

Proxy-based TCP-friendly streaming over mobile networks

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Outline

- **Introduction**
- **TCP Friendly Rate Control**
- **TCP Proxy**
- **Variable rate constraints and buffering delays**
- **Simulations**
- **Conclusion**

Introduction

- Streaming media becoming more important over wireless links
- Specialized requirements
 - Timely delivery for playback
 - Limited losses acceptable
 - Receiver buffer used to compensate for data rate fluctuations
- Initial buffering delay
 - needed to adjust for data rate changes
- Buffer Size Factors
 - Encoding of content
 - Available bandwidth during transmission

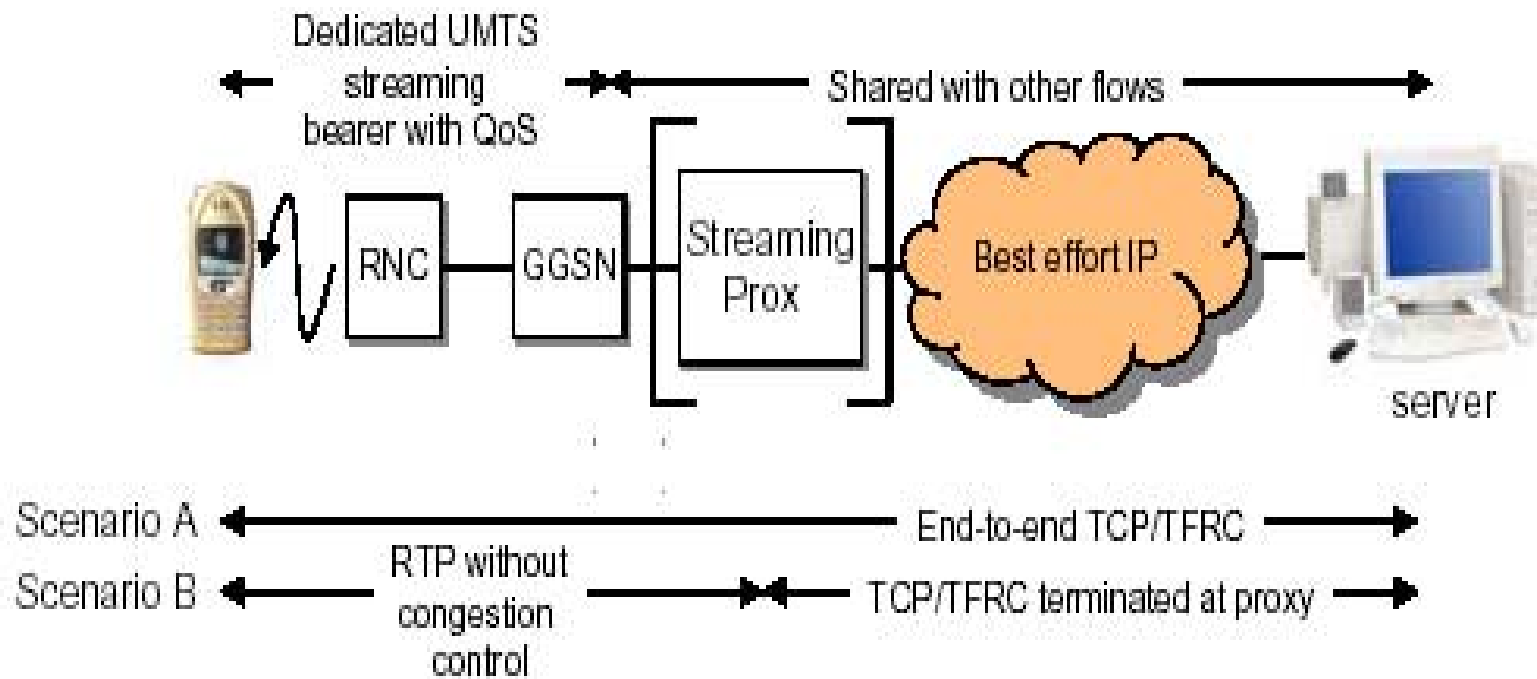
RTP Overview

- Internet-standard protocol for the transport of real-time data, including audio and video
- Data part of RTP is a thin protocol providing support for applications with real-time properties including timing reconstruction, loss detection, security and content identification
- UDP/IP is its initial target

WCMDA Overview

- (W-CDMA) Wideband Code-Division Multiple-Access implementation of third-generation (3G) cellular systems.
- Based on radio access technique proposed by ETSI Alpha group and the specifications was finalized 1999.
- W-CDMA link-level simulations are over 10 times more compute-intensive than current second-generation simulations.
- W-CDMA supports different users simultaneously transmitting at different data rates and data rates can even vary in time.

