

画像情報特論 (2)

Advanced Image Information (2)

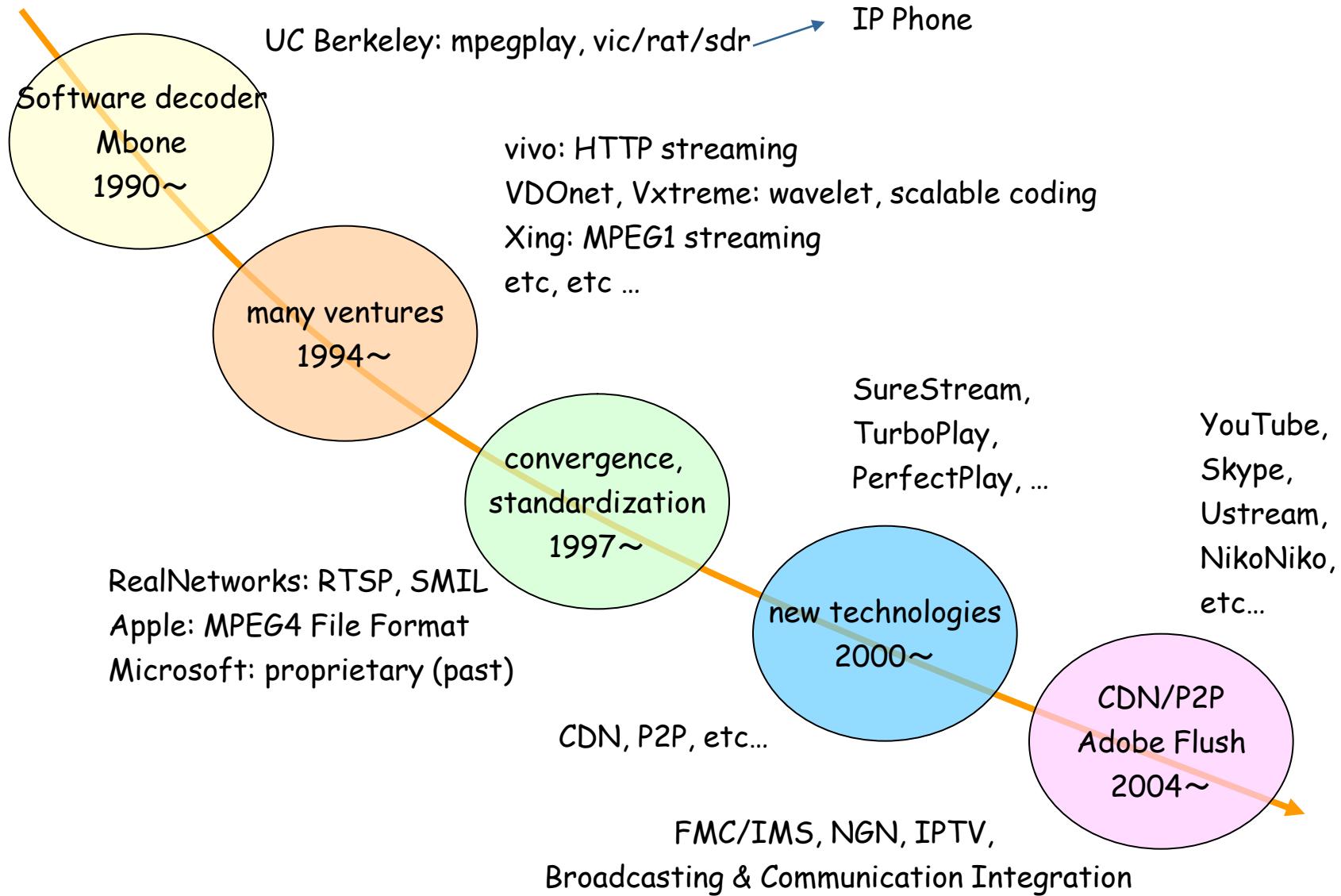
TCP/IP for Video Streaming

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Background

History of Streaming



Protocol Stack for Streaming

- Network Architecture

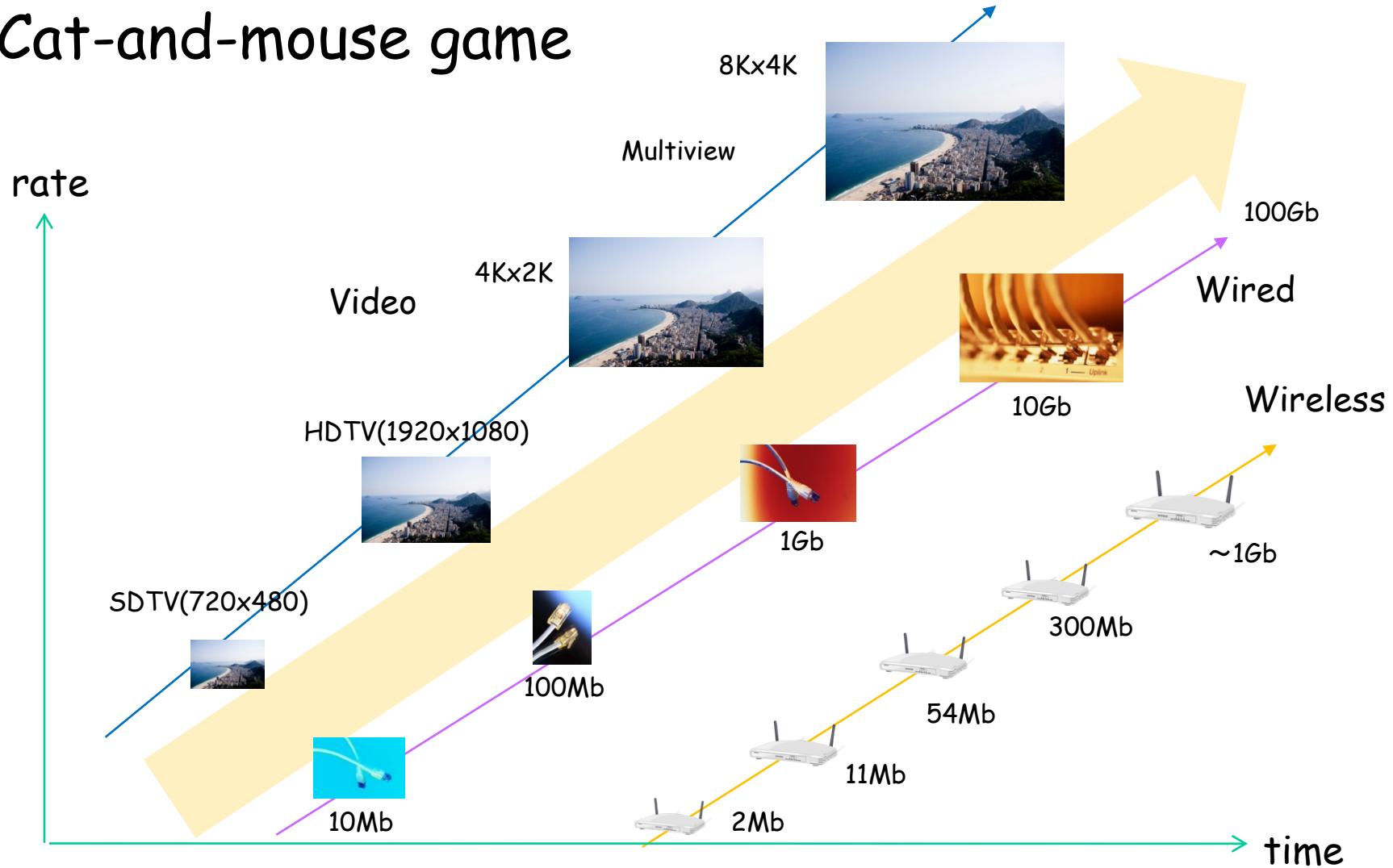
VoIP, IPTV and streaming shares almost common protocol stack

application (L7)	video (H.264 etc...)	audio	SDP	layout (HTML, SMIL)
adaptation	RTP / RTCP	RTSP, SIP, SAP*	HTTP	
transport (L4)	UDP / TCP / DCCP	TCP / UDP / SCTP		
network (L3)	IP (IPv4, IPv6, IP-multicast)			
datalink & physical (L2 & L1)	actual networks (802.3 (ethernet), 802.11 (WiFi), etc)			

* SAP: delivered by IP-multicast for program advertisement

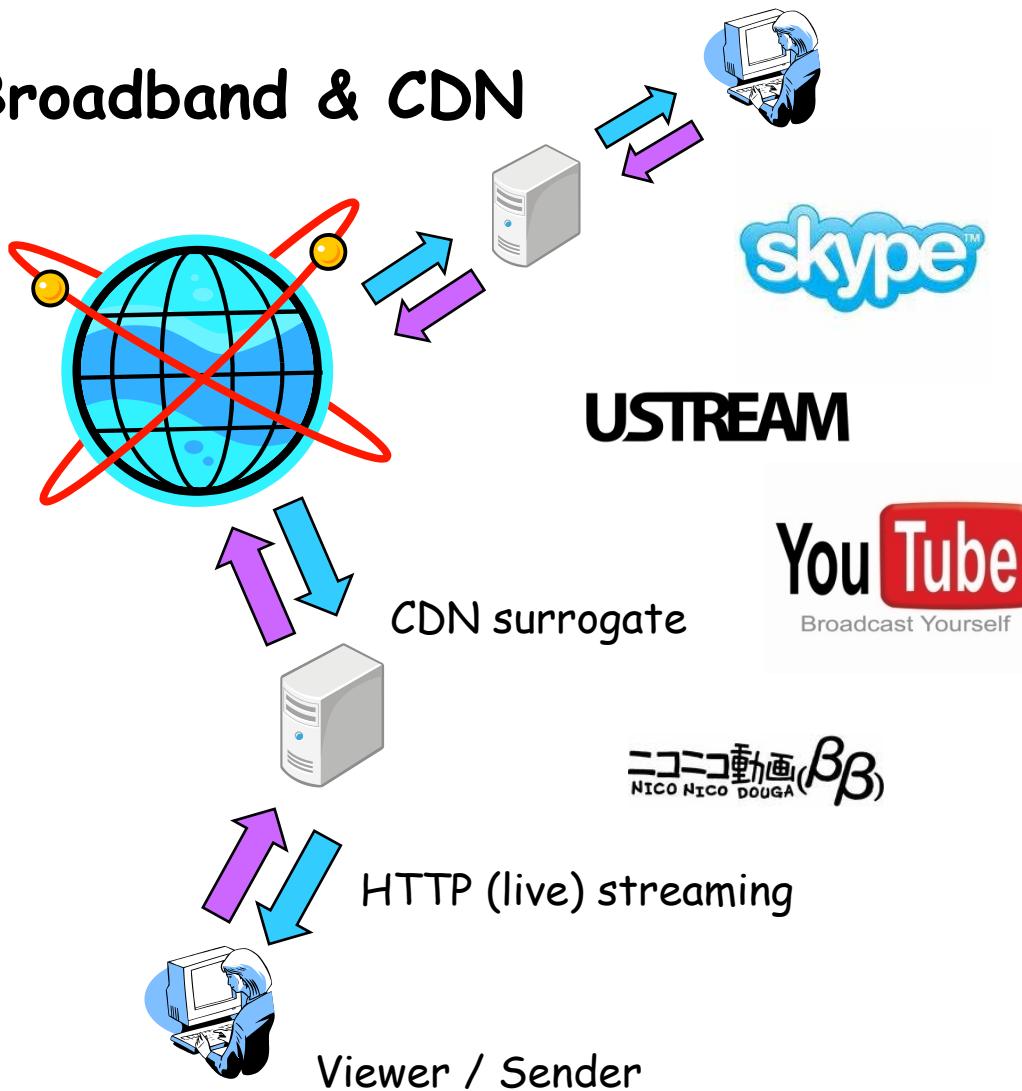
Networks and Multimedia

- Cat-and-mouse game



Wired Networks

Broadband & CDN



RTP/UDP & RTSP & TFRC

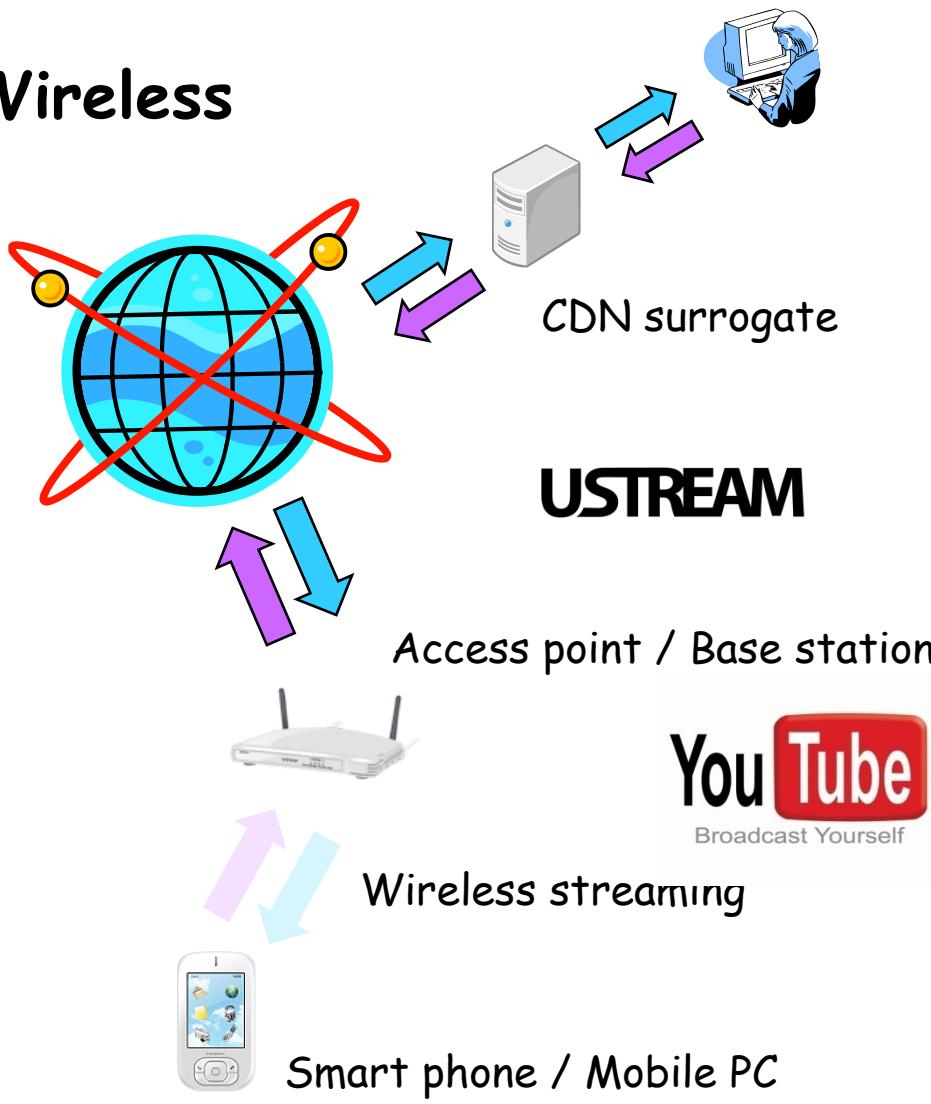
→ HTTP/TCP streaming

- Broadband
- CDN (Akamai, Lime Networks)
- Firewall (port 80)
- ...

One-way (on-demand / live)
Bi-directional (interactive)

Wireless Networks

Wireless



Wireless specific problems

- Wireless LAN: IEEE 802.11
- Cellular: 3G, LTE, 4G
- WiMAX: IEEE 802.16
- Home Networks: DLNA
- (Satellite)
- ...

