

# 無線網路多媒體系統

# Wireless Multimedia System

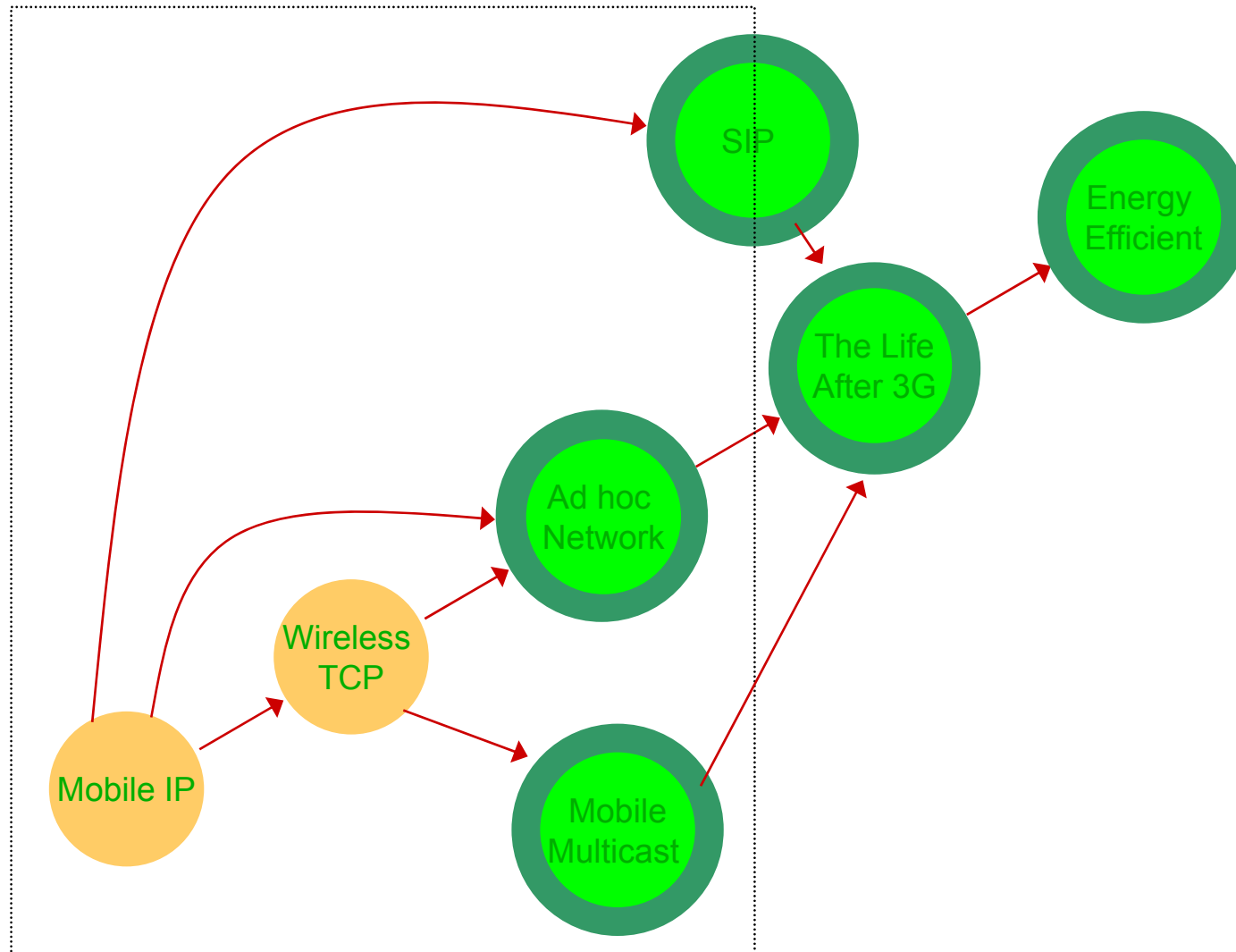
## Lecture 8: Wireless TCP

吳曉光博士

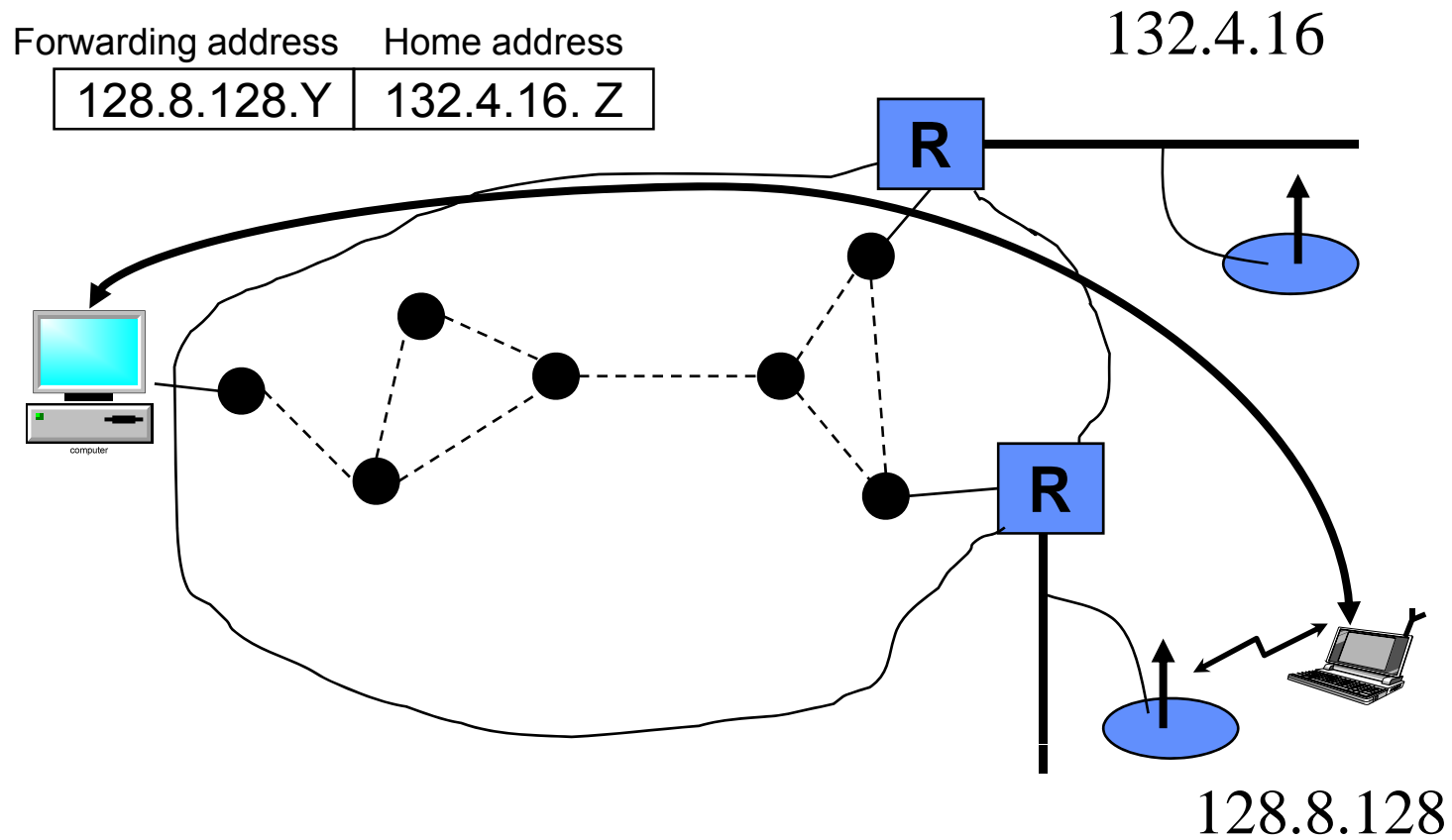
<http://wmlab.csie.ncu.edu.tw/wms>

We  
provide  
無線網路多媒體實驗室  
Wireless  
Wireless Network & Multimedia Laboratory  
Solution

