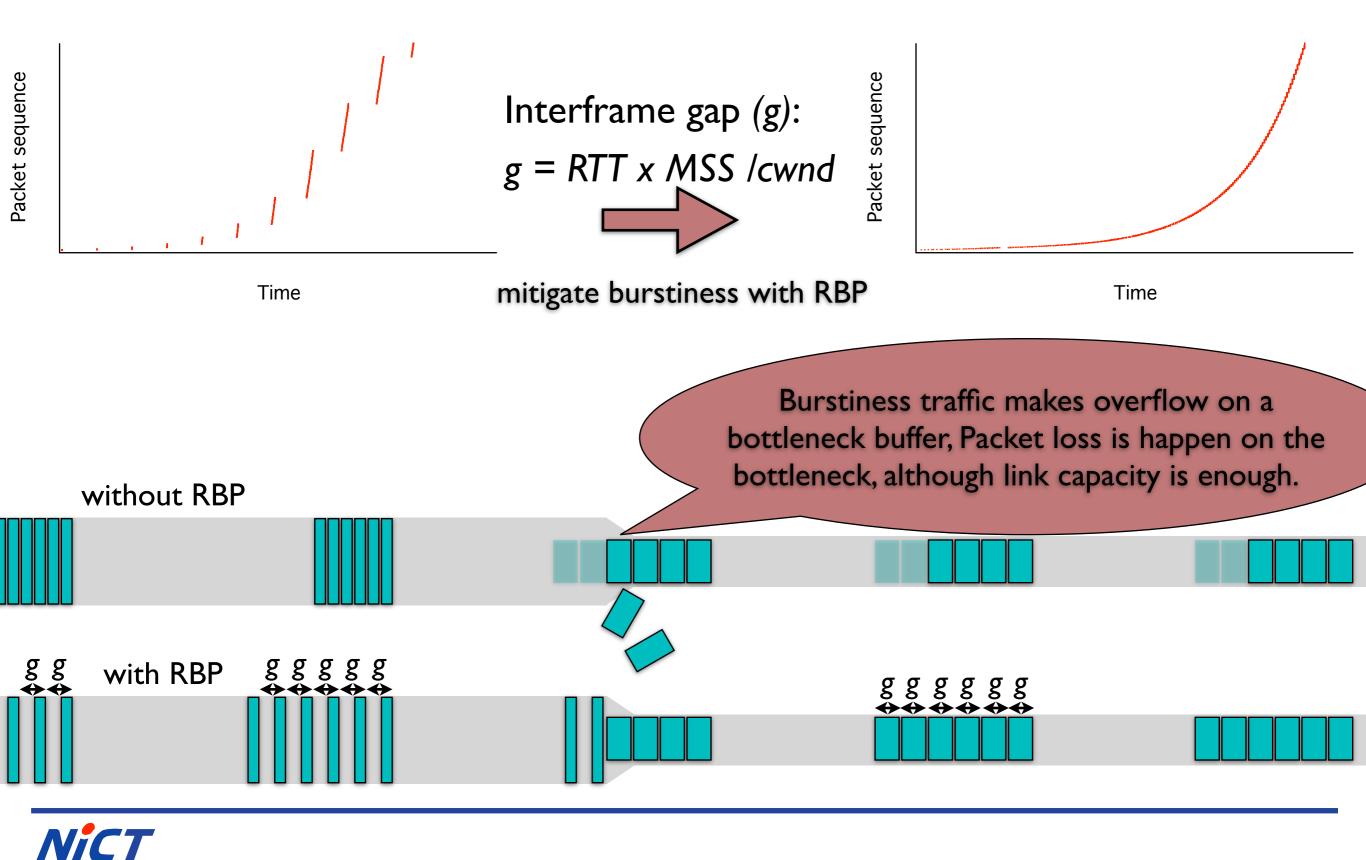
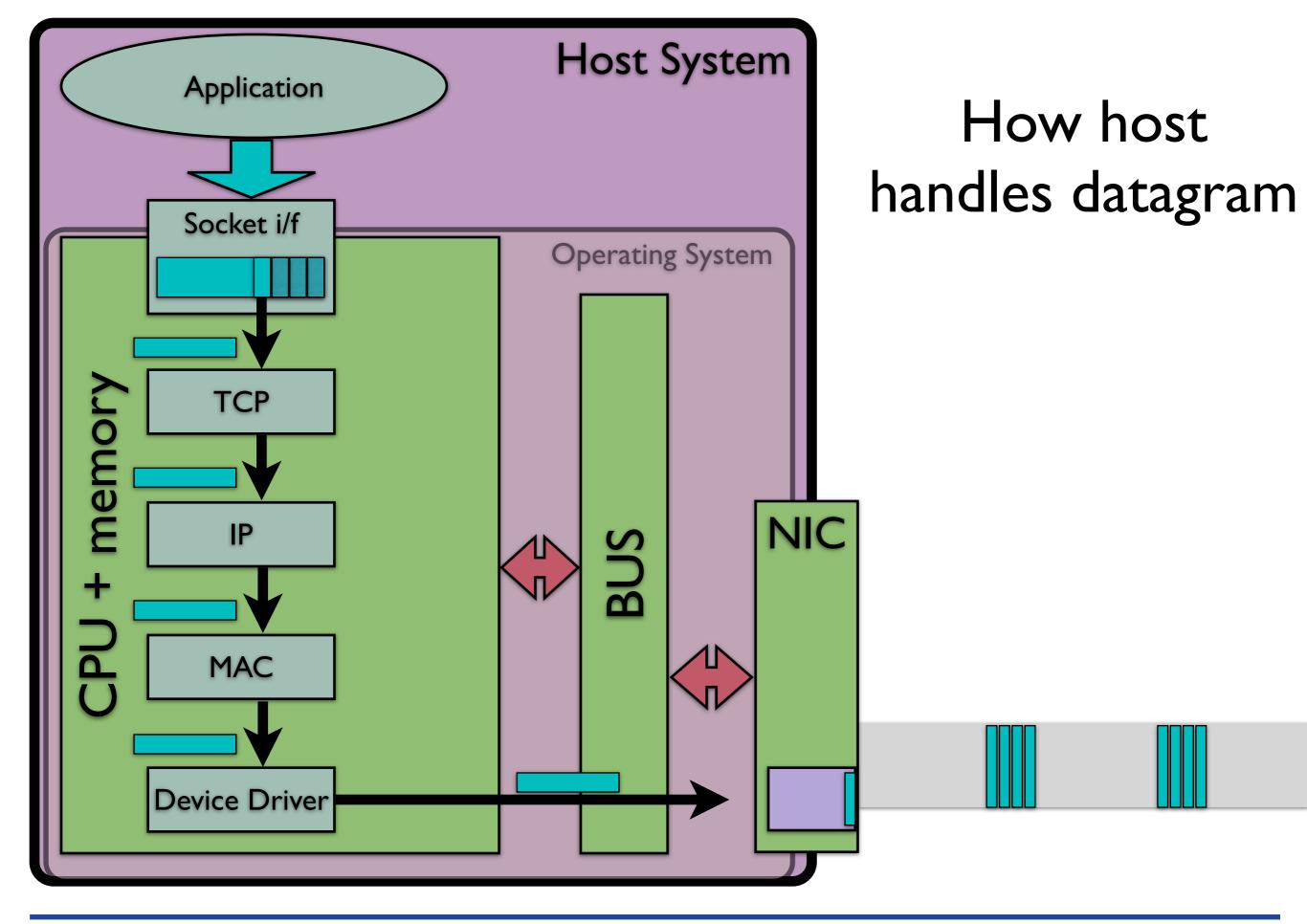
Transmission Timer Approach for Rate Based Pacing TCP with Hardware Support

