

# Chapter 2: outline

2.1 principles of network applications

2.2 Web and HTTP

2.3 FTP

2.4 electronic mail

- SMTP, POP3, IMAP

2.5 DNS

2.6 P2P applications

2.7 video streaming and content distribution networks

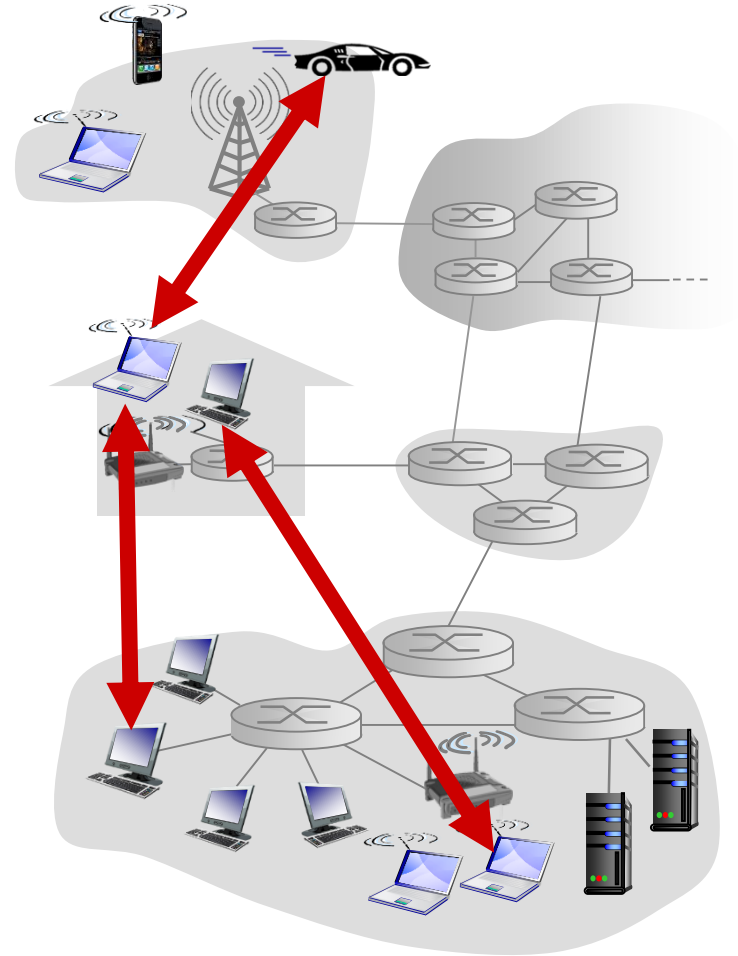
2.8 socket programming with UDP and TCP

# Pure P2P architecture

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

## *examples:*

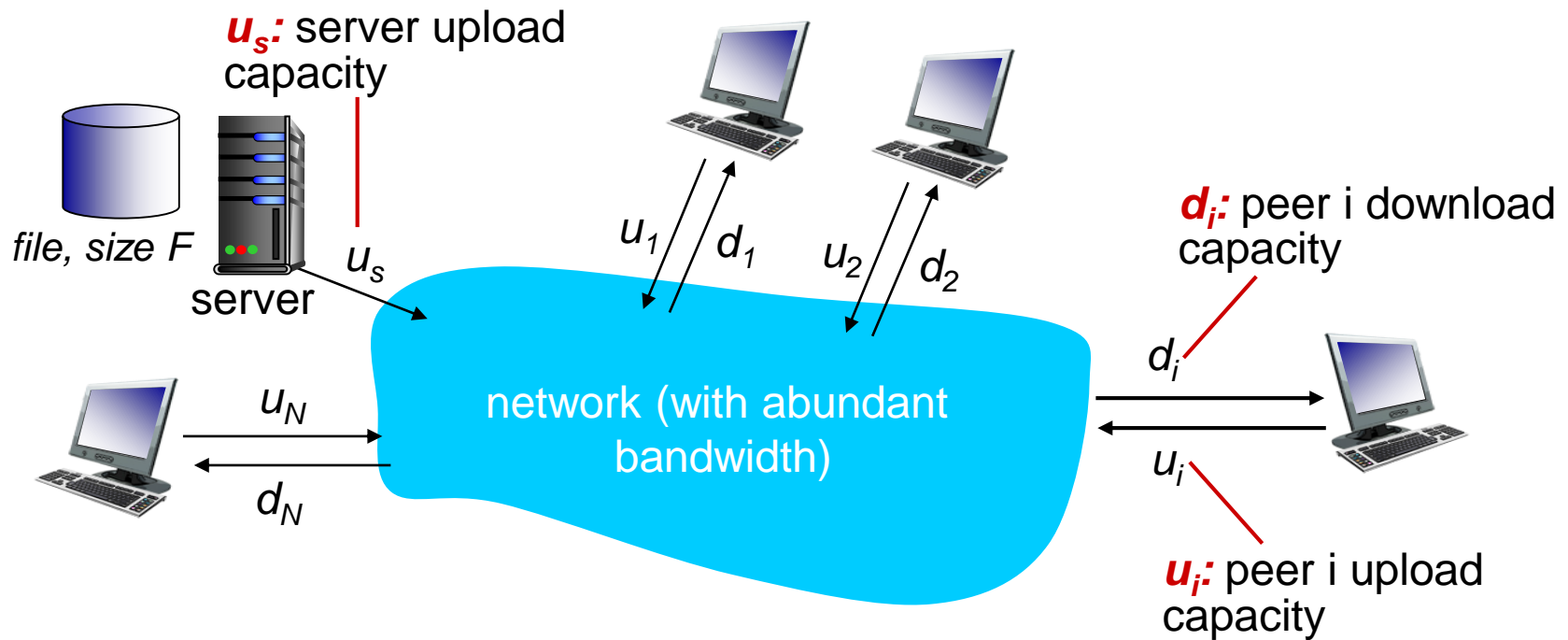
- file distribution (BitTorrent)
- Streaming (KanKan)
- VoIP (Skype)