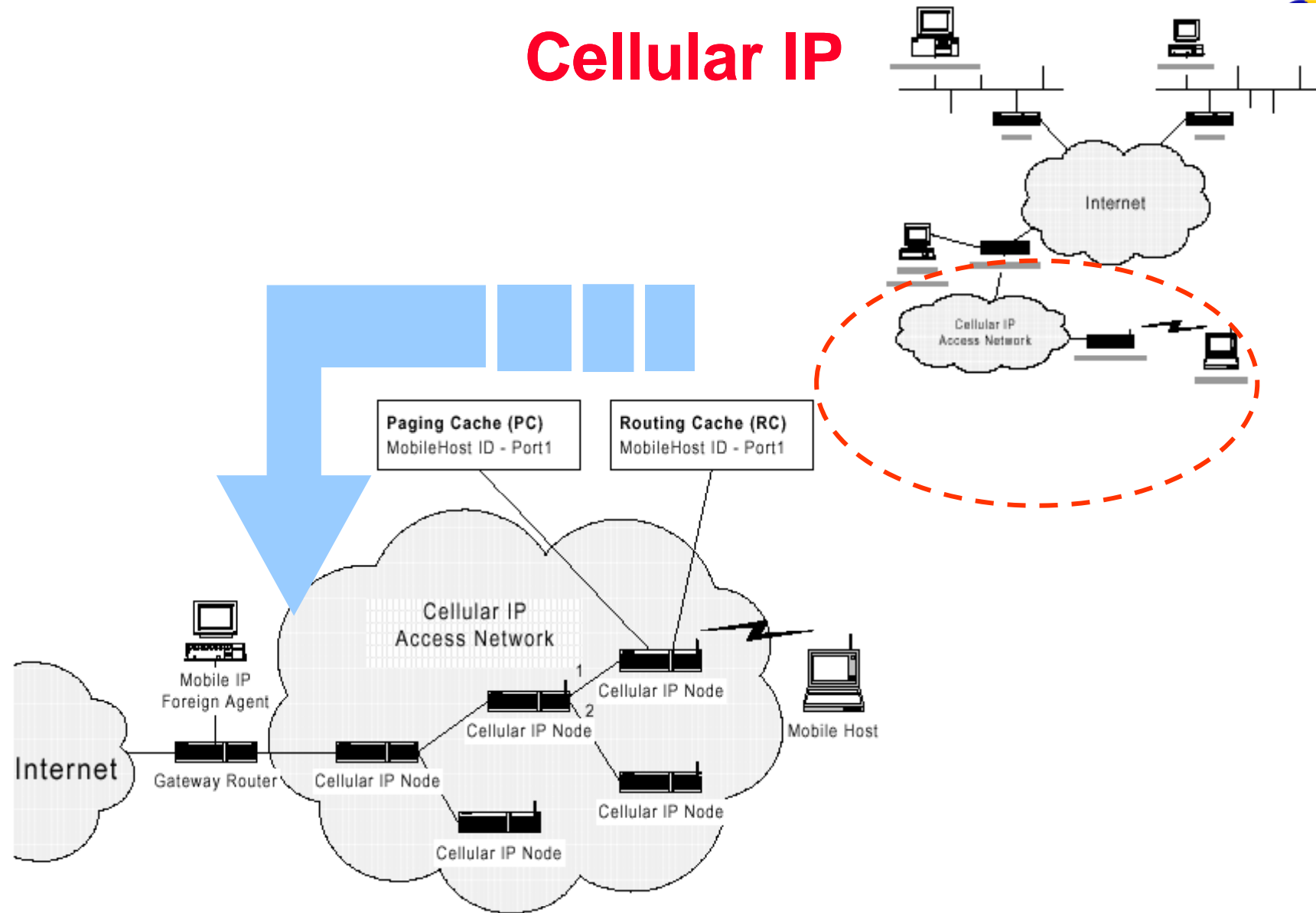
# Coming Issues



# Mobile IP

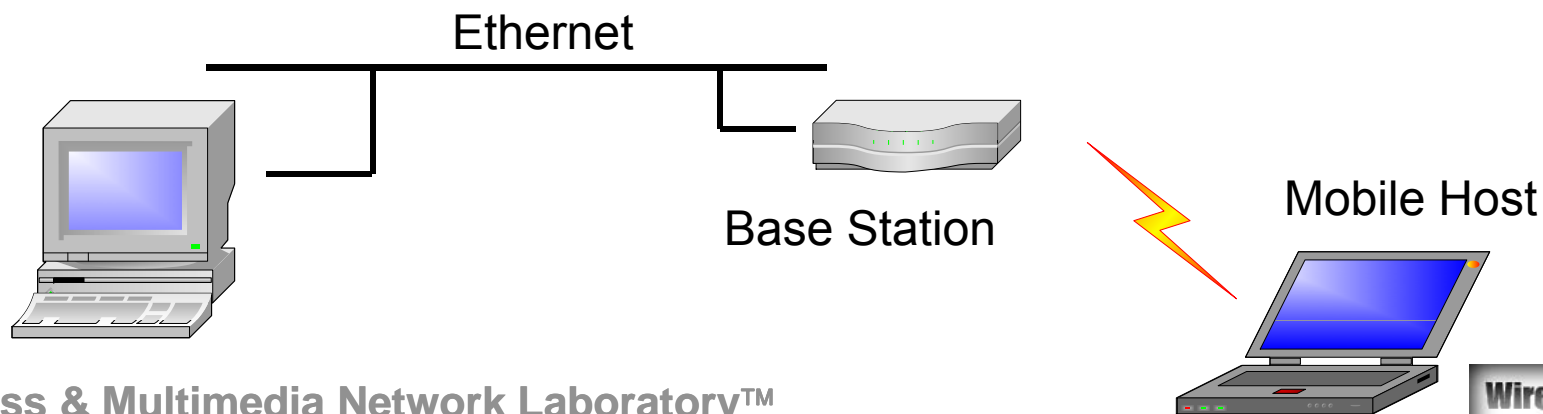


# Cellular IP



# Wireless TCP

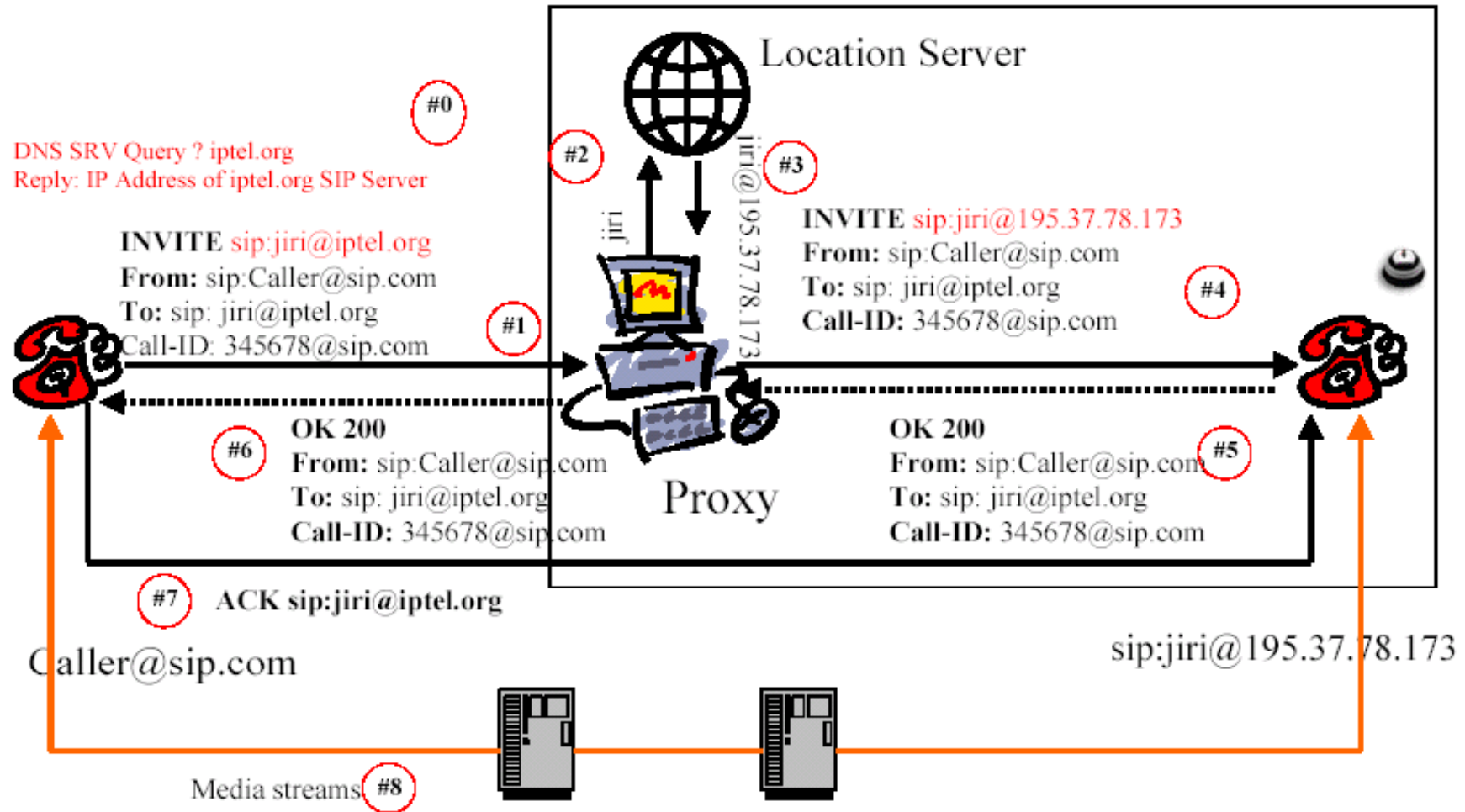
- ◆ TCP turned to perform well in traditional network where the packet losses occur mostly because of congestion.
- ◆ In the wireless environment
  - **Non-congestion** losses caused by wireless link
  - The degraded performance of TCP is mostly due to **mistaking wireless losses for congestion**.



# SIP Protocol



# Mobility support



# Mobile Multicast

- ◆ Mobile Network~ Mobile IP
- ◆ Application Requirements: updates to replicated databases, Inter-process communication among cooperating processes
- ◆ Resource Conservations~ Single Copy in...Multicast IP