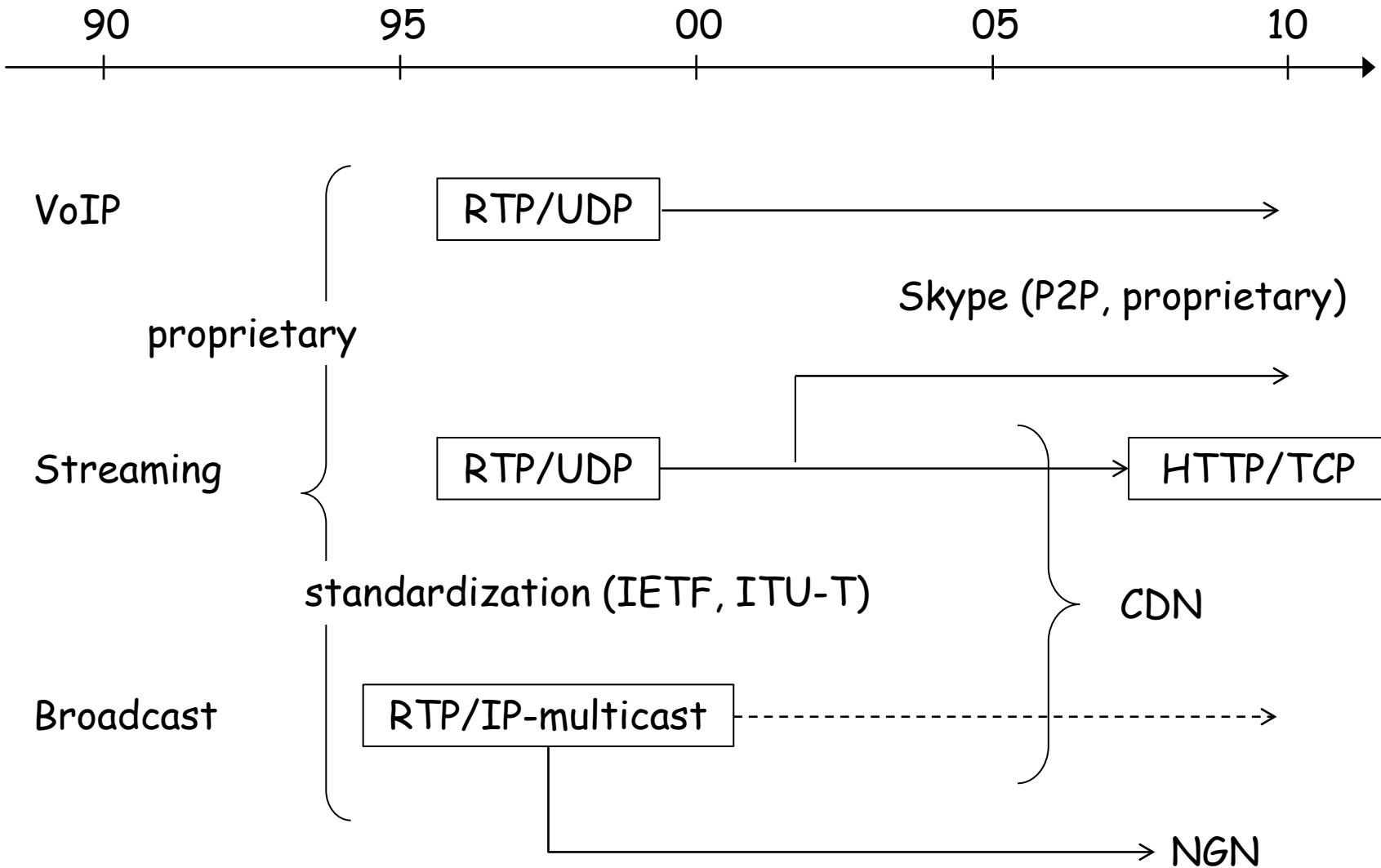
- **Wireless issues**

random errors, collisions,
interference, delay increase

- **Multi-hop issues**

severe interference, lower
throughput ⁷ and higher delay

Protocol Transition

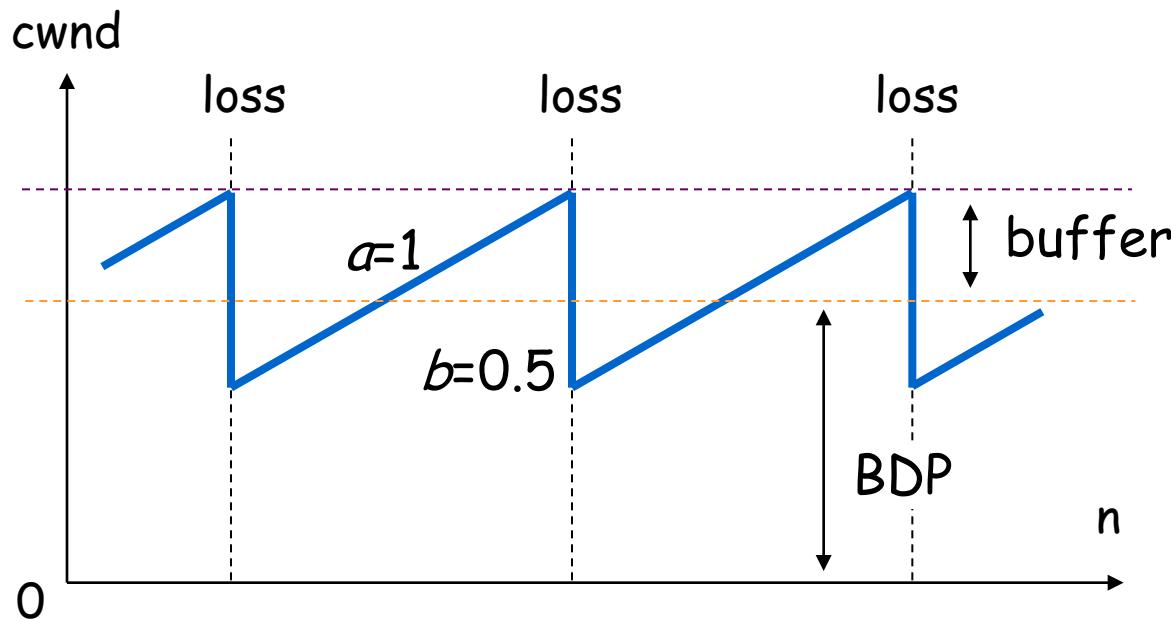


Overview

- Multimedia Delivery Techniques over IP Networks
- Broadband Wired Network
 - Transport techniques with efficiency, friendliness and low-delay properties
 - TCP and TFRC
- Wireless Network
 - Transport techniques towards low-delay transmission over wireless networks
 - Mobility
 - Multi-hop
 - Underwater sensor networks
- Network Simulators and Emulators

TCP Variants

TCP-Reno (loss-based)

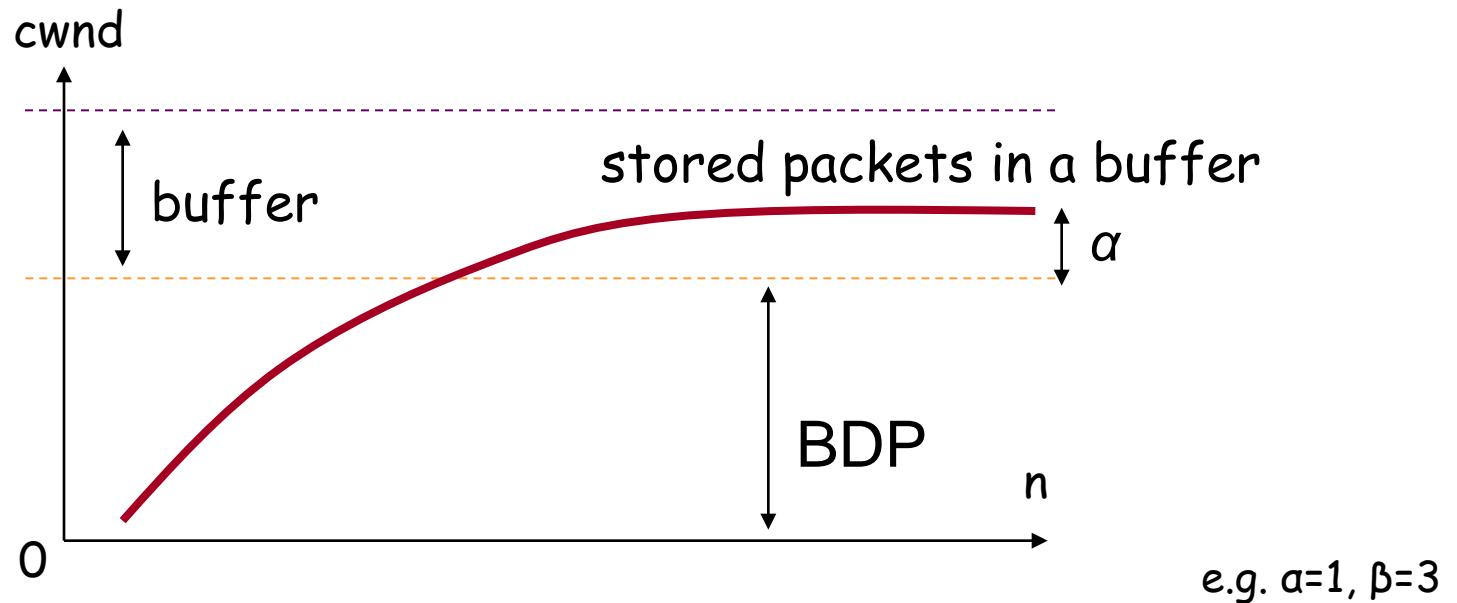


increase: $cwnd = cwnd + 1/cwnd$

decrease: $cwnd = cwnd / 2$

AIMD: additive increase multiplicative decrease

TCP-Vegas (delay-based)



$$diff = \left(\frac{cwnd}{RTT_{min}} - \frac{cwnd}{RTT} \right) \cdot RTT_{min}$$

stored packets in a buffer

increase:

$$cwnd = \begin{cases} cwnd + 1 & diff < \alpha \\ cwnd & otherwise \\ cwnd - 1 & diff > \beta \end{cases}$$

decrease:

$$cwnd = cwnd * 0.75$$

TCP problems ten years ago

- broadband wired networks
 - slow window increase (inefficiency)
 - deployment of wireless networks
 - cannot distinguish wireless errors and buffer overflow
-
- TCP-Reno (NewReno, SACK) problem
 - Reno expels Vegas (unfriendliness)

TCP Variants in the 21th century

- **Loss-driven (AIMD)**
 - TCP-Reno / NewReno / SACK
 - High-Speed TCP (IETF RFC 3649, Dec 2003)
 - Scalable TCP (PFLDnet 2003)
 - BIC-TCP / **CUBIC-TCP** (IEEE INFOCOM 2004, PFLDnet 2005) ... Linux
 - H-TCP (PFLDnet 2004)
 - TCP-Westwood (ACM MOBICOM 2001)
- **Delay-driven (RTT Observation)**
 - TCP-Vegas (IEEE JSAC, Oct 1995)
 - FAST-TCP (INFOCOM 2004)
- **Hybrid**
 - Gentle High-Speed TCP (PfHSN 2003)
 - TCP-Africa (IEEE INFOCOM 2005)
 - **Compound TCP** (PFLDnet 2006) ... Windows
 - Adaptive Reno (PFLDnet 2006)
 - YeAH-TCP (PFLDnet 2007)
 - TCP-Fusion (PFLDnet 2007)

Loss-based TCPs

	<i>a</i>	<i>b</i>
Variants	Increase / Update	Decrease
TCP-Reno	1	0.5
aggressive	HighSpeed TCP (HS-TCP)	$a(w) = \frac{2w^2 \cdot b(w) \cdot p(w)}{2 - b(w)}$ <p>e.g. 70 (10Gbps, 100ms)</p> $b(w) = \frac{\log(w) - \log(W_{low})}{\log(W_{high}) - \log(W_{low})} (b_{high} - 0.5) + 0.5$ <p>e.g. 0.1 (10Gbps, 100ms)</p>
	Scalable TCP (STCP)	0.01 (per every ACK)
	BIC-TCP	$\begin{cases} \text{additive increase (fast)} \\ \text{binary search (slow)} \\ \text{max probing (fast)} \end{cases}$
	CUBIC-TCP	$w = 0.4(t - \sqrt[3]{2W_{max}})^3 + W_{max}$
adaptive	H-TCP	$\alpha \leftarrow 2(1 - \beta)\{1 + 10.5 \cdot (t - TH)\}$ $\beta \leftarrow \begin{cases} 0.5 & \text{for } \left \frac{B(k+1) - B(k)}{B(k)} \right > 2 \\ \frac{RTT_{min}}{RTT_{max}} & \text{otherwise} \end{cases}$
	TCP-Westwood (TCPW)	1 $\begin{cases} RE * RTT_{min} / PS & (\text{not congested}) \\ BE * RTT_{min} / PS & (\text{congested}) \end{cases}$

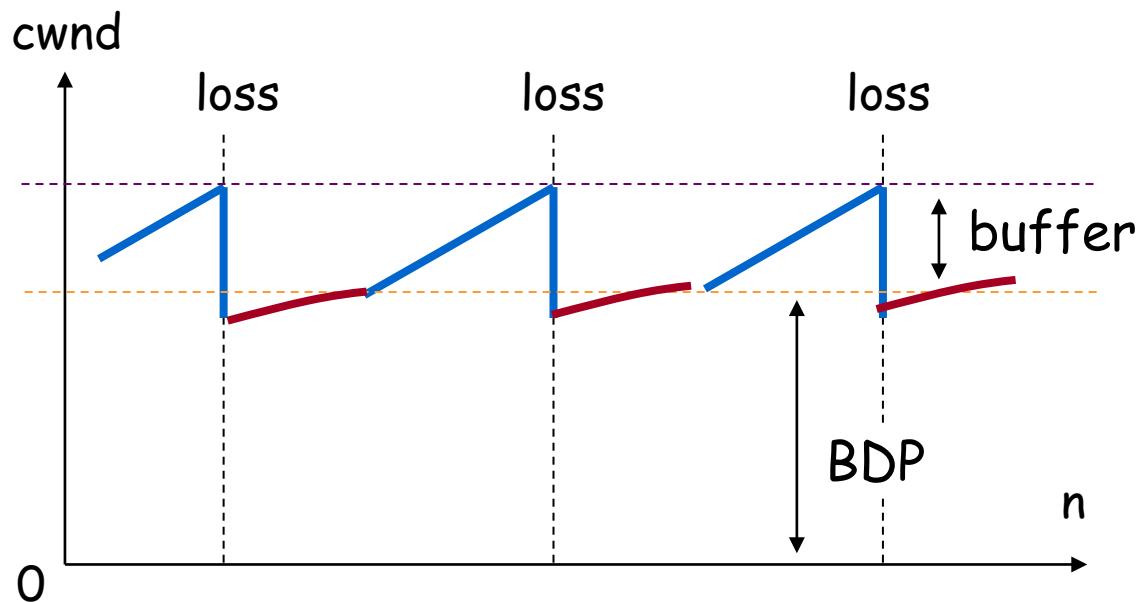
Delay-based TCPs

a

b

Variants	Update	Decrease
TCP-Vegas	$w \leftarrow \begin{cases} w+1 & (\text{no congestion}) \\ w & (\text{stable}) \\ w-1 & (\text{early congestion}) \end{cases}$	0.75
FAST-TCP	$w \leftarrow \min \left\{ 2w, (1-\gamma)w + \gamma \left(\frac{RTT_{\min}}{RTT} w + \alpha \right) \right\}$	0.5 (?)

Hybrid TCP



- RTT increase: loss mode \Rightarrow improvement of friendliness
- no RTT increase: delay mode \Rightarrow improvement of efficiency

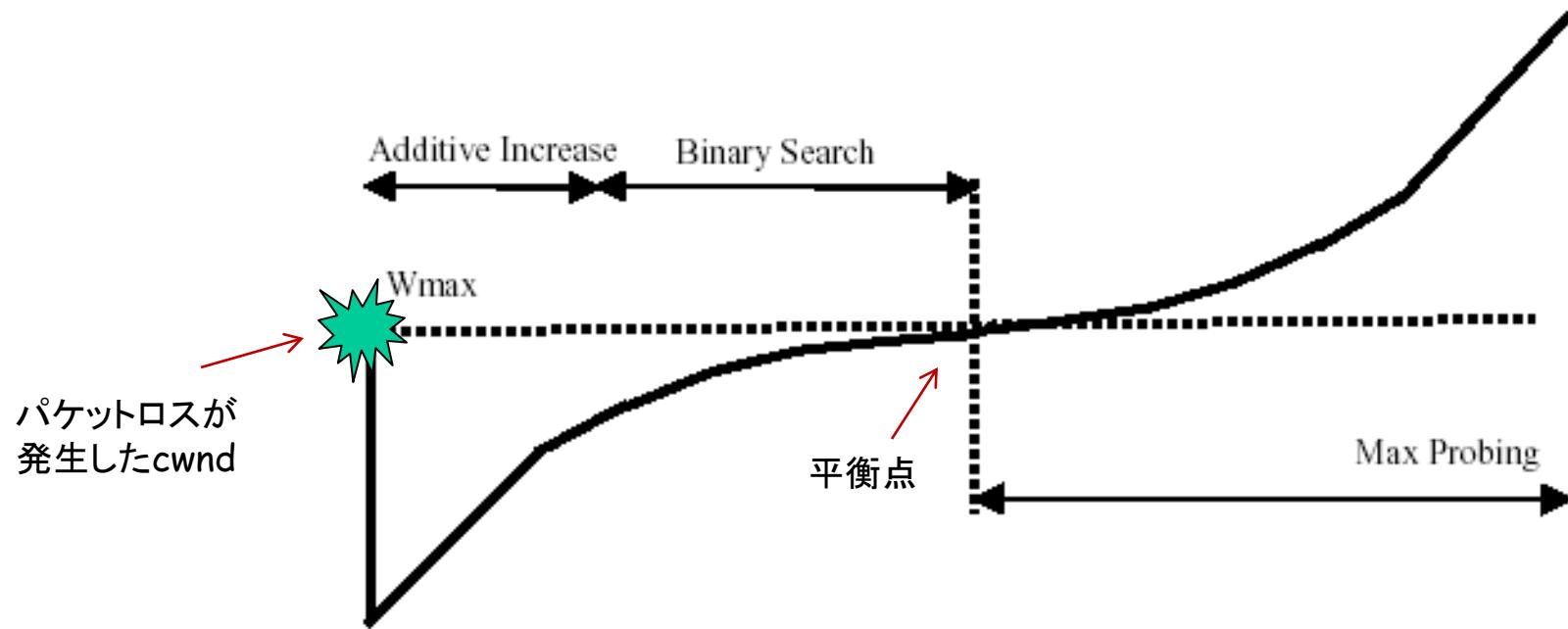
Hybrid TCPs

	<i>a</i>	<i>b</i>
Variants	Increase	Decrease
simple	Gentle HS-TCP TCP-Africa Compound TCP (CTCP)	HS-TCP / Reno HS-TCP / Reno $0.125 \cdot cwnd^{0.75} / \text{Reno}$
adaptive	Adaptive Reno (ARENO) YeAH-TCP TCP-Fusion	$\begin{cases} 1 & (\text{non congestion loss}) \\ 0.5 & (\text{congestion loss}) \end{cases}$ $\max\left(\frac{RTT_{\min}}{RTT}, 0.5\right)$ $\max\left(\frac{RTT_{\min}}{RTT}, 0.5\right)$

CUBIC-TCP

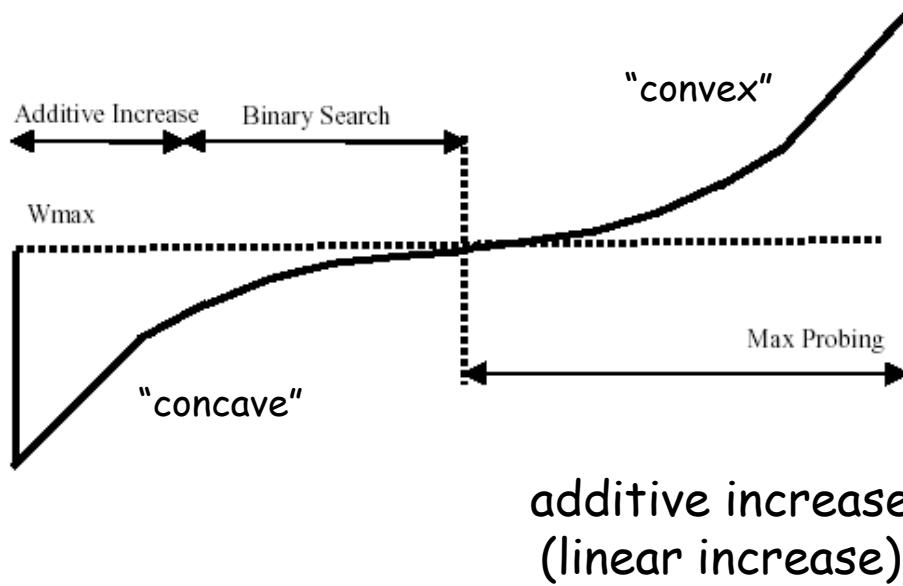
BIC-TCP (1)