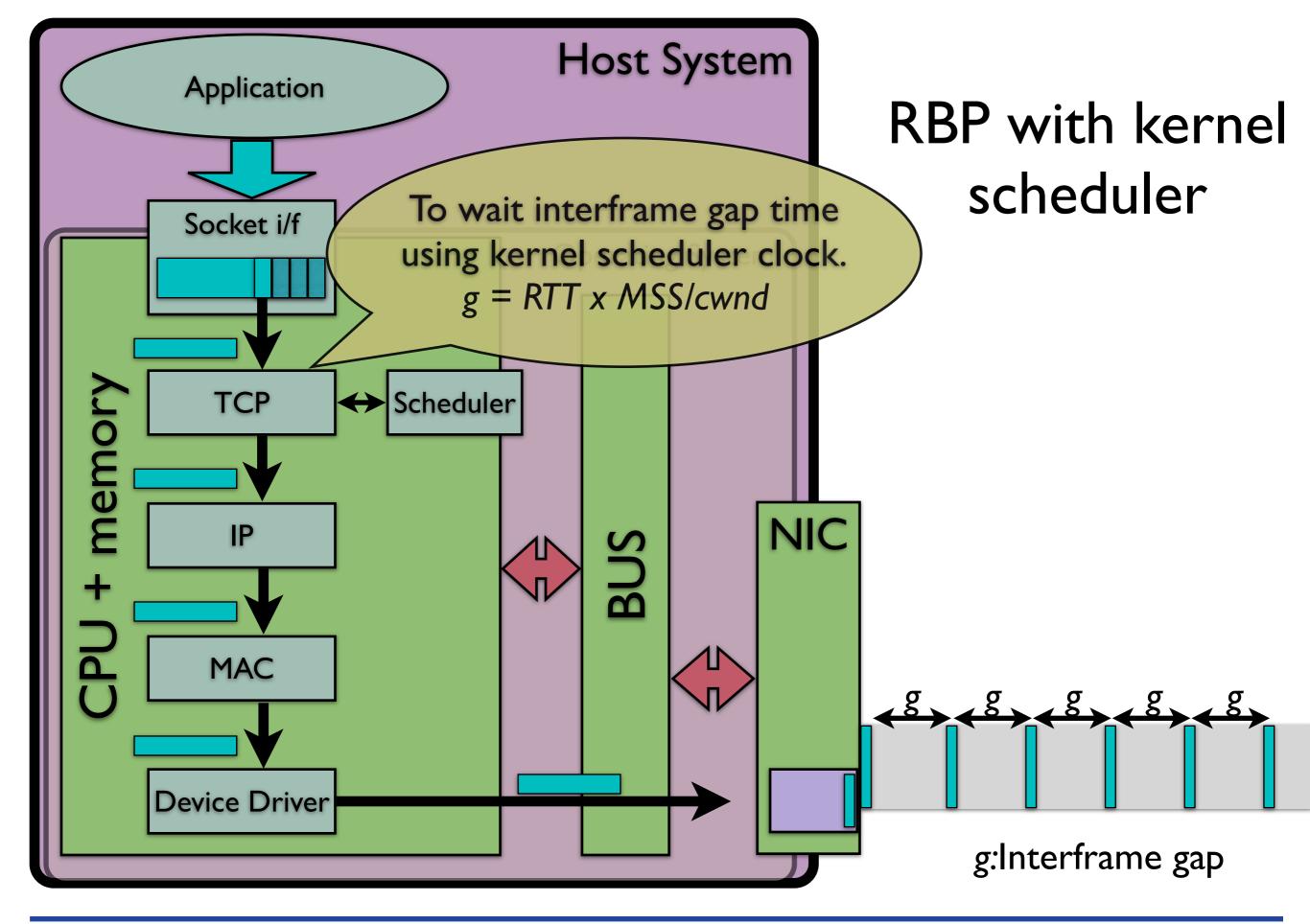
Katsushi Kobayashi <u>ikob@koganei.wide.ad.jp</u> NICT, JAPAN