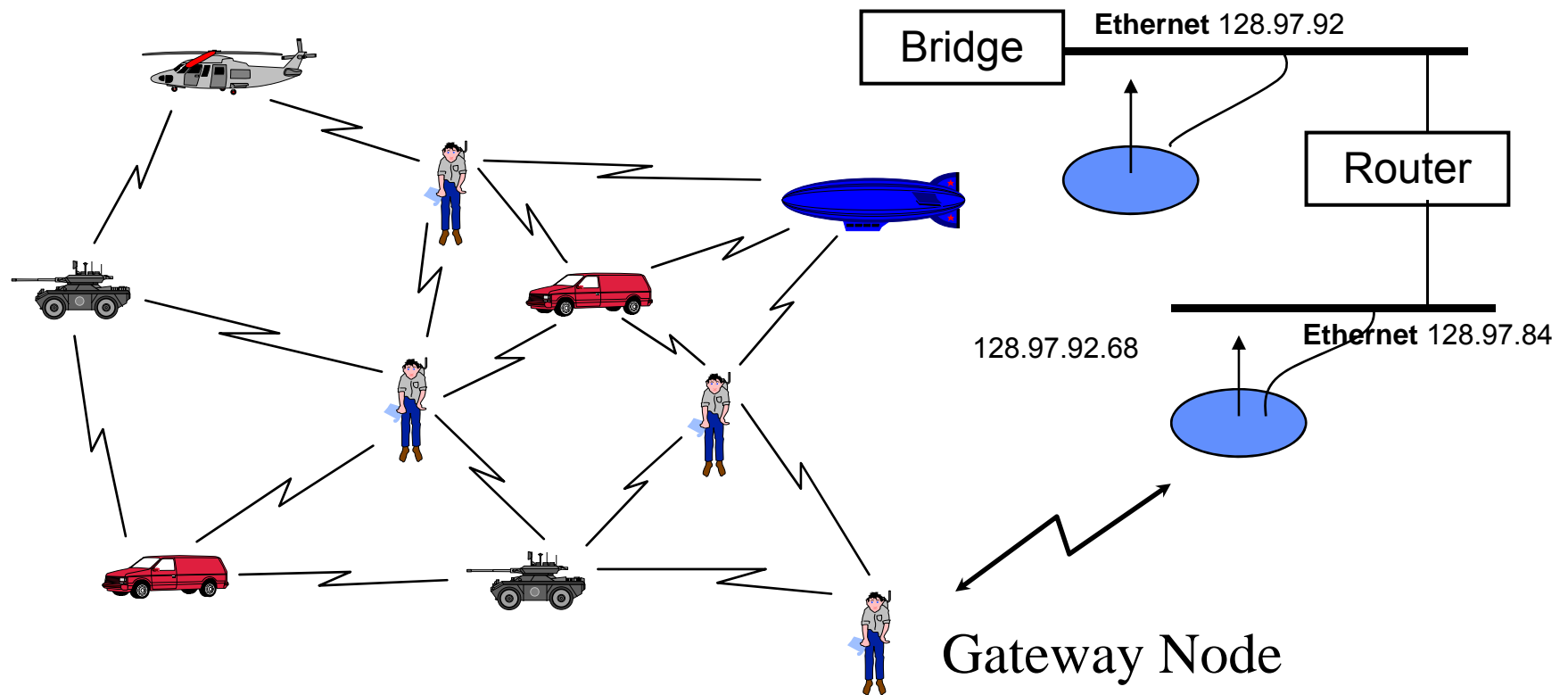
# Mobile Multicast

One to Many Mobile Multicasting Services

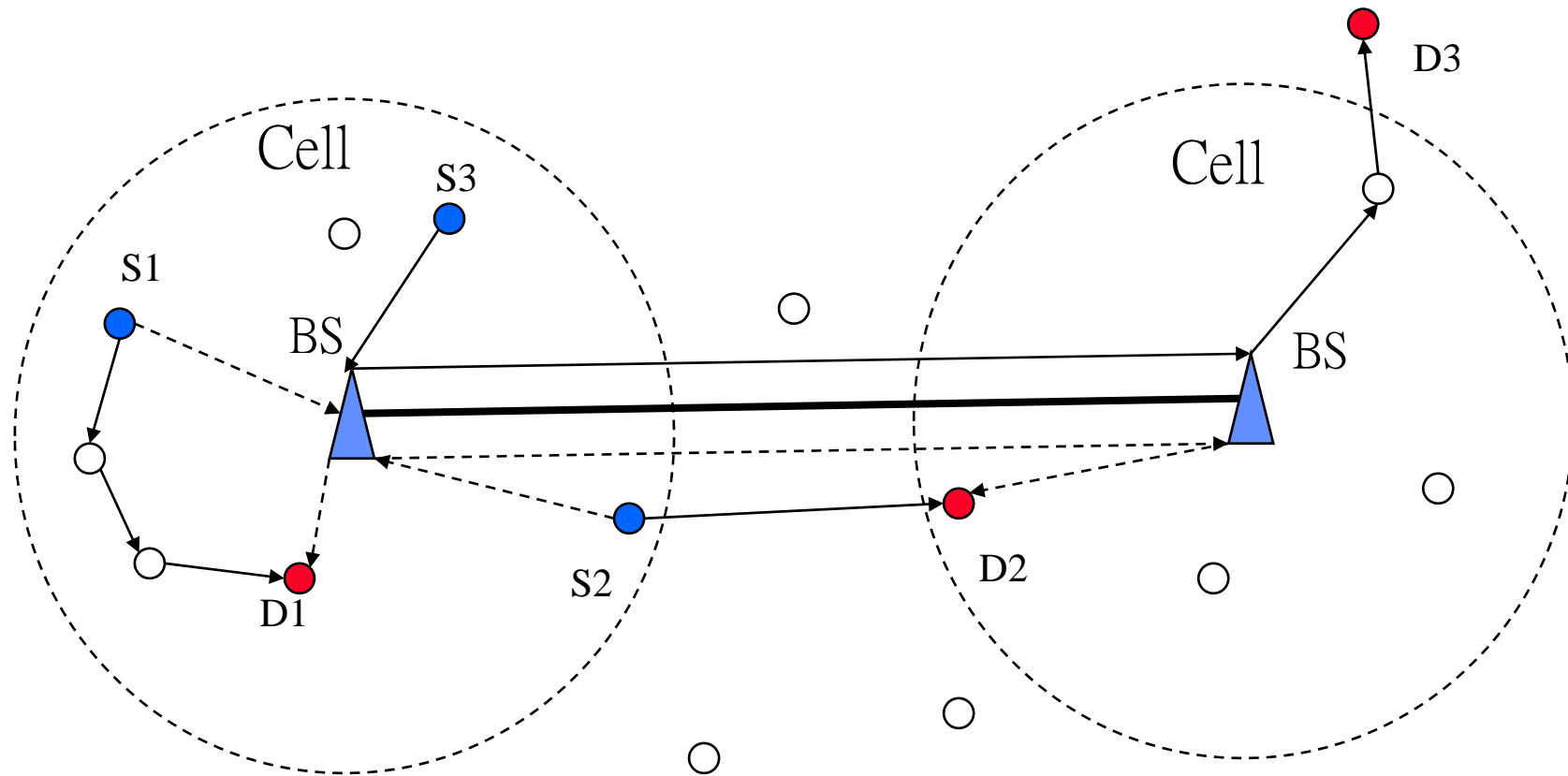


# Internet Interconnection and Mobile IP

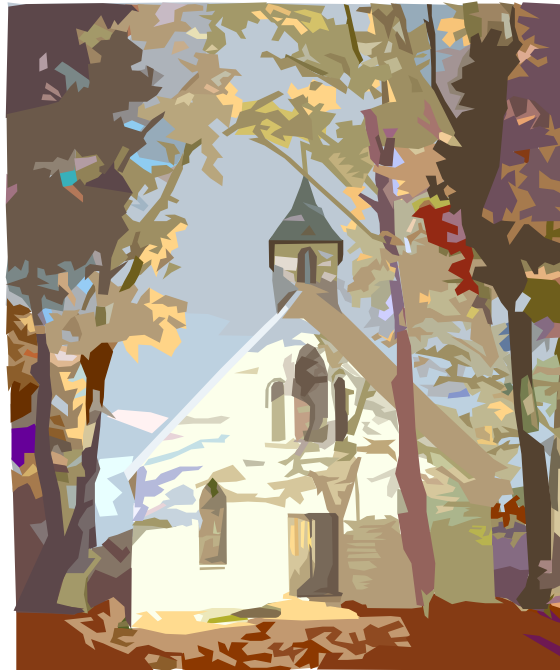
- ◆ DSR support the seamless interoperation between an ad hoc network and the Internet



# Ad hoc & Cellular System



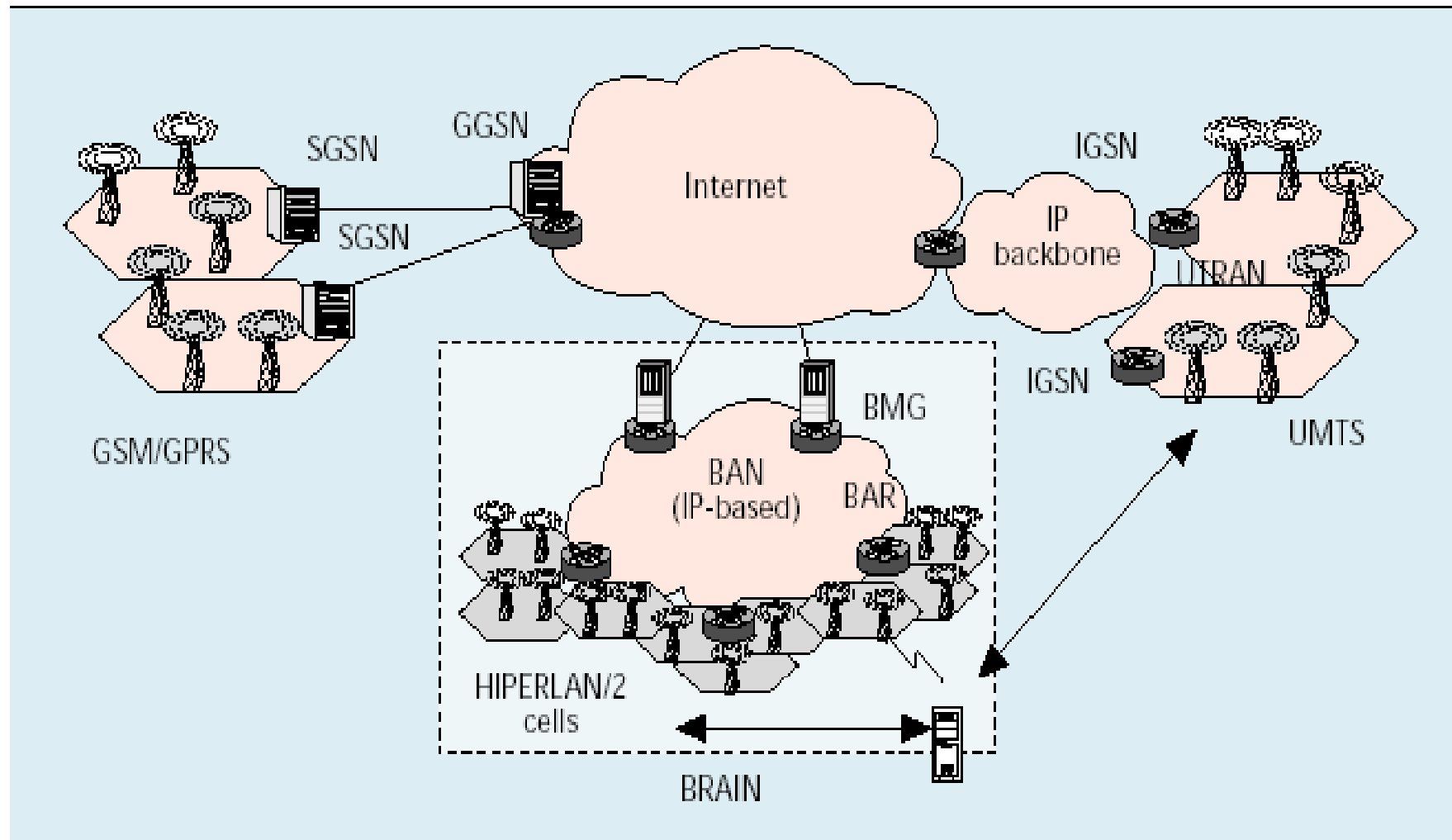
# QoS Support for an All-IP System Beyond 3G



# BRAIN

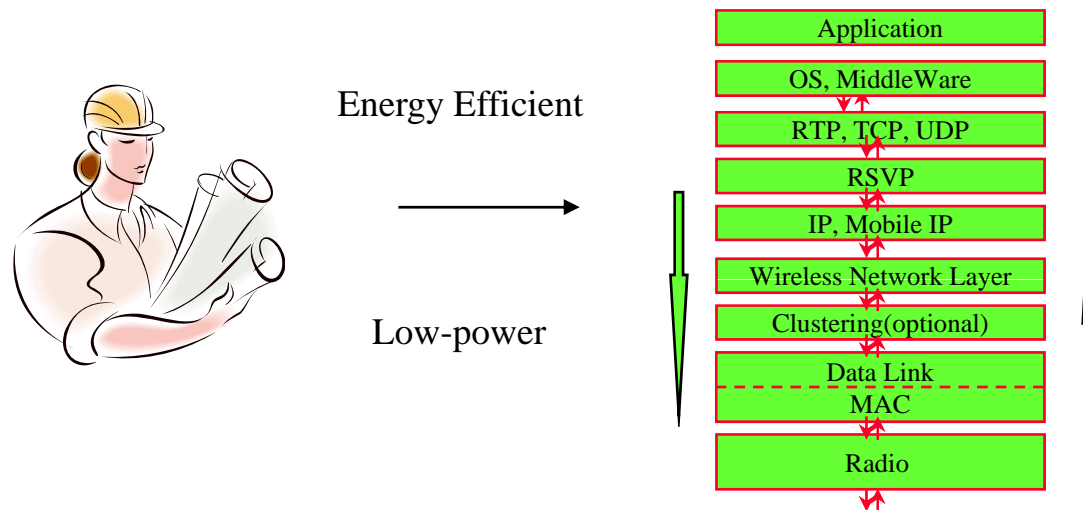
- ◆ Broadband Radio Access for IP-Based Networks
  - Cellular systems, fixed networks, and wireless LANs
  - Personal mobility, adapted for the terminal and link bandwidth
  - End-to-end QoS
  - A new QoS model for applications (BRENTA)
  - The radio link improvements
- ◆ IP-aware RAN (Radio Access Network)
  - Better support to IP applications
  - IP infrastructure will be widely available
- ◆ Protocol must be redesigned
  - Resource Management
  - Terminal mobility
  - RAN and terminal must have IP Stack

# BRAIN (Broadband Radio Access for IP-based Network)



# Energy and Power Efficient

- ◆ As wireless networks become an integral component of the modern communication infrastructure, **energy efficiency** will be an important design consideration due to the limited battery life of mobile terminals.
- ◆ This paper presents a comprehensive summary of recent work addressing energy efficient and low-power design within **all layers** of the wireless network protocol stack.



# Agenda

- ◆ Basic TCP
- ◆ Impact of Mobility & Wireless on TCP performances
- ◆ Solutions for Wireless TCP
- ◆ Midterm (next week)



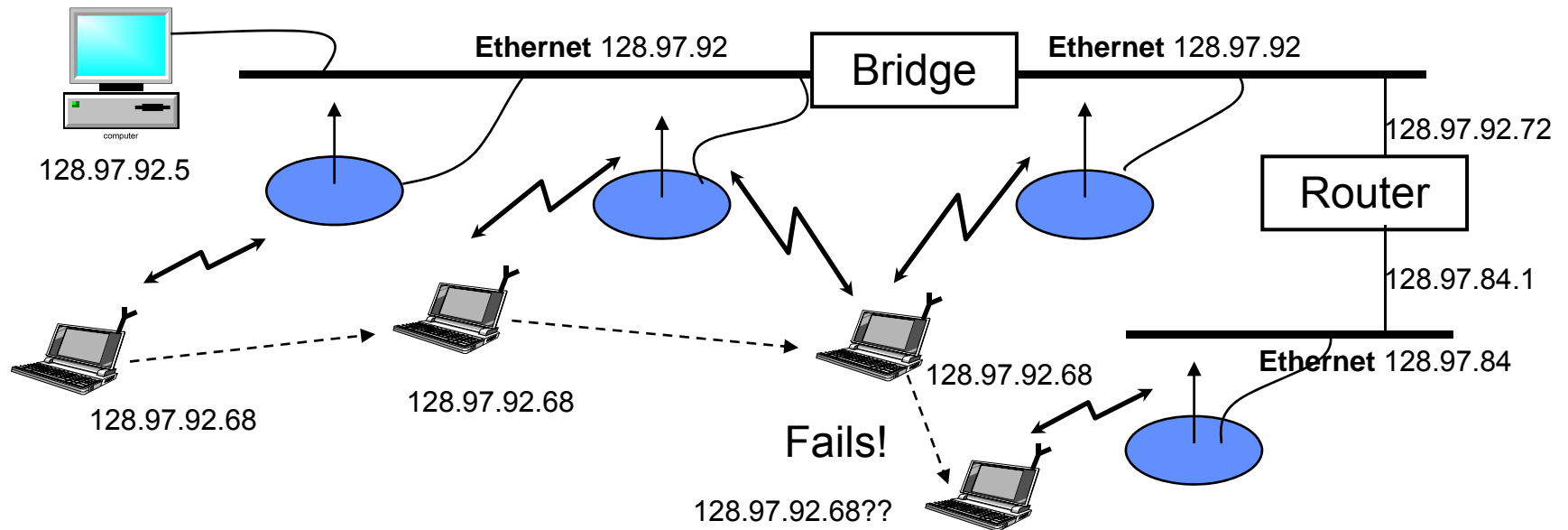


# Reading

- ◆ [Balakrishnan95], Harri Balakrishnan, Srinivasan Seshan, Elan Amir and Randy H. Katz, “Improving TCP/IP Performance over Wireless Networks”, ACM Mobicom95
- ◆ [Balarkrishnan97], Harri Balarkrishna, Venkat N, Padmanabhan, Srinivasan Seshan and Randy Katz, “A Comparison of Mechanisms for Improving TCP Performance over Wireless Links”, IEEE JSAC 97.
- ◆ Reference: [Mario2001], Saverio Mascolo, Claudio Casetti, Mario Gerla, Renwang “TCP Westwood: Bandwidth Estimation for Enhanced Transport over Wireless Links”, Mobicom2001

