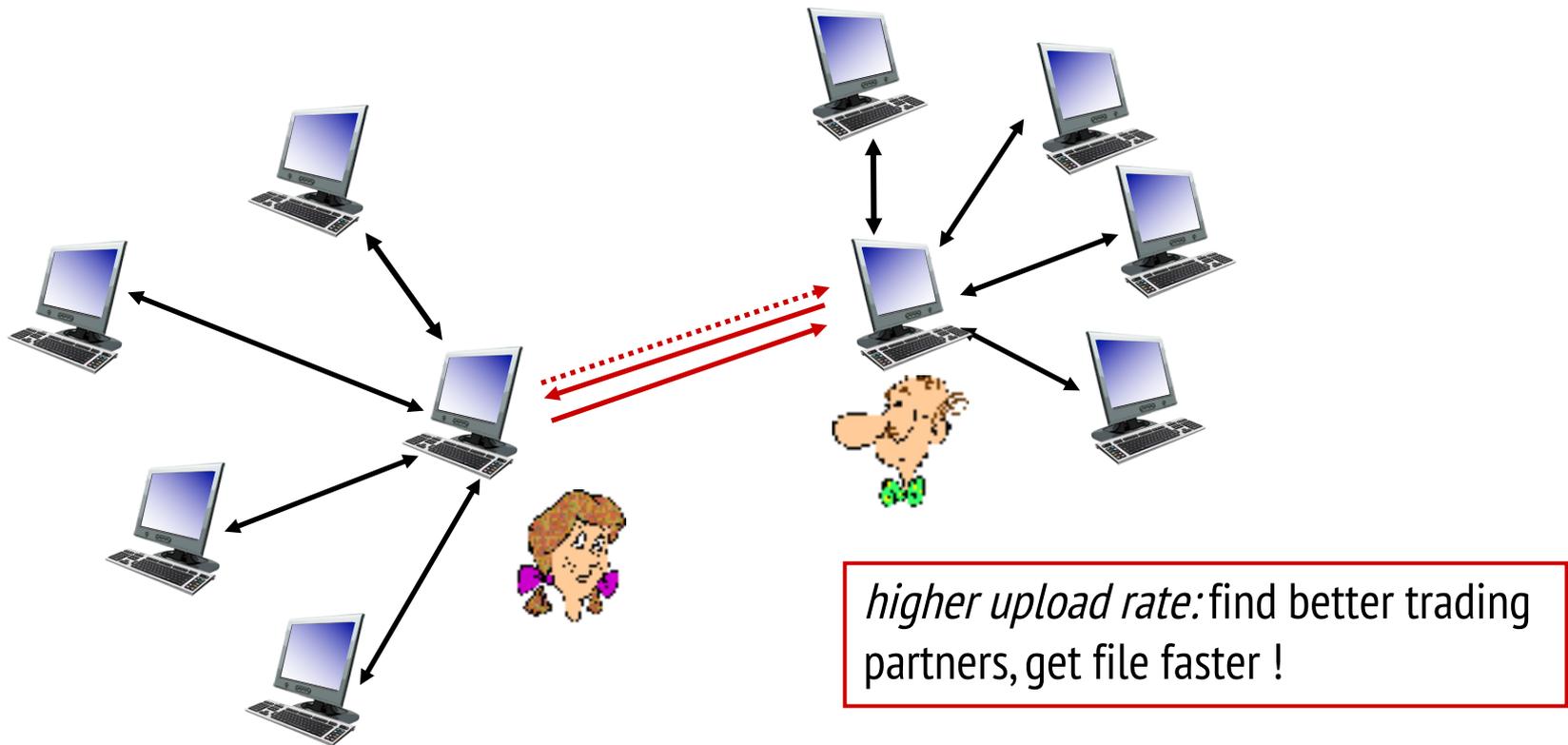
- at any given time, different peers have different subsets of file chunks
- periodically, Alice asks each peer for list of chunks that they have
- Alice requests missing chunks from peers, rarest first