# 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



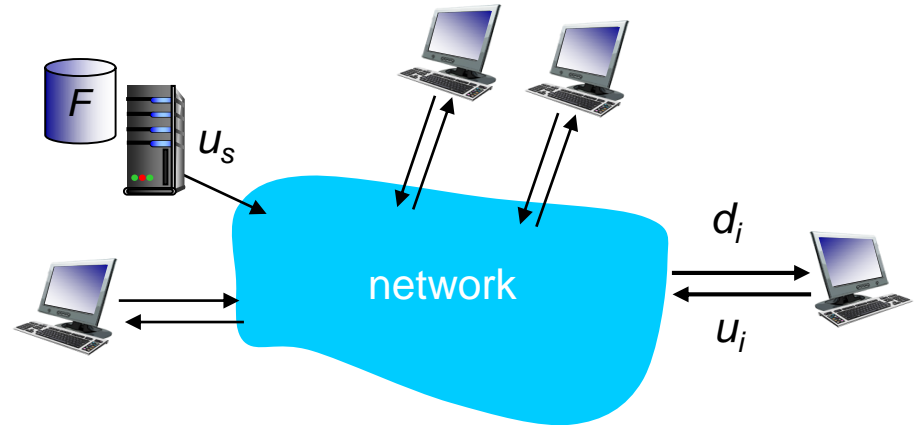
# File distribution time: client-server

- ❖ **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:  $F/d_{\min}$



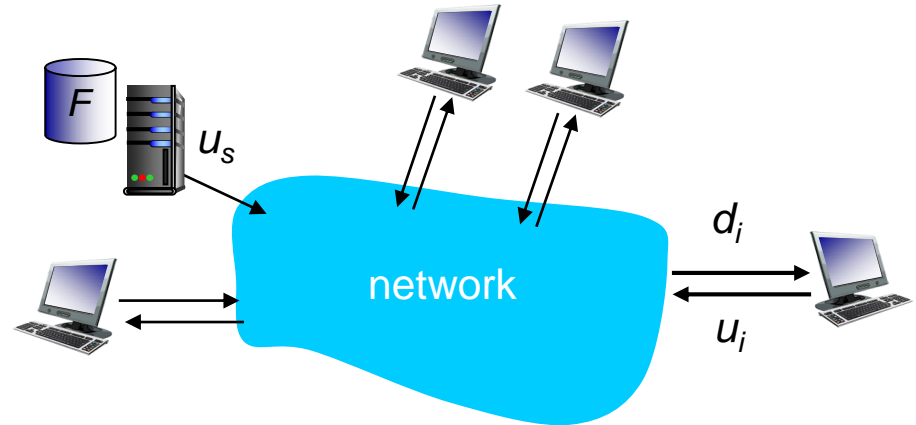
*time to distribute  $F$   
to  $N$  clients using  
client-server approach*

$$D_{c-s} \geq \max\{NF/u_s, F/d_{\min}\}$$

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*

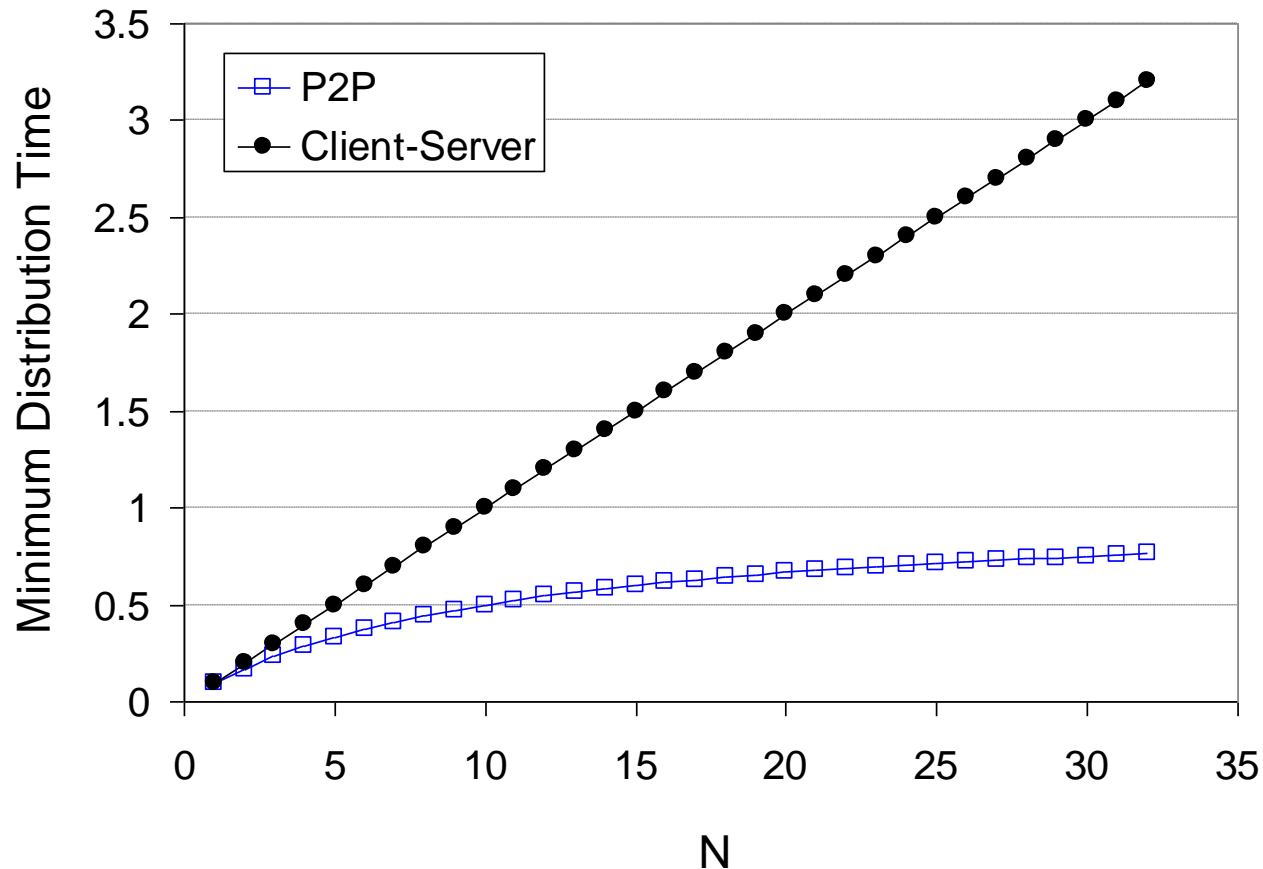
$$D_{P2P} \geq \max\{F/u_s, F/d_{\min}, NF/(u_s + \sum u_i)\}$$

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,  $u_s = 10u$ ,  $d_{min} \geq u_s$