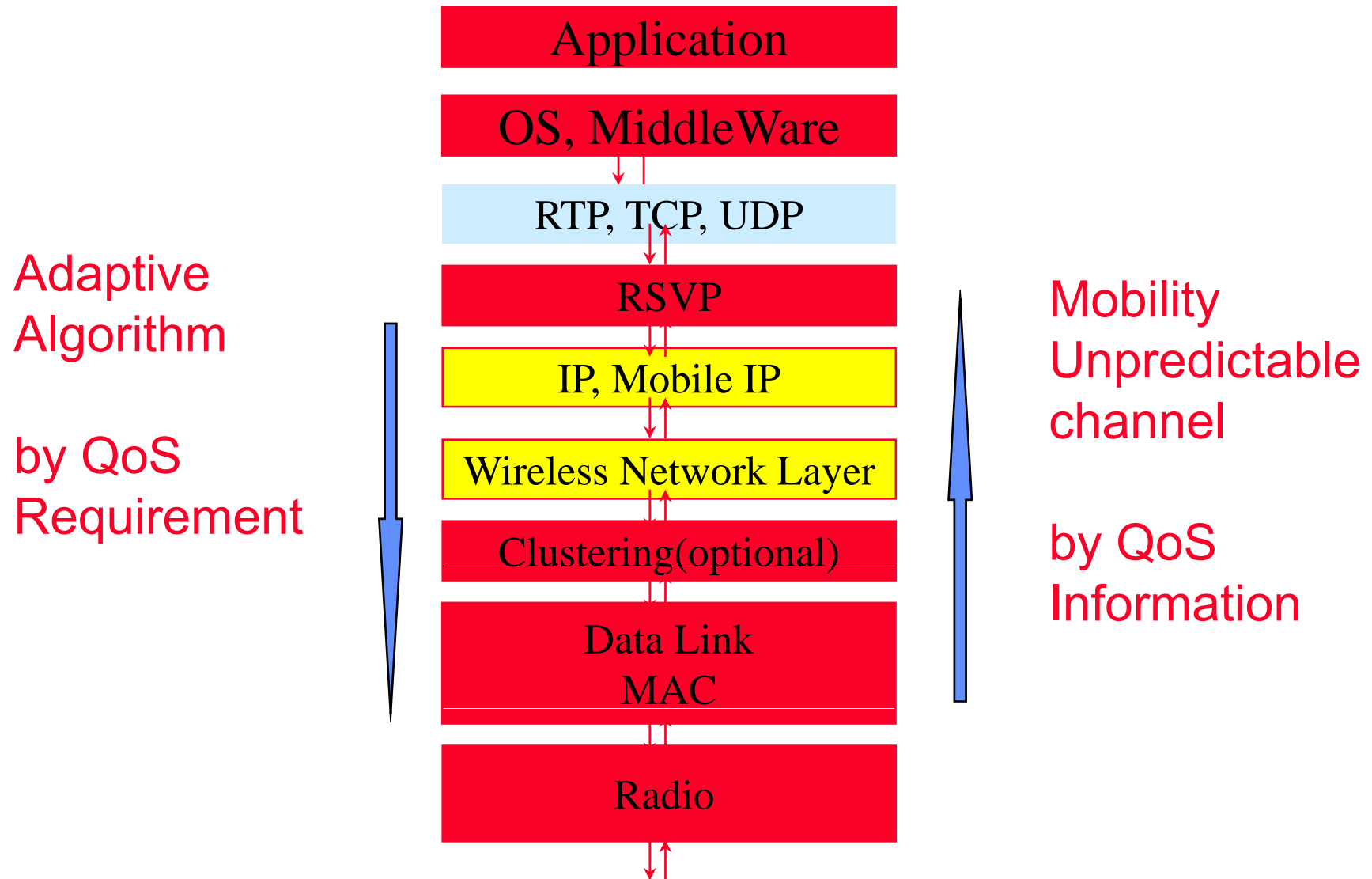


# Mobility in Wireless LANs: Basestation as Bridges



- ◆ Basestations are bridges(layer 2) – i.e. they relay MAC frames
  - Smart bridges avoid wasted bandwidth
- ◆ Works the within an ethernet(or other broadcast LAN)
  - Fails across network boundaries, and in switched LANs(e.g. ATM)

# QoS and Multimedia Traffic Support



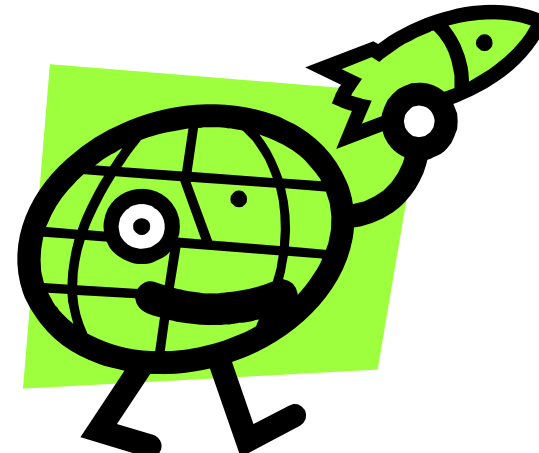
# Background

- ◆ With the growth of wireless device, wireless network access will become popular, but...
- ◆ Import the protocol from the wire network to wireless network...
- ◆ Packet losses occur in wireless due to the lossy links, not network congestion
- ◆ In traditional TCP, it can not distinguish the difference between that lossy link and network congestion

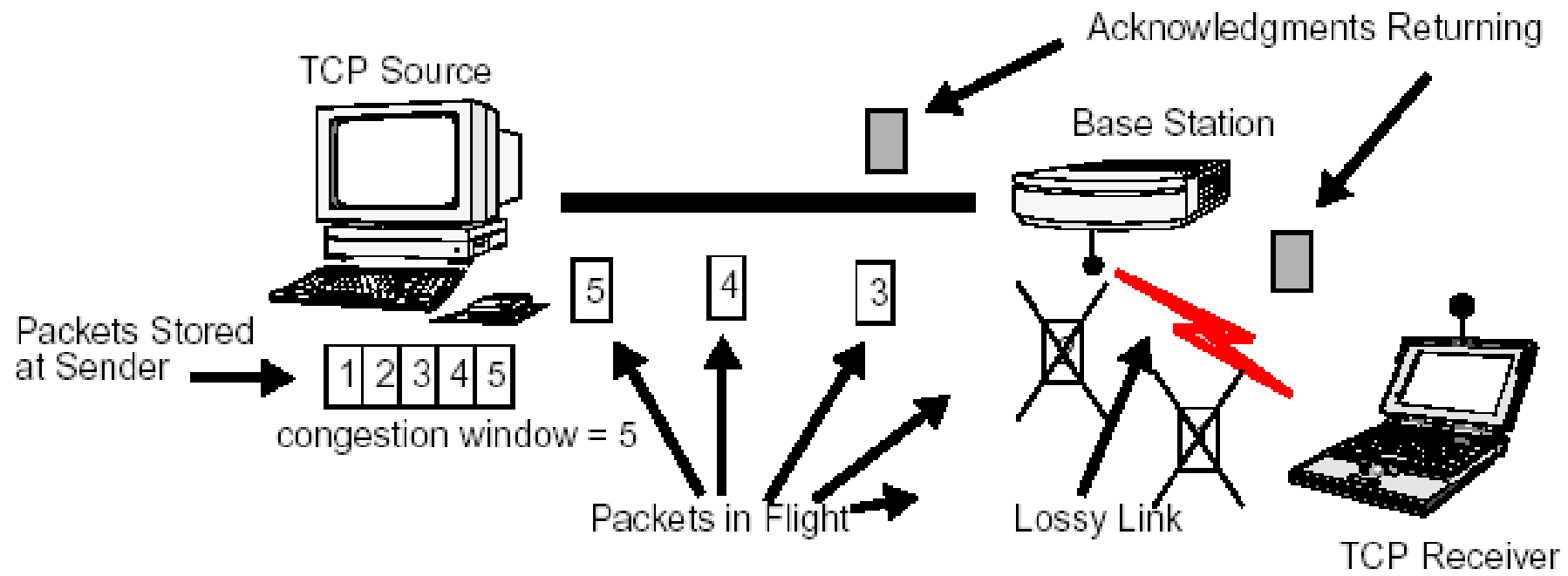
# Characteristics of Wireless & Mobility

- ◆ Limited Bandwidth
  - Small frame sizes (MTU) to keep latency small
- ◆ High bit error rates
  - Small frame sized to keep packet loss probability small
- ◆ Time varying bit error rate
  - Fading, frequency collisions etc.
- ◆ QoS (loss rate, delay) degradation during hand-off
  - Due to network layer rerouting
  - Due to link layer procedures
- ◆ QoS degradation after hand-offs
  - Lack of resource at new basestation
  - Less optimal route

# Basic End-to-End Control (Transport)



# Typical loss situation



# UDP (Connectionless, Unreliable)



Possible Multicast, Real Time Traffic, TCP-Friendly



# Impact on Connectionless, Unreliable Transport Protocol



- ◆ Example: effect on UDP applications
- ◆ Increase in end-to-end packet losses
  - Error on wireless link
  - Packet loss during hand-offs
- ◆ Drop in application throughput
  - Errors on wireless link
  - Packet loss during hand-off
- ◆ Pauses in interactive applications
  - Burst errors on wireless link
  - Packet loss during hand-off
  - Delay increase due to buffering & re-sequencing during hand-offs
- ◆ Application level impact is much more complex!

# TCP (Connection Oriented, Reliable)



Data Transmission, WWW, flow control, error control

# TCP Basics

- ◆ Sliding window protocol: Go-Back N ARQ
  - Transfers a byte stream in “segments”, not fixed user blocks, logical timer associated with each segment that is sent
  - 32-bit sequence number indicated byte number in stream
    - ◆ Window is max number of outstanding unACK'ed bytes in network
- ◆ Cumulative acknowledgement scheme (original TCP)
  - Ack's all bytes up through n
  - Piggybacked on data packets in reverse direction
- ◆ Control of sender's window size
  - Min (receiver's advertized window, congestion window)
  - Three goals
    - ◆ Flow control to avoid receiver buffer overflow
    - ◆ Congestion control to react to congestion in network layer & below
    - ◆ Congestion avoidance
- ◆ Segment loss is assumed to be a result of congestion in routers
  - Reasonable for wired network since BER on fiber is better than  $10^{-12}$

# TCP's End to End Congestion Control

- ◆ Window-based congestion control
  - Cwnd: congestion window size
  - Ssthresh: slow start threshold (for slow down of increase)
- ◆ Timeout is an indicator of segment loss
- ◆ Timeout value
  - Using estimated average of ACK delay and expected deviation
- ◆ On timeout
  - Segment is assumed lost and is attributed to congestion
  - One-half of current window is recorded in ssthresh
  - Cwd is reduced to 1
  - Timeout value is increased in case packet was delayed

# TCP's End-to-end Congestion Control

- ◆ On new ACK
  - Everything okay, so allow larger congestion window
  - Two ways of increasing cwnd
    - ◆ Phase1: slow start until  $cwnd \leq ssthresh$ 
      - Fast (exponential) increase of cwnd
    - ◆ Phase2: congestion avoidance
      - Slow (additive) increase of cwnd
  
- ◆ Duplicate ACKs
  - Two causes: lost segment, misordered segment
  - $\geq 3$  duplicate ACKs in a row are a good indication of a lost segment but data is still flowing
  - Fast Retransmit and Fast Recovery
    - ◆ Missing segment is retransmitted without waiting for timeout
    - ◆ One half of current window is recorded in ssthresh
    - ◆ Congestion avoidance is done but not slow start

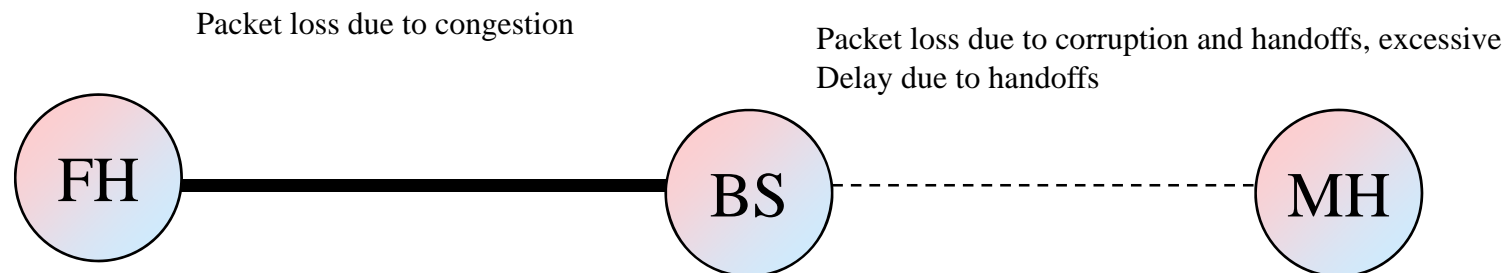
# Challenges of Mobility and Wireless on Network Performance