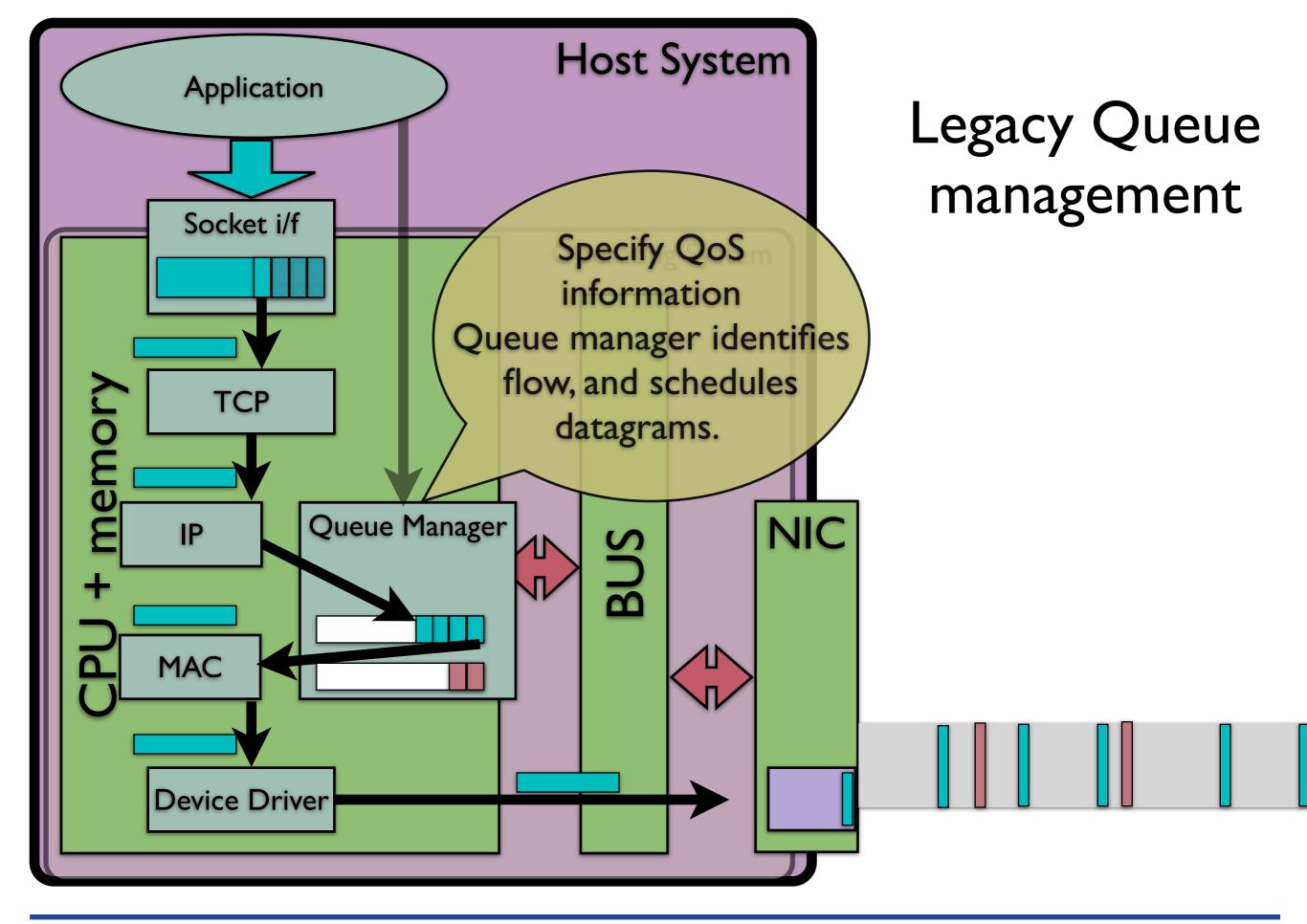


TCP slow-start burstiness and RBP







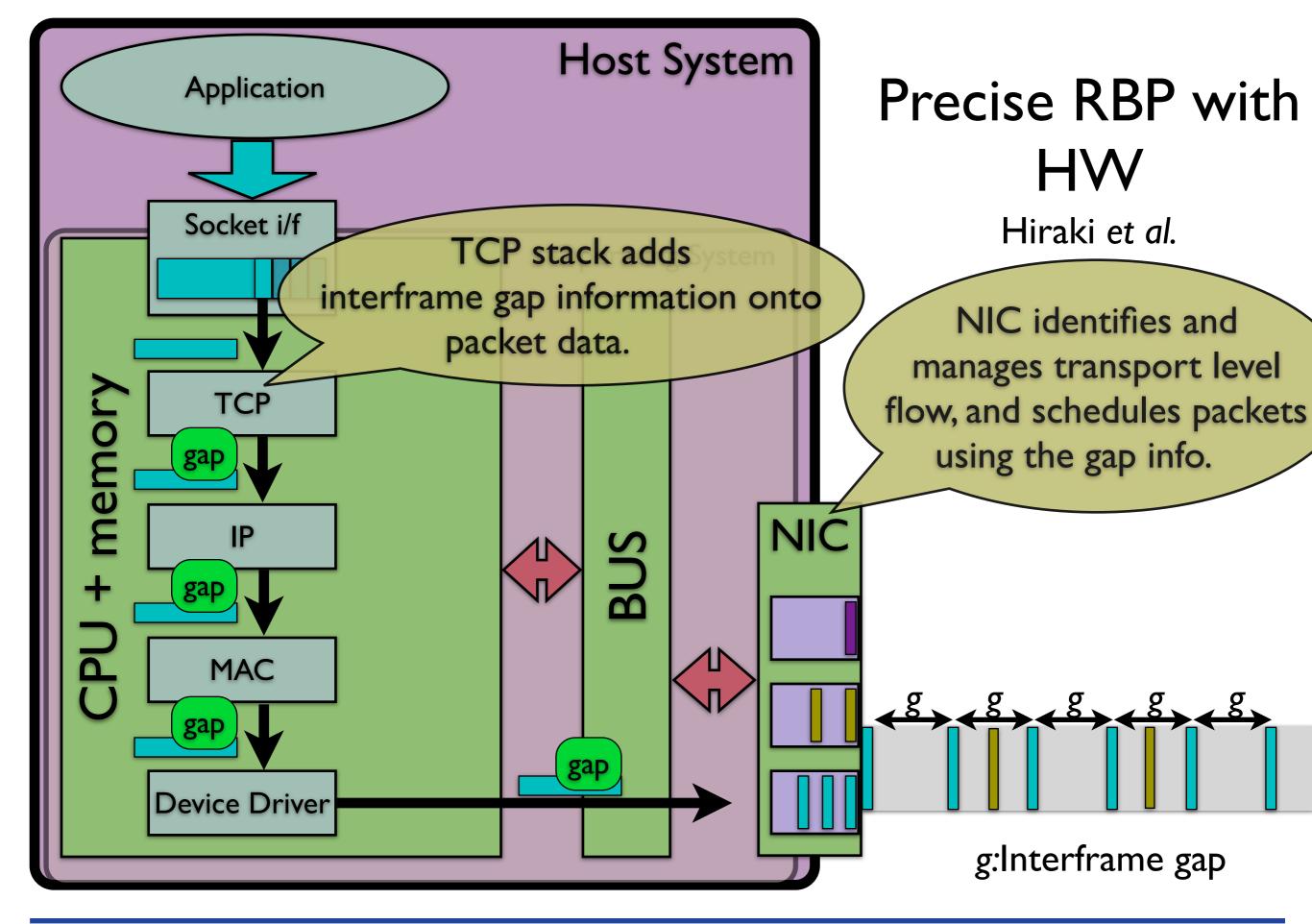


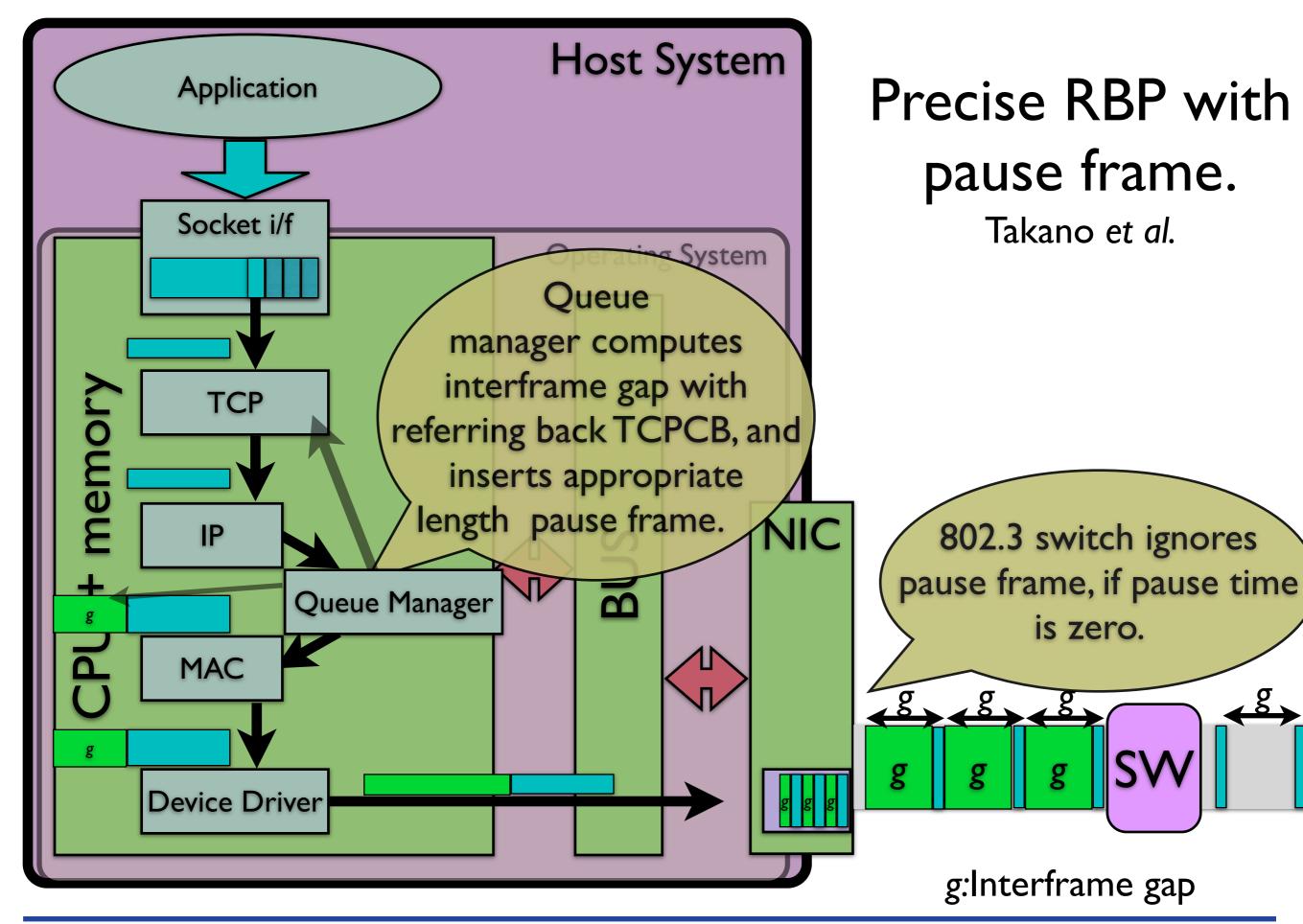
RBPTCP with OS scheduler

Interframe gap@1500Bytes packet size

Bandwidth	Interframe gap	
100Kbps	120 msec.	
IMbps	12 msec.	