Two Scenarios – Proxy and End to End



TCP - Friendly Rate Control

- Goal – achieve fairness among TCP and non - TCP flows (improve utilization)
- Limit the drastic responses to congestion control for streaming
- Two major choices:
 - Mimic AIMD
 - Use throughput equation (less fluctuation)

TCP throughput Eq

- TCP throughput =
$$\frac{MTU}{RTT \sqrt{\frac{2p}{3}} + T_0 \sqrt{\frac{27p}{8}} p(1 + 32p^2)}$$
- Upper bound $\leq \frac{1.22MTU}{RTT \sqrt{p}}$

$$\text{TFRC} \quad t_{inter\text{-}packet} = \frac{s \times \sqrt{R_0}}{\text{Throughput} \times M}$$

- Equation-based congestion control
- Assigns network estimation to receiver
- Uses Weighting method to average p over number of loss intervals
- Uses EWMA (exponential weighted moving average) to calculate RTT
- Creates less fluctuation

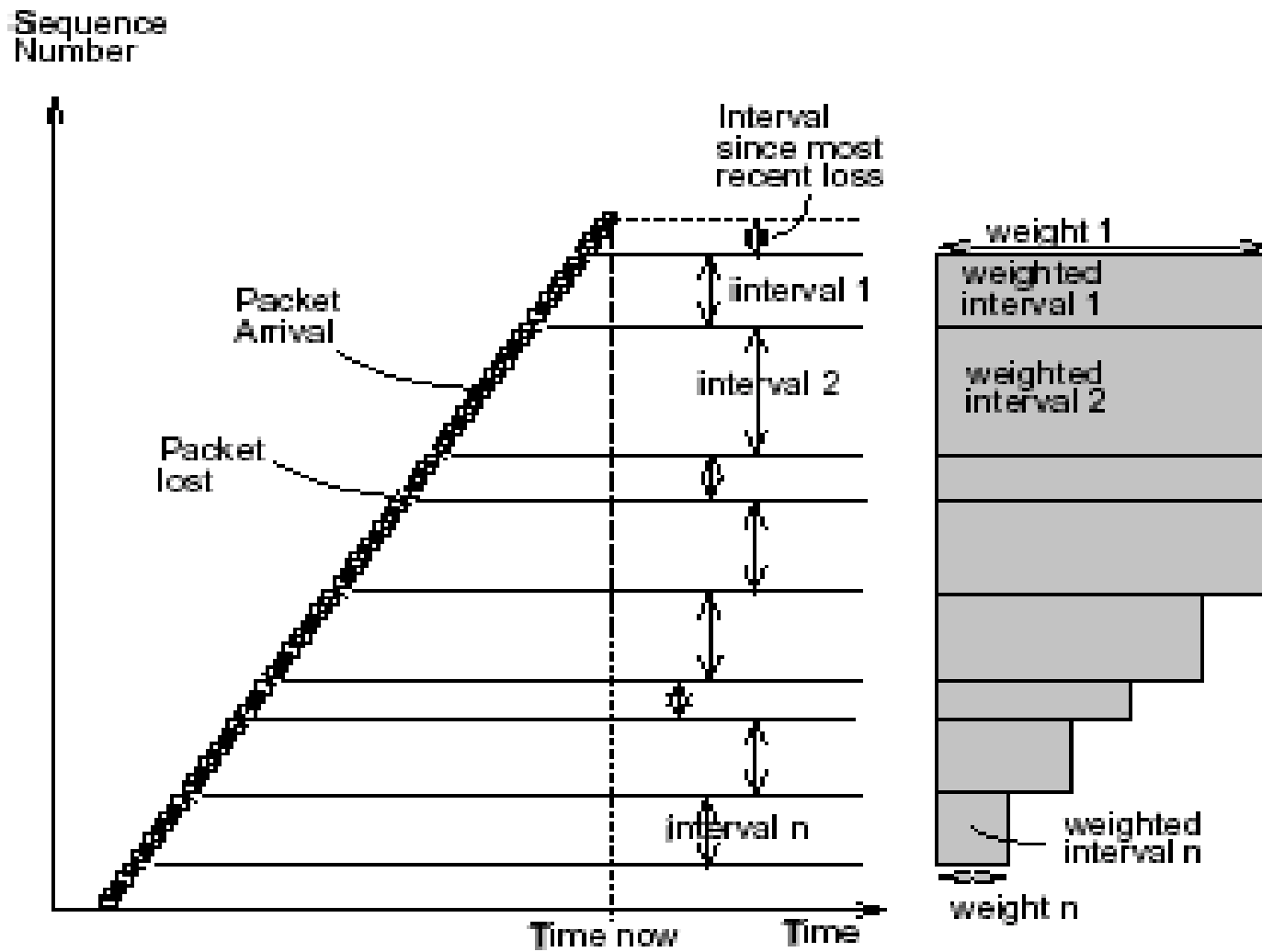


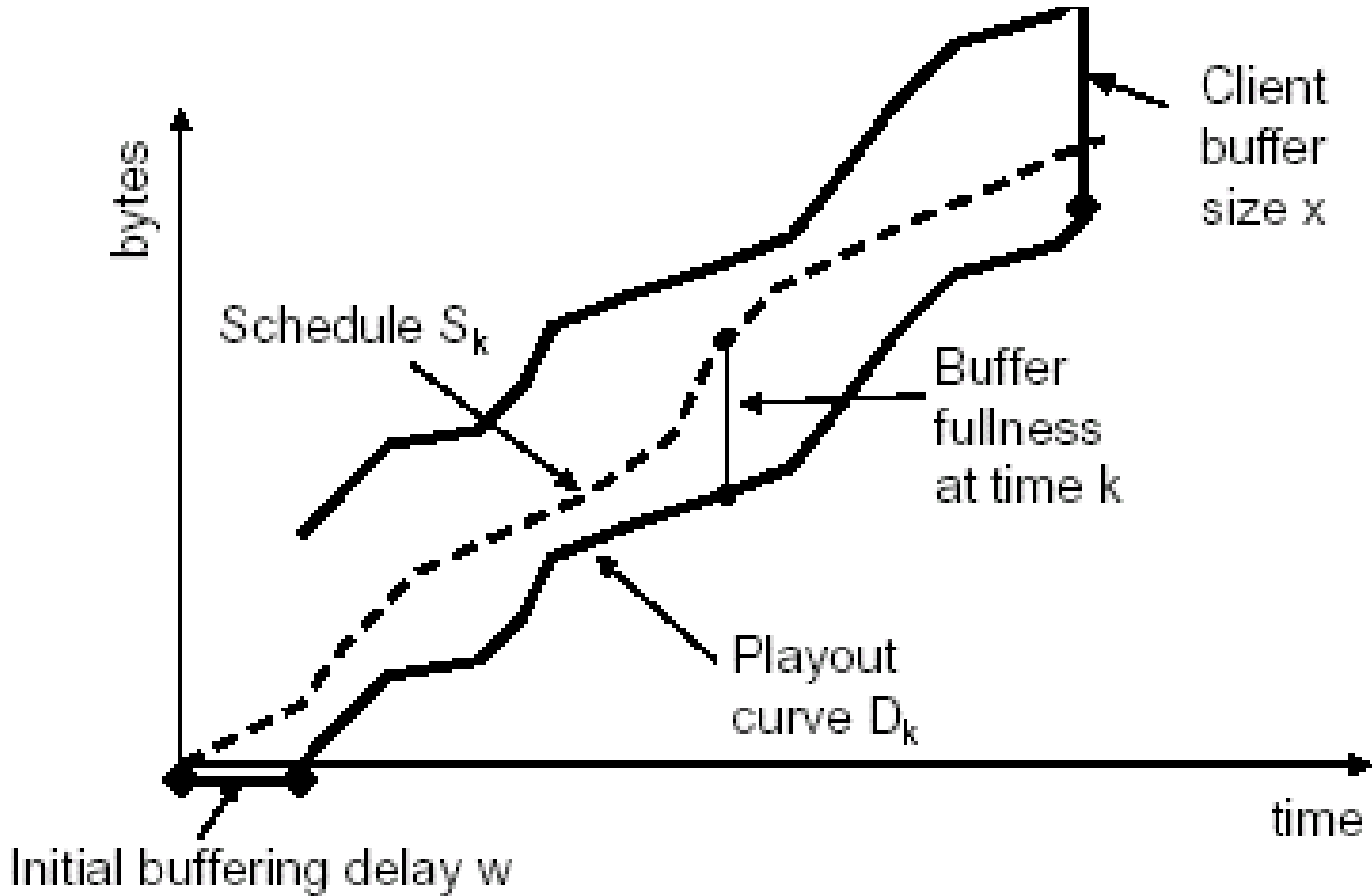
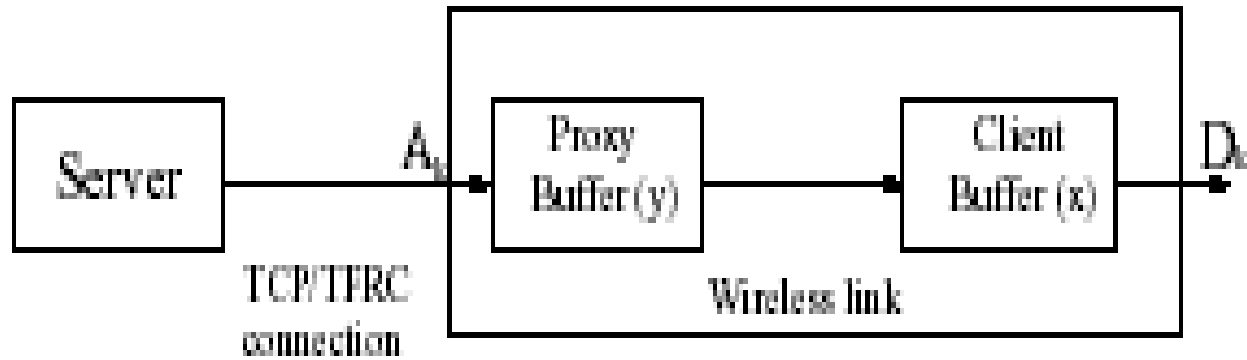
Figure 1: Weighted intervals between loss used to calculate loss probability.