TCP Performance

# The Problem

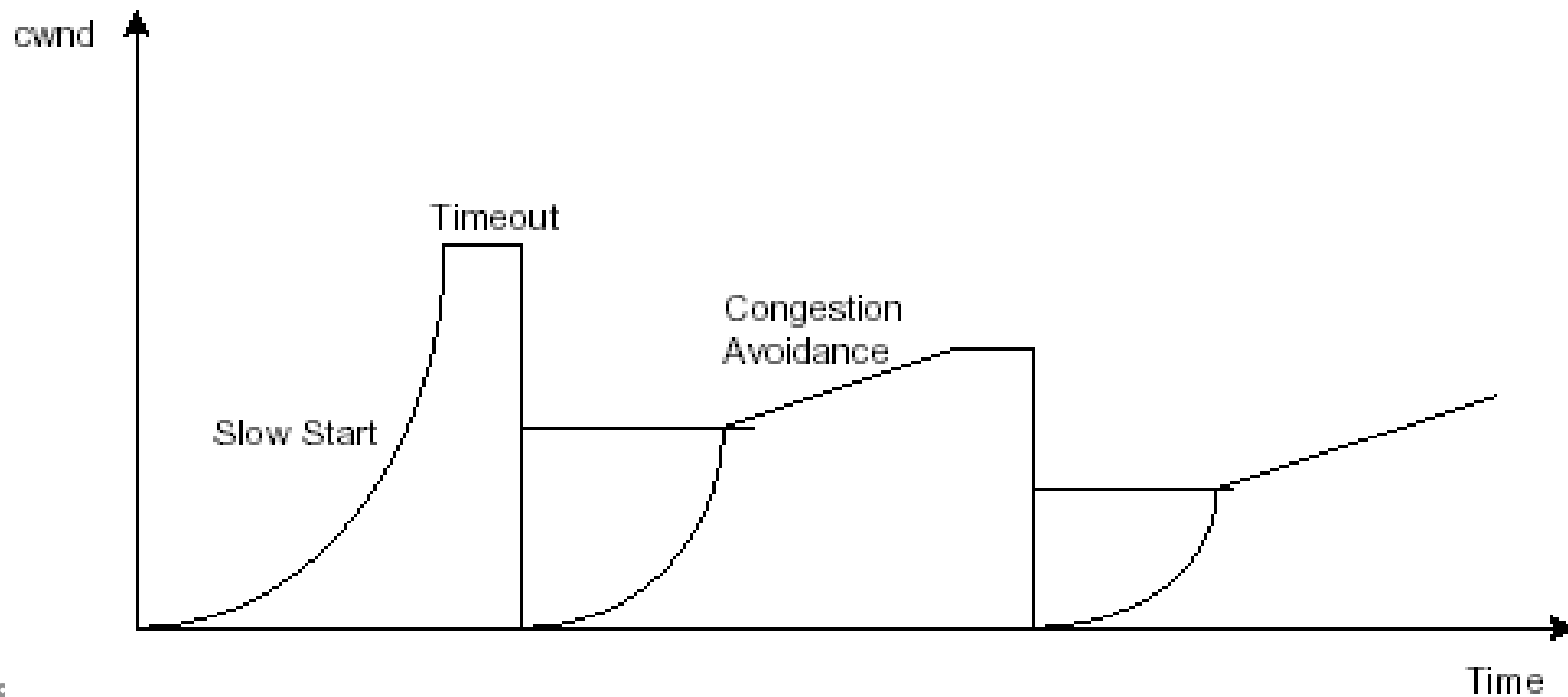
- ◆ In Wireless and mobile networks, segment loss is likely not due to congestion
  - Packet corruption due to high BER on wireless link (noise, fading)
  - Packet delay and losses during handoffs
- ◆ But, TCP invokes congestion control nevertheless
- ◆ Mistaking wireless errors and handoffs for congestion causes
  - Significant reductions in throughput (window size decreases, slow start)
  - Unacceptable delays (low resolution TCP times ~500ms, back-off)



# Example graph

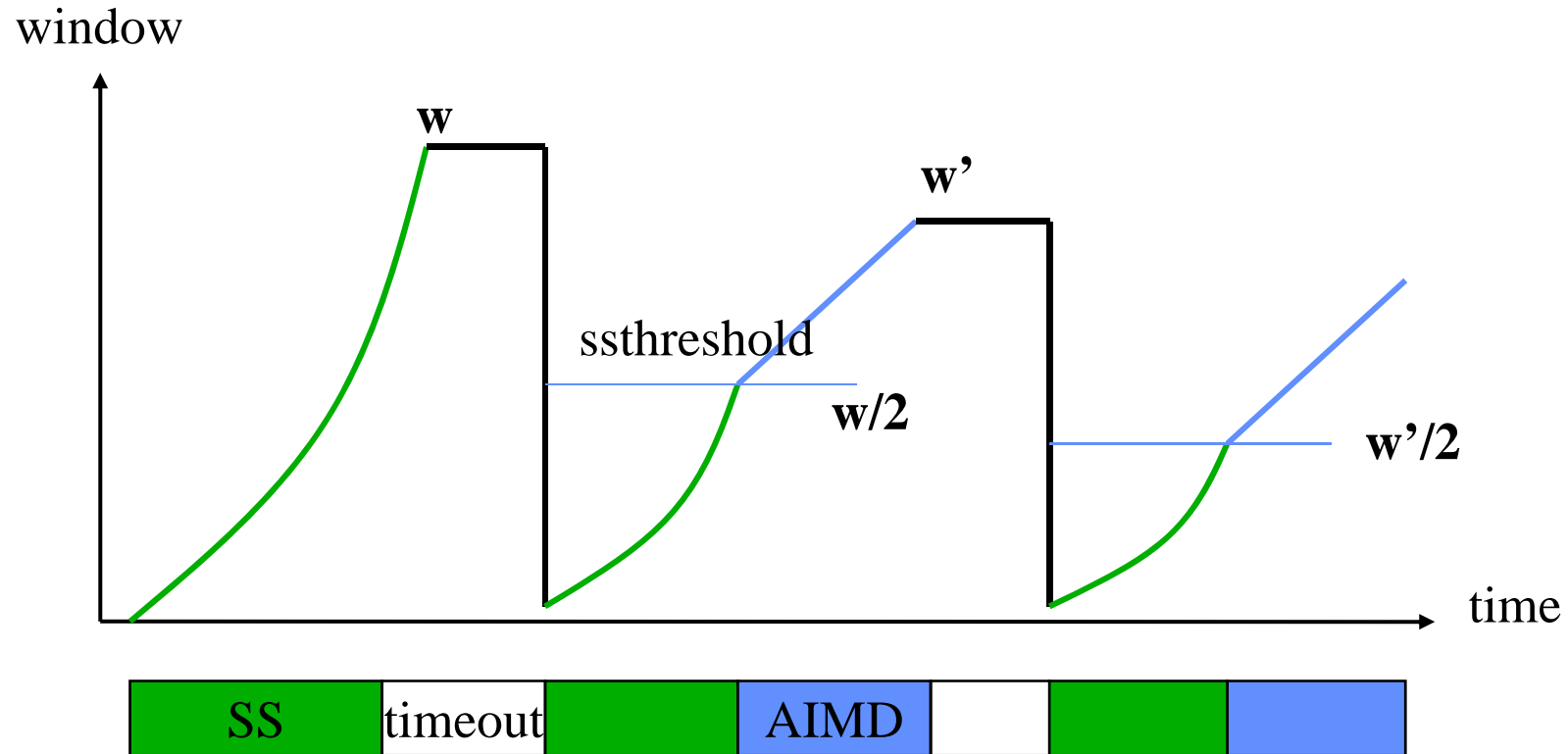
$cwnd \leq ssthresh \rightarrow$  slow start

$cwnd > ssthresh \rightarrow$  congestion avoidance



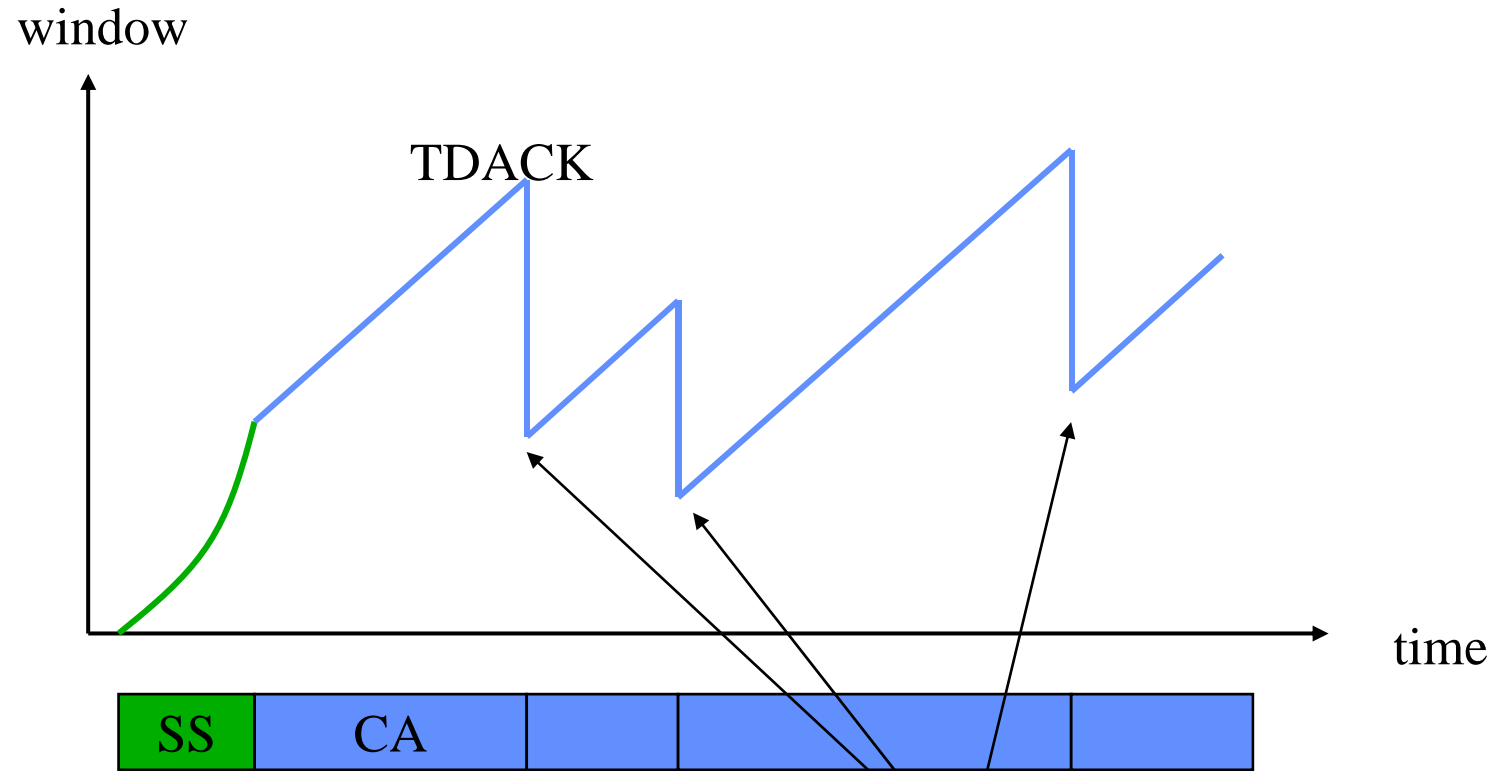


# Slow Start of TCP Reno



ssthreshold : slow-start threshold

# Congestion Avoidance of TCP Reno



SS: slow start  
 CA: congestion avoidance

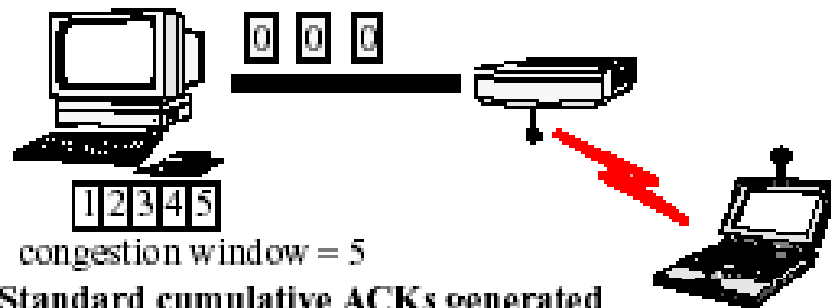
Fast retransmission / Fast recovery

# Fixes?

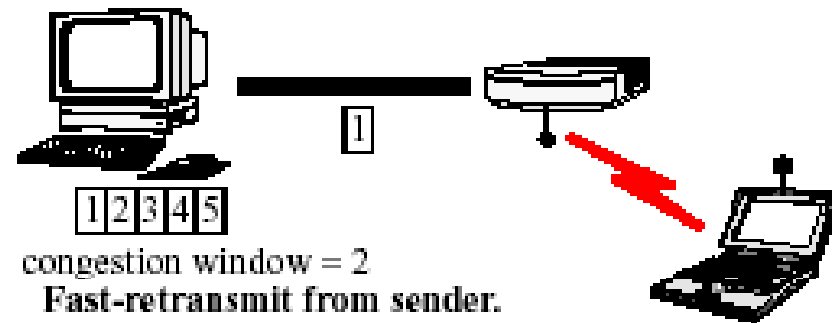
- ◆ Fix TCP
  - TCP really a hack in many ways..
  - Separate congestion control from error control
  - Move away from cumulative ACK
- ◆ Fix lower layer to make TCP work better
  - Improve the wireless link
- ◆ Use something different
  - Something totally new
  - Something different for the wireless part