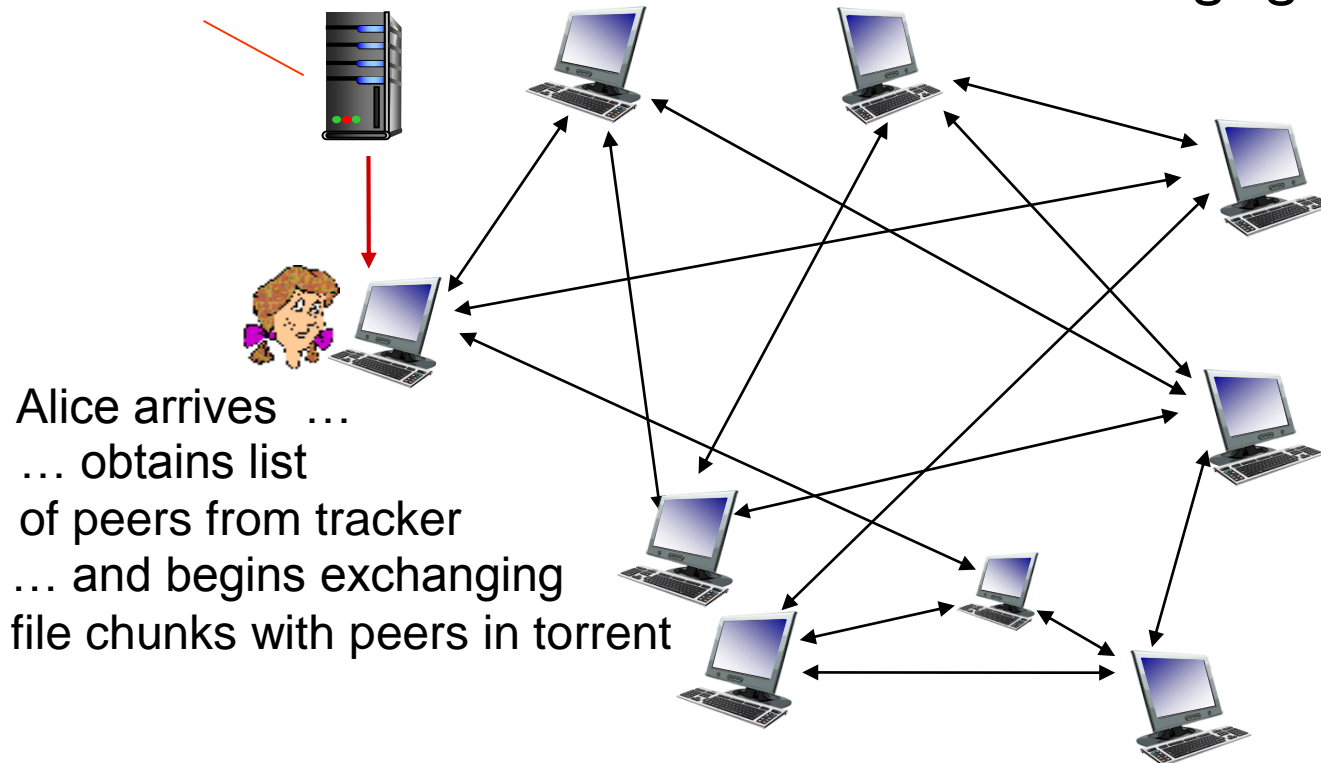


# P2P file distribution: BitTorrent

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

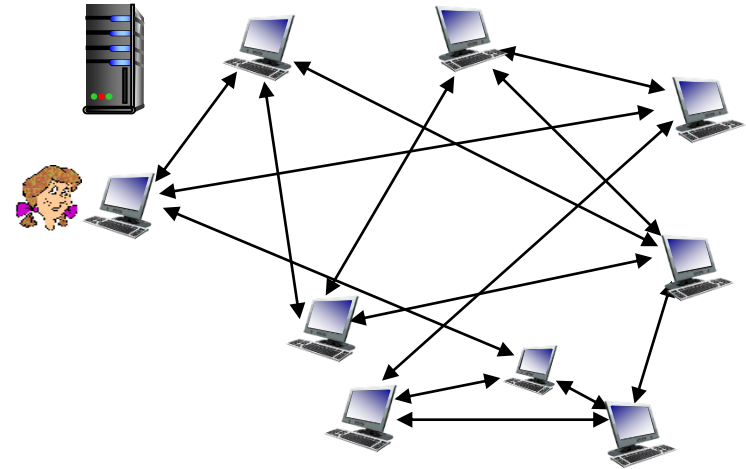
**tracker:** tracks peers participating in torrent

**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:*

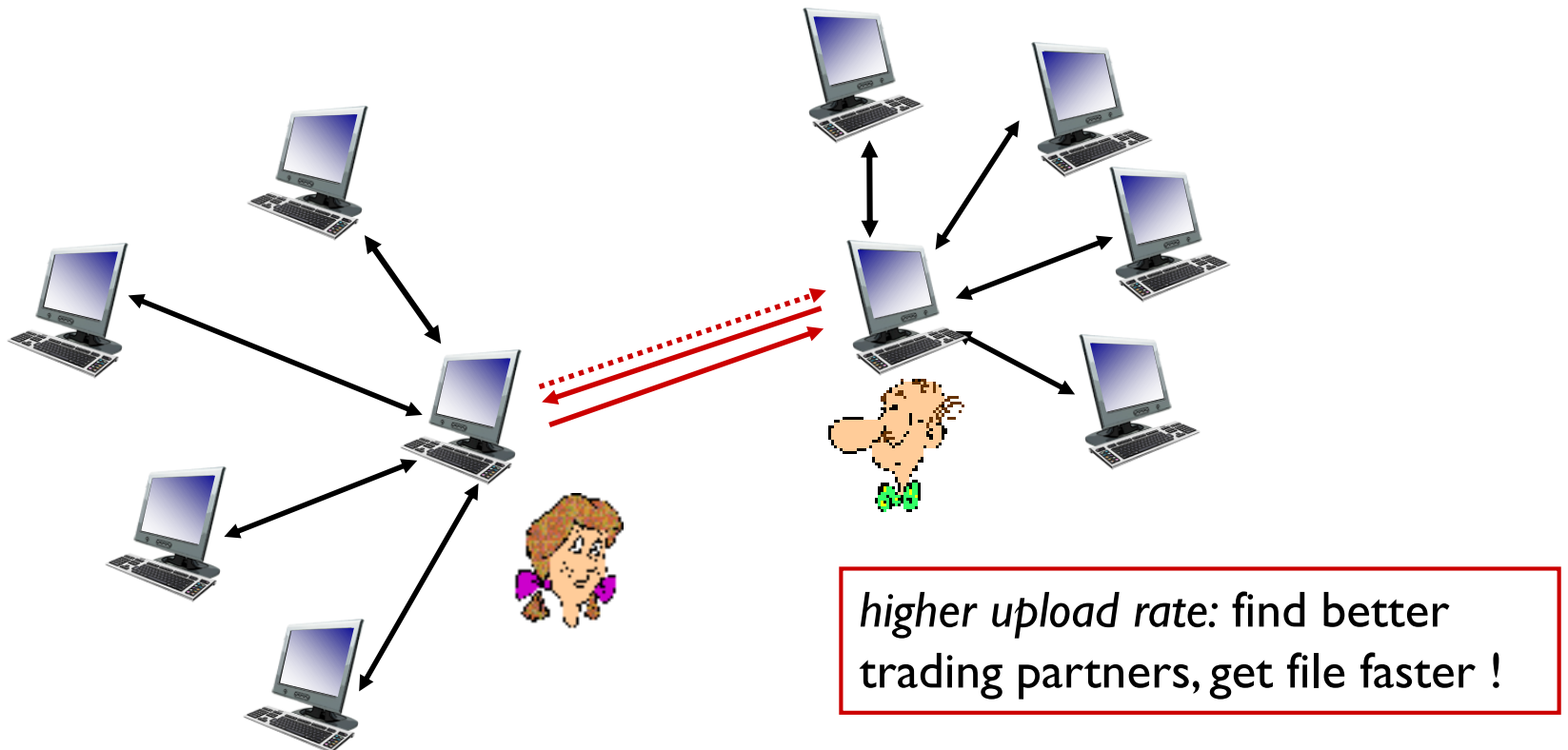
- ❖ at any given time, different peers have different subsets of file chunks
- ❖ periodically, Alice asks each peer for list of chunks that they have
- ❖ Alice requests missing chunks from peers, **rarest first**

## *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



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# Video Streaming and CDNs: context

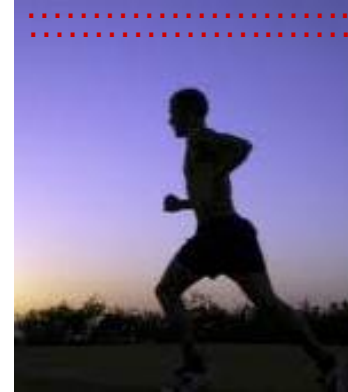
- video traffic: major consumer of Internet bandwidth
  - Netflix, YouTube: 37%, 16% of downstream residential ISP traffic
  - ~1B YouTube users, ~75M Netflix users
- challenge: scale - how to reach ~1B users?
  - single mega-video server won't work (why?)
- challenge: heterogeneity
  - different users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