TCP Proxy

- Needed because TCP has trouble fully utilizing mobile links
 - Large RTT
 - Non congestion packet loss
- Shields wireless link from Internet packet loss

Traffic Smoothing

- Video streaming uses constant quality encoding – results in variable rate streams
- To reduce fluctuations, server transmits frames into client playback buffer in advance of bursts
- Goal is to minimize required client buffer size and also don't over or under fill buffer



Minimizing Initial Buffering Delay

$$S_k^{late} = \begin{cases} D_n \dots \dots \dots (k=N) \\ \max\{S_{k+1}^{late} - R_k, D_k\} \dots \dots (k < N) \end{cases}$$

$$x^* = \max\{S_k^{late} - D_k\}$$

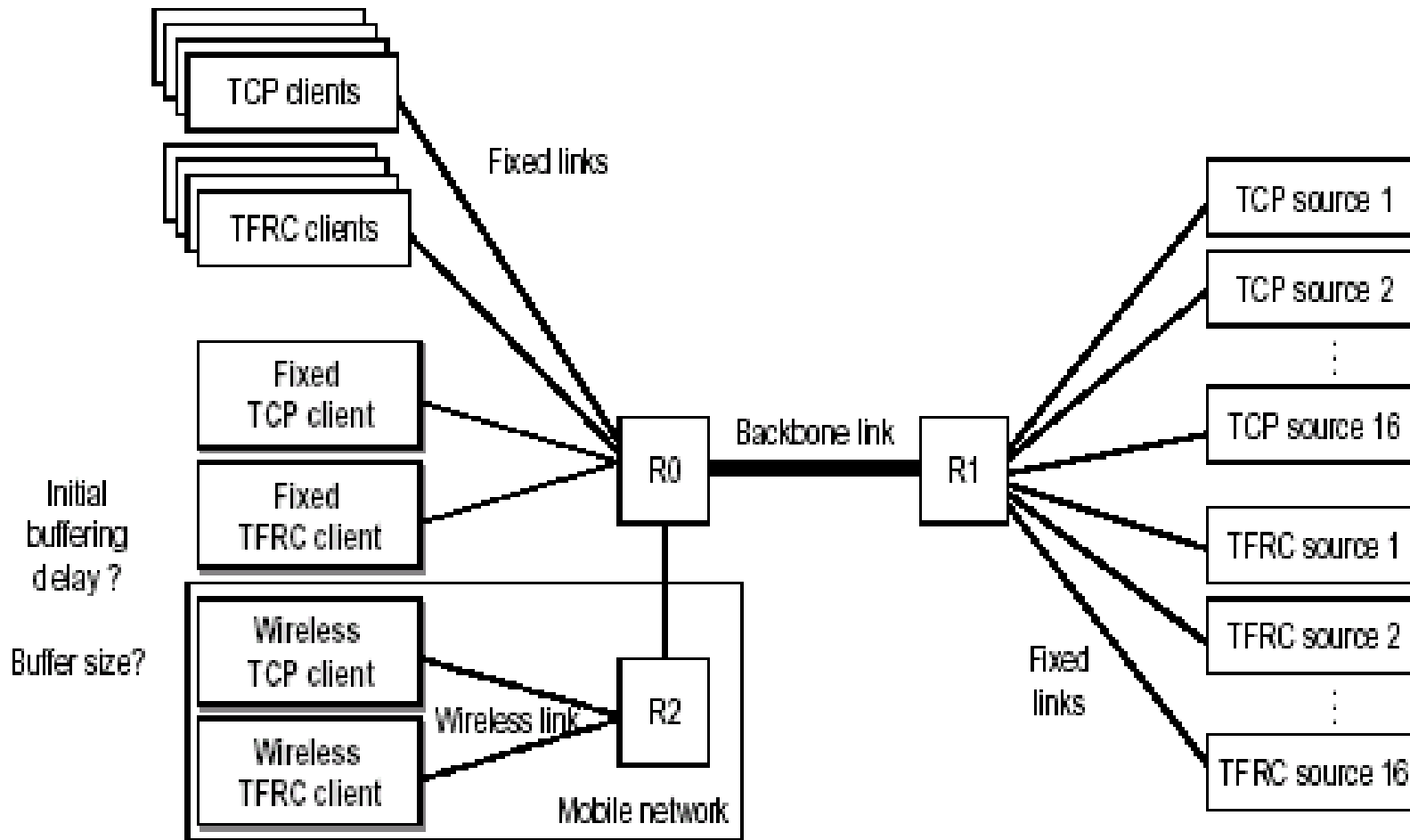
$$w^* = \min\{w \mid A_k - S_{k-w}^{late} \geq 0\}$$