10Mbps	I.2 msec.	
100Mbps	120 u sec.	
IGbps	12 u sec.	
10Gbps	I.2 u sec.	

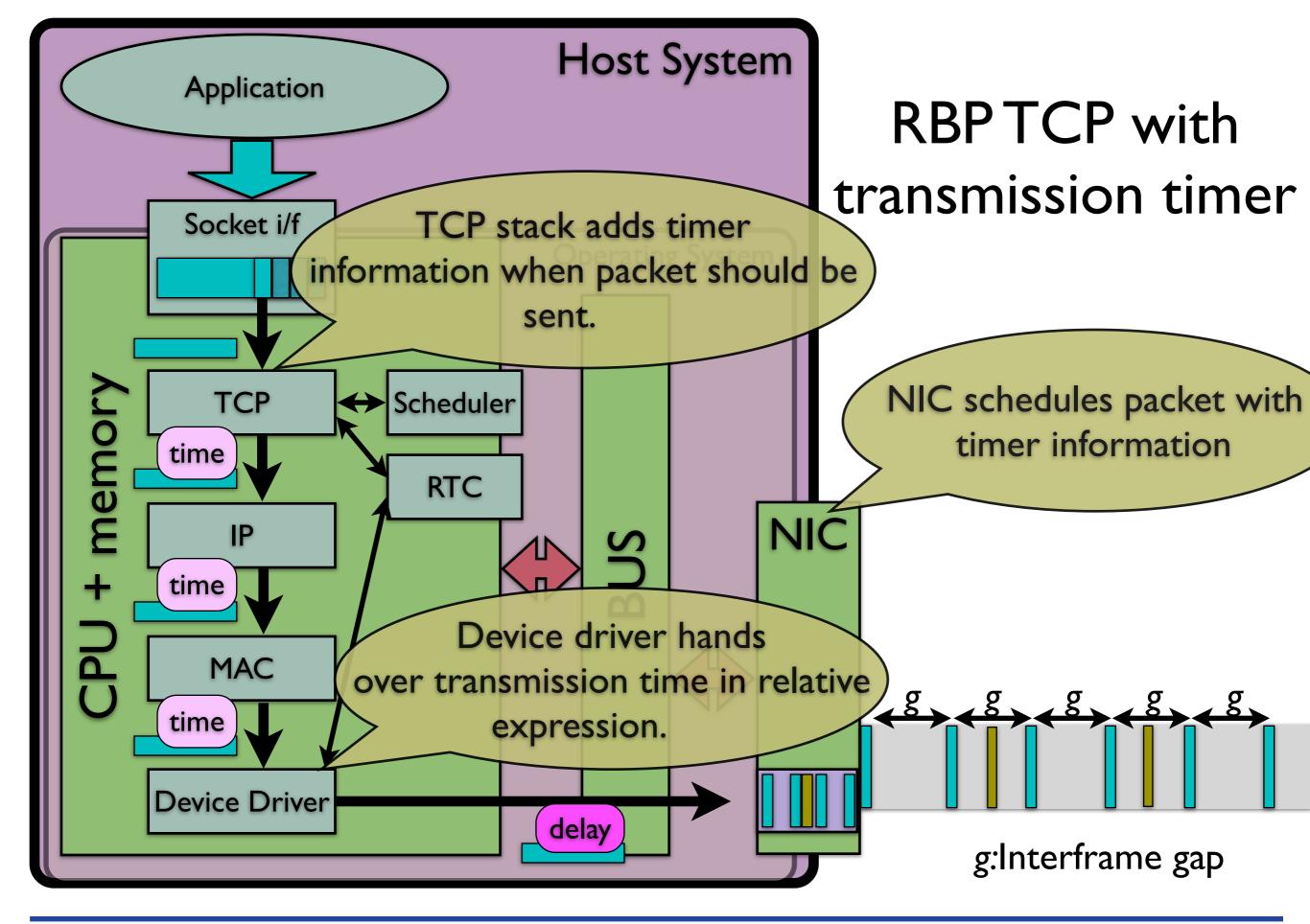
- Interframe gap@higher bandwidth communication
 < OS scheduler tick granularity as 1 - 10 m sec.
- Precise rate based pacing mechanism not to rely on kernel scheduler is required.





