



# VoIP Call Capacity

Group 2

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# Introduction

- Problem Overview
- Topology and Results Summary
- Attempted Improvements
- Conclusions

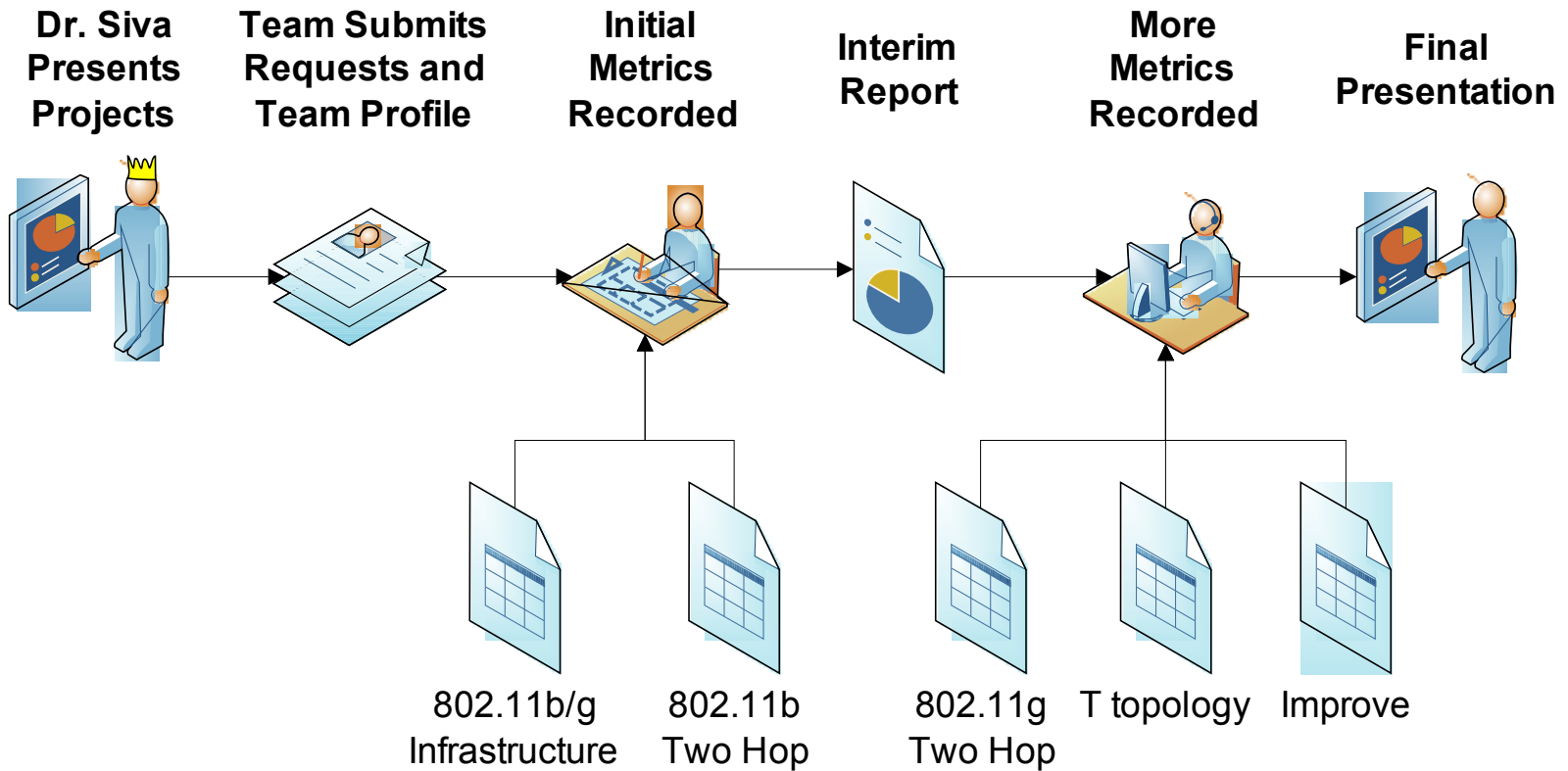
# Problem Statement

- Using open source tools, setup appropriate network topologies and measure VoIP call capacity for 802.11b and 802.11g networks
- Develop solutions to improve call capacity

# Motivation

- WLANs are widely deployed
- VoIP has a promising future as the dominant voice communication method
- WLAN ad-hoc networks are ideal for communicating in disaster areas
- Important to measure the quality/capacity of VoIP traffic over WLANs