# Normal TCP

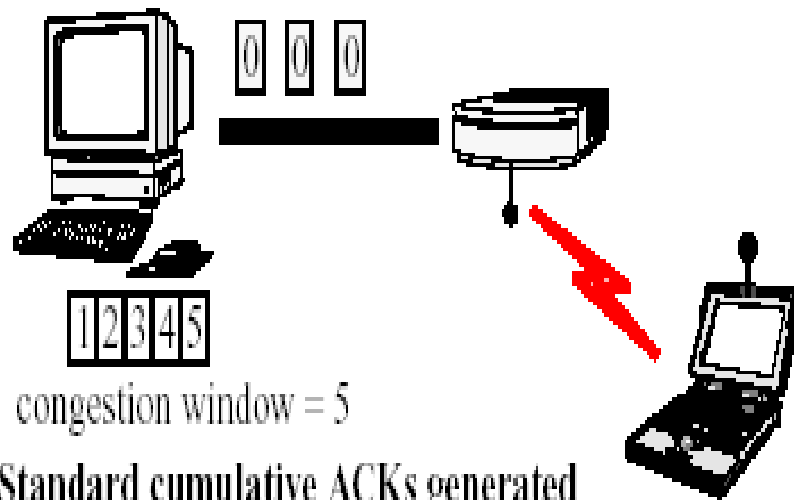


congestion window = 5  
 Standard cumulative ACKs generated  
 by TCP Reno receiver.

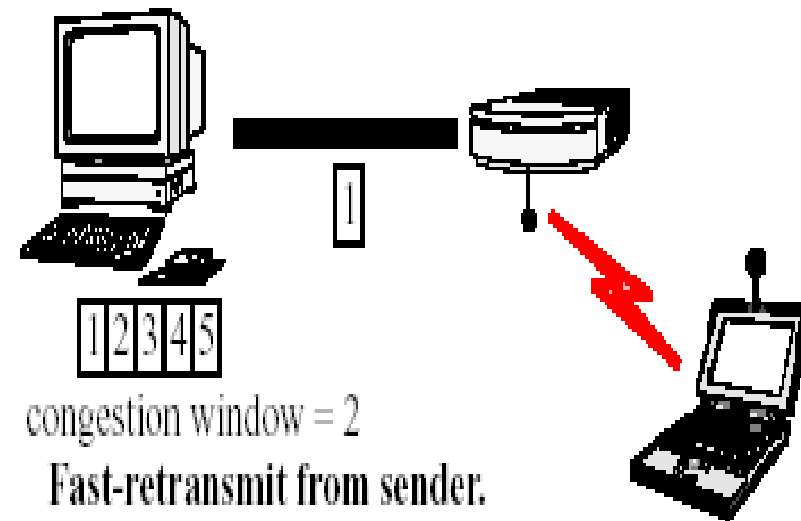


congestion window = 2  
 Fast-retransmit from sender.

# Fast-Retransmit Scheme



congestion window = 5  
 Standard cumulative ACKs generated  
 by TCP Reno receiver.



congestion window = 2  
 Fast-retransmit from sender.

# Solutions for WTCP (I)



Split the connection into two parts

# Split Connection Approaches

- ◆ Main Idea: split MH ↔ FH connection into two MH ↔ BS & BS ↔ FH
  - Separate flow control and reliable delivery mechanisms
  - Intermediate higher layer agent at the base-station
  - Session layer hides the split connection
  
- ◆ Two approaches:
  - Both FH ↔ BS & BS ↔ MH segments use TCP: Rutgers' Indirect-TCP
    - ◆ e.g. uses MTCP (Multiple TCP) over BS ↔ MH
  - BS ↔ MH uses specialized protocol
    - ◆ e.g. uses SRP (Selective Repeat) over BS ↔ MH
    - ◆ Error and flow control optimized for lossy wireless link

# Pros & Cons of Split-Connection Approaches

## ◆ Pros

- FH is shielded from wireless link behavior
- Handoff is transparent to FH
- Relative easy to implement
- Requires no modification to FH
- Can use specialized protocol over wireless link

## ◆ Cons

- Loss of end-to-end semantics
- Application relink with new library
- Software overhead: efficiency and latency
- Large handoff latency



## Solutions for WTCP (II)



Lower layer to make TCP work better

# Link-level Error Control

- ◆ FEC and ARQ on wireless link to increase its reliability
  - Improves performance independent of transport protocol
- ◆ Disadvantage
  - Coupling between link level and end-to-end retransmission may lead to degraded performance at high error rates
  - Does not address the delay and losses due to handoffs

## Solutions for WTCP (III)

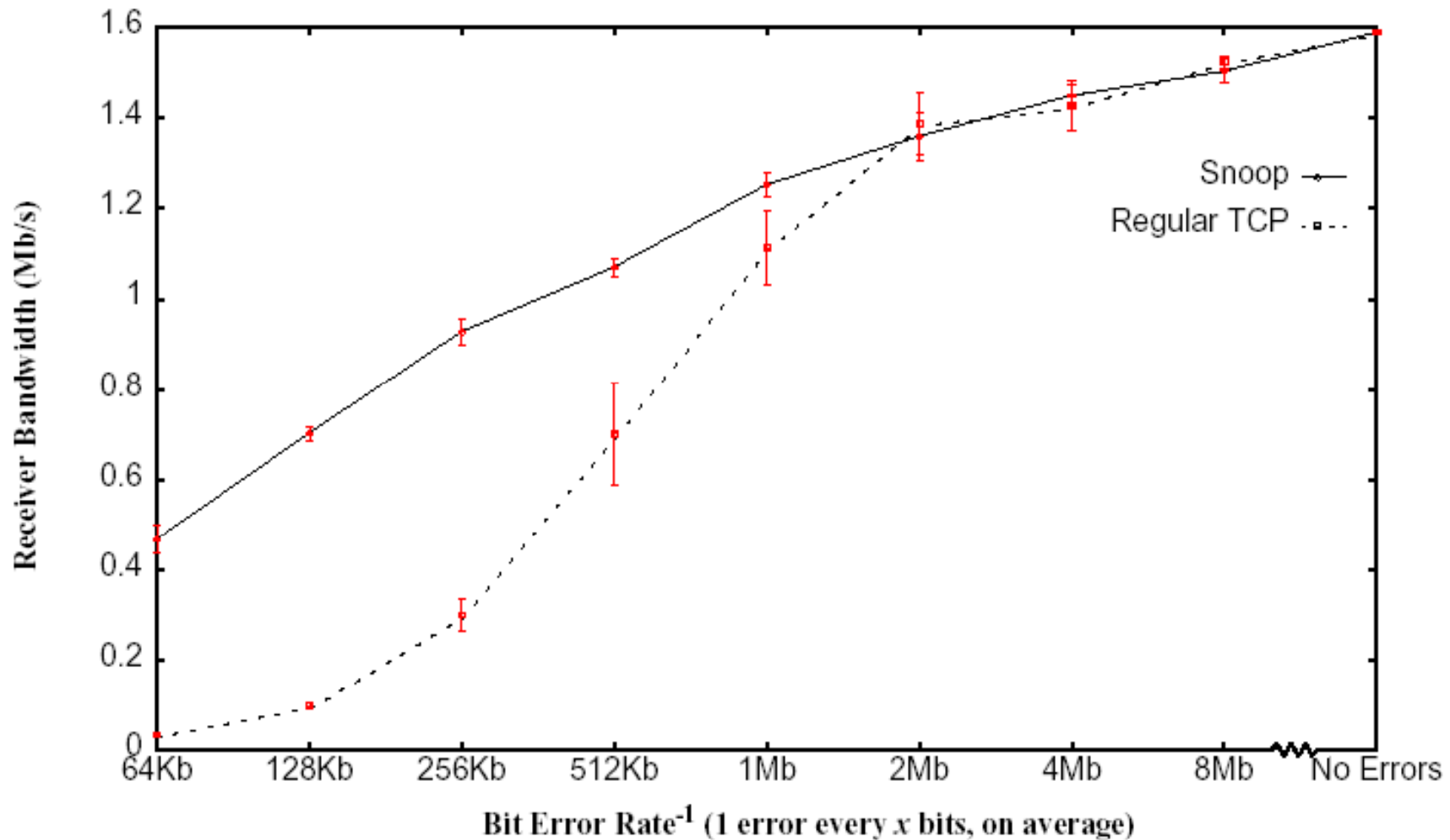


Snoop, Make it look like!