Requirement for precise packet pacing

- Precise packet pacing more than OS scheduler granularity
- Follow frequent interframe gap (rate) changing
- Re-order packet at sender host even on the same connection stream.
- Free from upper layer
- Not depend specific transport protocol.



NIC implementation

- Intel IXP2400 NP (Radisys ENP2611) act as PCI GbE NIC
 - 3 output queues: transmission timer, high priority, bast effort
 - timer value (delay) specified in DMA descriptor by device driver
- up to 100 milli second delay using 2 level scheduler
 - 4,096 ring bin slots, 27 micro sec./slot
 - NP scheduler, 26.7 nano sec. granularity

RBPTCP implementation

- FreeBSD 4.10
- Add transmission time attribute to mbuf
- Add transmission time field of last segment to tcpcb
- Precise RTT measurement for computing interframe_gap : interframe_gap = RTT x MSS/cwnd;
- Estimate transition_time for each segment: tranmission_time
 - = interframe_gap + transmission_time of previous segment;
- Decide whether send it, or postpone next events:

if (transmission_time > current_time() + scheduler_tick)

tsleep(); sleep until next scheduler interrupt **APP with Software**

else{

update mbuf with transmission_time;

ip_output();

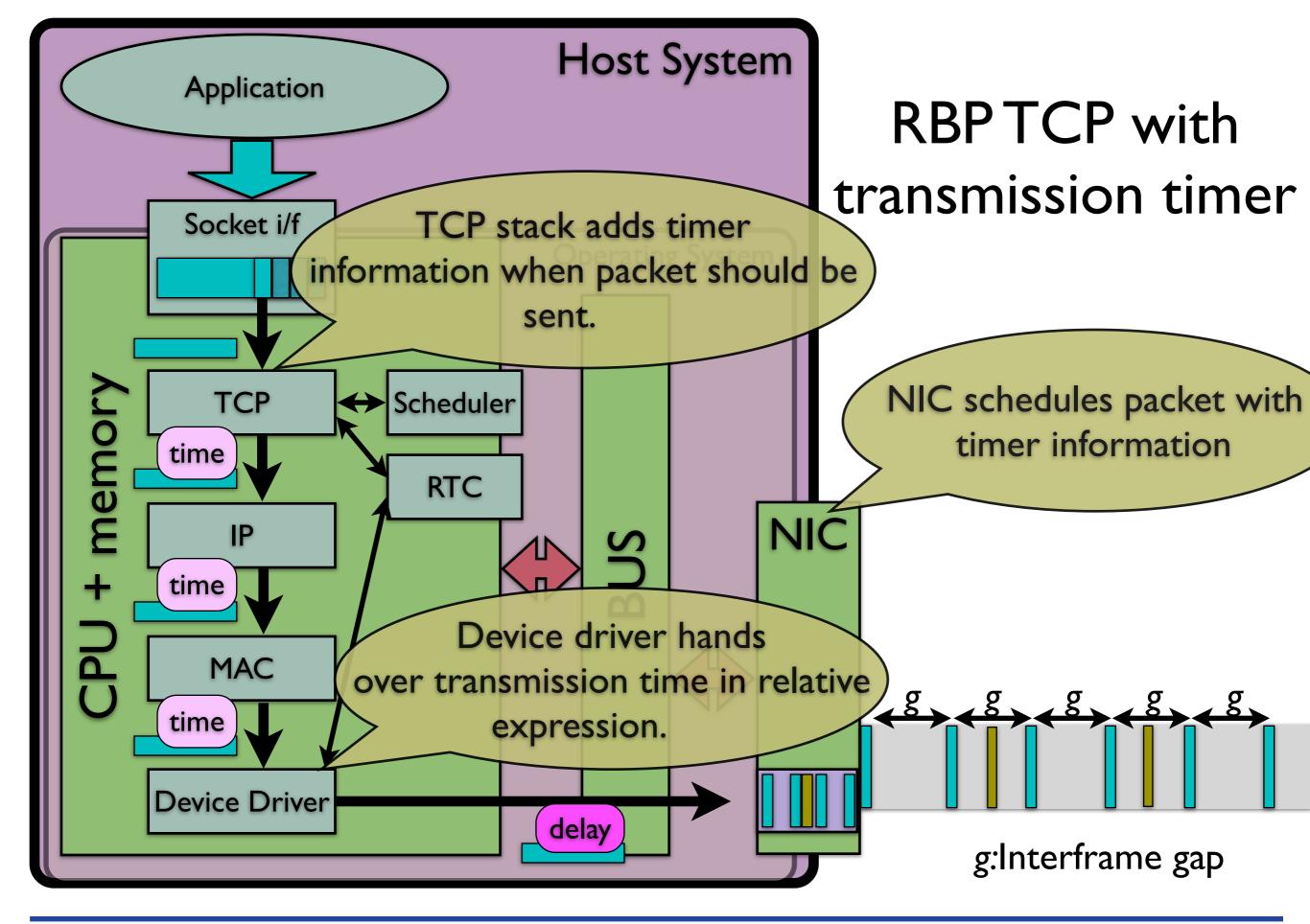




RBPTCP implementation (cont'd)