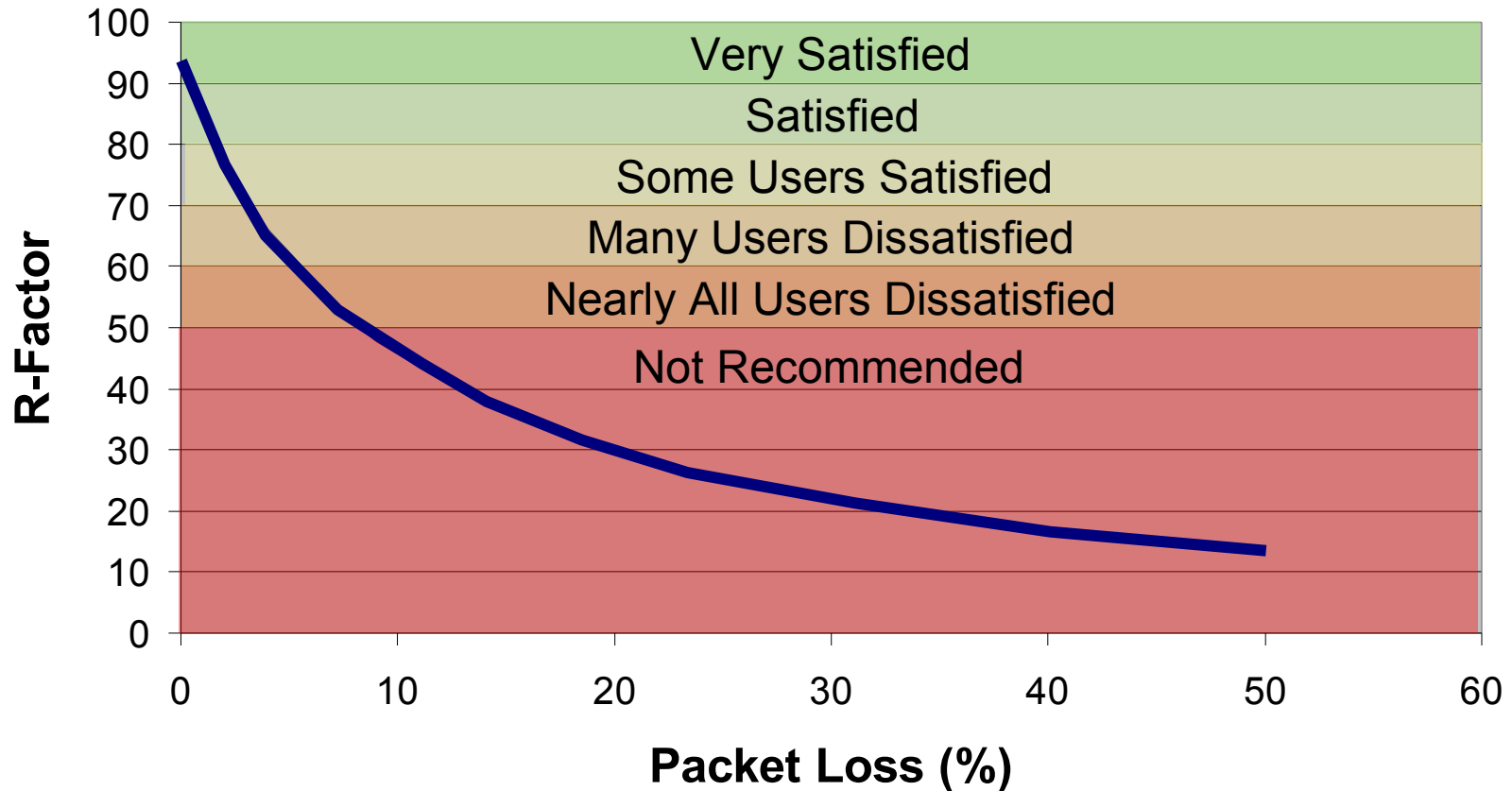
# Milestones



# R-Factor

ITU Recommendation G107 Introduced E-model which outputs R-factor based on perceptual and equipment impairments in its calculation.