# Multimedia: video

- ❖ video: sequence of images displayed at constant rate
  - e.g., 24 images/sec
- ❖ digital image: array of pixels
  - each pixel represented by bits
- ❖ coding: use redundancy *within* and *between* images to decrease # bits used to encode image
  - spatial (within image)
  - temporal (from one image to next)

*spatial coding example:* instead of sending  $N$  values of same color (all purple), send only two values: color value (*purple*) and number of repeated values ( $N$ )



frame  $i$

*temporal coding example:* instead of sending complete frame at  $i+1$ , send only differences from frame  $i$

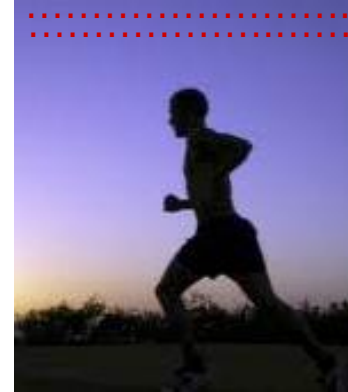


frame  $i+1$

# Multimedia: video

- **CBR: (constant bit rate):**  
video encoding rate fixed
- **VBR: (variable bit rate):**  
video encoding rate changes  
as amount of spatial,  
temporal coding changes
- **examples:**
  - MPEG I (CD-ROM) 1.5 Mbps
  - MPEG2 (DVD) 3-6 Mbps
  - MPEG4 (often used in Internet, < 1 Mbps)

*spatial coding example:* instead of sending  $N$  values of same color (all purple), send only two values: color value (purple) and number of repeated values ( $N$ )



frame  $i$

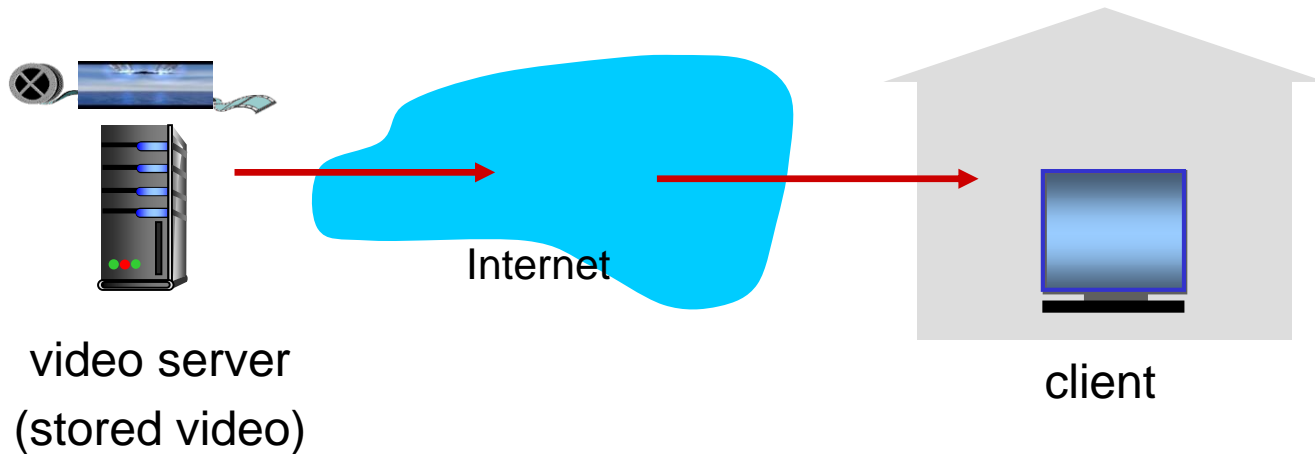
*temporal coding example:*  
instead of sending complete frame at  $i+1$ ,  
send only differences from frame  $i$



frame  $i+1$

# Streaming stored video:

simple scenario:



# Streaming multimedia: DASH

❖ *DASH*: *D*ynamic, *A*daptive *S*treaming over *H*TTP

❖ *server*:

- divides video file into multiple chunks
- each chunk stored, encoded at different rates
- *manifest file*: provides URLs for different chunks

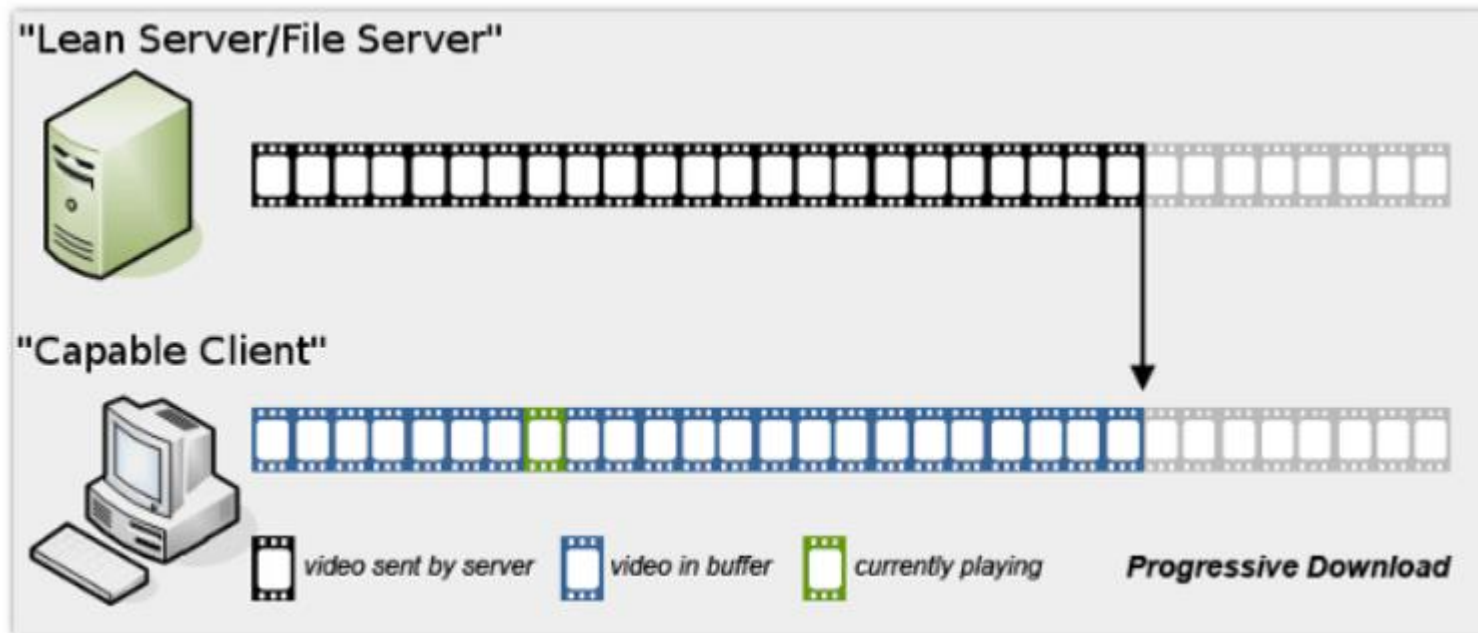
❖ *client*:

- 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)



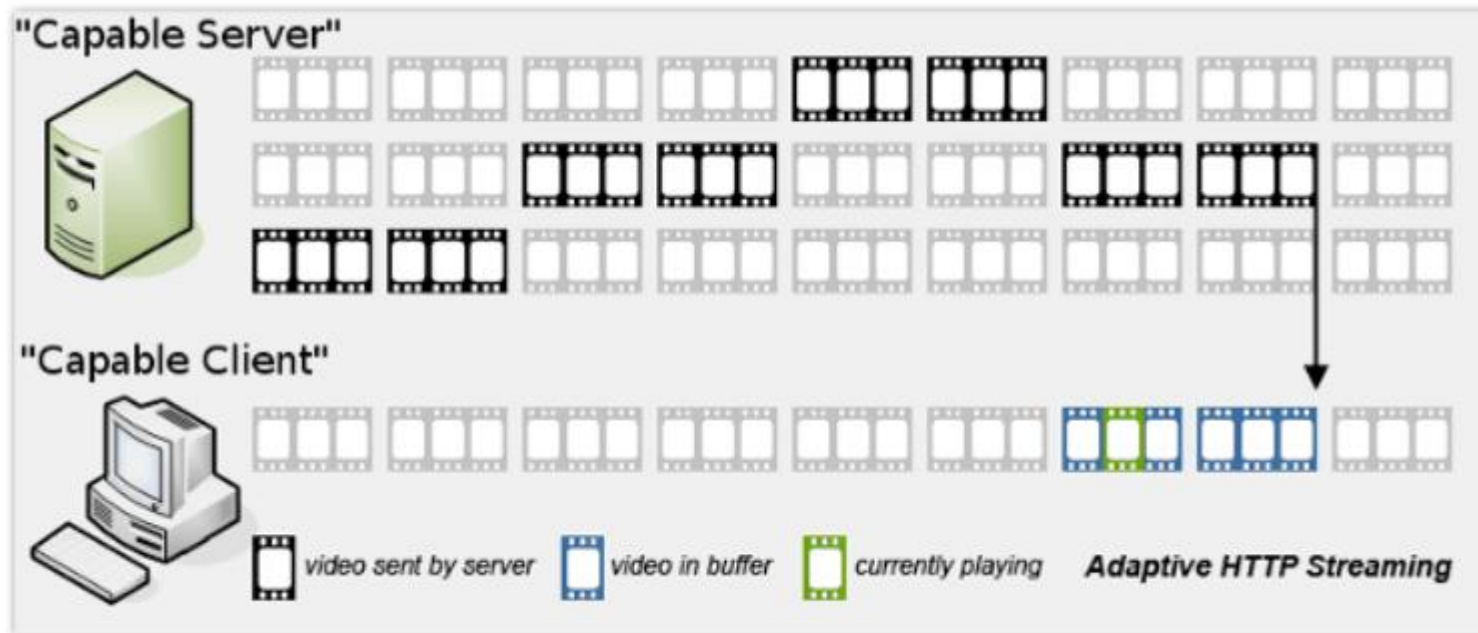
# Streaming multimedia: DASH

## Progressive Download



# Streaming multimedia: DASH

## Adaptive HTTP Streaming



# 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

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❖ *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*