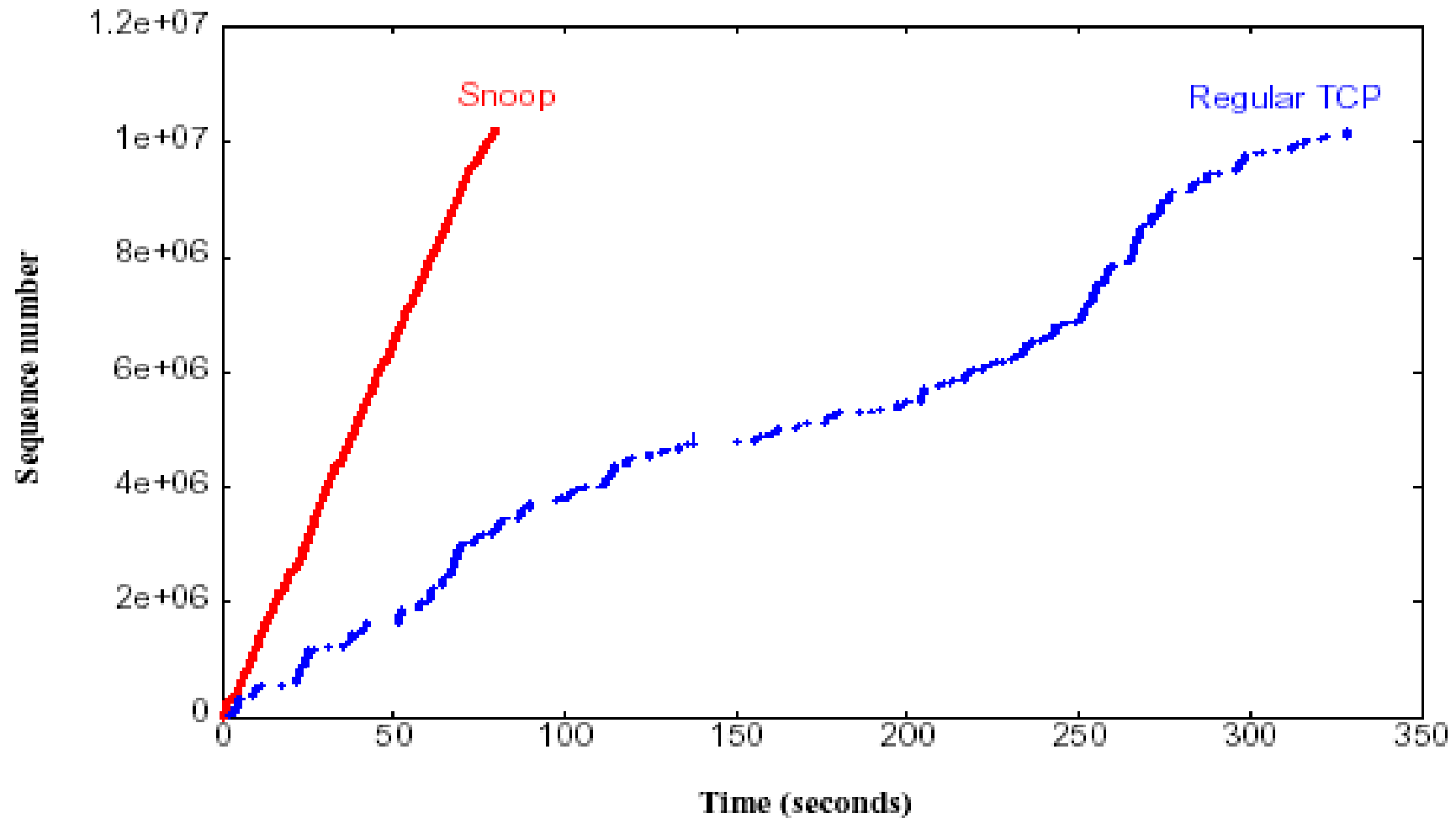
# Snoop TCP

- ◆ Basic Idea for transfer of data to MH
  - Snoop Module: Modify network layer routing code at BS
  - Cache un-acknowledged TCP data going to MH at BS
  - Perform local retransmissions over wireless link
    - ◆ Policies to deal with ACKs from MH and timeout
    - ◆ Used duplicate ACKs to identify packet losses
  - Shields sender from wireless link
    - ◆ Transient conditions of high BER, temporary disconnection
- ◆ Basic idea for transfer of data from MH
  - BS detects missing packets and generated NACKs for MH, exploits SACK option for TCP
  - MH re-sends the packets, requires modifying TCP code at MH
- ◆ Features
  - Speedups of up to x20 over regular TCP depending on bit error rate
  - Maintain end-to-end semantics
  - Does not address the handoff problem

# Performance of the Snoop Mechanism



# Performance of the Snoop Mechanism



# Comparison of Wireless TCP Techniques

- ◆ End-to-End proposals
  - Selective ACKs
    - ◆ Allows sender to recover from multiple packet losses without resorting to course timeout
  - Explicit Loss Notification (ELN)
    - ◆ Allow sender to distinguish between congestion vs. other losses
- ◆ Split-connection proposal
  - Separate reliable connection between BS & MH
    - ◆ May use standard TCP or, special techniques such as SACK, or NACK
- ◆ Link-layer proposal
  - Hide link-layer losses via general local retransmission and FEC
  - Make link-layer TCP aware
    - ◆ Snoop agent to suppress duplicate ACKs

# Main Conclusions of [Balakrishnan97]

- ◆ Simple link layers do not quite work
  - Adverse interaction of times is actually a minor problem
  - Fast retransmission and associated congestion control gets triggered and cause performance loss
- ◆ Reliable link layer with TCP knowledge works well
  - Shielding sender from duplicate ACKs due to wireless losses improves throughput by 10-30%
- ◆ No need to split end-to-end connections
  - I-TCP does as bad because sender stalls due to buffer space limit at BS
  - Using SAK or BS-MH link works well
- ◆ SACK and ELN helps significantly
  - Help avoid timeous
  - e.g. ELN helped throughput by x2 over vanilla TCP-Reno
  - But still do 15% to 35% worse than TCP-aware link layer schemes



# Introduction

- ◆ TCP Westwood (TCPW) is a sender-side modification of TCP Reno in wire as well as wireless network
- ◆ TCPW can estimate the E2E b/w and the improvement is most significant in wireless network with lossy links
- ◆ TCPW sender monitors the ACK reception and from it estimates the data rate
- ◆ The sender uses the b/w estimate to properly set the cwin and ssthresh

## Filtering the ACK reception rate

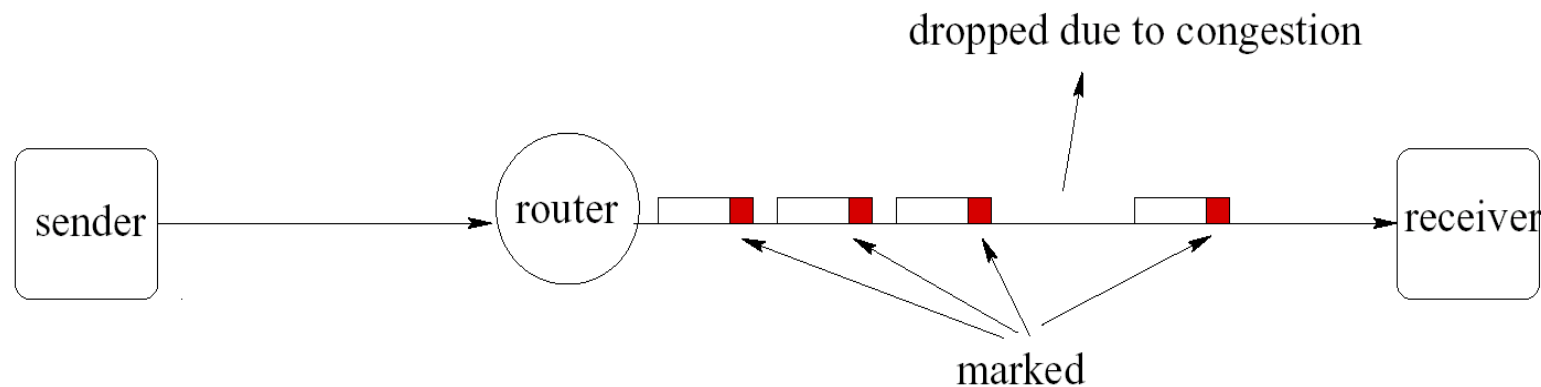
- ◆ Sample of bandwidth

- ◆ We measure  $b_k = \frac{d_k}{t_k - t_{k-1}}$  to average sampled

# Congestion Coherence

Chnlei Liu, and Raj Jain, “*Requirements and Approaches of Wireless TCP Enhancements*,”.

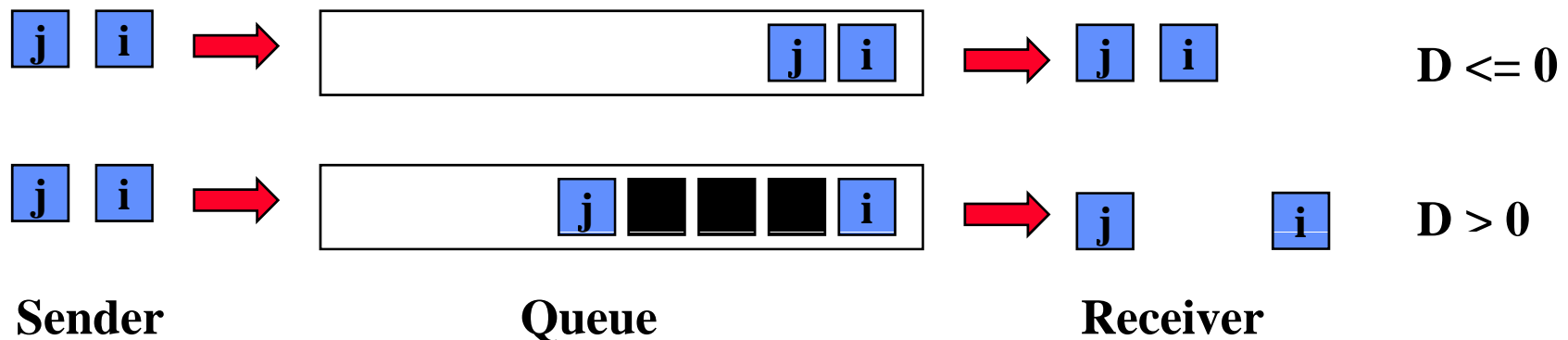
- ◆ This paper proposes a new enhancement approach that use Explicit Congestion Notification (ECN) to signal network congestion and use the sequential coherence of ECN marks to distinguish wireless and congestion losses.



# inter-arrival jitter

- ◆ [RFC 1889] The difference  $D$  is packet spacing at the receiver compared to the sender for a pair of packets.
- ◆ The  $D$  (sec) is called inter-arrival jitter.

$$D(i, j) = (R_j - R_i) - (S_j - S_i) = (R_j - S_j) - (R_i - S_i)$$



# Jitter ratio

Shi-Yang Chen, Eric Hsiao-Kuang Wu, and Mei-Zhen Chen, “A New Approach Using Time-Based Model for TCP-Friendly Rate Estimation”, 2002.

The ratio of packet queued at the router is

$$\frac{\left[ \frac{1}{t_A} - B \right]}{\frac{1}{t_A}} = \frac{\left[ \frac{1}{t_A} - \frac{1}{t_D} \right]}{\frac{1}{t_A}} = \frac{t_D - t_A}{t_D}$$

$$\approx \frac{(R_j - R_i) - (S_j - S_i)}{R_j - R_i} = \frac{D}{R_j - R_i} \quad \Rightarrow \quad \text{Jitter ratio} \quad Jr = \frac{D}{R_j - R_i}$$