Minimizing Initial Buffering Delay

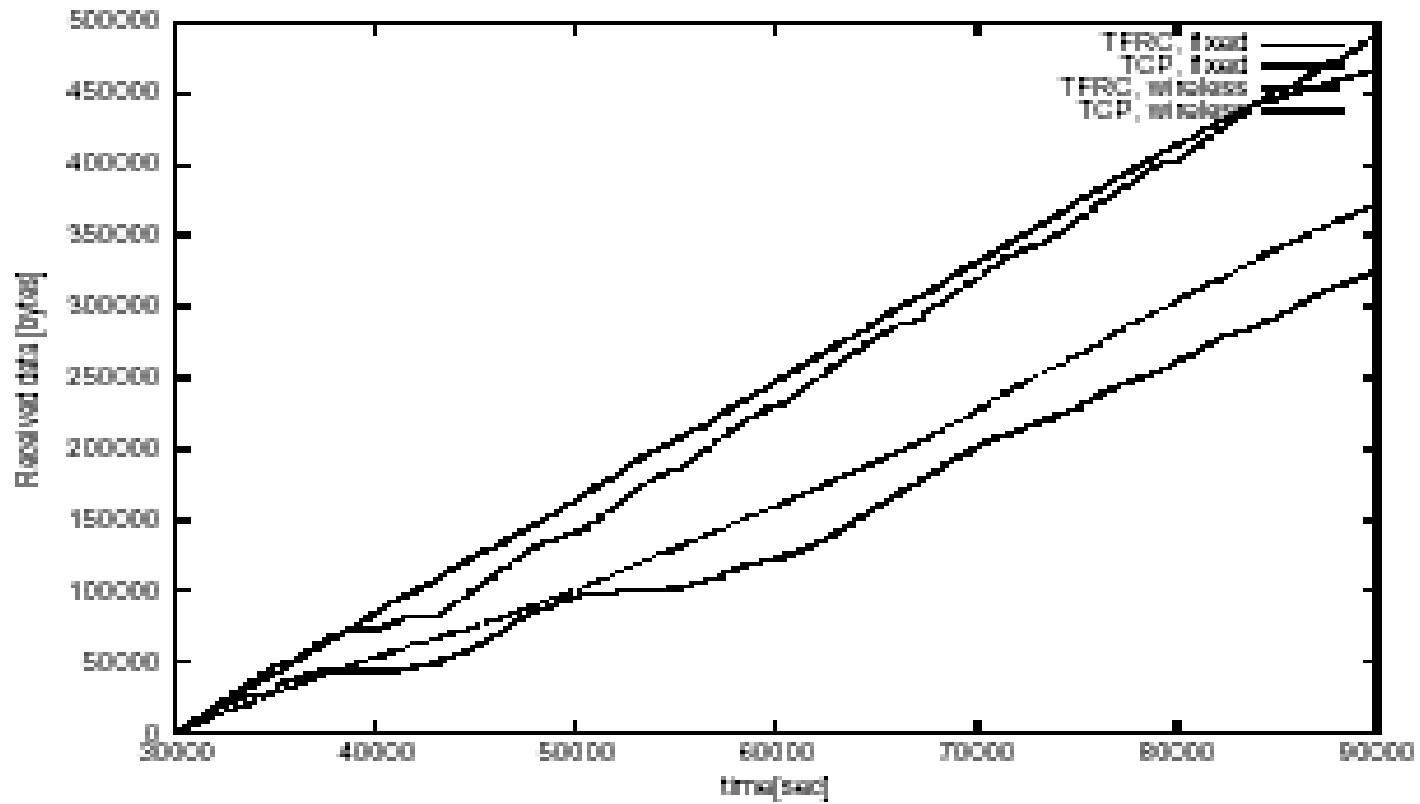
Proxy case

$$S_k^{early} = \begin{cases} 0 \dots \dots \dots (k=0) \\ \max\{S_{k-1}^{early} + R_k, A_k\} \dots \dots (k > N) \end{cases}$$