sending chunks: tit-for-tat

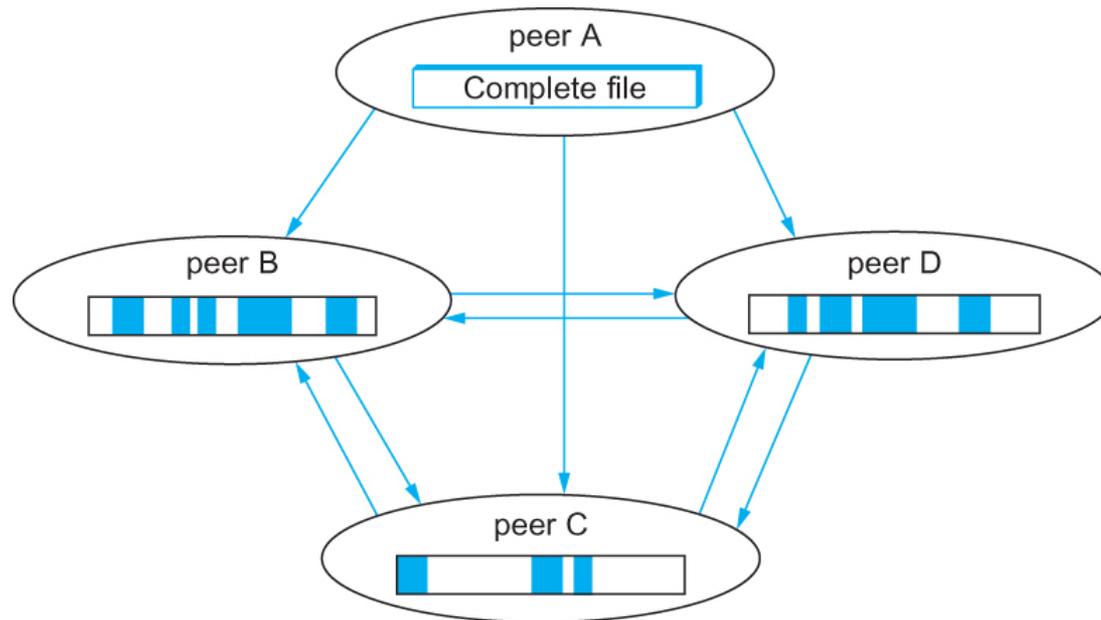
- Alice sends chunks to those four peers currently sending her chunks *at highest rate*
 - other peers are choked by Alice (do not receive chunks from her)
 - re-evaluate top 4 every 10 secs
- every 30 secs: randomly select another peer, starts sending chunks
 - “optimistically unchoke” this peer
 - newly chosen peer may join top 4

BitTorrent: tit-for-tat

- (1) Alice “optimistically unchokes” Bob
- (2) Alice becomes one of Bob’s top-four providers; Bob reciprocates
- (3) Bob becomes one of Alice’s top-four providers



BitTorrent: another aspect



Peers in a BitTorrent swarm download from other peers that may not yet have the complete file