- Binary Increase Congestion Control



BIC-TCP (2)

- Window Increase



W_{max} : cwnd when a last loss happened

S_{max} : maximum increase rate (e.g. 32)

S_{min} : minimum increase rate (e.g. 0.01)

```

binary search
if (cwnd < Wmax )
  Winc = (Wmax - cwnd) / 2;
else
  Winc = (cwnd - Wmax) / 2;

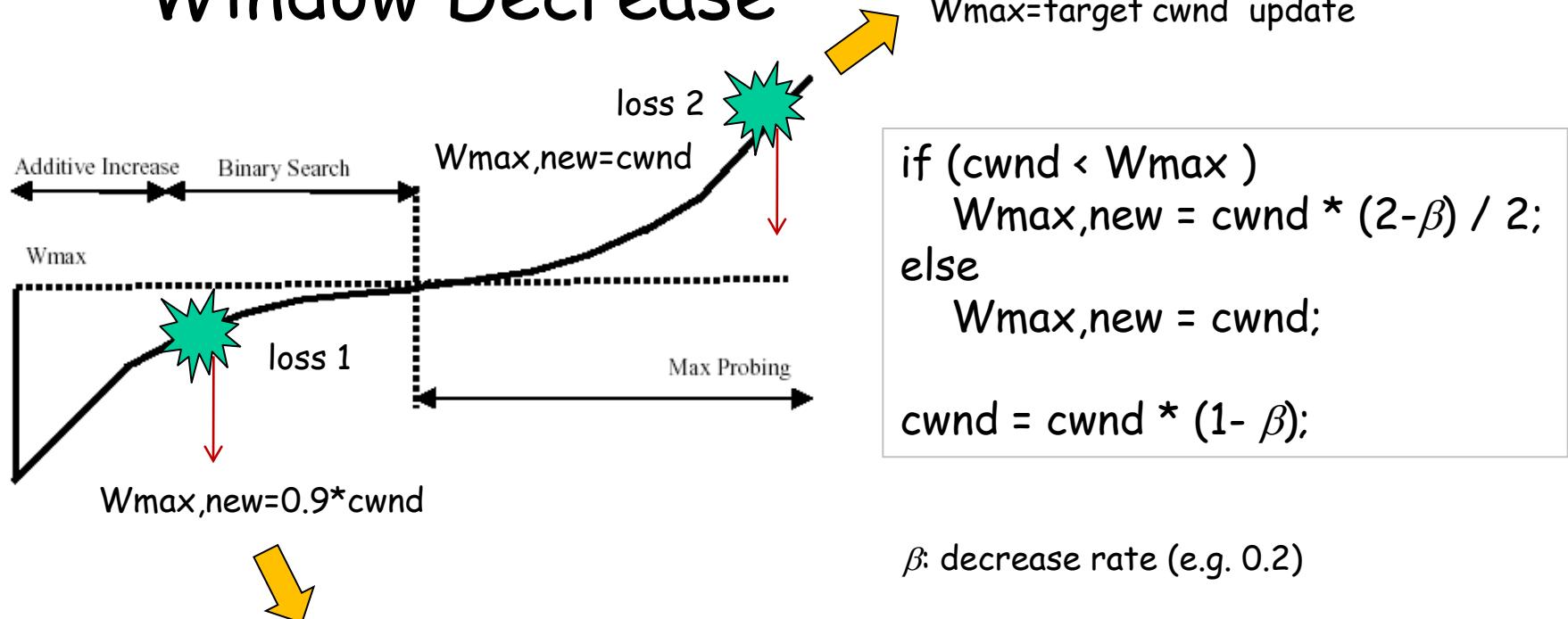
if (Winc > Smax)
  Winc = Smax;
elseif (Winc < Smin)
  Winc = Smin;

cwnd = cwnd + Winc / cwnd;

```

BIC-TCP (3)

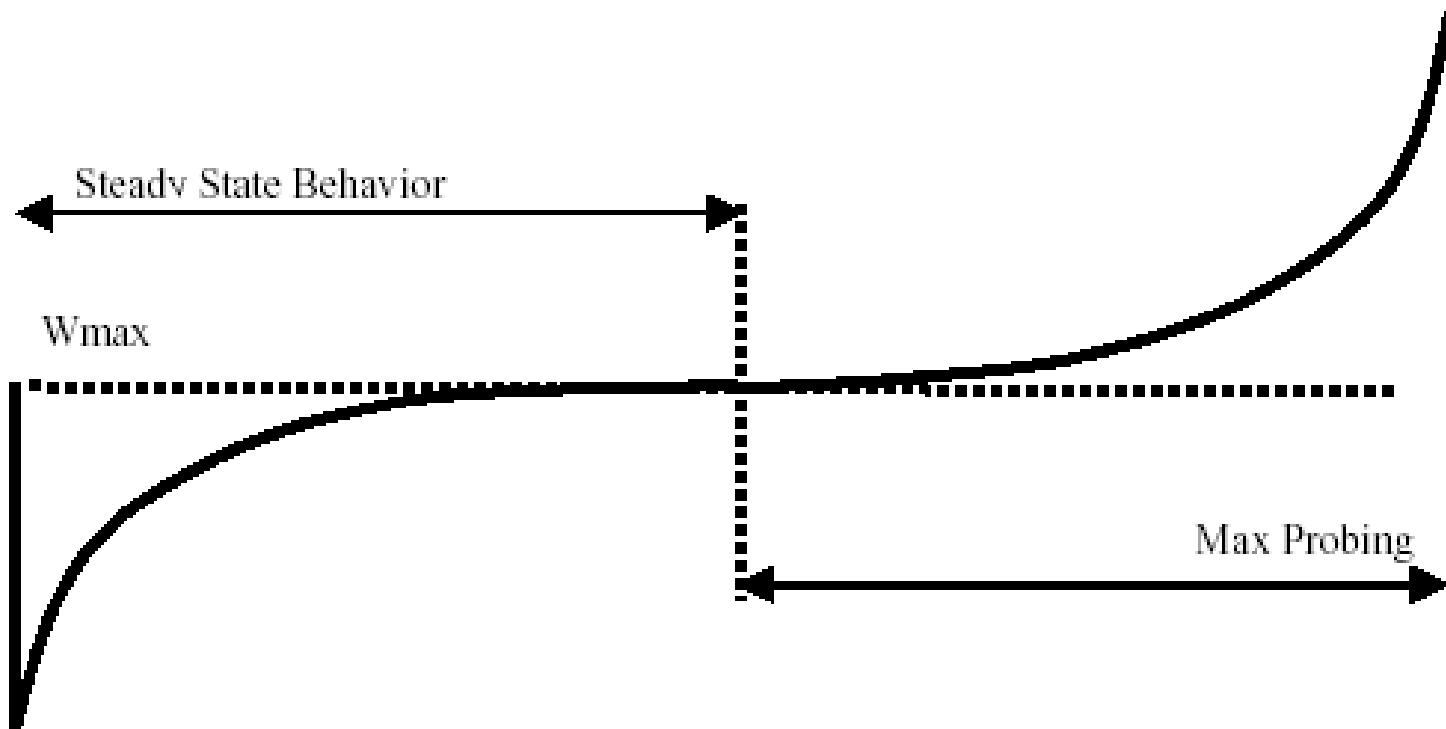
- Window Decrease



*0.9: give bandwidth to newly-coming flows
... "Fast Convergence"

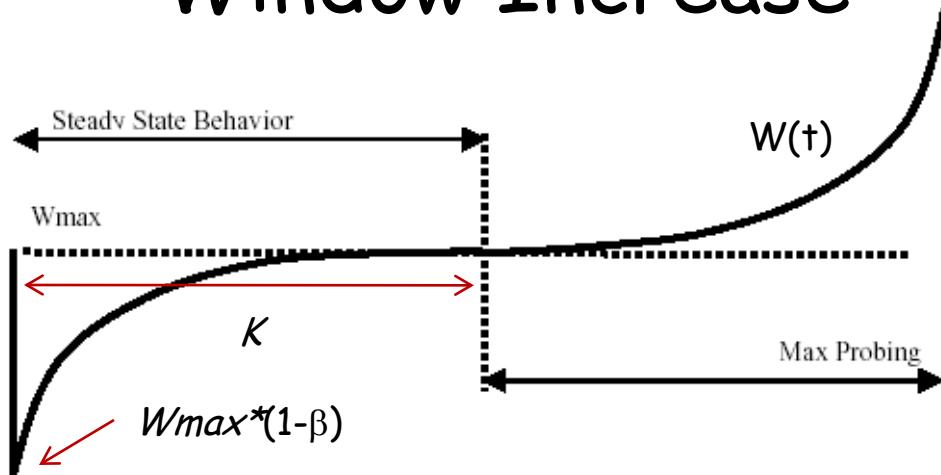
CUBIC-TCP (1)

- Cubic approximation of BIC-TCP



CUBIC-TCP (2)

- Window Increase



$$W(t) = C * (t - K)^3 + W_{\max}$$

$$K = \sqrt[3]{\frac{W_{\max} \beta}{C}}$$

$$W_{tcp}(t) = W_{\max} (1 - \beta) + 3 \frac{\beta}{2 - \beta} \frac{t}{RTT}$$

equivalent to Reno



```
/* cubic function */
Winc = W(t+RTT) - cwnd;
```

```
cwnd = cwnd + Winc / cwnd;
```

```
/* TCP mode */
if ( Wtcp > cwnd )
    cwnd = Wtcp;
```

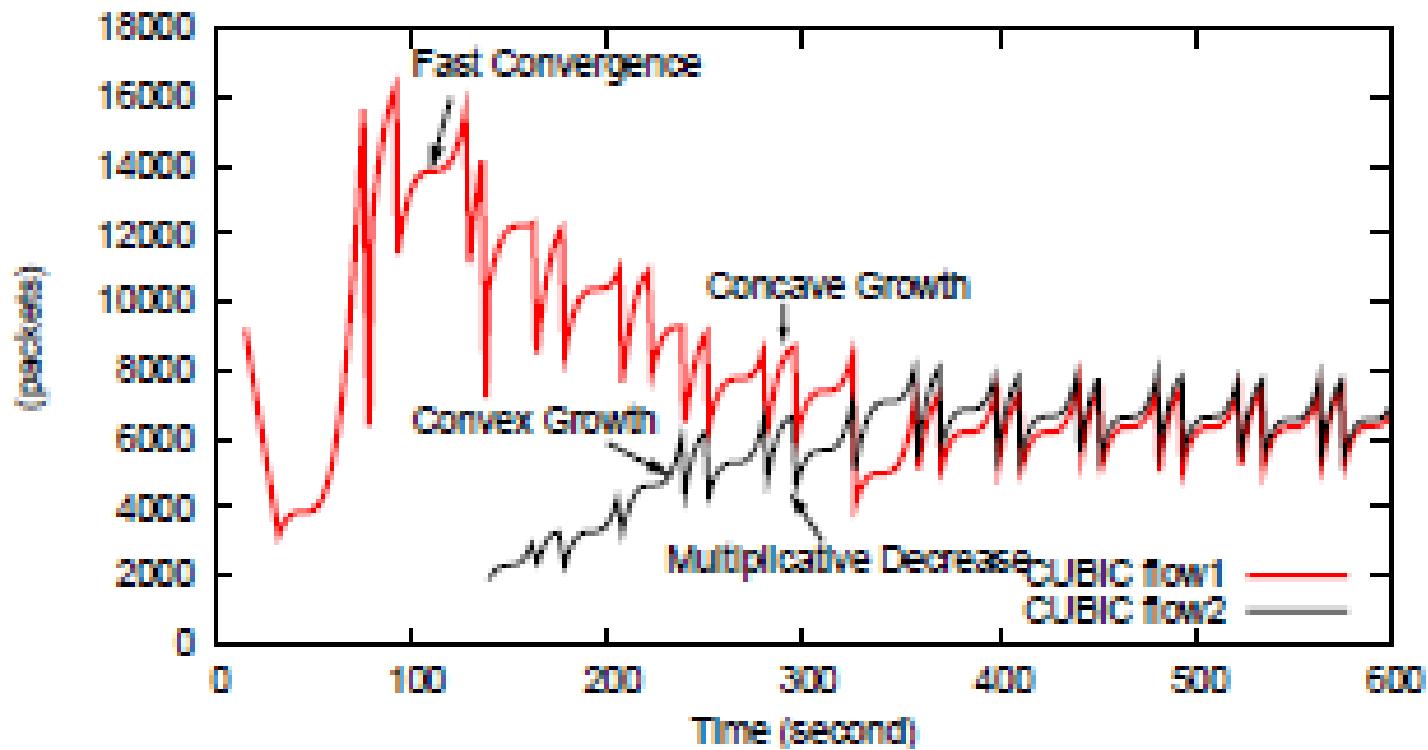
※ window decrease is the same as BIC

β : decrease rate (e.g. 0.2)

C : constant (e.g. 0.4)

CUBIC-TCP (3)

- CUBIC's cwnd behavior



CUBIC-TCP (4)

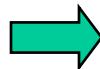
- Advantages
 - stability
 - “intra-protocol fairness” among multiple CUBIC flows
- Disadvantages
 - heavy buffer occupancy and delay increase (\Leftrightarrow delay-based)
 - “inter-protocol unfairness” against other TCP flows
 - “Linux beats Windows!” (vs. Compound TCP)

Hybrid TCPs (and its Performance Analysis)

Hybrid TCP (1)

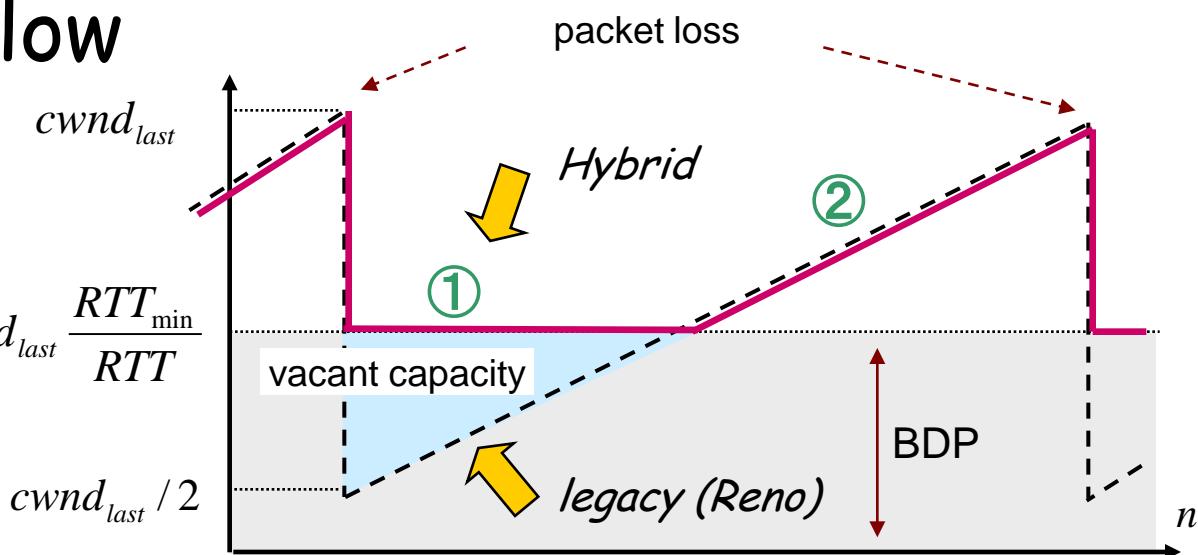
- single flow

clearing buffer
(TCPW)



$$cwnd_{last} \frac{RTT_{min}}{RTT}$$

$$cwnd_{last} / 2$$

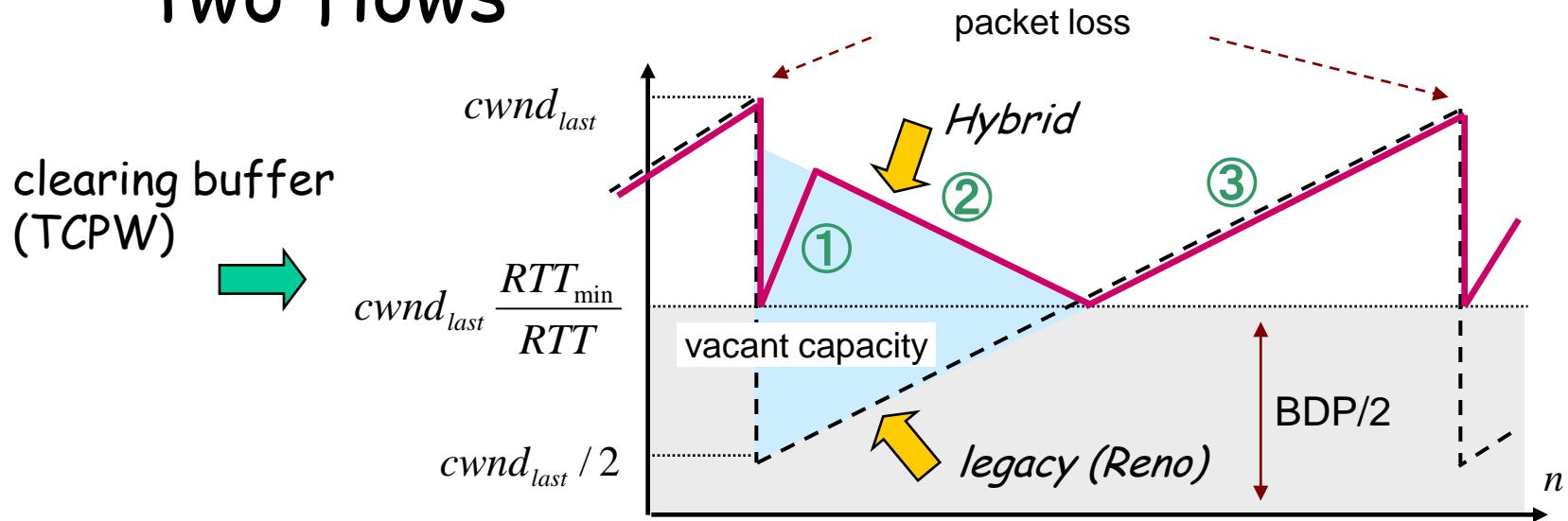


adaptive switching of two modes (loss & delay):