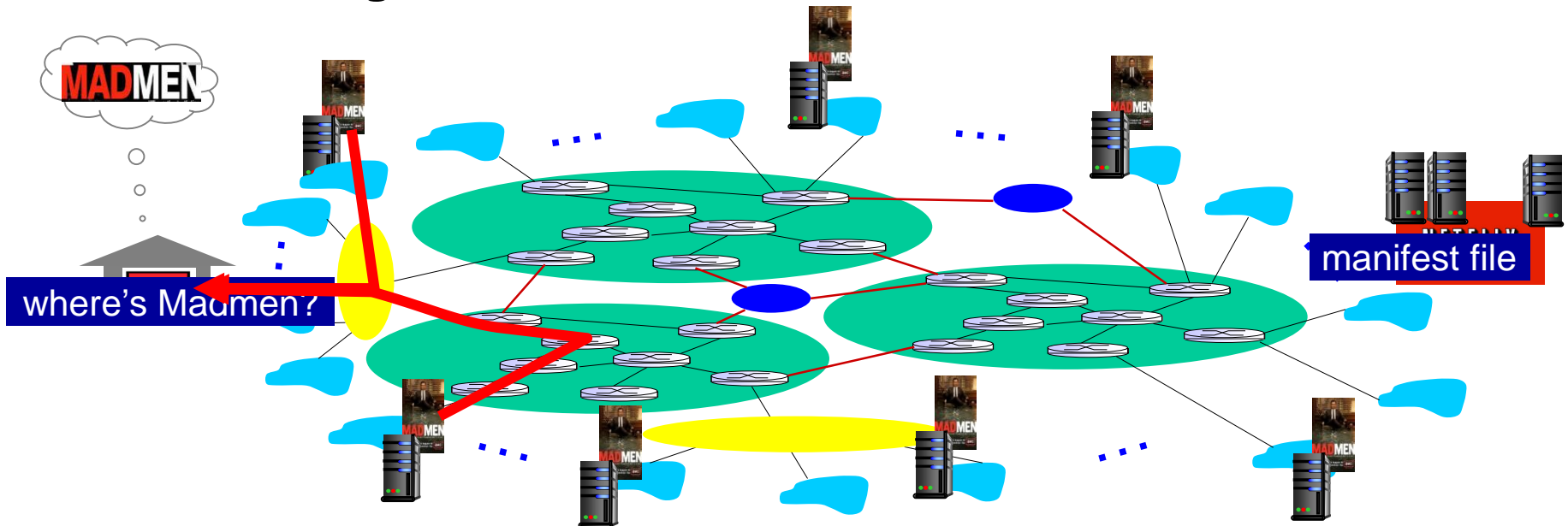
# Content distribution networks

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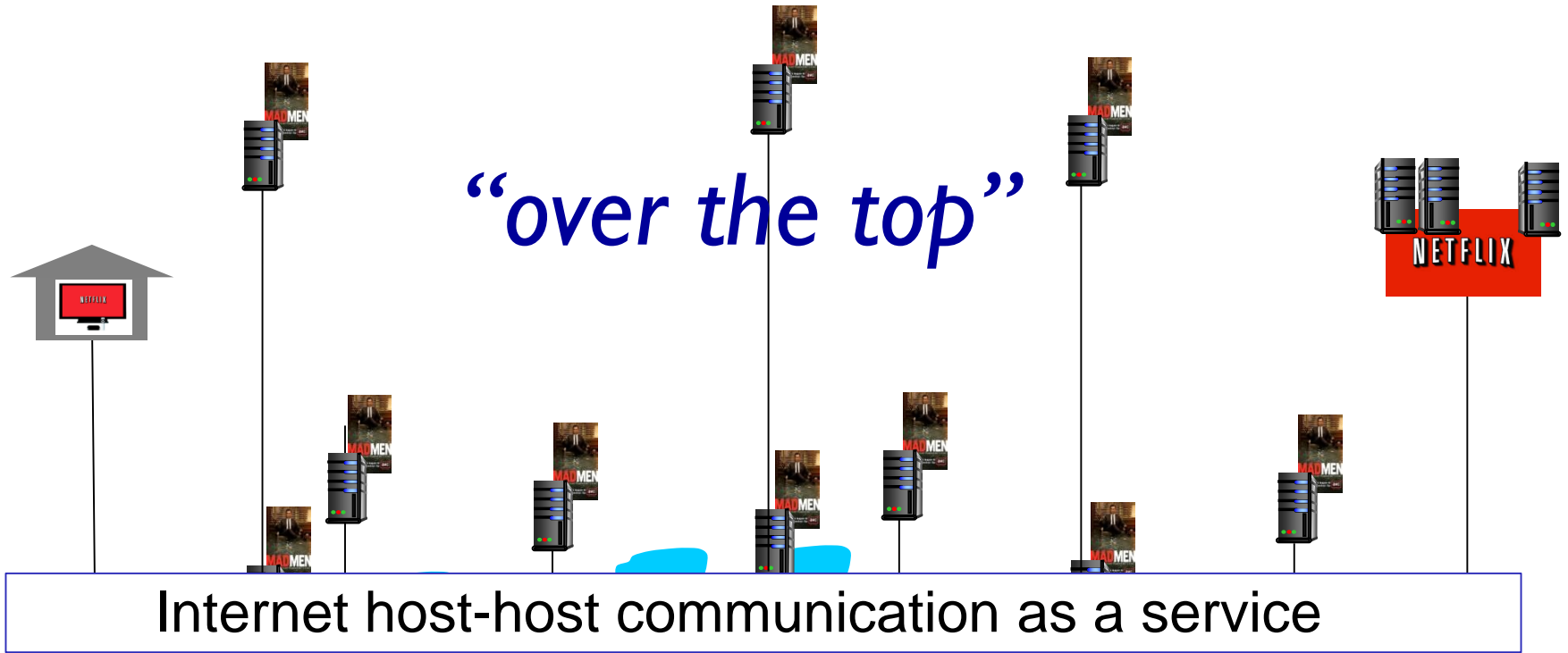
- ❖ *challenge*: how to stream content (selected from millions of videos) to hundreds of thousands of simultaneous users?
- ❖ *option 2*: store/serve multiple copies of videos at multiple geographically distributed sites (*CDN*)
  - *enter deep*: push CDN servers deep into many access networks
    - 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)

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



# Content Distribution Networks (CDNs)



**OTT challenges:** coping with a congested Internet

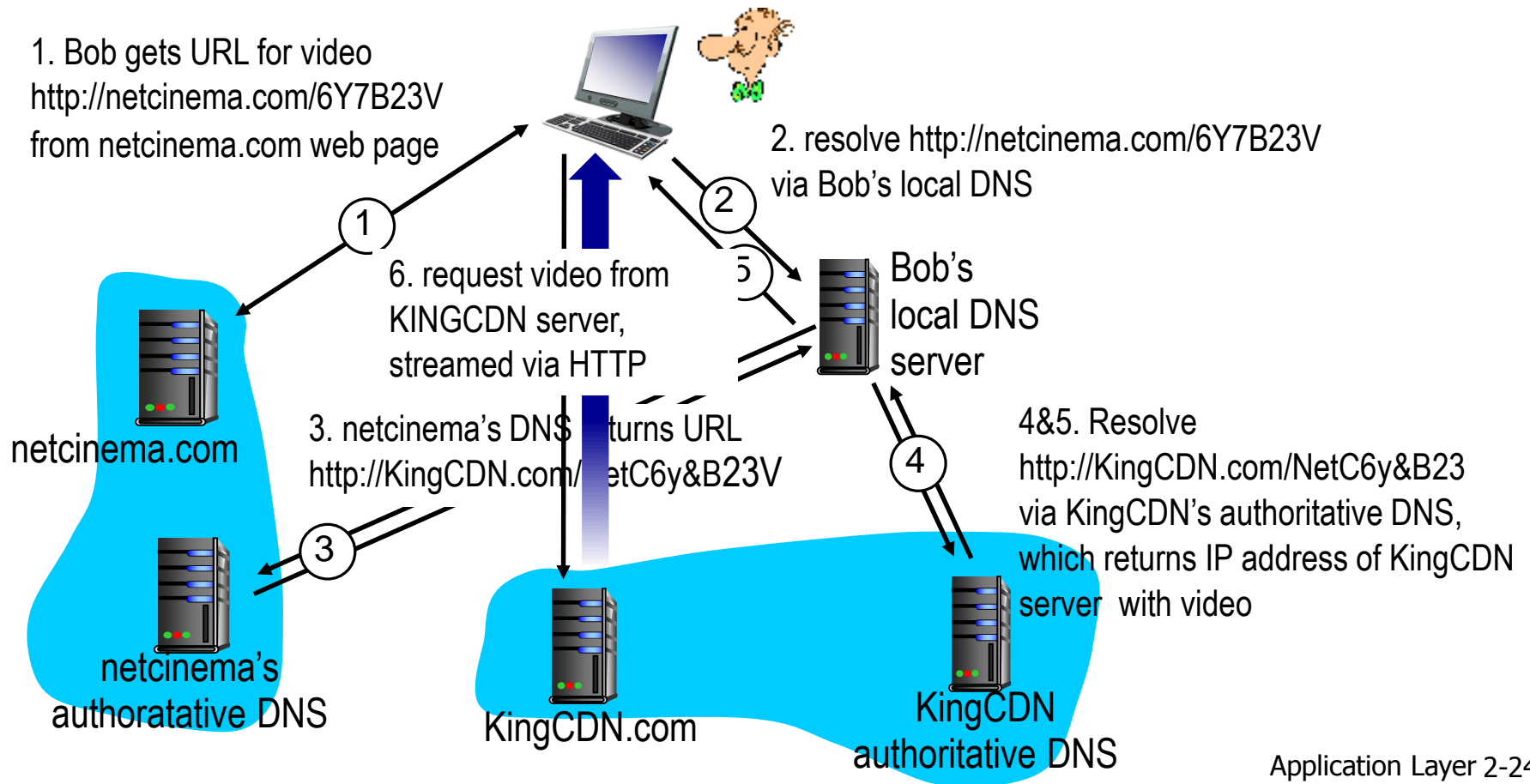
- from which CDN node to retrieve content?
- viewer behavior in presence of congestion?
- what content to place in which CDN node?