Simulation

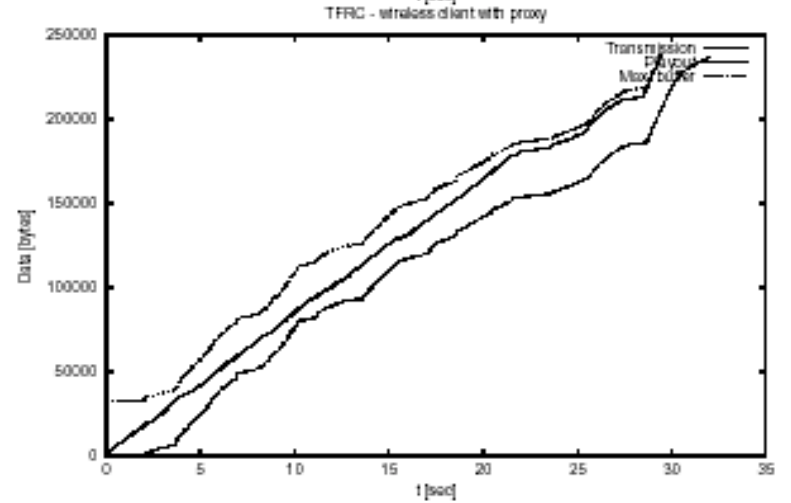
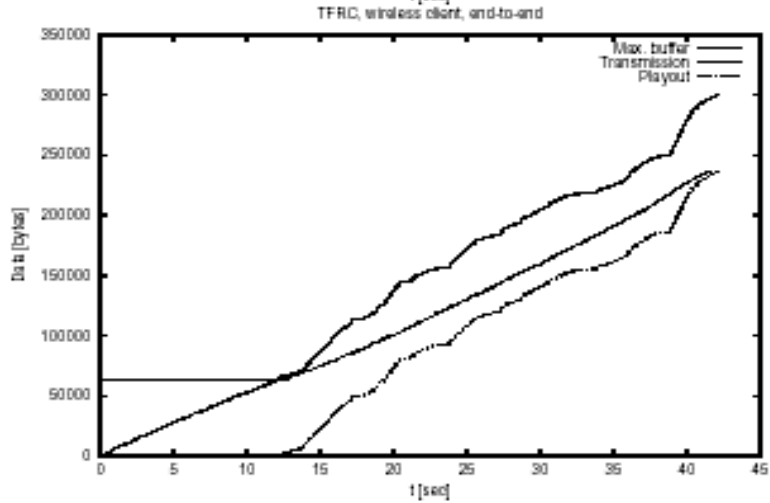
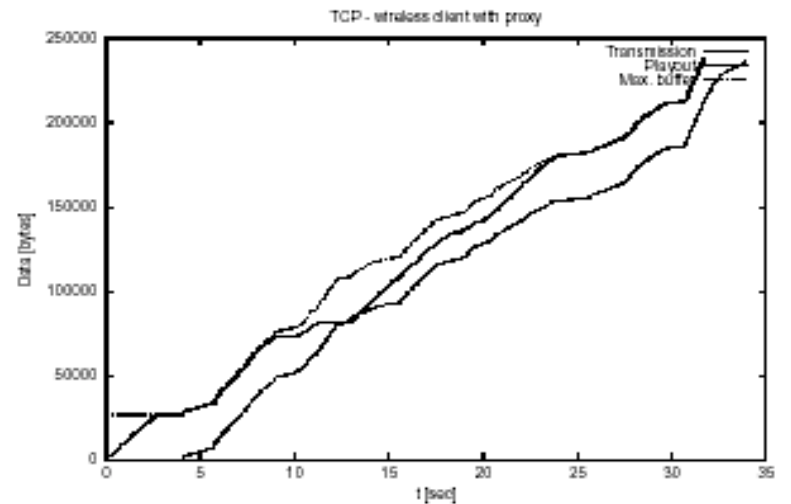
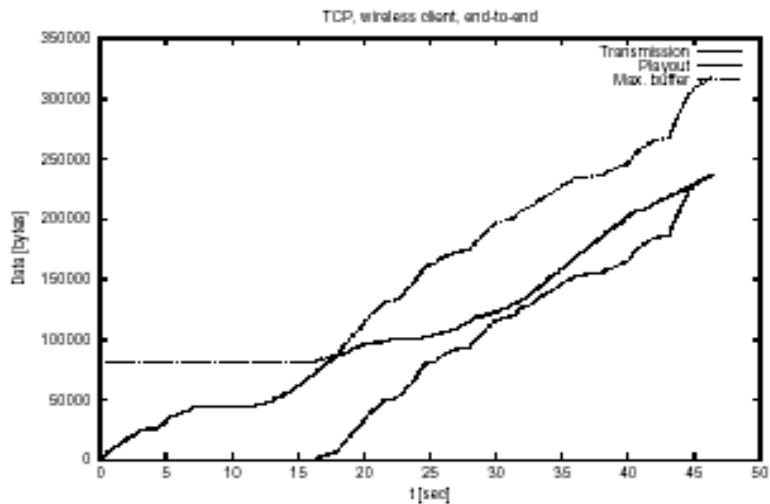


TCP TFRC Transmission behavior



Simulation Results

Case	R [kbps]	W [sec]	X [bytes]
TCP, wired ?	65.6	1.7	26567
TFRC, wired ?	62.2	2.8	32016
TCP, wireless	43.3	13.9	76840
TFRC, wireless	49.7	9.0	58320



Simulation Results (Proxy)

Without proxy			
Case	R [kbps]	W [sec]	X [bytes]
TCP	16.5	81030	n.a.
TFRC	12.5	63833	n.a.
With Proxy			
TCP	4.1	26924	26924
TFRC	2.2	26924	32602

Conclusions

- Use of Proxy resulted in decrease of initial delay buffer and decreased client size buffer
- Also showed increased utilization and average throughput
- Many aspects still to consider in the future

Questions?