- constant rate until RTT increases (delay mode) : "efficiency" and "low delay"
- performs as Reno when RTT increases (loss mode) : "friendliness"

Hybrid TCP (2)

- two flows

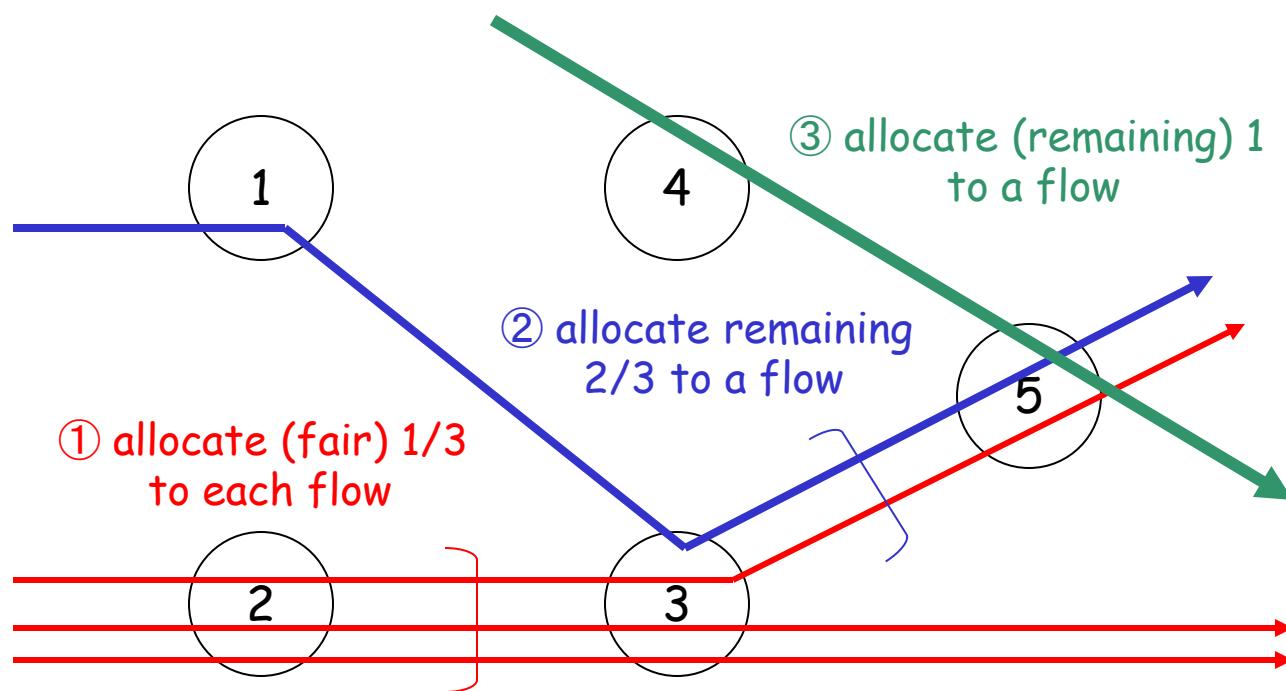


adaptive switching of two modes (loss & delay):

- ① fast cwnd increase (delay ... "efficiency")
- ② mild cwnd decrease (delay ... congestion avoidance)
- ③ performs as Reno when RTT increases (loss ... "friendliness")

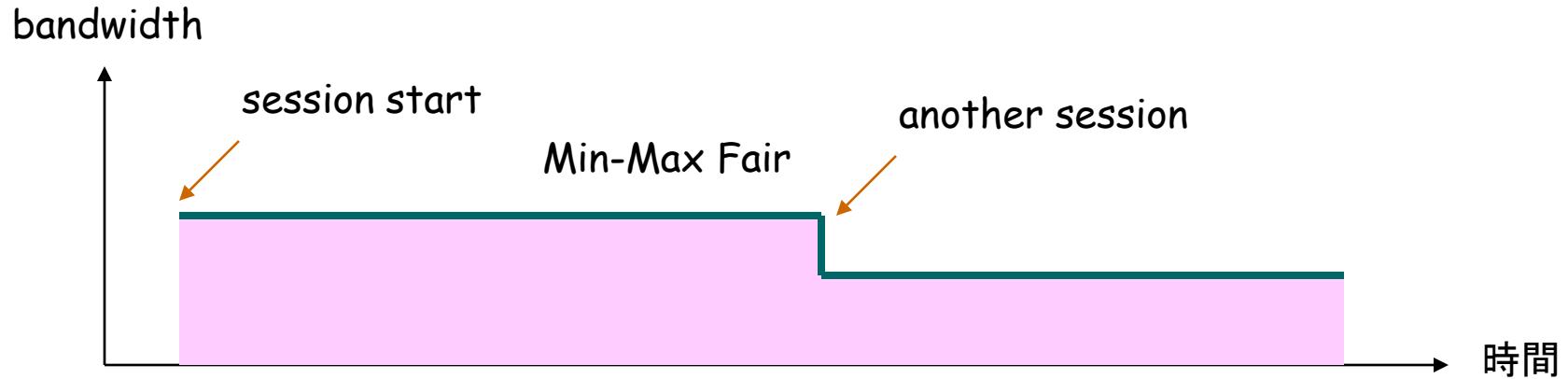
Min-Max Fair (ideal case)

- Min-Max-Fair: allocate “maximum bandwidth” to a user who has “minimum bandwidth”

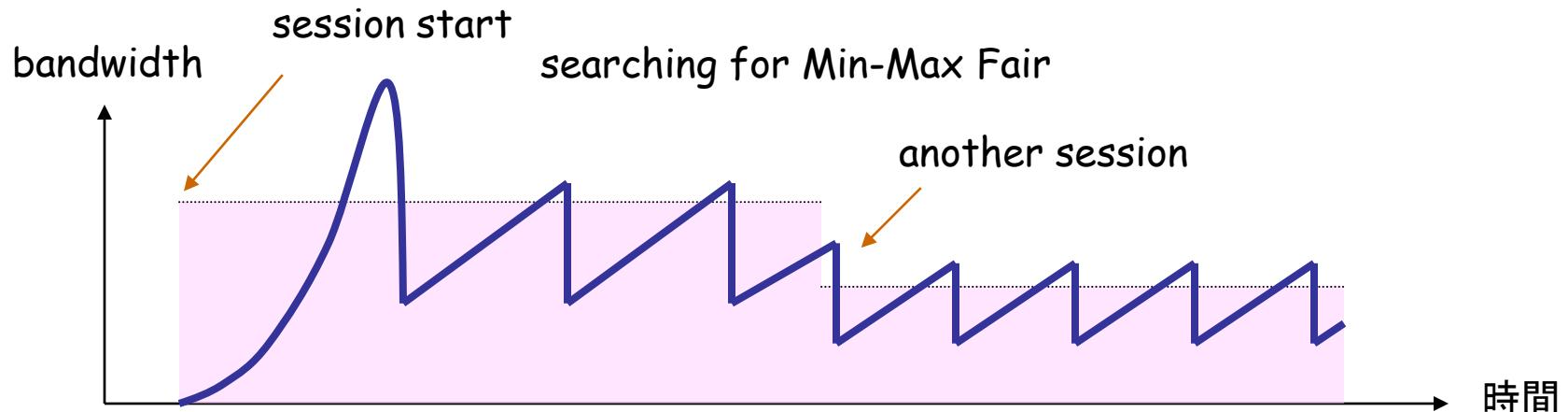


Ideal:

TCP's objective



TCP Reno



TCP behavior model (1)

- model definition
 - Loss-mode (TCP-Reno) :
 - cwnd += 1 (per "RTT round")
 - cwnd *= 1/2 (when a packet loss is detected)
 - Delay-mode :
 - fill a "pipe" (fully utilize a link) without causing RTT increase
 - Hybrid :
 - works in delay mode when RTT is not increased
 - works in loss mode when RTT is increases (i.e. when packets are buffered)
 - mode selection: $cwnd = \max(cwnd_{loss}, cwnd_{delay})$

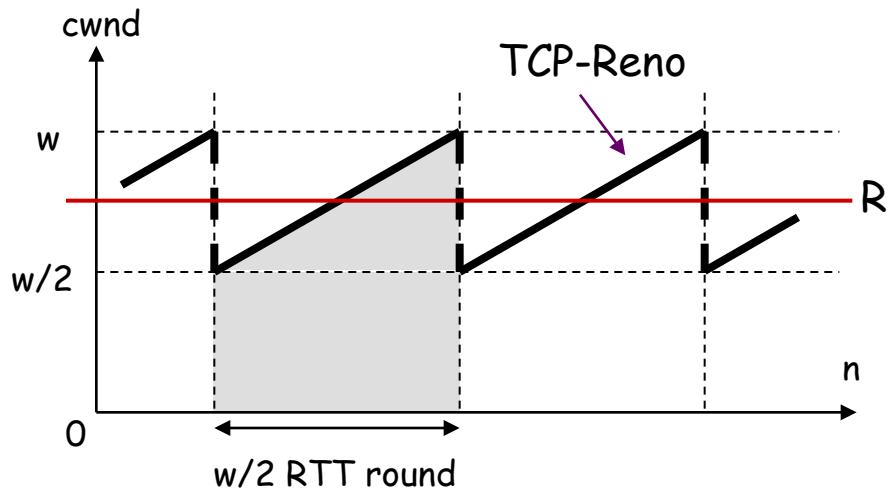
TCP behavior model (2)

- parameter definition
 - w : cwnd when a packet loss is detected
 - W : cwnd which just fills a pipe \sim BDP
 - p : packet loss rate
- assumption
 - packet loss due to buffer overflow is equivalent to packet loss due to random error

$$p = \frac{8}{3w^2} \quad (\text{in case of TCP-Reno})$$

TCP behavior model (3)

- TCP friendly model

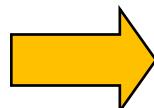


w: cwnd when a packet loss is detected
p: packet loss rate
RTT: round trip time

R: TCP equivalent rate

of transmitted packets until a packet loss is detected
= area of a trapezoid

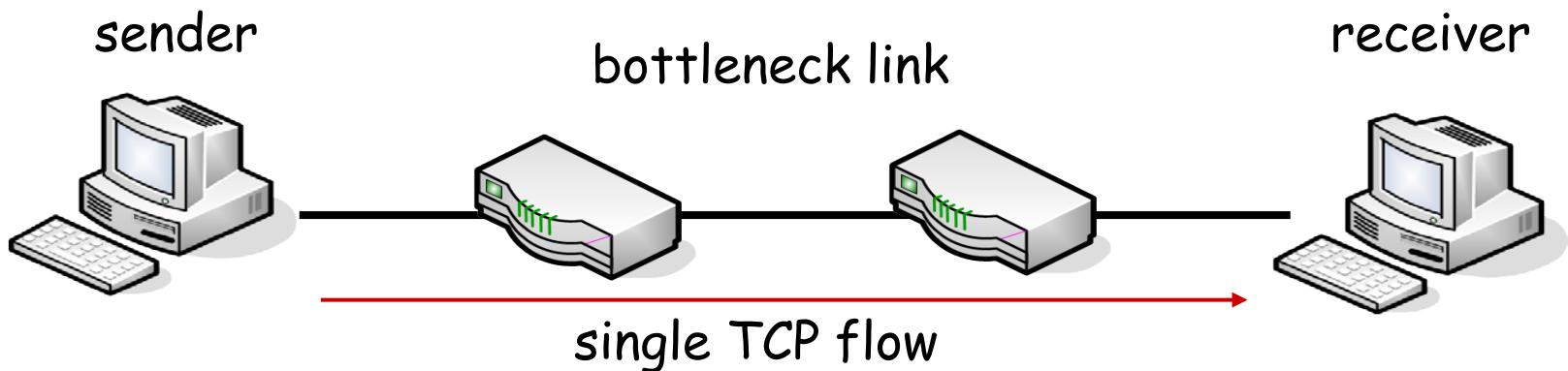
$$\frac{1}{2} \cdot \left(\frac{w}{2} + w \right) \cdot \frac{w}{2} = \frac{3w^2}{8}$$



$$\begin{cases} p = \frac{8}{3w^2} \\ R = \frac{PS}{RTT} \cdot \sqrt{\frac{3}{2p}} \end{cases}$$

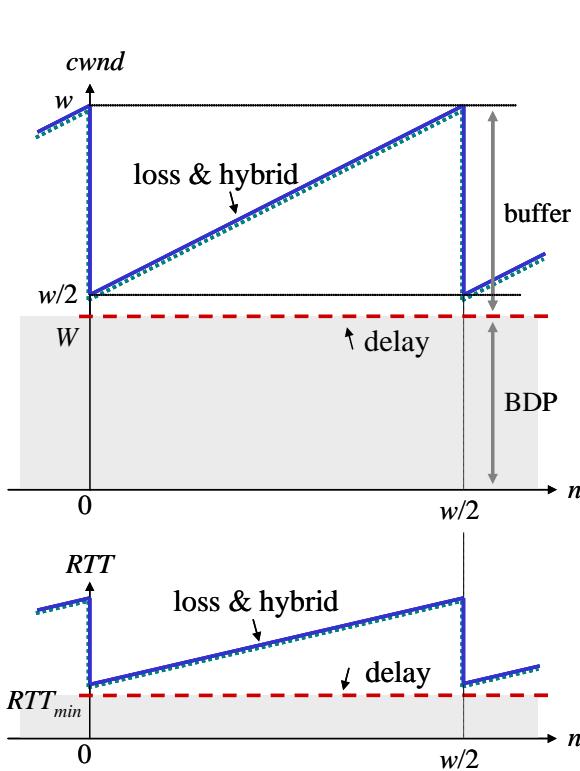
TCP behavior model (4)

- single flow



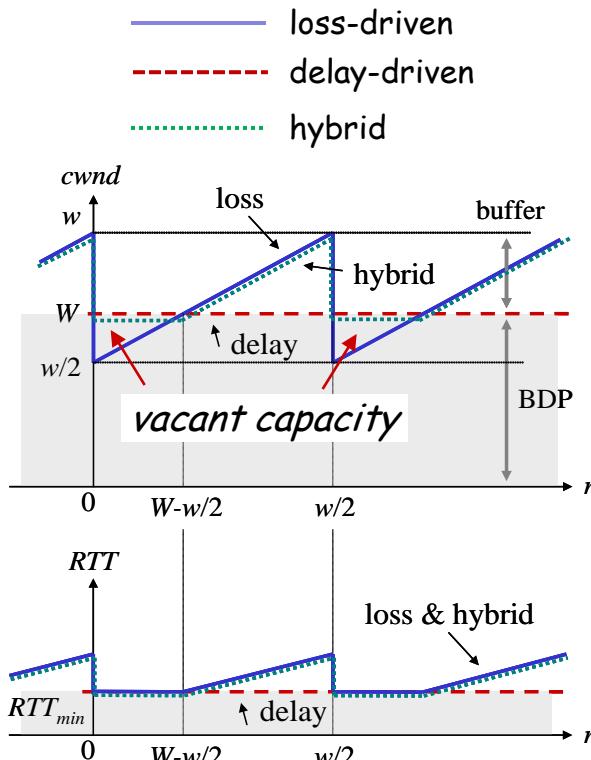
TCP behavior model (5)

- cwnd & RTT behaviors of ideal models (single flow case)



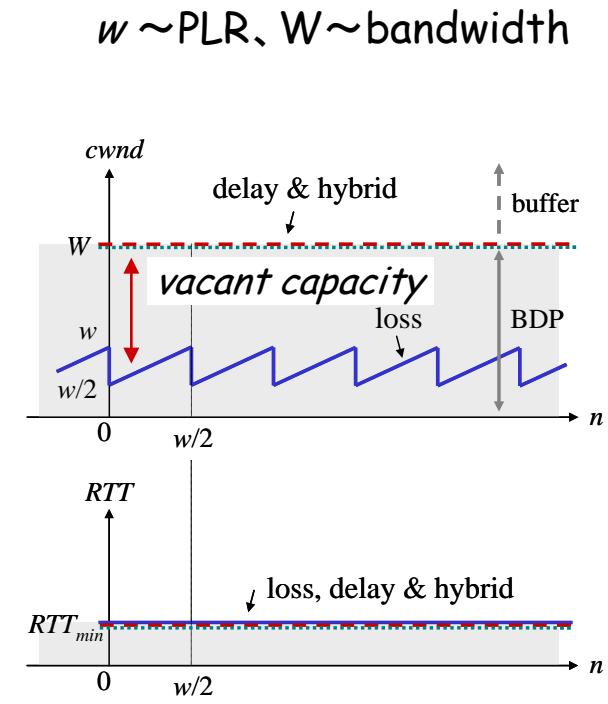
(i) $W < w/2$

large buffer, small PLR
(always loss-mode)



(ii) $w/2 < W < w$

small buffer, medium PLR
(delay ↔ loss)



(iii) $w < W$

large PLR, always vacant
(always delay-mode)

$w \sim \text{PLR}, W \sim \text{bandwidth}$

TCP behavior model (6)

- formulation

TCP	CA round	(i) $W < w/2$	(ii) $w/2 \leq W < w$	(iii) $w \leq W$
Loss	transmitted packets	$\frac{3}{8}w^2$	$\frac{3}{8}w^2$	$\frac{3}{8}w^2$
	elapsed time	$\frac{1}{2}w \cdot RTT_{\min} + \frac{1}{8}(3w^2 - 4wW) \cdot \frac{PS}{B}$	$\frac{1}{2}w \cdot RTT_{\min} + \frac{1}{2}(w-W)^2 \cdot \frac{PS}{B}$	$\frac{1}{2}w \cdot RTT_{\min}$
Delay	transmitted packets	$\frac{1}{2}w \cdot W$	$\frac{1}{2}w \cdot W$	$\frac{1}{2}w \cdot W$
	elapsed time	$\frac{1}{2}w \cdot RTT_{\min}$	$\frac{1}{2}w \cdot RTT_{\min}$	$\frac{1}{2}w \cdot RTT_{\min}$
Hybrid	transmitted packets	$\frac{3}{8}w^2$	$\frac{1}{2}w \cdot W + \frac{1}{2}(w-W)^2$	$\frac{1}{2}w \cdot W$
	elapsed time	$\frac{1}{2}w \cdot RTT_{\min} + \frac{1}{8}(3w^2 - 4wW) \cdot \frac{PS}{B}$	$\frac{1}{2}w \cdot RTT_{\min} + \frac{1}{2}(w-W)^2 \cdot \frac{PS}{B}$	$\frac{1}{2}w \cdot RTT_{\min}$