- Mainly modify tcp_output(), not to modify IP layer.
- Less than 1,000 line hacking in TCP
 - more than 30,000 is TCP code in FreeBSD
- FreeBSD TCP stack does not record the buffer pointer last transmitted.
 - Buffer traverse process is required at every packet.
 - Suppress TCP throughput on large *cwnd*.
 - borrowed from NetBSD 2.0 code.
- Too many referring RTC
 - Suppress TCP throughput with original microtime()





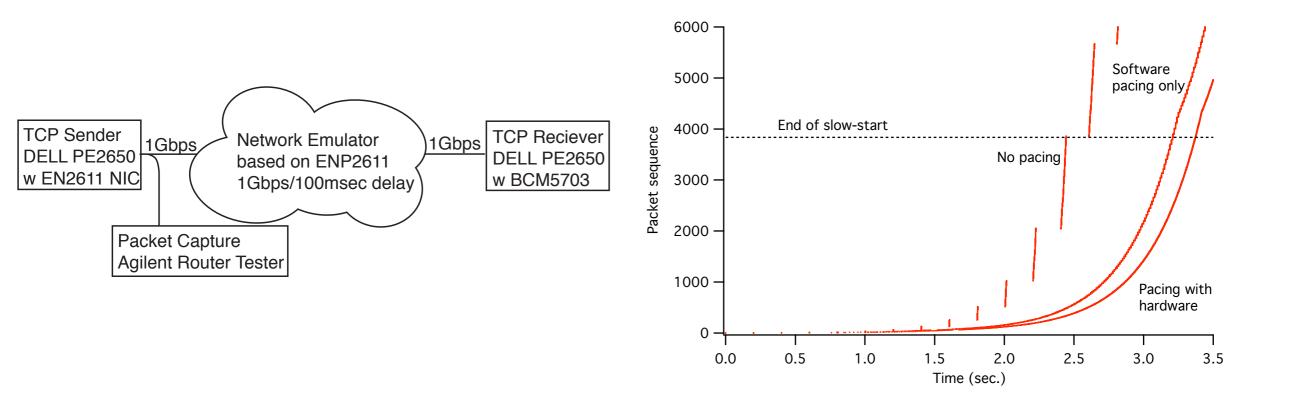
RTC performance

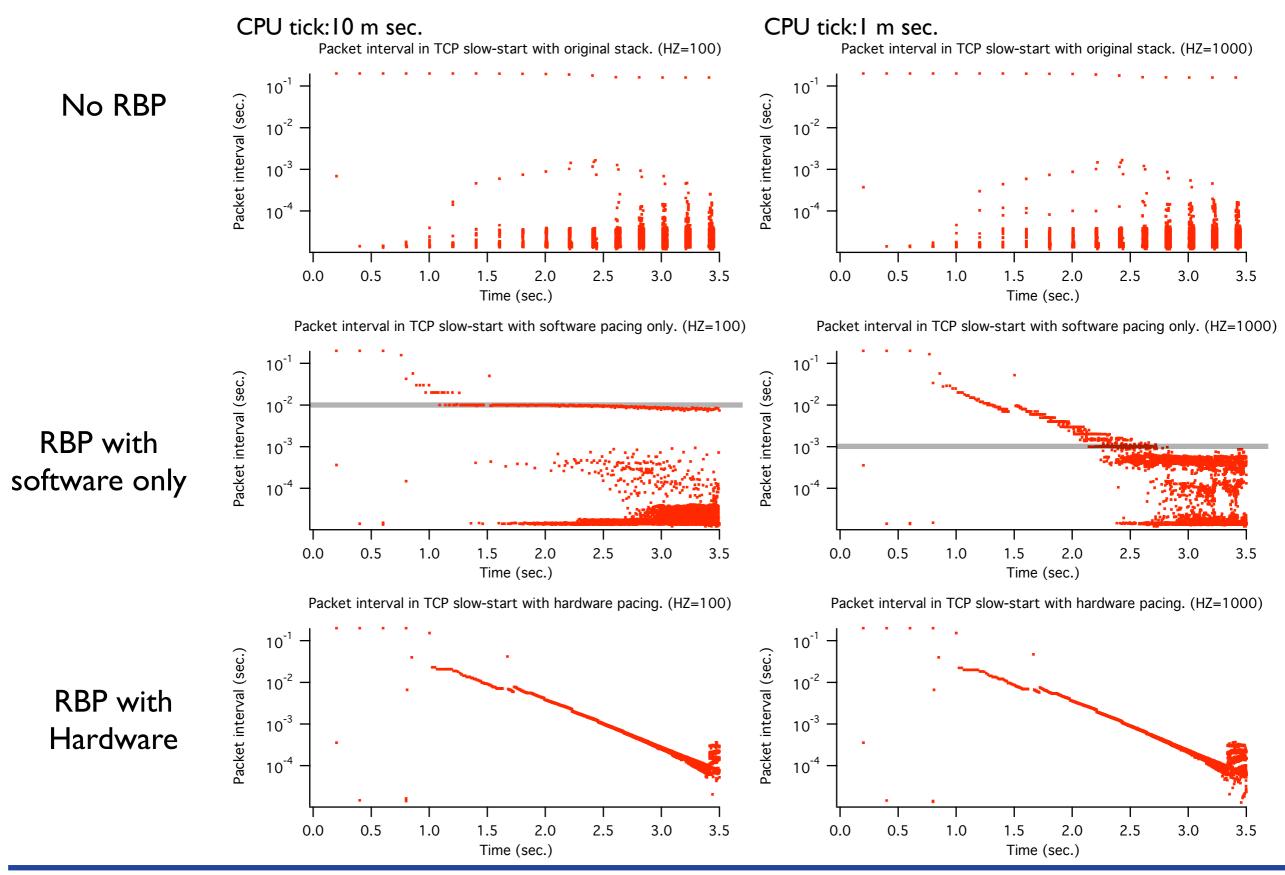
- 2 function is provided by original FreeBSD
 - microtime(): accesses RTC, higher accuracy result.
 - getmicrotime(): returns cached value that is updated each scheduler interrupt
- light microtime(): to correct getmicrotime() result with CPU TSC (Time Stamp Counter) on x86
 - TSC counts CPU clock cycle
 - Used for performance evaluation for program code.
 - x 40 faster than original microtime() with enough accuracy.