*more .. in chapter 7*

# CDN content access: a closer look

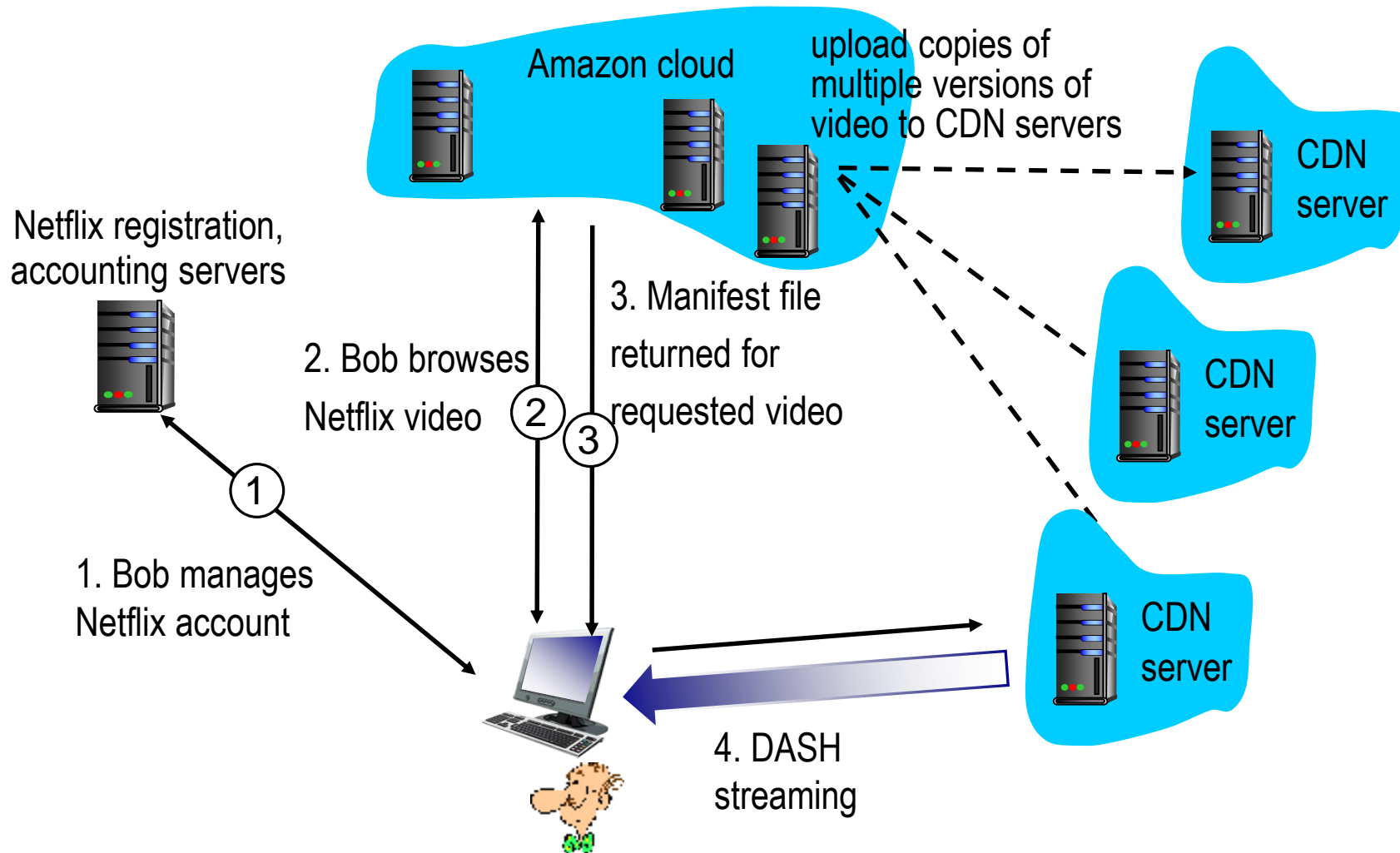
Bob (client) requests video `http://netcinema.com/6Y7B23V`

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





# Case study: Netflix



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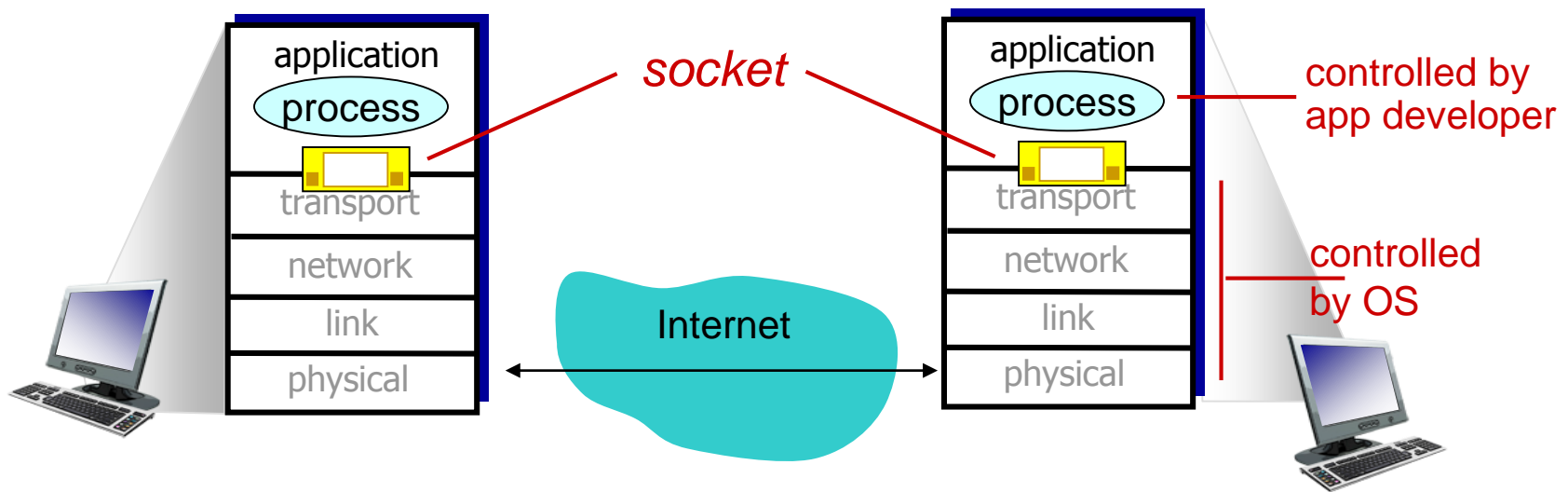
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2.8 socket programming with UDP and TCP

# 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 the data to the server.
2. The server receives the data and converts characters to uppercase.
3. The server sends the modified data to the client.
4. The client receives the modified data and displays the line on its screen.

# Socket programming *with* UDP

UDP: no “connection” between client & server

- ❖ no handshaking before sending data
- ❖ sender explicitly attaches IP destination address and port # to each packet
- ❖ rcvr 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

# Client/server socket interaction: UDP

## server (running on serverIP)

create socket, port= x:  
`serverSocket =  
socket(AF_INET,SOCK_DGRAM)`

↓  
read datagram from  
`serverSocket`

↓  
write reply to  
`serverSocket`  
specifying  
client address,  
port number

## client

create socket:  
`clientSocket =  
socket(AF_INET,SOCK_DGRAM)`

↓  
Create datagram with server IP and  
port=x; send datagram via  
`clientSocket`

↓  
read datagram from  
`clientSocket`

↓  
close  
`clientSocket`

# Example app: UDP client

## *Python UDPClient*

include Python's socket  
library

→ from socket import \*

serverName = '192.168.xxx.xxx'

serverPort = 12000

create UDP socket for  
server

→ clientSocket = socket(AF\_INET,  
SOCK\_DGRAM)

get user keyboard  
input

→ m = input('Input lowercase sentence:')