PS: Packet size, B: Link bandwidth

TCP behavior model (7)

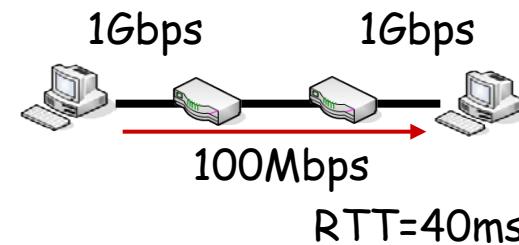
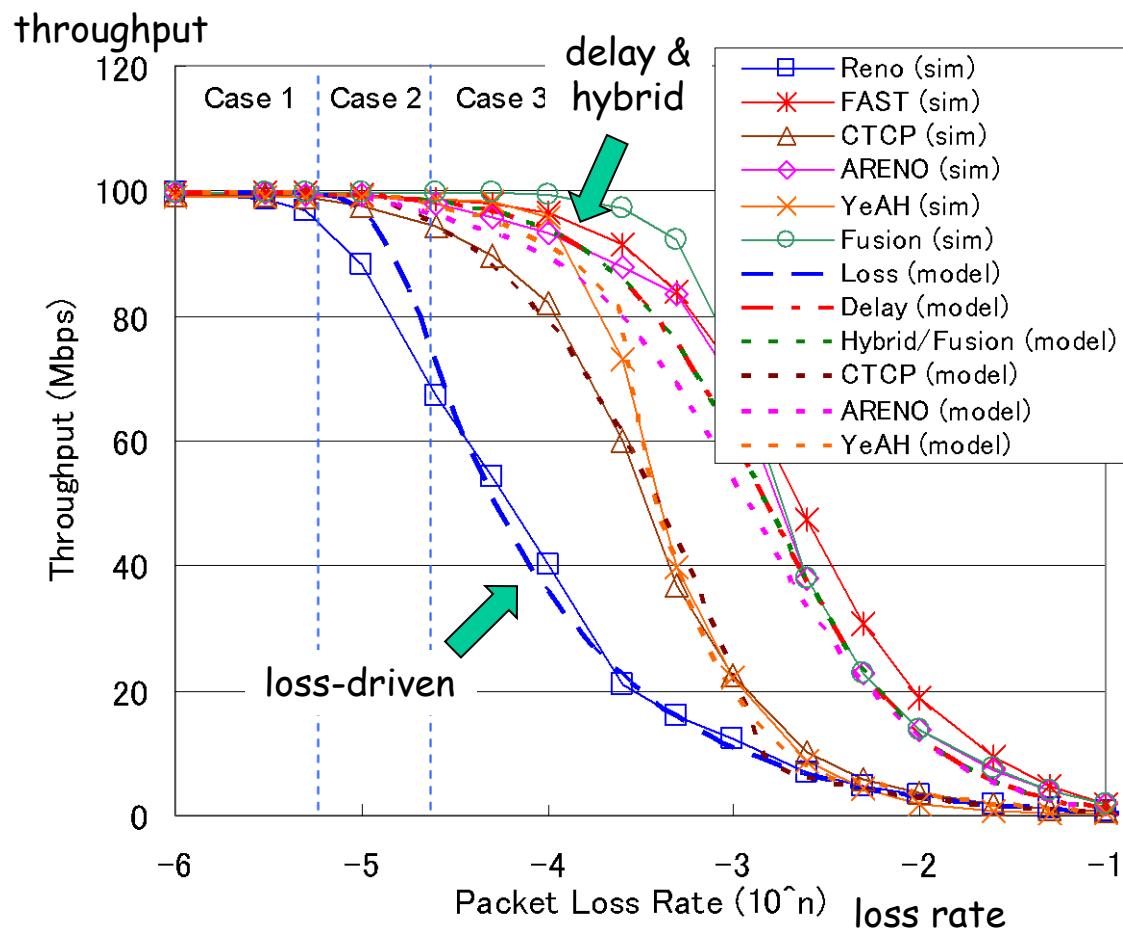
- abstraction of actual hybrids

Hybrids	Window increase	Window decrease
Compound TCP	$0.125 * \text{cwnd}^{0.75}$	1/2
ARENO	$B/10\text{Mbps}$	1/2~1
YeAH-TCP	Scalable TCP (1.01)	$1/2, \text{RTT}_{\min}/\text{RTT}, 7/8$
TCP-Fusion	$B * D_{\min} / (N * PS)$	$\text{RTT}_{\min}/\text{RTT}$

D_{\min} : timer resolution, N: # of flows

TCP behavior model (8)

- evaluation by models and simulations



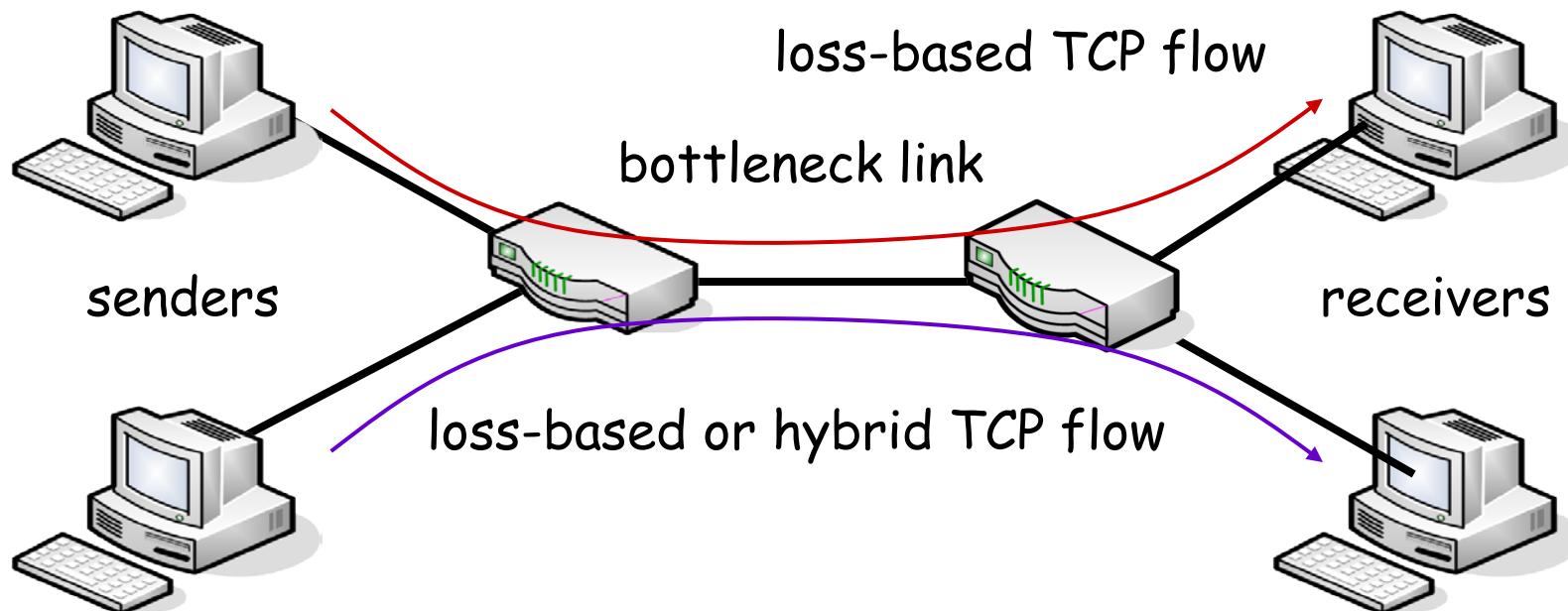
buffer size = BDP (constant)
Packet loss rate : variable

when PLR is large ($w/2 < W$),
throughputs of delay & hybrid are much larger than
that of loss-mode (i.e. efficiency)

degradation of Compound &
YeAH is due to fixed window
decrease

TCP behavior model (9)

- two flows (competing)

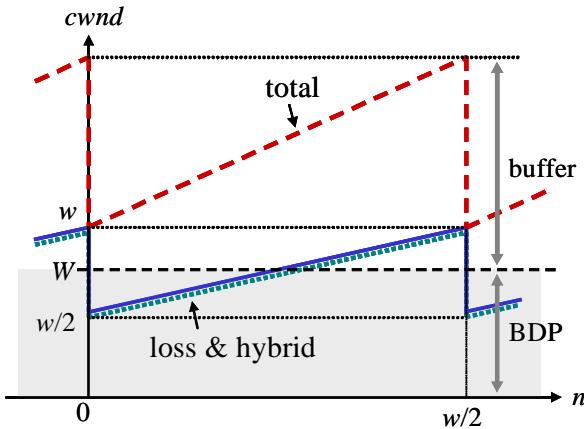


TCP behavior model (10)

- cwnd behavior of ideal models (two flow case)

— loss-driven
 - - - hybrid
 - - - total (loss + hybrid)

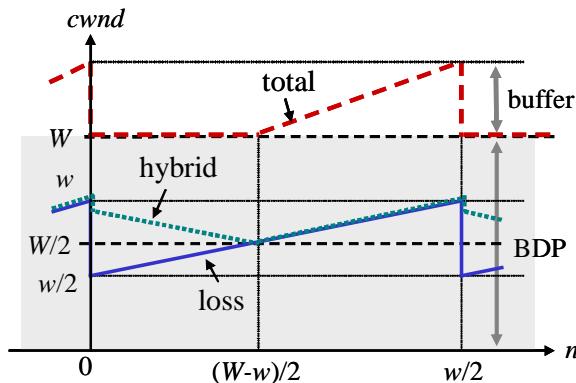
$w \sim \text{PLR}$, $W \sim \text{bandwidth}$



(i) $W < w$ (low PLR)

always buffered
(loss mode)

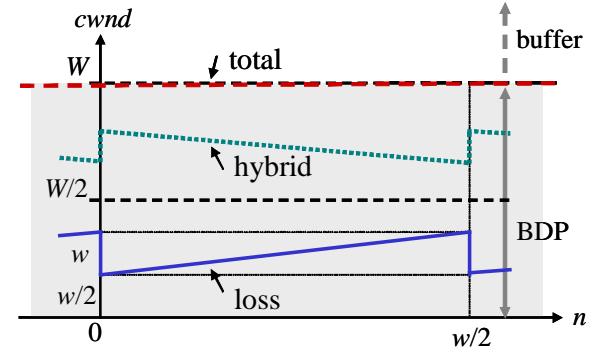
large buffer, small PLR



(ii) $w < W < 2w$ (medium PLR)

vacant \rightarrow buffered
(delay \rightarrow loss)

small buffer, medium PLR



always vacant
(delay mode)

large PLR, always vacant

TCP behavior model (11)

- formulation

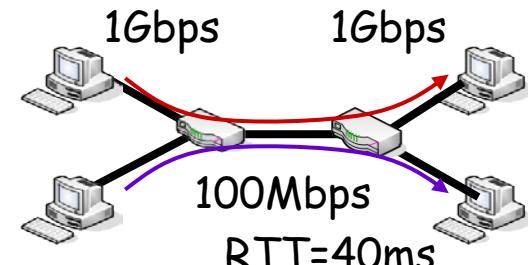
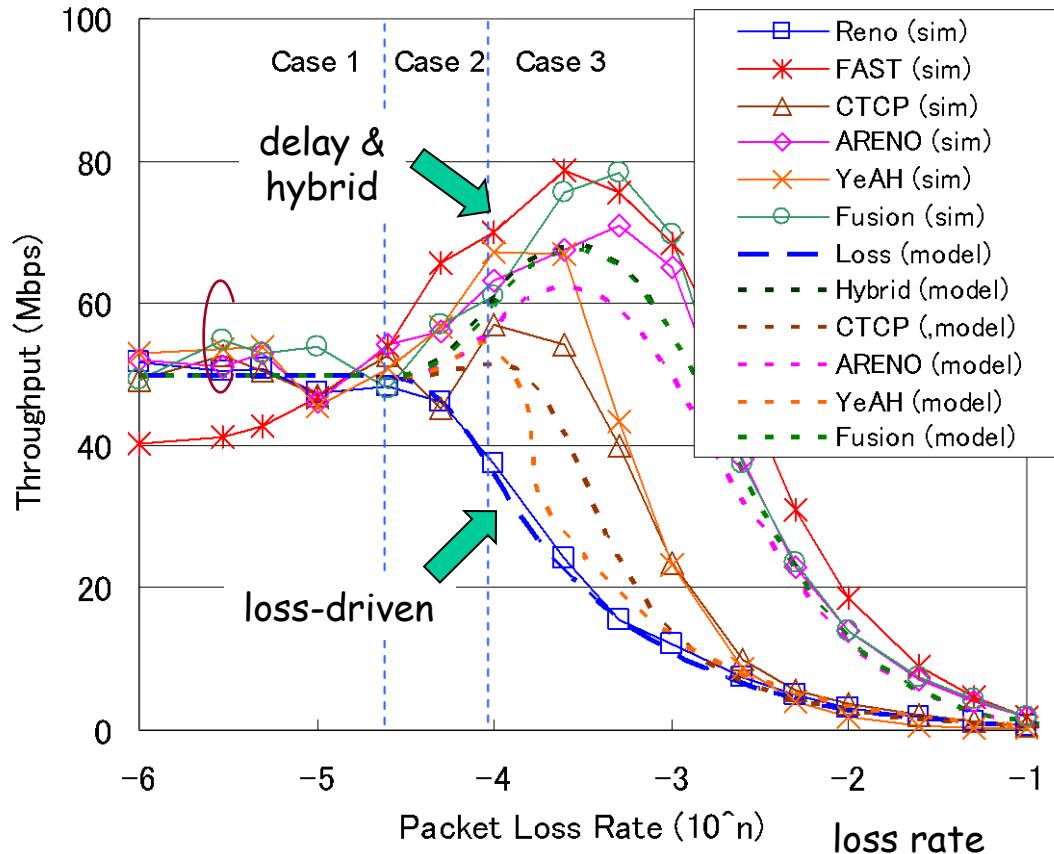
TCP	CA round	(i) $W < w$	(ii) $w \leq W < 2w$	(iii) $2w \leq W$
Loss	transmitted packets	$\frac{3}{8}w^2$	$\frac{3}{8}w^2$	$\frac{3}{8}w^2$
Hybrid	transmitted packets	$\frac{3}{8}w^2$	$\frac{3}{8}w^2 + \frac{1}{4}(W-w)^2$	$\frac{1}{2}w \cdot W - \frac{3}{8}w^2$
(common)	elapsed time	$\frac{1}{2}w \cdot RTT_{\min} + \frac{1}{4}w(3w-2W) \cdot \frac{PS}{B}$	$\frac{1}{2}w \cdot RTT_{\min} + \frac{1}{4}(2w-W)^2 \cdot \frac{PS}{B}$	$\frac{1}{2}w \cdot RTT_{\min}$

PS: Packet size, B: Link bandwidth

TCP behavior model (12)

- evaluation by models and simulations

throughput



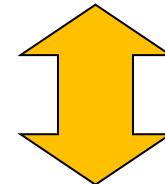
buffer size = BDP (constant)
Packet loss rate : variable

when PLR is large ($w < W$),
throughputs of delay &
hybrid are much larger than
that of loss-mode
(efficiency)

when PLR is low ($w > W$),
hybrid behaves similar to
loss-mode **(friendliness)**

TCP behavior model (13)

- Advantages of Hybrid TCP
 - when vacant capacity exists (or PLR is large), throughput efficiency is greatly improved (advantage of delay-mode)
 - when no vacant capacity exists (or buffer size is large), friendliness to legacy TCP (i.e. Reno) is achieved (advantage of loss-mode)
- Disadvantages of Hybrid TCP
 - when buffer size is large, delay-mode is never activated ...