	elapsed TSC elapsed time	accuracy of result	packet throughput limit due to RTC overhead
microtime()	16,000 count 5 micro sec.	micro sec.	200K packets/sec. 2.4Gbps@1,500 Bytes/packet
getmicrotime()	89 count < 30 n sec.	I-10 m sec. (depend on scheduler)	330M packets/sec. 3Tbps@1,500 Bytes/packet
light microtime()	360 count 120 n sec.	micro sec.	83 M packets/sec. ITbps@I,500Bytes/packet

RBPTCP experiment

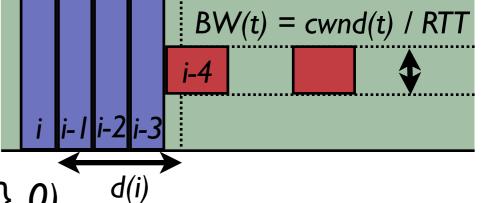
- IGbps/100 m sec. delay (200 m sec.RTT) between sender and receiver
 - no bottleneck between sender and receiver
 - no competitive flow
 - Agelent Router Tester captured packets on sender side.





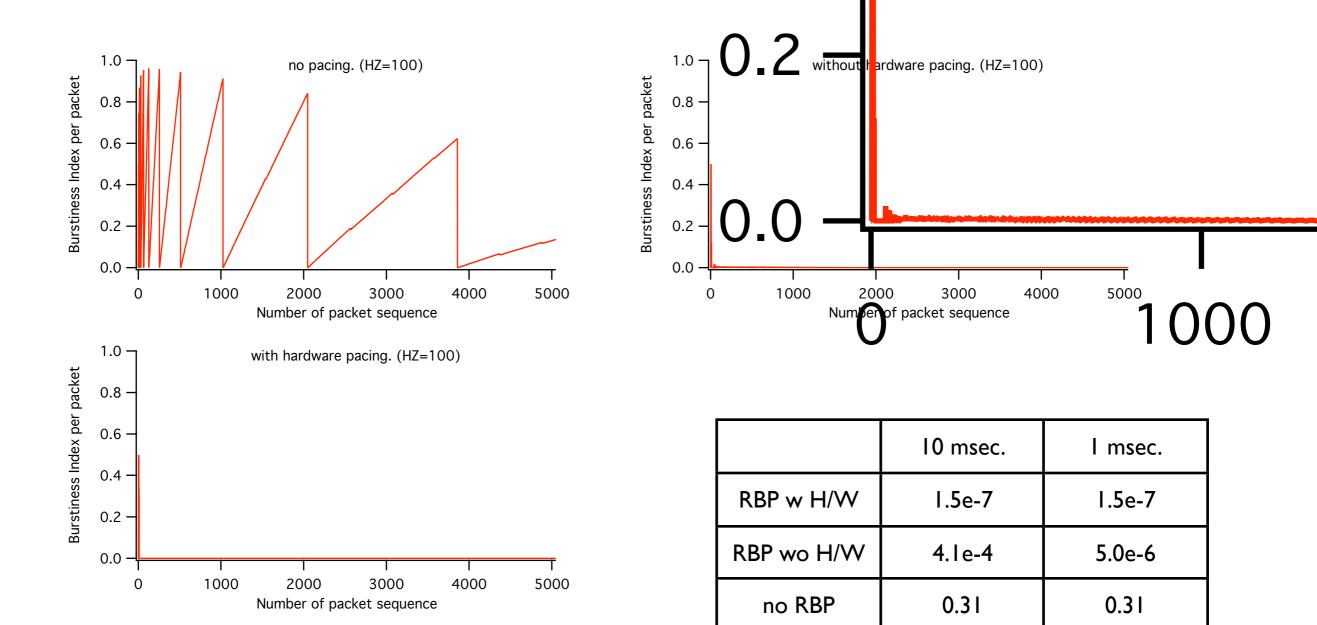
Buestiness index on slow-start

- Bandwidth BW(t) and serialization delay sigma(t) derived as:
 - BW(t) = cwnd(t) / RTT, sigma(t) = RTT/cwnd(t)
- *cwnd(t)* is assumed in exponential growth:
 - $cwnd(t) = 2^{t} (t / RTT) \times MS$
- Packet delay *d*(*i*) at bottleneck:
 - d(i) = max(d(i 2) + sigma(t_i) t_i t_{i-1}, 0)