$t_A$  : the packet - by - packet delay of the packets arrival at the router

$t_D$  : the delay of the packets departure from the router

$B$  : the service rate of the router

# Example

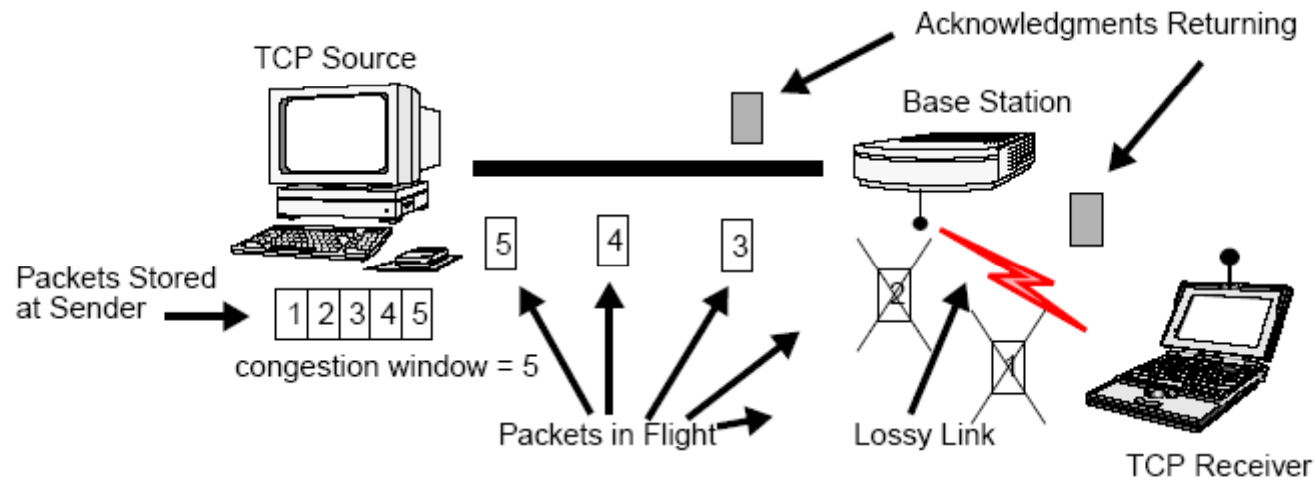


Figure 1. A typical loss situation

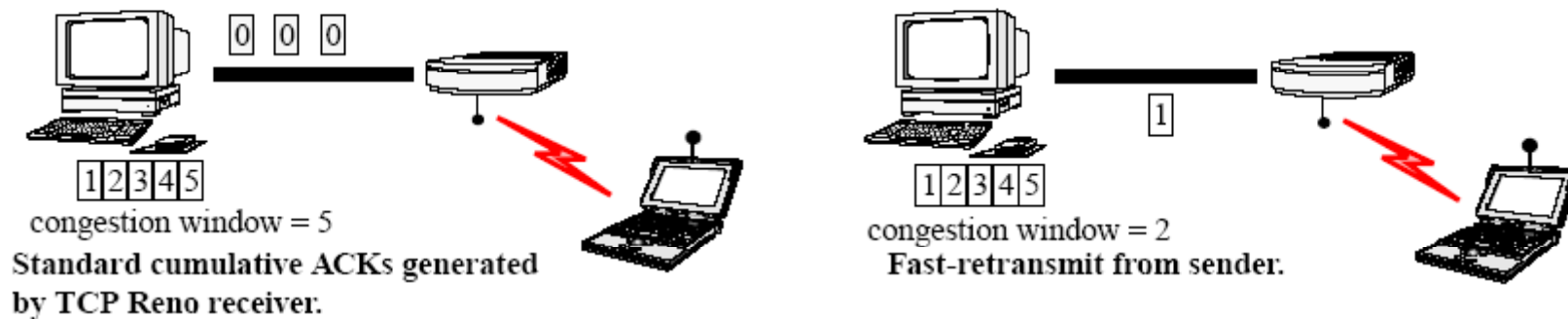


Figure 2. Normal TCP

# Enhanced Solution

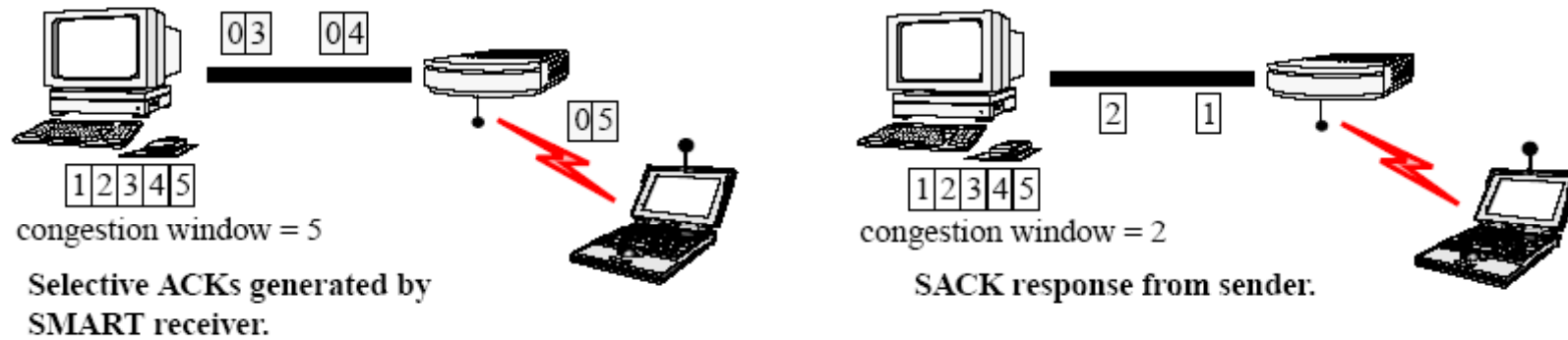


Figure 3. TCP with SMART-based selective acknowledgements

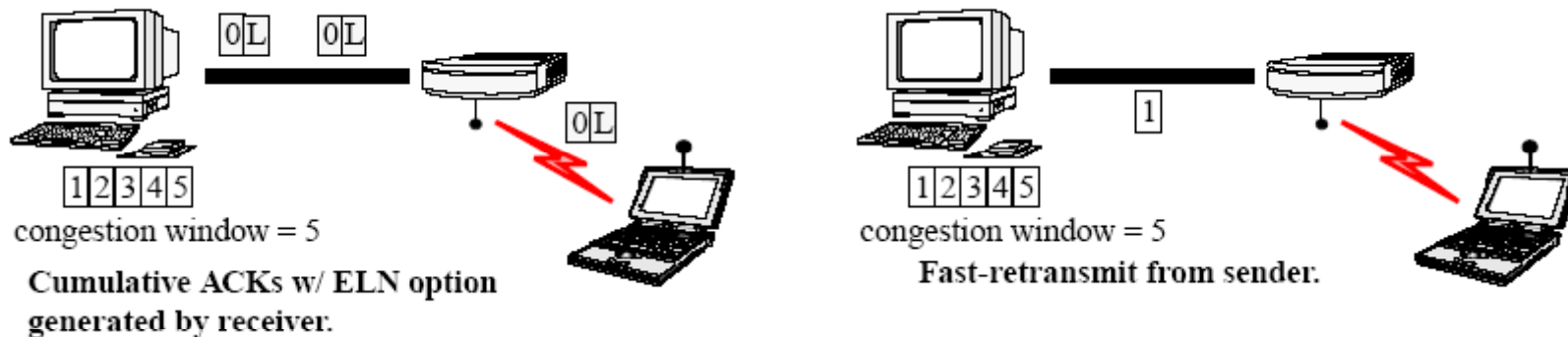


Figure 4. TCP with ELN

# Enhanced Solution

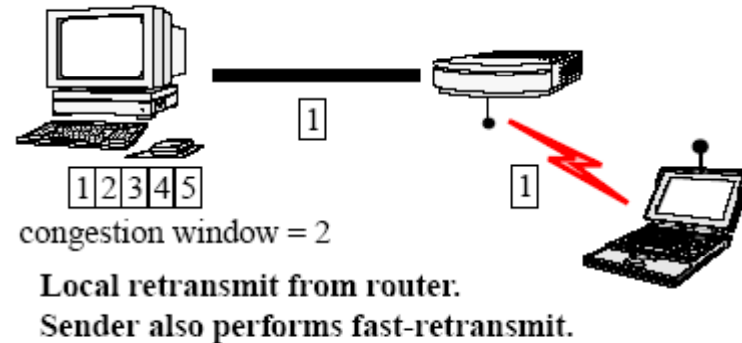
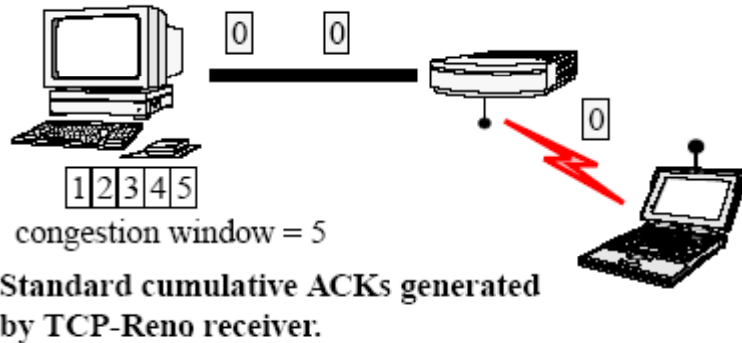


Figure 5. Basic Link-Layer protocol (LL)

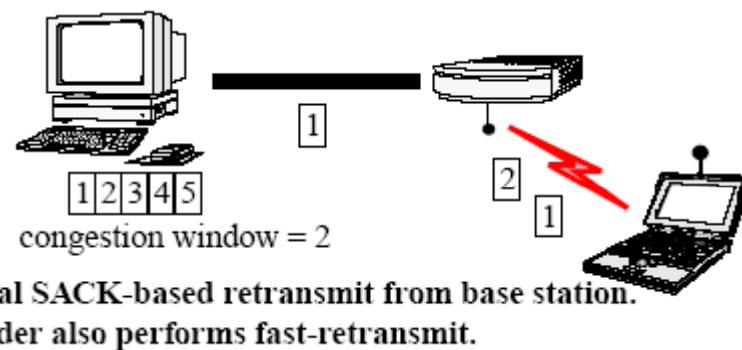
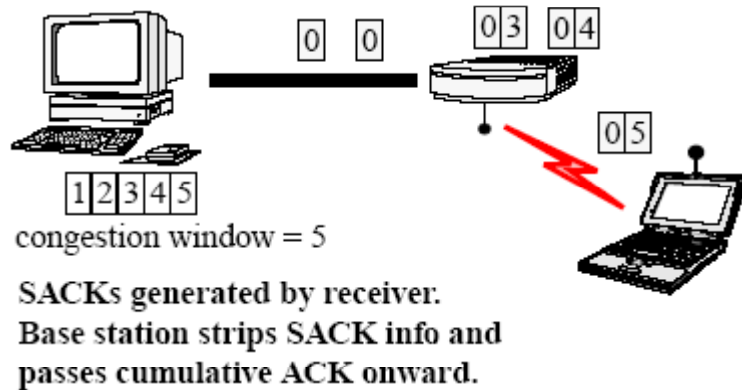


Figure 6. Link-Layer with SMART-based selective acknowledgments



# Enhanced Solution

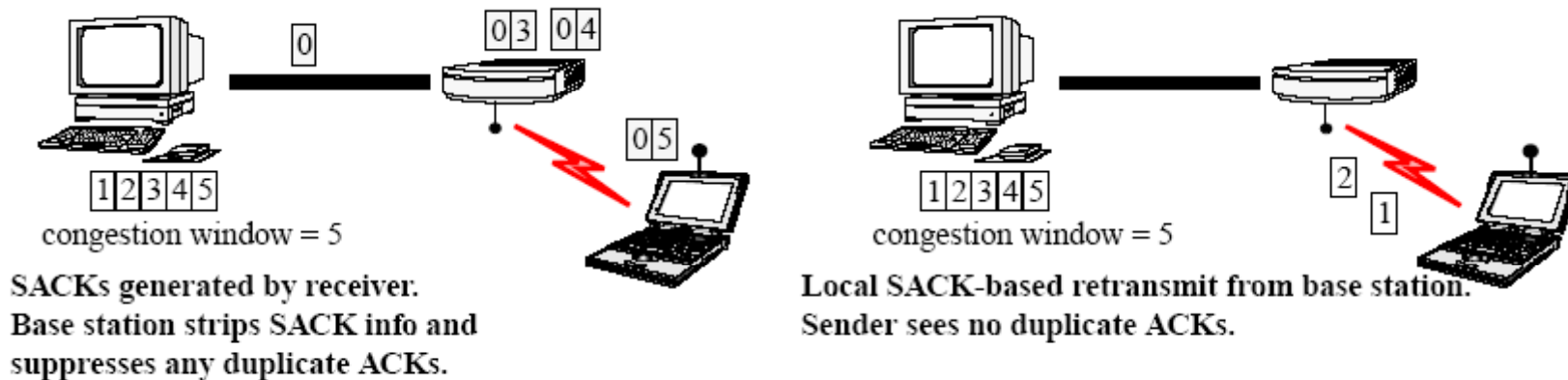


Figure 7. Link-Layer with SMART-based selective acknowledgments and TCP awareness

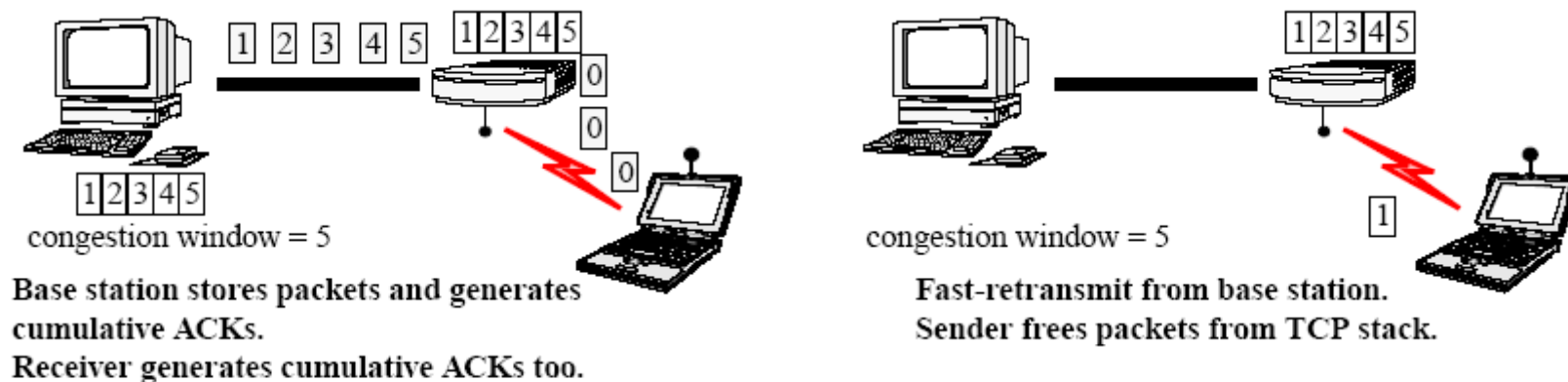


Figure 8. Split-Connection

# TCP Fairness over 802.11