Summary of Hybrid TCP

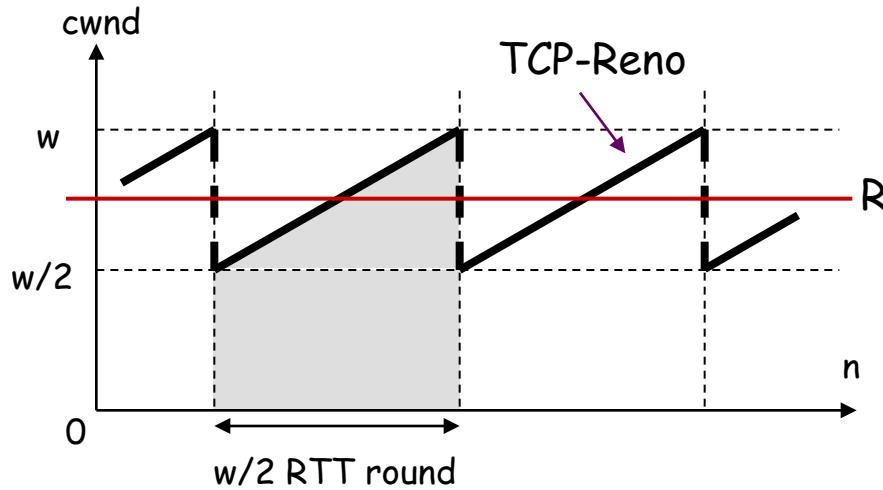
Hybrid TCP

- “Efficiency”, “Friendliness” and “Low delay”
 - can be applied to real-time streaming and large file download
 - might be effective in wireless networks
 - friendliness to CUBIC-TCP or Compound-TCP
 - CUBIC-TCP: Linux default
 - Compound-TCP: Windows
 - other metrics
 - RTT fairness, mice/elephant (short-lived or long-lived), convergence speed, etc...
 - efficiency is brought by delay-mode

TCP Equations

TCP Modeling

- TCP-Reno Equivalent Rate



w: cwnd when packet is lost
 p: PLR
 RTT: round trip time
 R: equivalent rate
 b: # of delayed ACKs

with timeout consideration

$$\rightarrow \left\{ \begin{array}{l} p = \frac{8}{3w^2} \\ R = \frac{PS}{RTT} \cdot \sqrt{\frac{3}{2p}} \end{array} \right. \rightarrow$$

$$R_{loss} = \frac{PS}{RTT \sqrt{\frac{2bp}{3}} + t_{RTO,loss} \cdot 3 \sqrt{\frac{3bp}{8}} \cdot p(1+32p^2)}$$

TCP Westwood

- Duplicate ACKs

FSE: Fair Share Estimates

$$ssthresh = FSE * RTT_{min}$$

$$\text{if } (cwnd > ssthresh) \text{ } cwnd = ssthresh$$

- Timeout

in TCP-Reno case

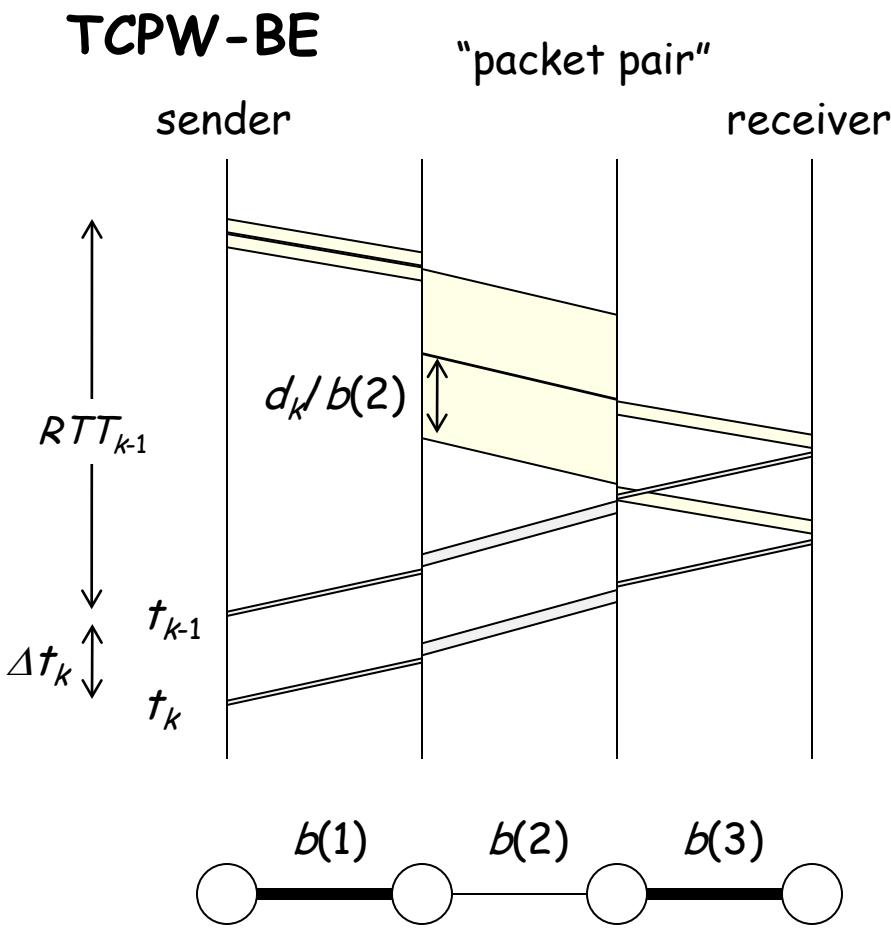
$$ssthresh = cwnd / 2$$

$$ssthresh = FSE * RTT_{min}$$

$$cwnd = 1$$

- multiple versions according to FSE estimation methods

Bandwidth Share Estimation



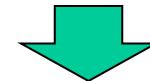
Bandwidth share: $b = \min_j(b(j))$

t_k : ack arrival time of the k -th packet

d_k : size of the k -th packet



$$\Delta t_k = t_k - t_{k-1} \approx \frac{d_k}{b}$$



$$b_k \approx \frac{d_k}{\Delta t_k}$$



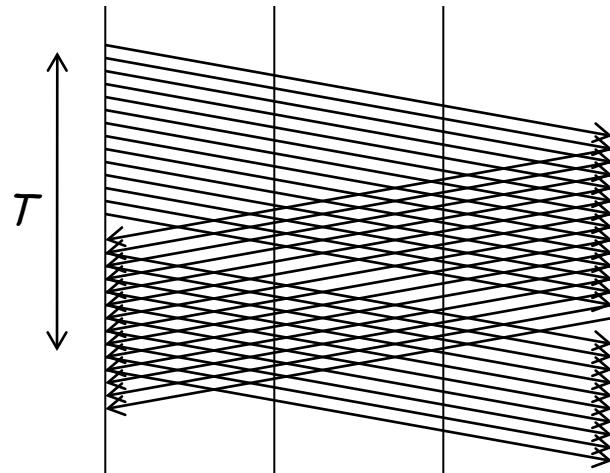
moving average: $\hat{b}_k \rightarrow FSE$

Rate Estimation

(reference) TCP-Vegas

$$diff = \left(\frac{cwnd}{RTT_{min}} - \frac{cwnd}{RTT} \right) \cdot RTT_{min}$$

↑ ↑
expect rate actual rate

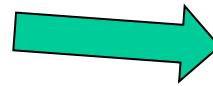


TCPW-RE:

$$RE_k = \frac{\sum d_j}{T}$$

↓

$$\begin{aligned} cwnd &\Rightarrow S = \sum d_k \\ RTT &\Rightarrow T = \sum \Delta t_k \end{aligned}$$



$$\hat{RE}_k \approx \frac{\sum d_k}{\sum \Delta t_k}$$

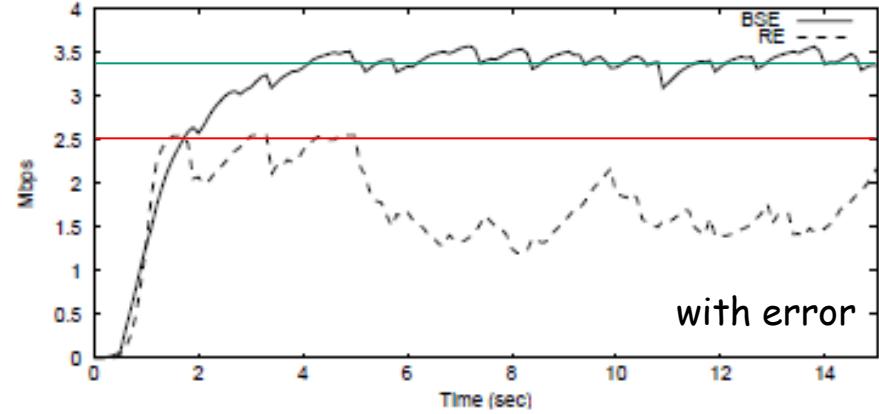
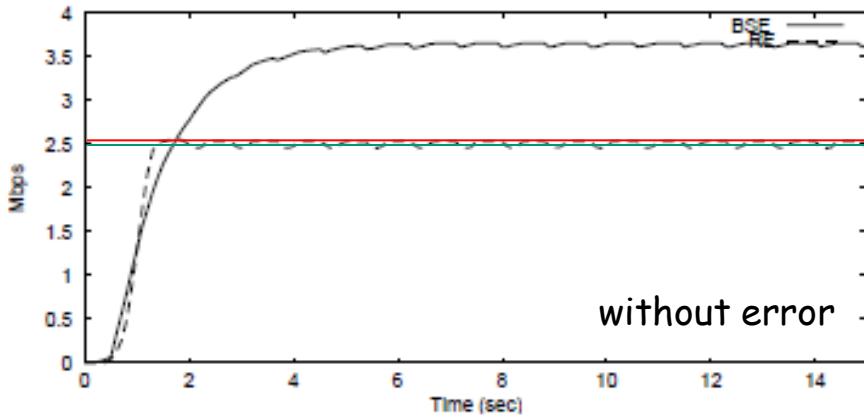
moving average:

$$\hat{RE}_k \rightarrow FSE$$

$$T = n \cdot RTT \quad (\text{e.g. } n=4)$$

Comparison of BSE and RE

solid: BSE, dashed: RE, red: fair share, green: capacity



- BSE tends to overestimate (due to burstiness)
- RE tends to underestimate when losses occur

Adaptive Bandwidth Share Estimation

- BSE: overestimation, RE: underestimation
- difference lies in sampling period T



- large T when congested (BSE), small T when not congested (RE)

TCPW-ABSE:

$$T_k = \max\left(T_{\min}, RTT \cdot \left(1 - \frac{\hat{Th} \cdot RTT_{\min}}{cwnd}\right)\right)$$

T_{\min} : ACK arrival interval

$\hat{Th} \cdot RTT_{\min} < cwnd$ \rightarrow congested \rightarrow larger T_k

多数のパケットを送っても実レートが上がらない

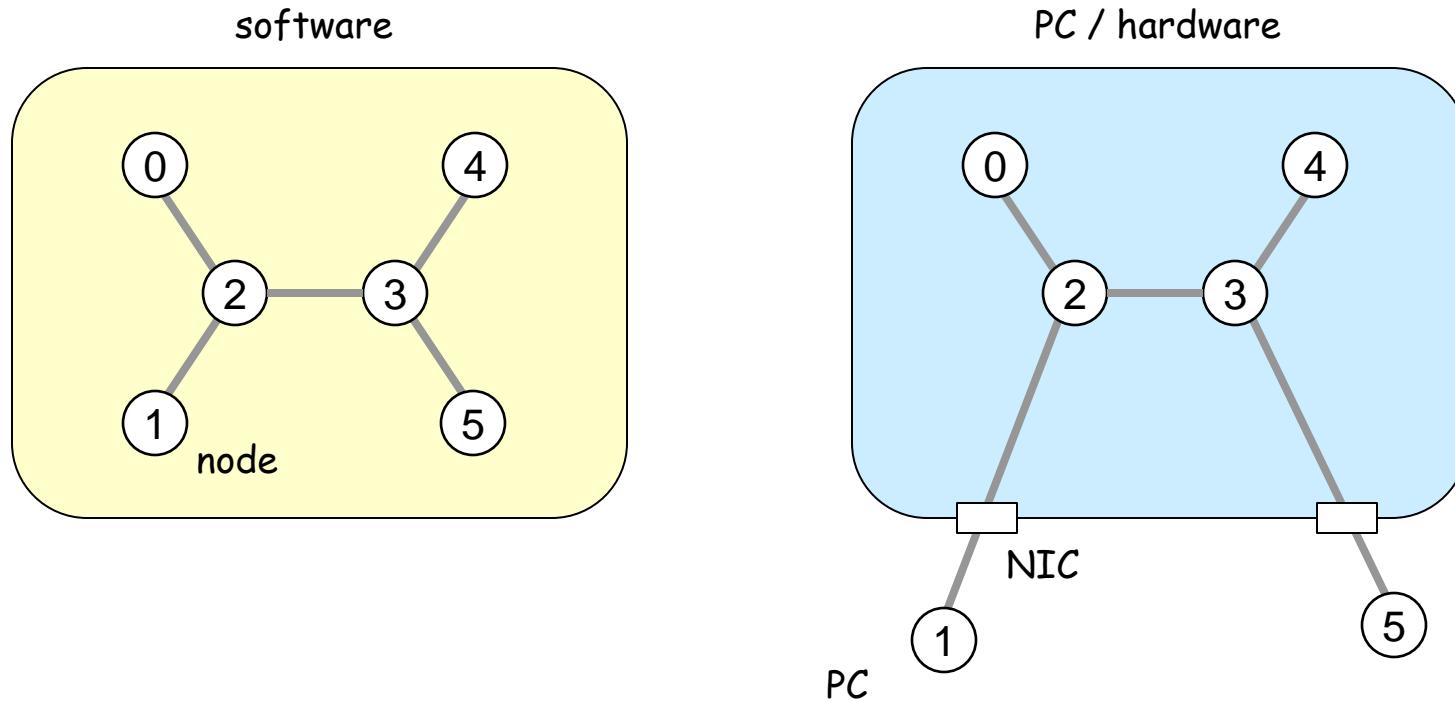
Network Simulation & Emulation

Networking Research

- Algorithm
- Theory (model)
- Simulation
- Emulation
- Implementation

Simulator & Emulator (1)

- simulation
- emulation



Simulator & Emulator (2)

simulator	emulator	URL
ns-2 (ns)	(nse)	http://www.isi.edu/nsnam/ns/
ns-3	nsc	http://www.nsnam.org/
OPNET		http://www.opnet.com/
Qualnet, GloMoSim	EXata	http://www.scalable-networks.com/
Scenargie		http://www.spacetimetime-eng.com/
	PacketStorm	http://www.packetstorm.com/
	Cloud	http://www.shunra.com/ve-cloud.php

ns-2

Ns-2 (1)

- <http://www.isi.edu/nsnam/ns/>

The screenshot shows a Microsoft Internet Explorer window with the title bar "The Network Simulator - ns-2". The address bar contains the URL "http://www.isi.edu/nsnam/ns/". The page content is as follows:

The Network Simulator - ns-2

Note: The project is migrating these web pages to a new [wiki](#). This page can now be found [here](#).

Ns is a discrete event simulator targeted at networking research. Ns provides substantial support for simulation of TCP, routing, and multicast protocols over wired and wireless (local and satellite) networks.

Ns began as a variant of the [REAL network simulator](#) in 1989 and has evolved substantially over the past few years. In 1995 ns development was supported by DARPA through the [VINT project](#) at LBL, Xerox PARC, UCB, and USC/ISI. Currently ns development is supported through DARPA with [SAMAN](#) and through NSF with [CONSER](#), both in collaboration with other researchers including [ACIRI](#). Ns has always included substantial contributions from other researchers, including wireless code from the UCB Daedelus and CMU Monarch projects and Sun Microsystems. For documentation on recent changes, see the version 2 [change log](#).

Read this first:

While we have considerable confidence in ns, ns is not a polished and finished product, but the result of an on-going effort of research and development. In particular, bugs in the software are still being discovered and corrected. Users of ns are responsible for verifying for themselves that their simulations are not invalidated by bugs. We are working to help the user with this by significantly expanding and automating the [validation tests and demos](#).

Similarly, users are responsible for verifying for themselves that their simulations are not invalidated because the model implemented in the simulator is not the model that they were expecting. The ongoing [Ns Manual](#) should help in this process.

Links to help getting started

- **Getting ns and avoiding problems**
 - [Download and Build ns](#)
 - [Installation Problems and Bug Fixes](#)
 - [Validation Tests and Demos](#)
 - [Copyright statement](#)
 - [Bug report form](#)
- **Documentation:**
 - [core documentation:](#)
 - [ns frequently asked questions](#)
 - limitations: [limitations and assumptions of ns, advice on running simulations](#)
 - introductory: [Marc Greis's tutorial](#) (now maintained by VINT group), also see "[NS for Beginners](#)" by Altman and Jimenez
 - reference: [Ns Manual](#) (formerly called "ns Notes and Documentation")
 - errata: [Installation Problems and Bug Fixes](#)
 - [development help:](#)
 - [NS-related Mailing Lists](#)
 - [debugging tips](#)

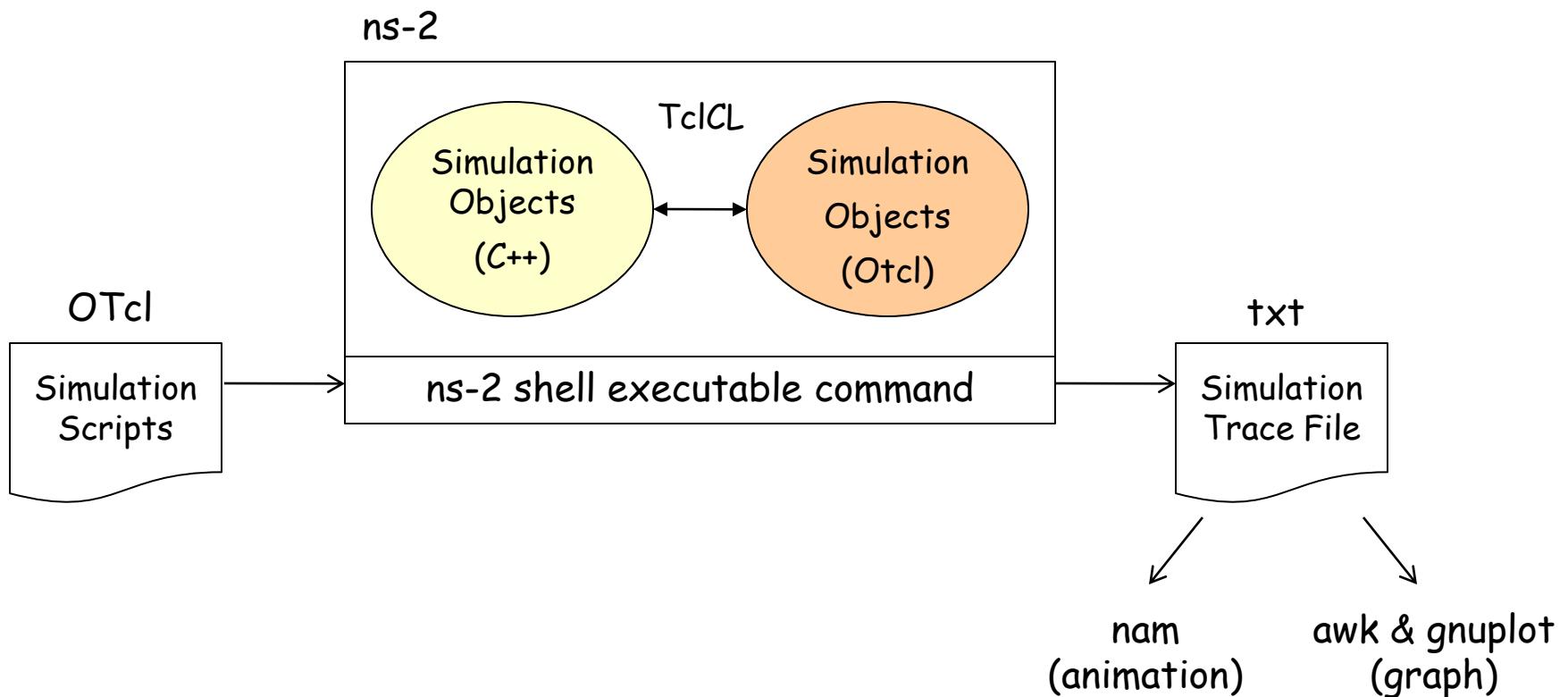
Ns-2 (2)