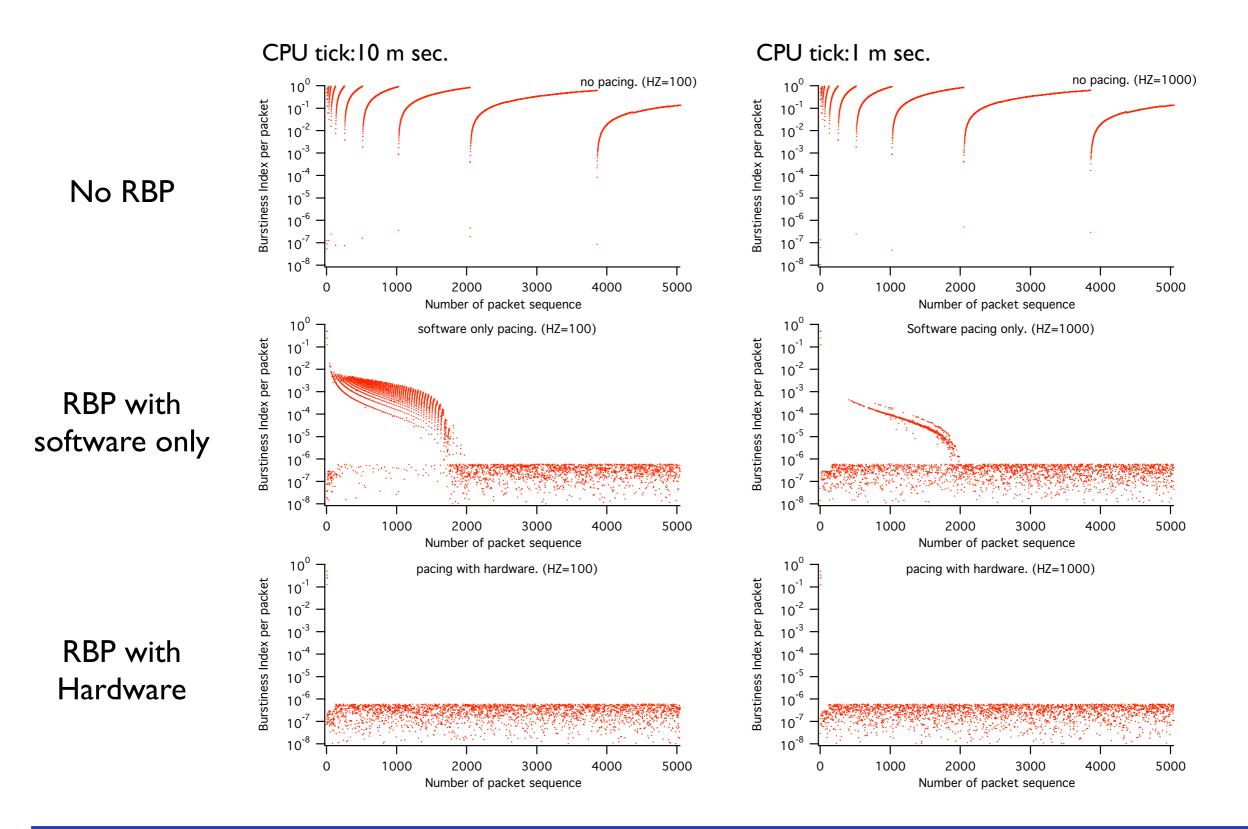


- Burstiness index b is derived normalizing with RTT:
 - b(i) = d(i)/RTT, B = sum(b(i))
- *b(i)* represents how much buffer is required at bottleneck by the stream itself.
 - If b(i) = I: delay-bandwidth product size
 - If b(i) = 0: no buffer, complete RBP.

Buestiness index result (1)

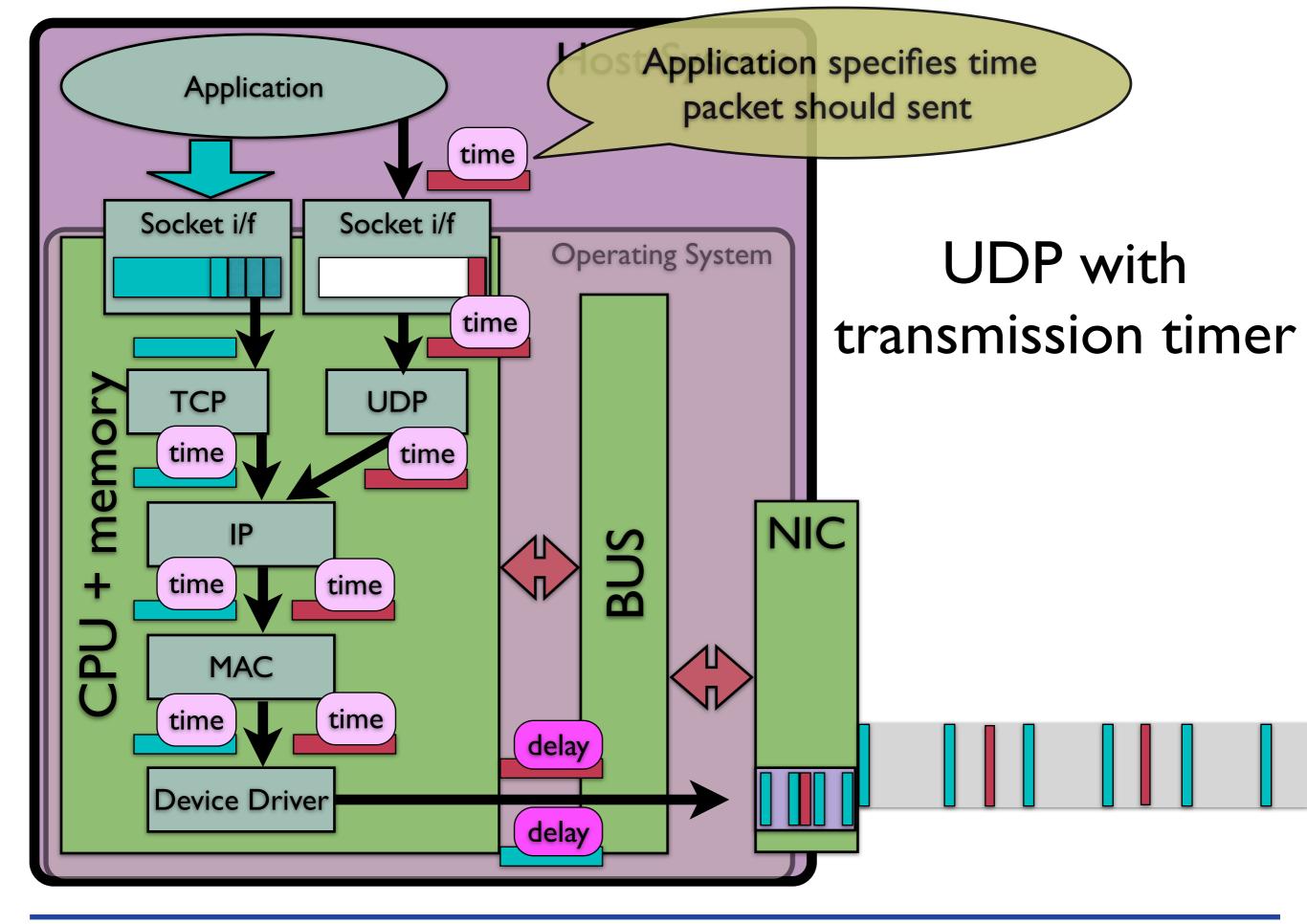


Buestiness index result (2)



Transmission Timer for UDP

- SO_TIMESTAMP is provided to obtain packet received time information.
 - struct timeval data accommodated as ancillary data with recvmsg() system call.
 - Receive only.
- To extend SO_TIMESTAMP at sending UDP datagram to specify the time that the application expects.



Conclusion

- Transmission timer approach for RBP
 - Framework
 - Implementation
 - Result
- Is our approach including interlayer interaction worse ?
- Does NIC vender interest this kind approach ?