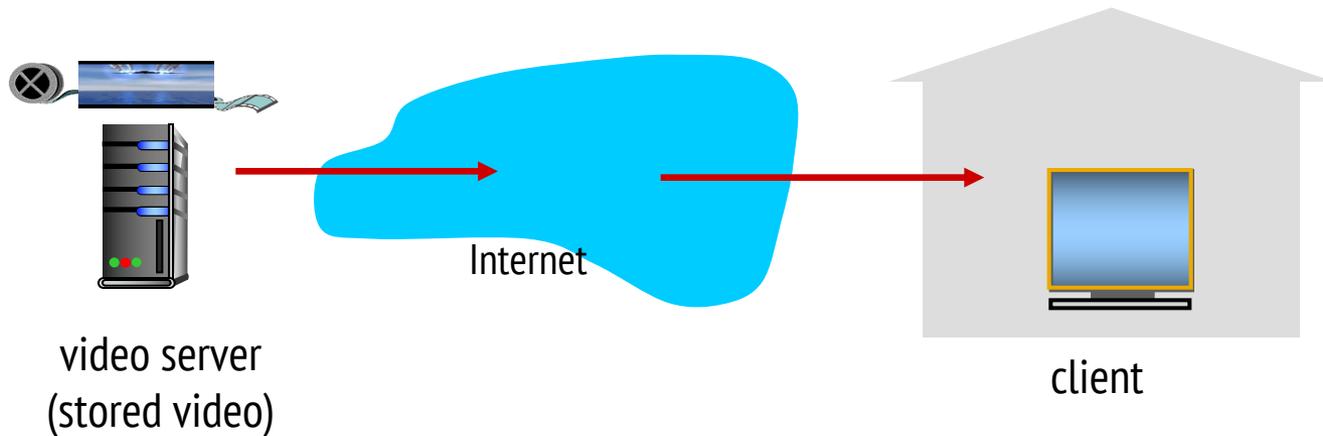
Video Streaming and CDNs: context

- video traffic: major consumer of Internet bandwidth
 - Netflix, YouTube: 37%, 16% of downstream residential traffic
 - ~1B YouTube users, ~75M Netflix users
- challenge: scale - how to reach ~1B users?
 - single mega-video server won't work (why?)
- challenge: heterogeneity
 - different users have different capabilities (e.g., wired versus mobile; bandwidth rich versus bandwidth poor)
- *solution*: distributed, application-level infrastructure



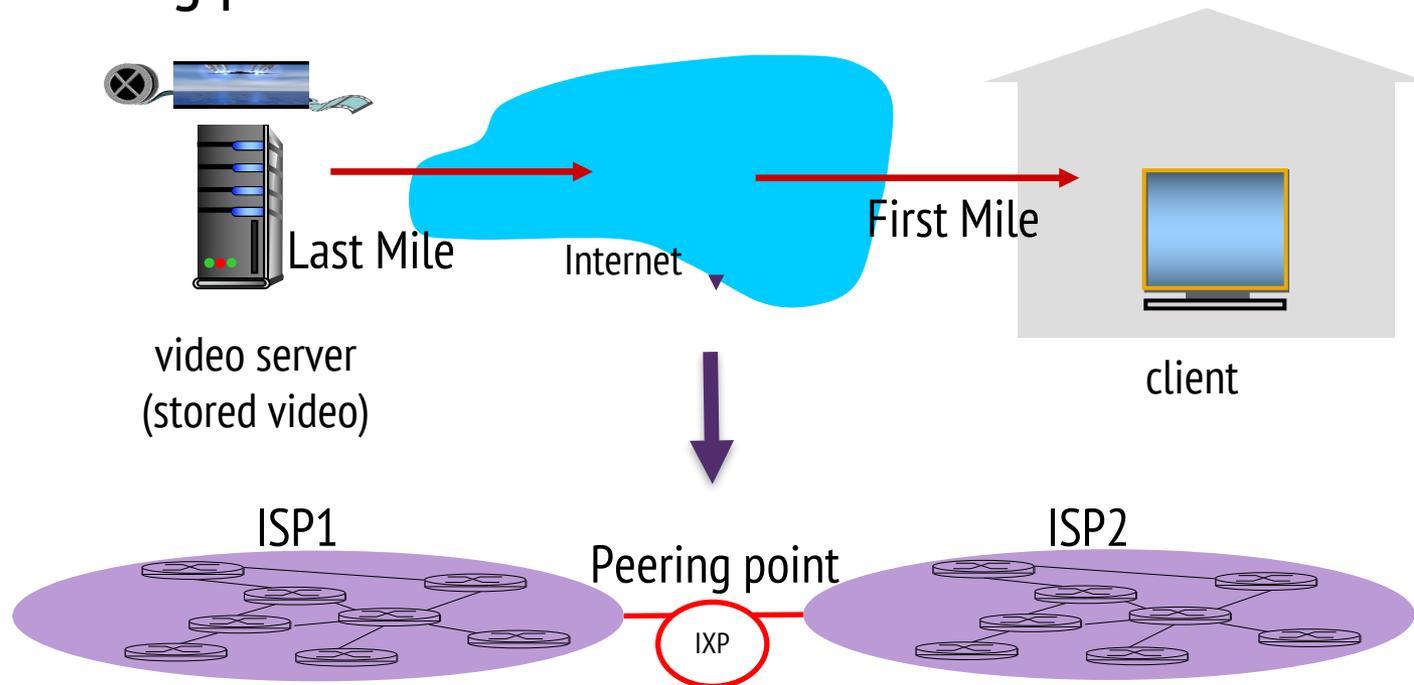
Streaming stored video:

simple scenario:



Bottlenecks in the system

- The first mile
- The last mile
- The server itself
- Peering points



Streaming multimedia: DASH

- *DASH: Dynamic, Adaptive Streaming over HTTP*
- *server:*
 - divides video file into multiple chunks
 - each chunk stored, encoded at different rates
 - *manifest file:* provides URLs for different chunks
- *client:*
 - periodically measures server-to-client bandwidth
 - consulting manifest, requests one chunk at a time
 - chooses maximum coding rate sustainable given current bandwidth
 - can choose different coding rates at different points in time (depending on available bandwidth at time)

Streaming multimedia: DASH

- *DASH: Dynamic, Adaptive Streaming over HTTP*
- “*intelligence*” at client: client determines
 - *when* to request chunk (so that buffer starvation, or overflow does not occur)
 - *what encoding rate* to request (higher quality when more bandwidth available)
 - *where* to request chunk (can request from URL server that is “close” to client or has high available bandwidth)

Content distribution networks

- *challenge*: how to stream content (selected from millions of videos) to hundreds of thousands of *simultaneous* users?
- *option 1*: single, large “mega-server”
 - single point of failure
 - point of network congestion
 - long path to distant clients
 - multiple copies of video sent over outgoing link

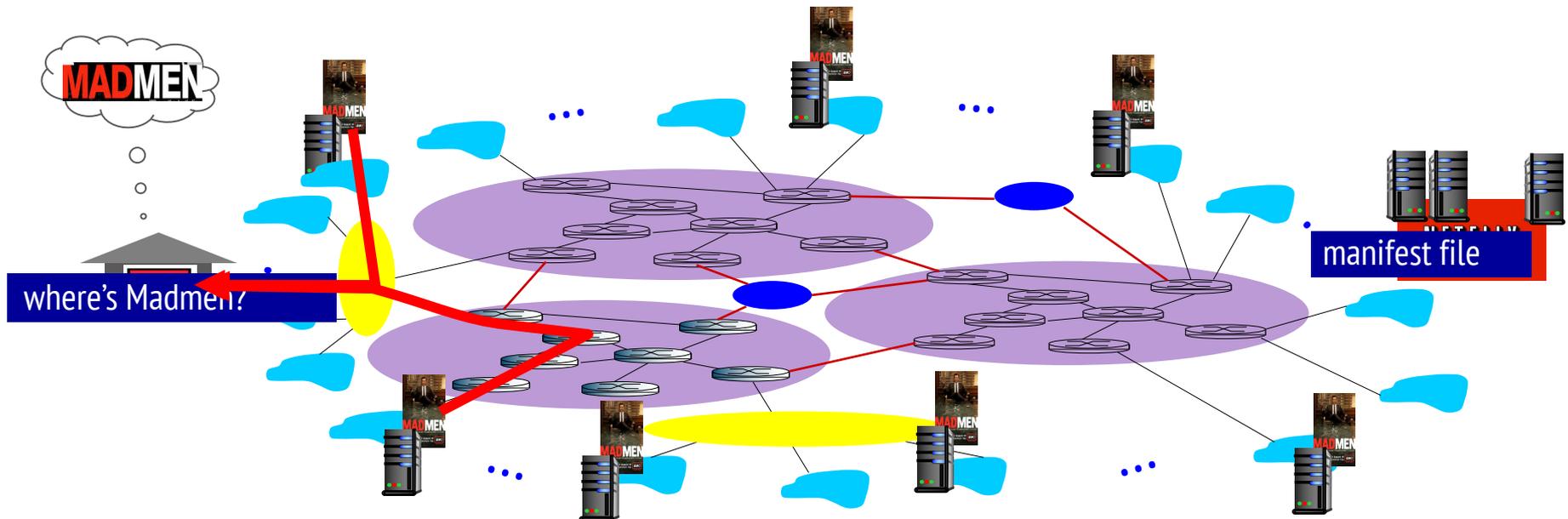
....quite simply: this solution *doesn't scale*