- download
 - 2.29 and later:
<http://sourceforge.net/projects/nsnam/>
 - before 2.28:
<http://www.isi.edu/nsnam/dist/>

Download "allinone", expand, configure, and make
(Tcl/Tk, Otcl, TclCL, ns, nam)

Ns-2 (3)

- ns-2 Architecture



Ns-2 (4)

- Simulation scripts (*.tcl)

```
# initialization
# Simulator object
set ns [ new Simulator ]
# network topology
# definition of agents and apps
# procedure definition (e.g. finish)
proc finish () ...
# event definition
$ns at 1.0 "$ftp start"
# simulation start
$ns run

set ns [new Simulator]
set f [open out.tr w]
$ns trace-all $f

set n0 [$ns node]
set n1 [$ns node]
$ns duplex-link $n0 $n1 100Mb 1ms DropTail

set udp0 [new Agent/UDP]
$ns attach-agent $n0 $udp0
set cbr0 [new Application/Traffic/CBR]
$cbr0 attach-agent $udp0
...

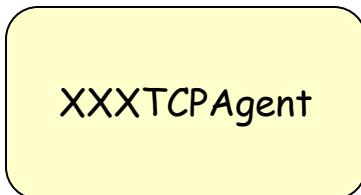
```

Ns-2 (5)

- Simulation Objects (C++/OTcl)

C++

```
static class XXXTcpClass : public TclClass {  
public:  
    XXXTcpClass() : TclClass("Agent/TCP/XXX") {}  
    TclObject* create(int, const char*const*) {  
        return (new XXXTcpAgent());  
    }  
} class_XXX;
```



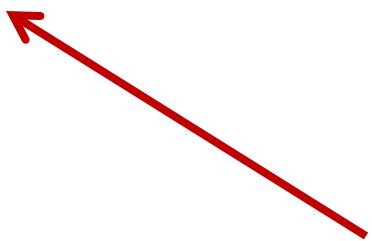
TclObject

OTcl

```
set tcp [new Agent/TCP/XXX]
```



TclClass

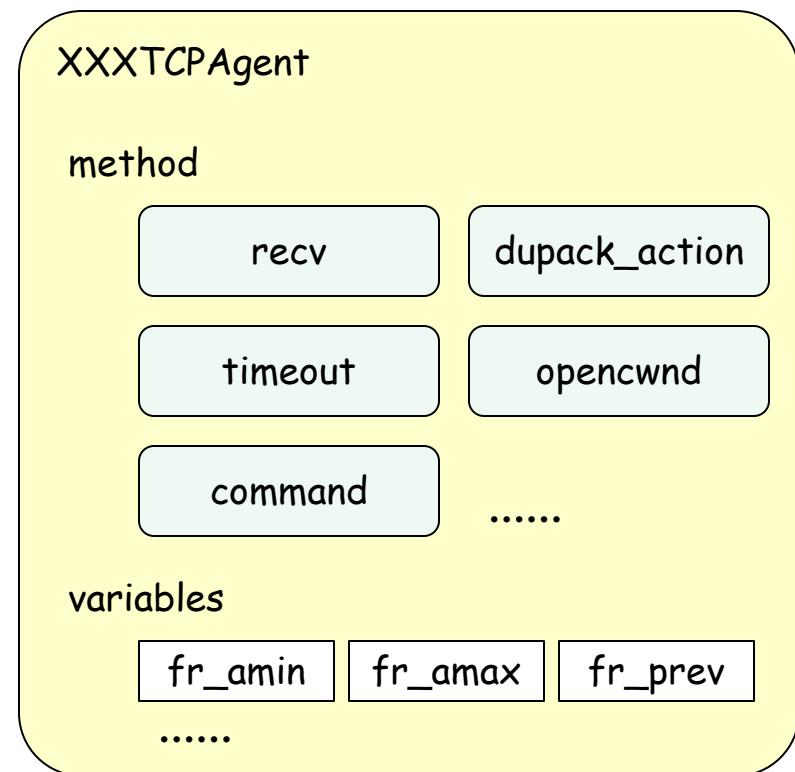


Ns-2 (6)

- Simulation Objects (C++/OTcl)

C++

```
class XXXTcpAgent : public TcpAgent {  
public:  
    XXXTcpAgent();  
    virtual void recv(Packet *pkt, Handler*);  
    virtual void dupack_action();  
    virtual void timeout (int tno);  
    virtual void opencwnd();  
    ...  
protected:  
    int command(int argc, const char*const* argv);  
  
    double fr_amin_;  
    double fr_amax_;  
    double fr_prev_;  
}
```



Ns-2 (7)

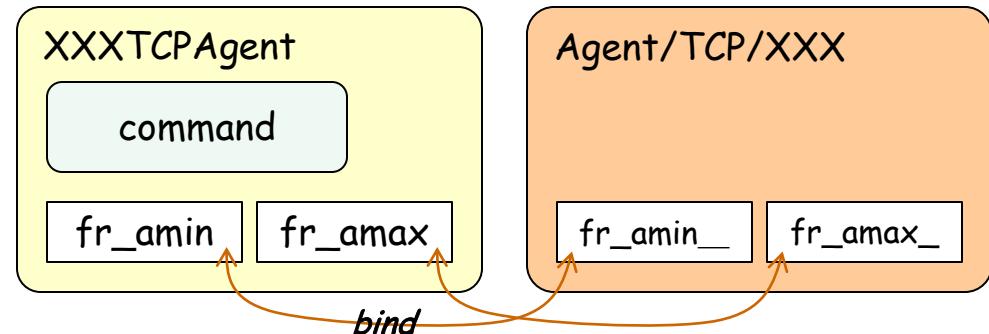
- Simulation Objects (C++/OTcl)

C++

```
XXXTcpAgent::XXXTcpAgent()
{
    bind("fr_amin_", &fr_amin_);
    bind("fr_amax_", &fr_amax_);
    ...
}
XXXTCPAgent::command(int argc, const char*const* argv)
{
    if (argc == 3) {
        if (strcmp(argv[1], "target") == 0 ) {
            ...
        }
    }
    return (NsObject::command(argc,argv));
}
```

OTcl

```
set tcp [new Agent/TCP/XXX]
$tcp set fid_1
$tcp set fr_amin_ 0.2
$tcp set fr_amax_ 0.8
$tcp target [new Agent/Null]
```



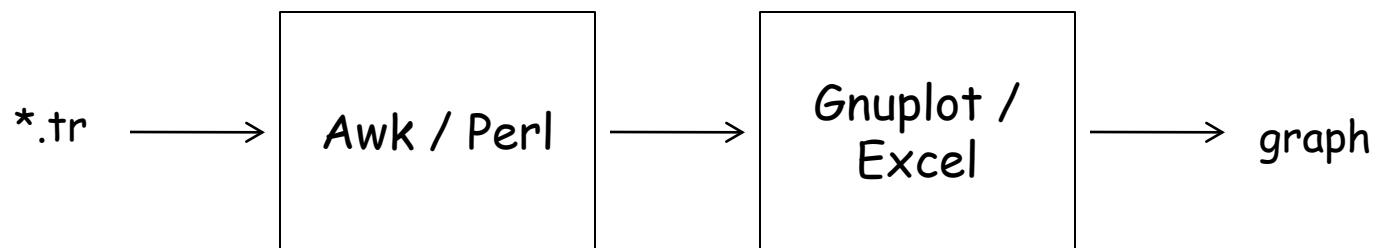
Ns-2 (8)

- Trace File (*.tr)

```
+ 1.84375 0 2 cbr 210 ----- 0 0.0 3.1 225 610
enqueue →
- 1.84375 0 2 cbr 210 ----- 0 0.0 3.1 225 610
dequeue →
r 1.84471 2 1 cbr 210 ----- 1 3.0 1.0 195 600
receive →
r 1.84566 2 0 ack 40 ----- 2 3.2 0.1 82 602
+ 1.84566 0 2 tcp 1000 ----- 2 0.1 3.2 102 611
- 1.84566 0 2 tcp 1000 ----- 2 0.1 3.2 102 611
r 1.84609 0 2 cbr 210 ----- 0 0.0 3.1 225 610
+ 1.84609 2 3 cbr 210 ----- 0 0.0 3.1 225 610
d 1.84609 2 3 cbr 210 ----- 0 0.0 3.1 225 610
drop →
- 1.84612 2 3 cbr 210 ----- 0 0.0 3.1 192 511
r 1.84612 3 2 cbr 210 ----- 1 3.0 1.0 196 603
```

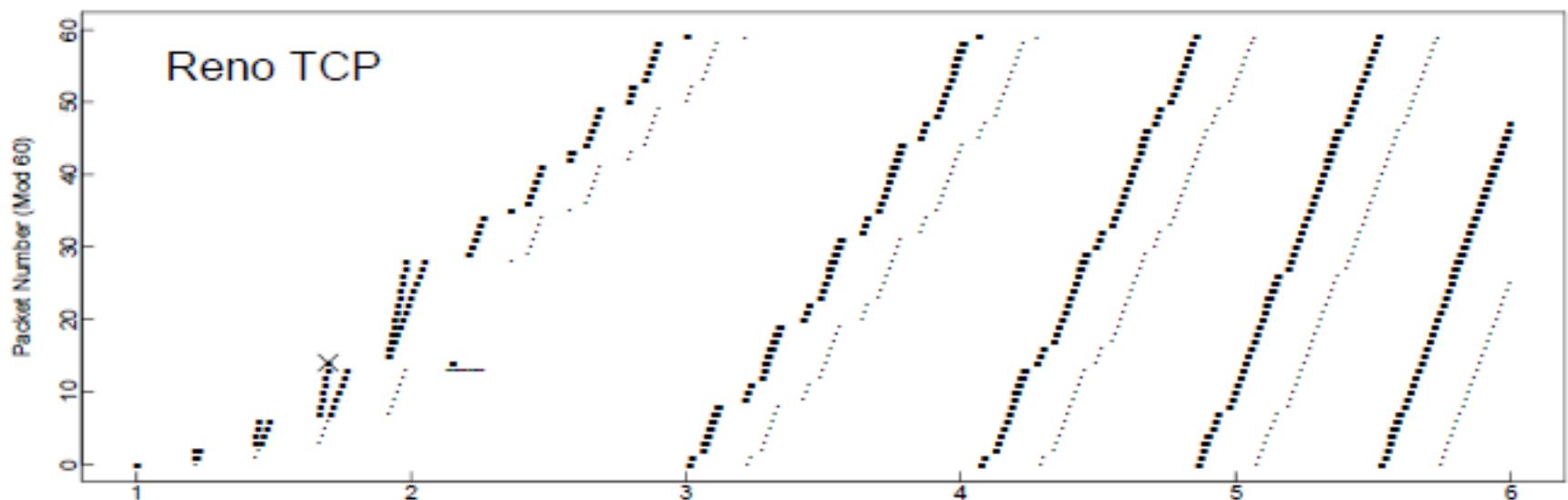
Ns-2 (9)

- Awk / Perl (script)
 - applied to trace files
 - generate graph files, e.g. for Gnuplot



Ns-2 (10)

- example



IEICE Network System Technical Group's Archive

- http://www.ieice.org/~ns/jpn/archive_s.html
 - 2009/8: ns-2 (summer school)
 - 2009/8: OPNET (summer school)
 - 2009/12: Qualnet (tutorial)

Googling "ns-2 tutorial" gives many sites
※ "ns-3" sites are increasing

ns-2 TCP-Linux

ns-2 TCP-Linux (1)

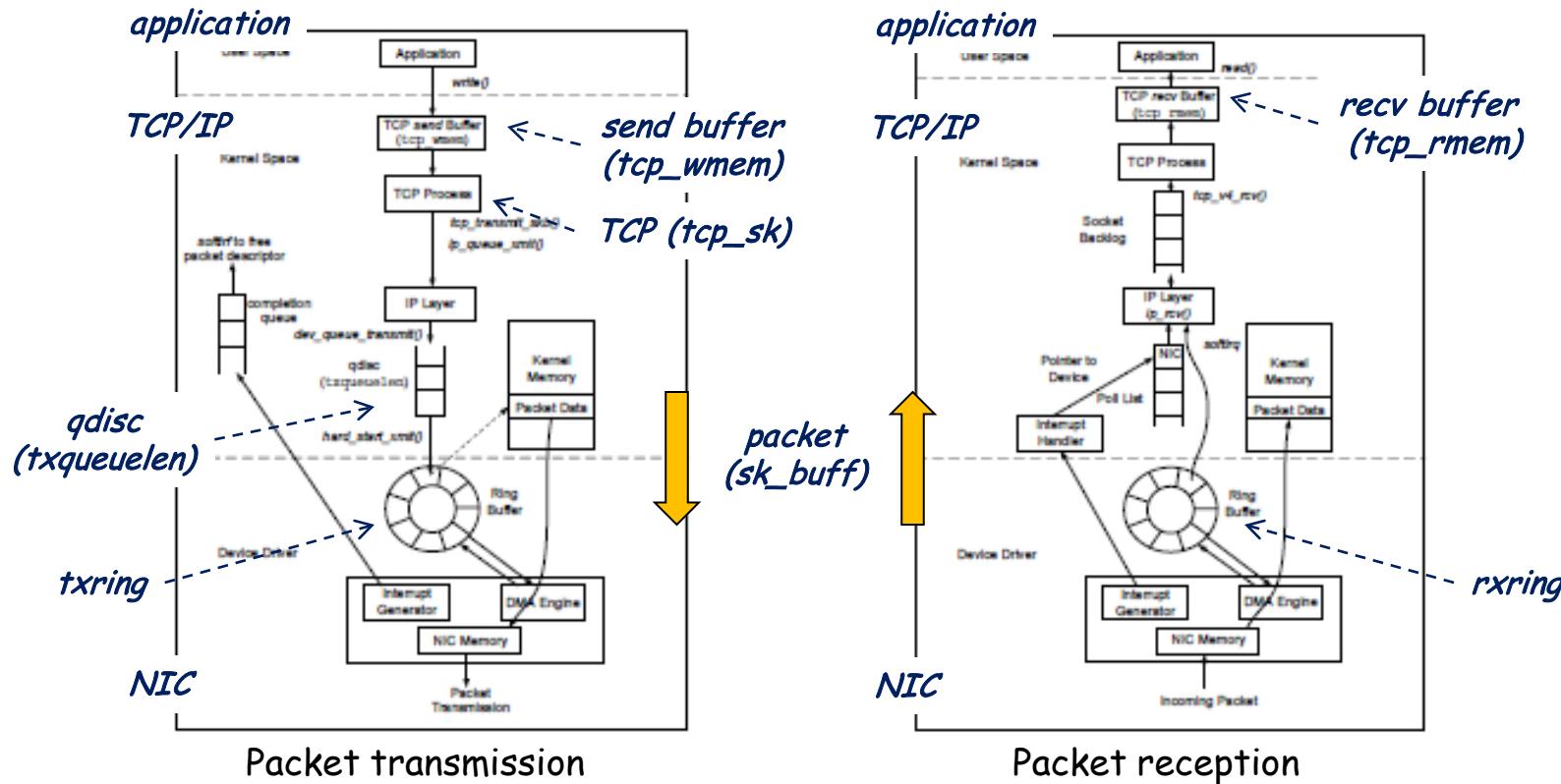
- ns-2 simulation using TCP implementation code in Linux kernel
 - bridge between implementations (Linux kernel) and simulations (ns-2)
 - fill a gap between implementation and simulation
 - verification of implementation codes

ns-2 TCP-Linux (2)

- TCPs implemented in Linux kernel (2.6.16-3)
 - TCP-Reno, TCP-Vegas, HighSpeed-TCP, Scalable-TCP, BIC-TCP, CUBIC-TCP (default), TCP-Westwood, H-TCP, TCP-Hybla
- TCPs to be implemented
 - TCP-Veno, TCP-LowPriority, Compound-TCP (Windows)

ns-2 TCP-Linux (3)

- TCP Implementation in Linux



ns-2 TCP-Linux (4)

- Variables in `tcp_sk`

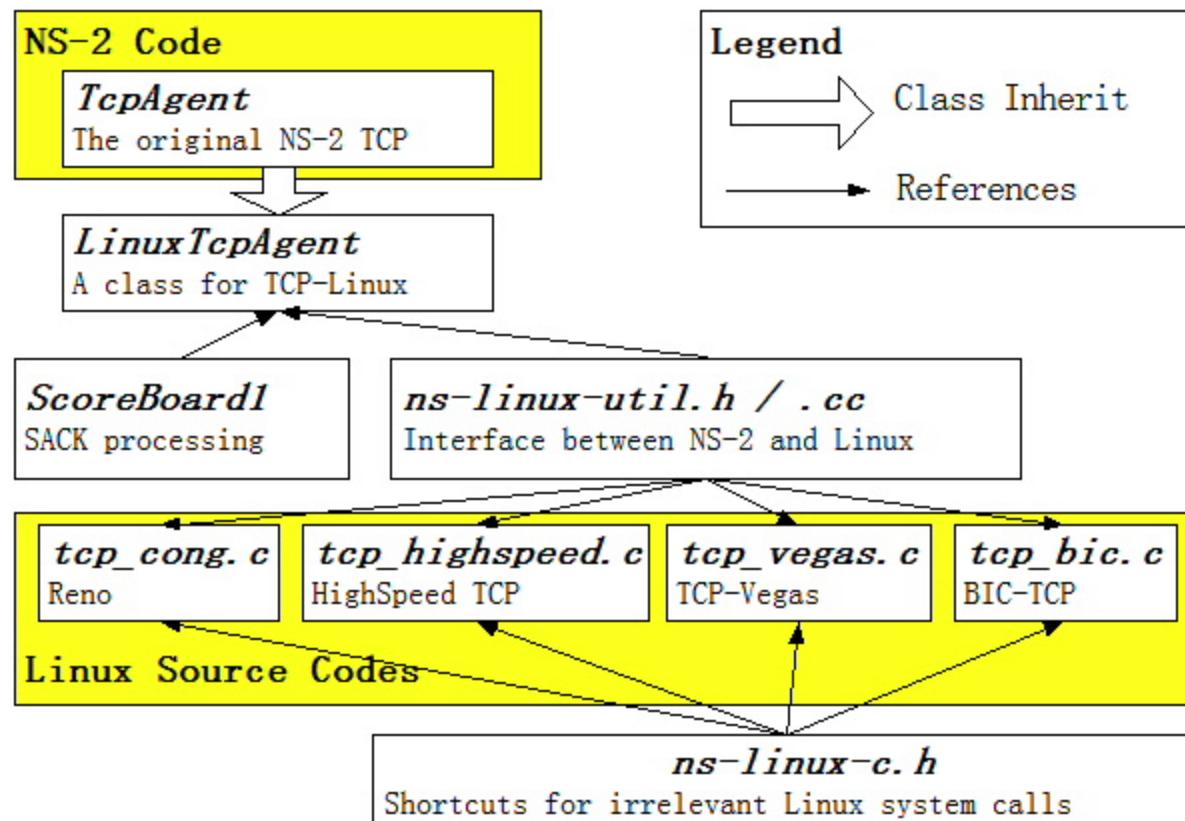
Name	Meaning	Equivalent in NS-2 <i>TCPAgent</i>
<code>snd_ssthresh</code>	slow-start threshold	<code>ssthresh_</code>
<code>snd_cwnd</code>	integer part of the congestion window	<code>trunc(cwnd_)</code>
<code>snd_cwnd_cnt</code>	fraction of congestion window	<code>trunc(cwnd_ ^ 2)</code> <code>%trunc(cwnd_)</code>
<code>icsk_ca_priv</code>	a 512-bit array to hold per-flow state for a congestion control algorithm	n/a
<code>icsk_ca_ops</code>	a pointer to the congestion control algorithm interface	n/a