Attach server name, port to  
message; send into socket

→ clientSocket.sendto(m.encode(),(serverName, serverPort))

read reply characters from  
socket into string

→ modifiedMessage, serverAddress =  
clientSocket.recvfrom(2048)

print out received string  
and close socket

→ print (modifiedMessage.decode())  
clientSocket.close()

# Example app: UDP server

## *Python UDPServer*

```
from socket import *
```

```
serverPort = 12000
```

create UDP socket → 

```
serverSocket = socket(AF_INET, SOCK_DGRAM)
```

bind socket to local port  
number 12000 → 

```
serverSocket.bind(("", serverPort))
```

```
print ("The server is ready to receive")
```

loop forever → 

```
while 1:
```

Read from UDP socket into  
message, getting client's  
address (client IP and port) → 

```
message, clientAddress = serverSocket.recvfrom(2048)
```

```
modifiedMessage = message.upper()
```

send upper case string  
back to this client → 

```
serverSocket.sendto(modifiedMessage, clientAddress)
```



# 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
  - source port numbers used to distinguish clients (more in Chap 3)

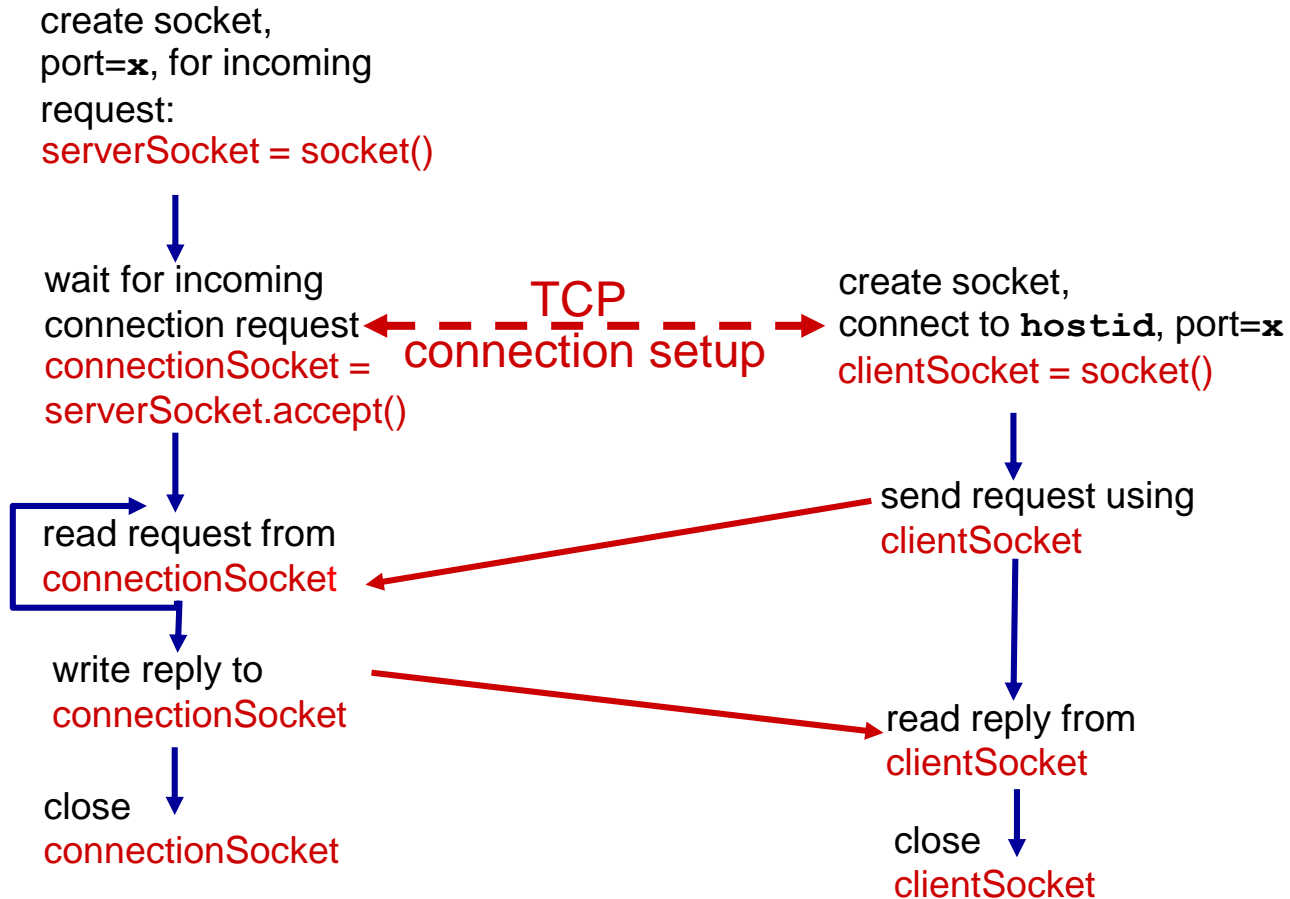
## **application viewpoint:**

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

# Client/server socket interaction: TCP

server (running on `hostid`)

client



# Example app:TCP client

## *Python TCPClient*

```
from socket import *
```

```
serverName = '192.168.xxx.xxx'
```

```
serverPort = 12000
```

create TCP socket for  
server, remote port 12000

```
→ clientSocket = socket(AF_INET, SOCK_STREAM)
```

```
clientSocket.connect((serverName,serverPort))
```

```
sentence = input('Input lowercase sentence:')
```

No need to attach server  
name, port

```
→ clientSocket.send(sentence.encode())
```

```
modifiedSentence = clientSocket.recv(1024)
```

```
print ('From Server: ', modifiedSentence.decode())
```

```
clientSocket.close()
```

# Example app: TCP server

## *Python TCPServer*

create TCP welcoming socket	→	<pre>from socket import * serverPort = 12000 serverSocket = socket(AF_INET, SOCK_STREAM) serverSocket.bind(('', serverPort)) serverSocket.listen(1) print('The server is ready to receive') while 1:     connectionSocket, addr = serverSocket.accept()     sentence = connectionSocket.recv(1024)     capitalizedSentence = sentence.upper()     connectionSocket.send(capitalizedSentence)     connectionSocket.close()</pre>
server begins listening for incoming TCP requests	→	
loop forever	→	
server waits on accept() for incoming requests, new socket created on return	→	
read bytes from socket (but not address as in UDP)	→	
close connection to this client (but <i>not</i> welcoming socket)	→	

# Chapter 2: summary

*our study of network apps now complete!*

- application architectures
  - client-server
  - P2P
- application service requirements:
  - reliability, bandwidth, delay
- Internet transport service model
  - connection-oriented, reliable: TCP
  - unreliable, datagrams: UDP
- specific protocols:
  - HTTP
  - FTP
  - SMTP, POP, IMAP
  - DNS
  - P2P: BitTorrent
- video streaming, CDNs
- socket programming:  
TCP, UDP sockets

# Chapter 2: summary

*most importantly: learned about protocols!*

- ❖ typical request/reply message exchange:
  - client requests info or service
  - server responds with data, status code
- ❖ message formats:
  - **headers**: fields giving info about data
  - **data**: info being communicated

## *important themes:*

- ❖ control vs. data msgs
  - in-band, out-of-band
- ❖ centralized vs. decentralized
- ❖ stateless vs. stateful
- ❖ reliable vs. unreliable msg transfer
- ❖ “complexity at network edge”