Content distribution networks

- *challenge*: how to stream content (selected from millions of videos) to hundreds of thousands of simultaneous users?
- *option 2*: store/serve multiple copies of videos at multiple geographically distributed sites (*CDN*)
 - *enter deep*: push CDN servers deep into many access networks
 - close to users
 - used by Akamai, 1700 locations
 - *bring home*: smaller number (10's) of larger clusters in POPs near (but not within) access networks
 - used by Limelight

Content Distribution Networks (CDNs)

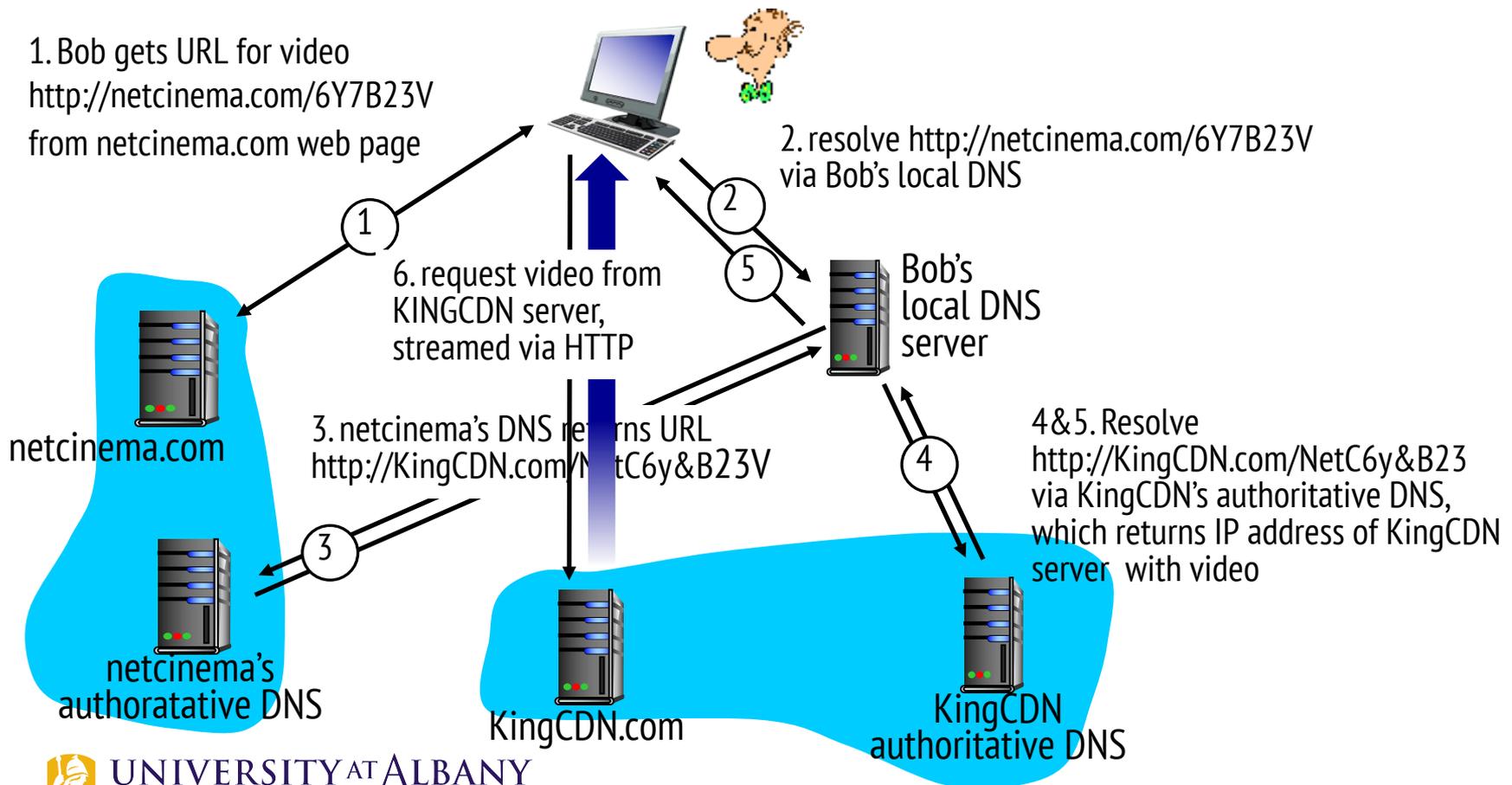
- subscriber requests content from CDN
- CDN: stores copies of content at CDN nodes
 - e.g. Netflix stores copies of MadMen
 - directed to nearby copy, retrieves content
 - may choose different copy if network path congested