Congestion control modules:

`cong_avoid`: slow start & congestion avoidance
`ssthresh`: loss event handling
`min_cwnd`: fast retransmission

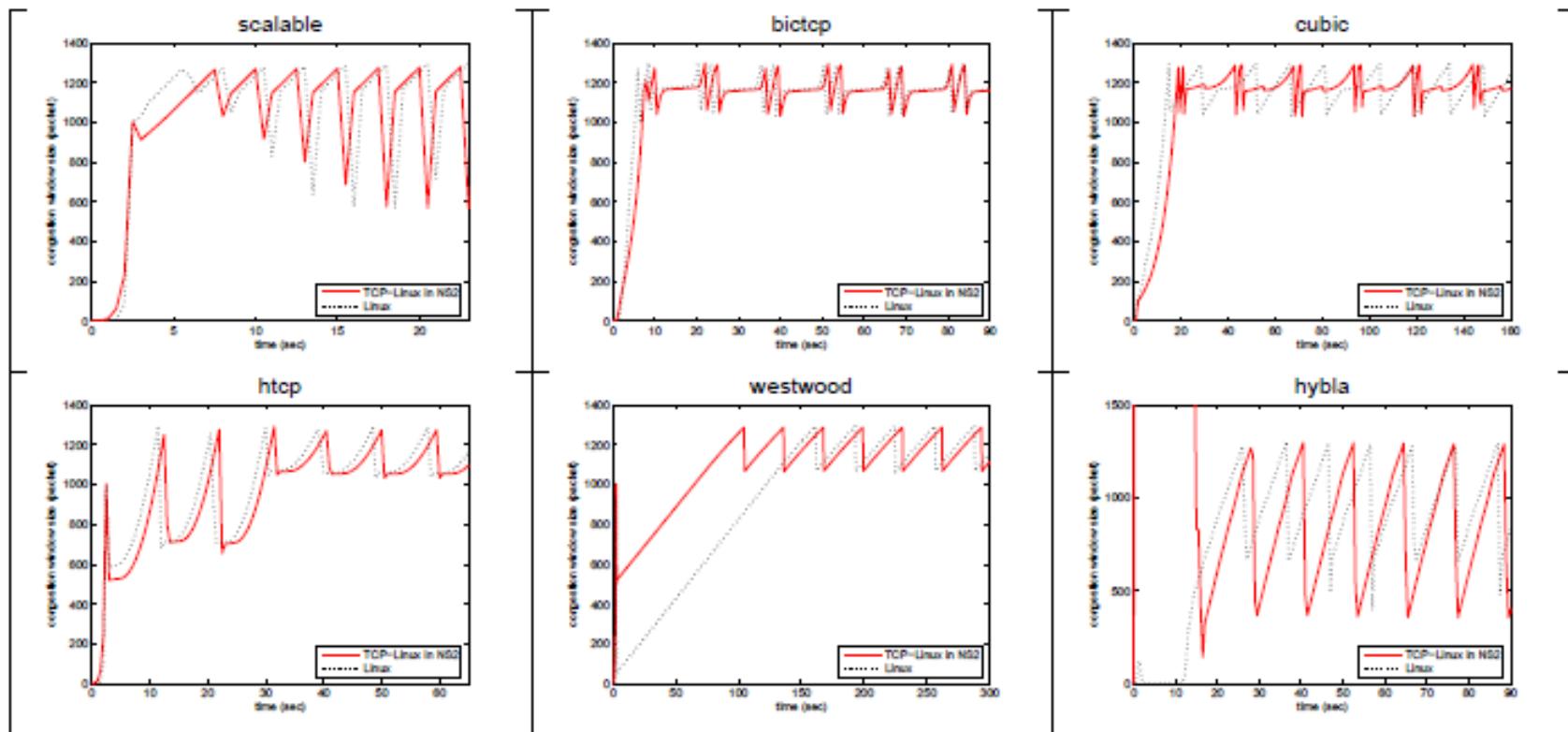
ns-2 TCP-Linux (5)

- Code structure



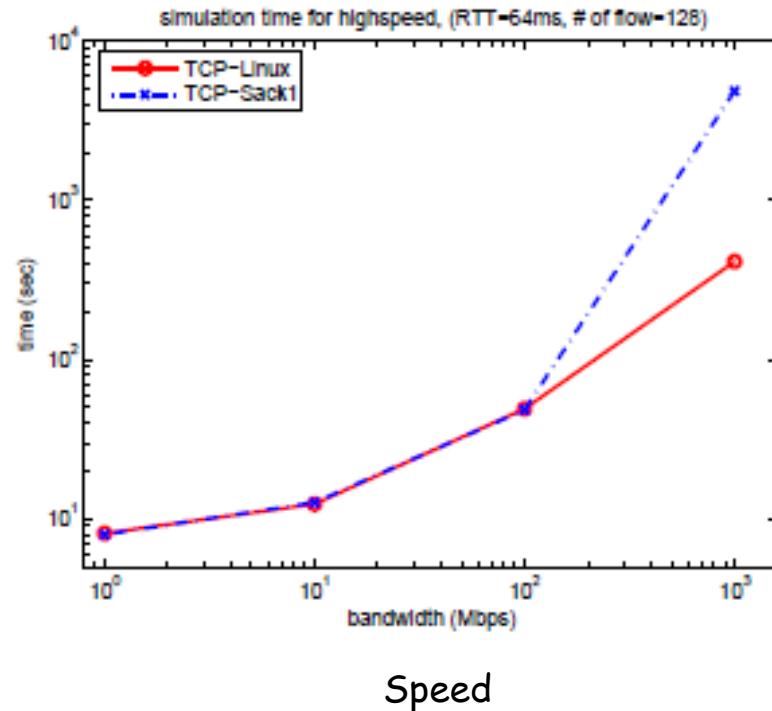
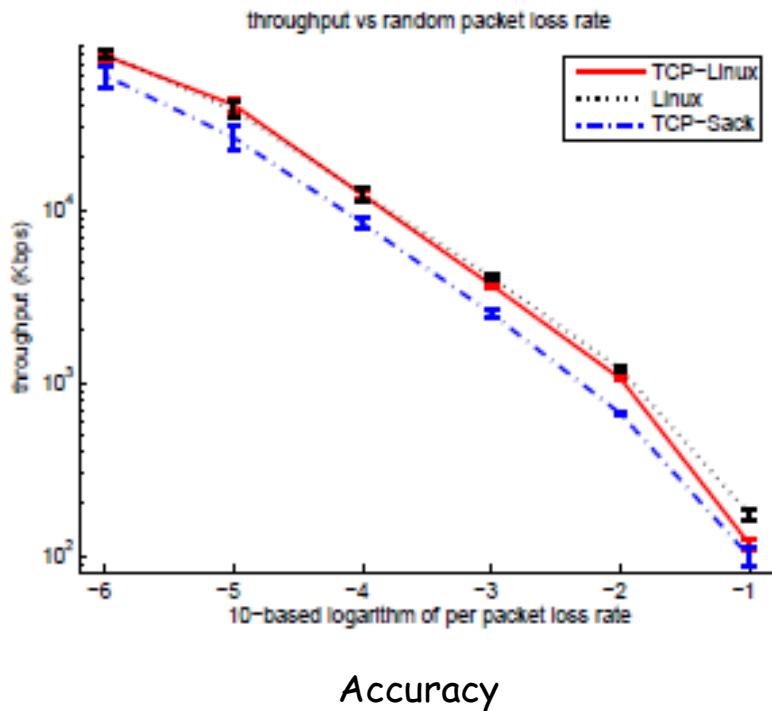
ns-2 TCP-Linux (6)

- Simulation (1) ns-2 & Linux



ns-2 TCP-Linux (7)

- Simulation (2) accuracy & speed



ns-3

ns-3 software overview

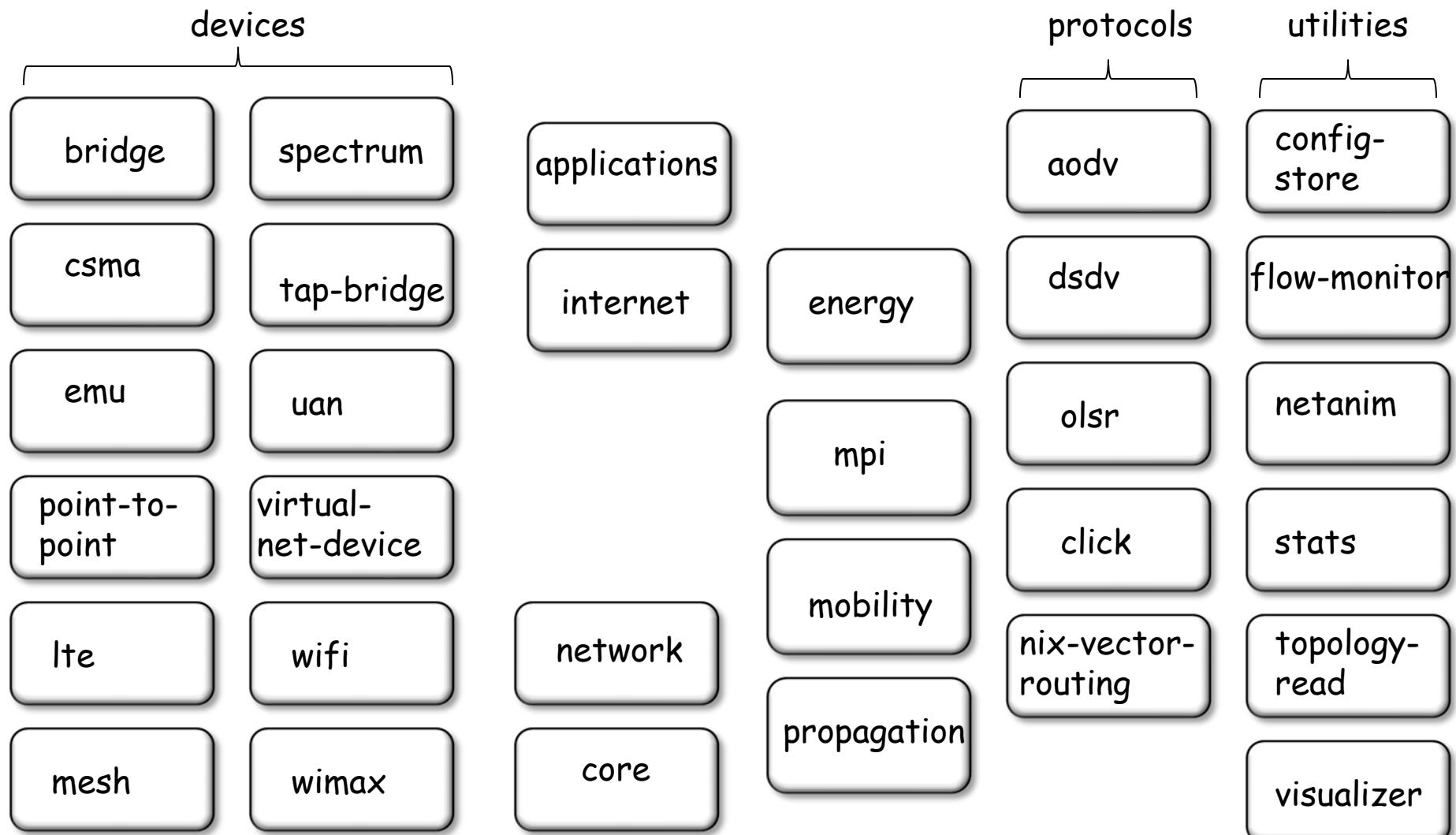
- ns-3 is written in C++, with bindings available for Python
 - simulation programs are C++ executables or Python programs
 - Python is often a glue language, in practice
- ns-3 is a GNU GPLv2-licensed project
- ns-3 lacks an integrated development / visualization environment (IDE)
- ns-3 is not backwards-compatible with ns-2

ns-3 development process

ns-3 is run as an open source project
backed by research funding

- GPLv2 licensing
- open mailing lists
- uses standard tools (Mercurial, Bugzilla,
Mediawiki, GNU/Linux development)
- ~20 maintainers worldwide

Available modules (ns-3.11 May 2011)



Analytics

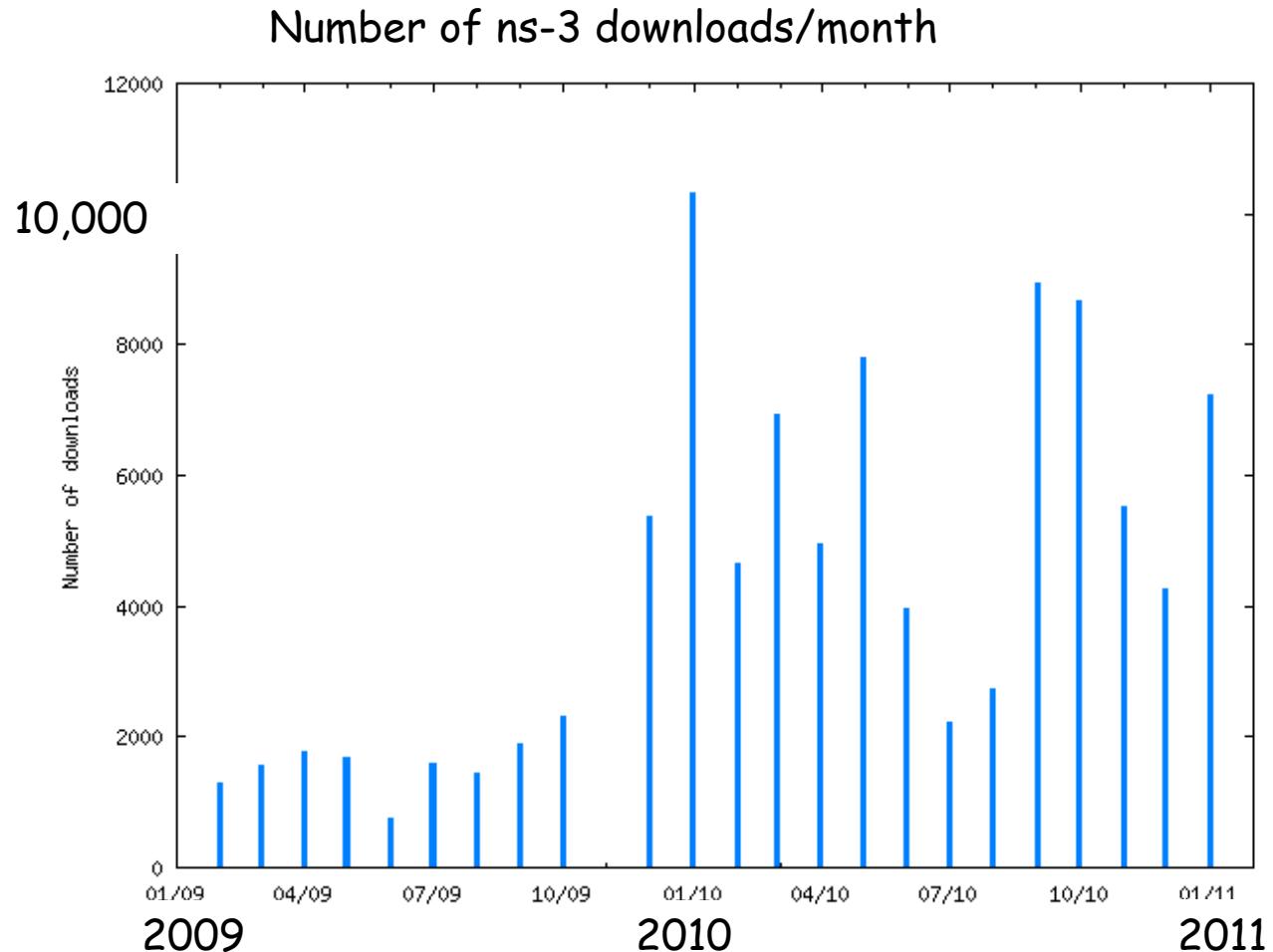
Mailing list

subscriptions:

- ns-3-users: 963
- ns-developers: 1176

Downloads:

- 6000/month in 2010



summary of ns-3 features

- modular, documented core
- C++ programs or (optionally) Python scripting
- alignment with real systems (sockets, device driver interfaces)
- emphasis on software integration
- virtualization and testbed integration are a priority (emulation modes)
- well-documented attribute system
- updated models

Resources

Web site:

<http://www.nsnam.org>

Mailing list:

<http://mailman.isi.edu/mailman/listinfo/ns-developers>

IRC: #ns-3 at freenode.net

Tutorial:

<http://www.nsnam.org/docs/tutorial/tutorial.html>

Code server:

<http://code.nsnam.org>

Wiki:

http://www.nsnam.org/wiki/index.php/Main_Page

"Architecture, design and source code comparison of ns-2 and ns-3 network simulators", 2010 Spring Simulation Multi-conference

<http://portal.acm.org/citation.cfm?id=1878651>