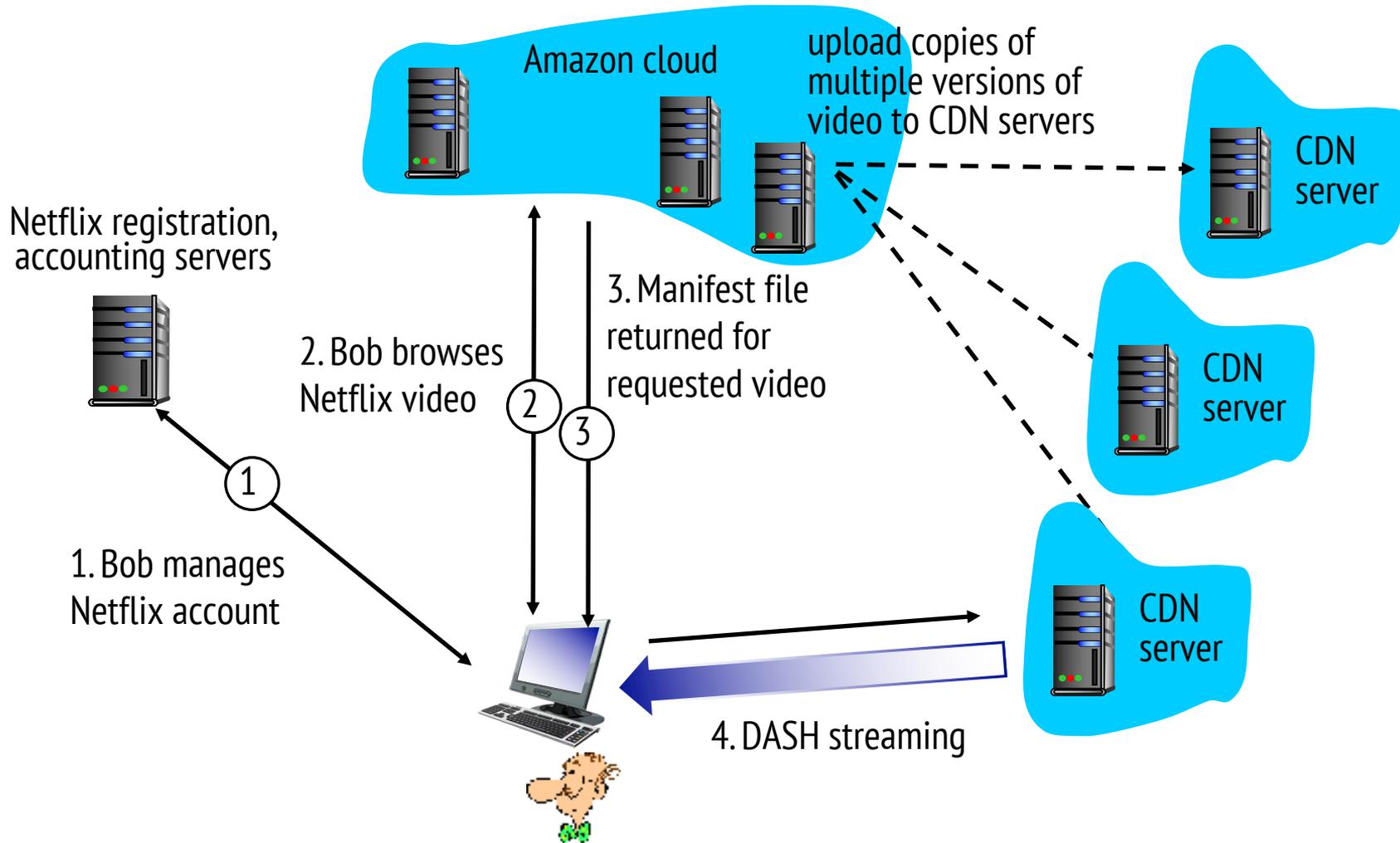
CDN content access: a closer look

Bob (client) requests video `http://netcinema.com/6Y7B23V`

- video stored in CDN at `http://KingCDN.com/NetC6y&B23V`



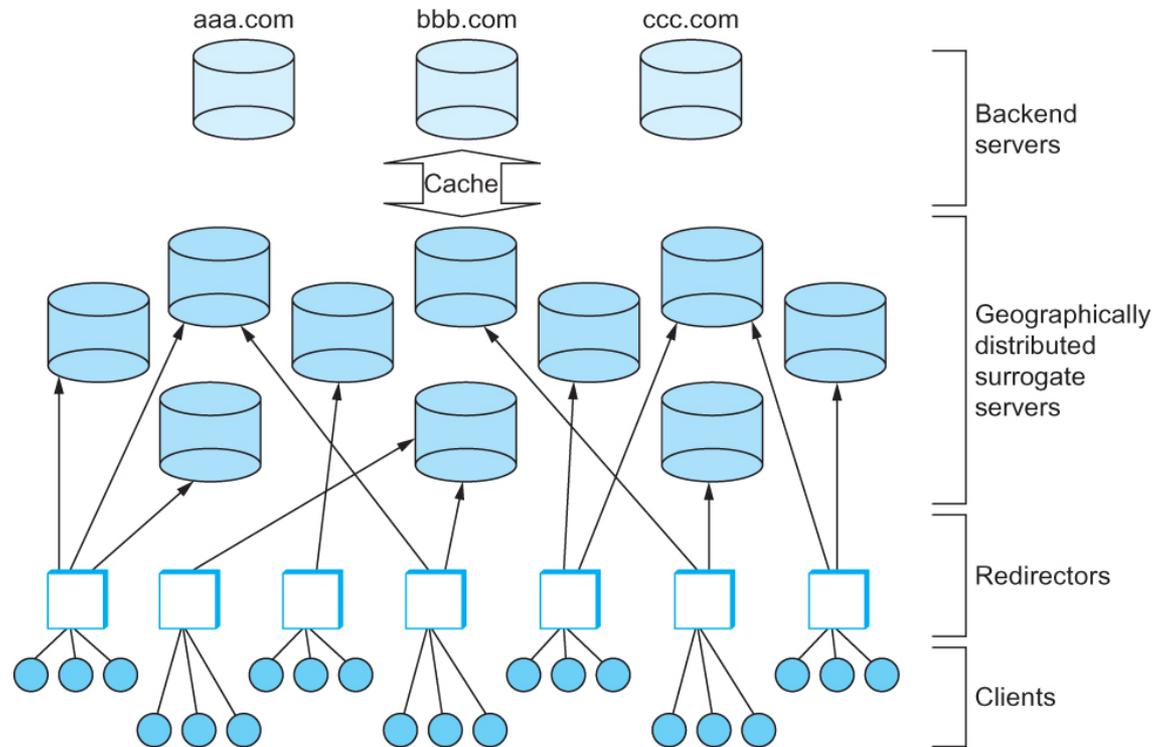
Case study: Netflix



CDN: In short

- The idea of a CDN is to geographically distribute a collection of *server surrogates* that cache pages normally maintained in some set of *backend servers*
 - Akamai operates what is probably the best-known CDN.
- Thus, rather than have millions of users wait forever to contact www.cnn.com when a big news story breaks—such a situation is known as a *flash crowd*—it is possible to spread this load across many servers.
- Moreover, rather than having to traverse multiple ISPs to reach www.cnn.com, if these surrogate servers happen to be spread across all the backbone ISPs, then it should be possible to reach one without having to cross a peering point.

CDN Components



Components in a Content Distribution Network (CDN).

Summary

- We have discussed some of the popular applications in the Internet
 - Electronic mail, World Wide Web
- We have discussed multimedia applications
- We have discussed infrastructure services
 - Domain Name Services (DNS)
- We have discussed overlay networks
- We have discussed content distribution networks