## Packet Loss and R-Factor

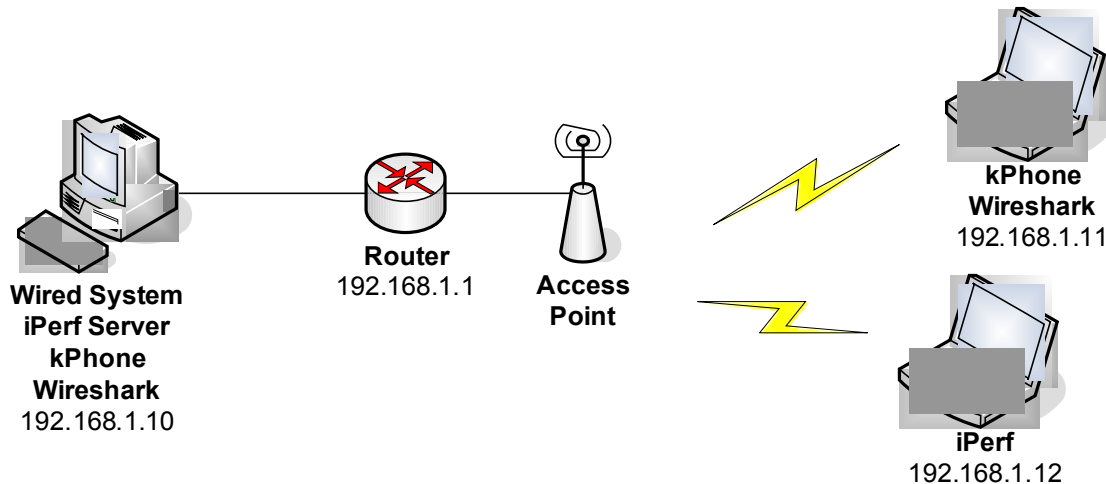




# Topologies Overview

- Infrastructure
- Two Hop
- Cross Traffic T
- Two Hop with Improvements

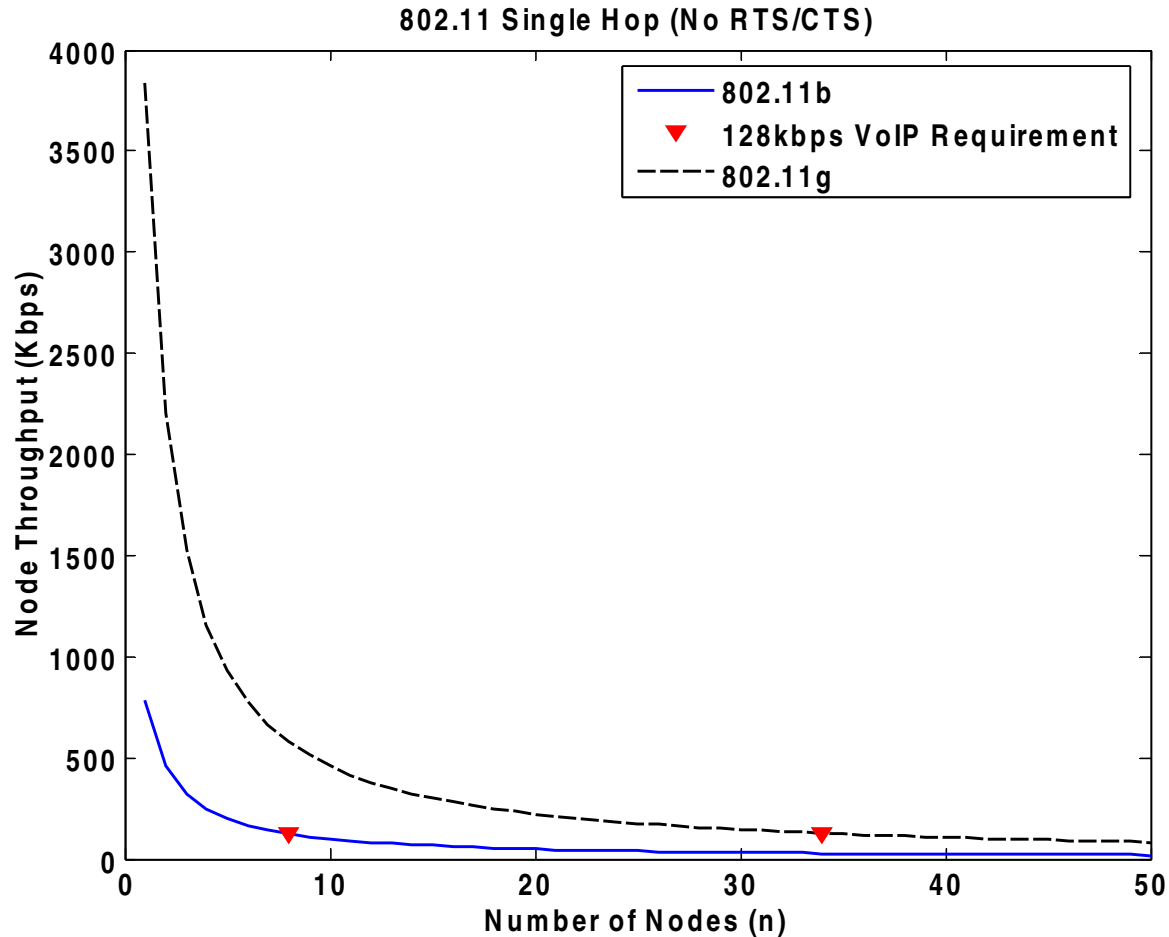
# Infrastructure



- Software: Linux, kPhone, iPerf, Wireshark
- Mix of lab and personal hardware
- Recording on *both* kPhone clients!



# Theoretical Node Throughput



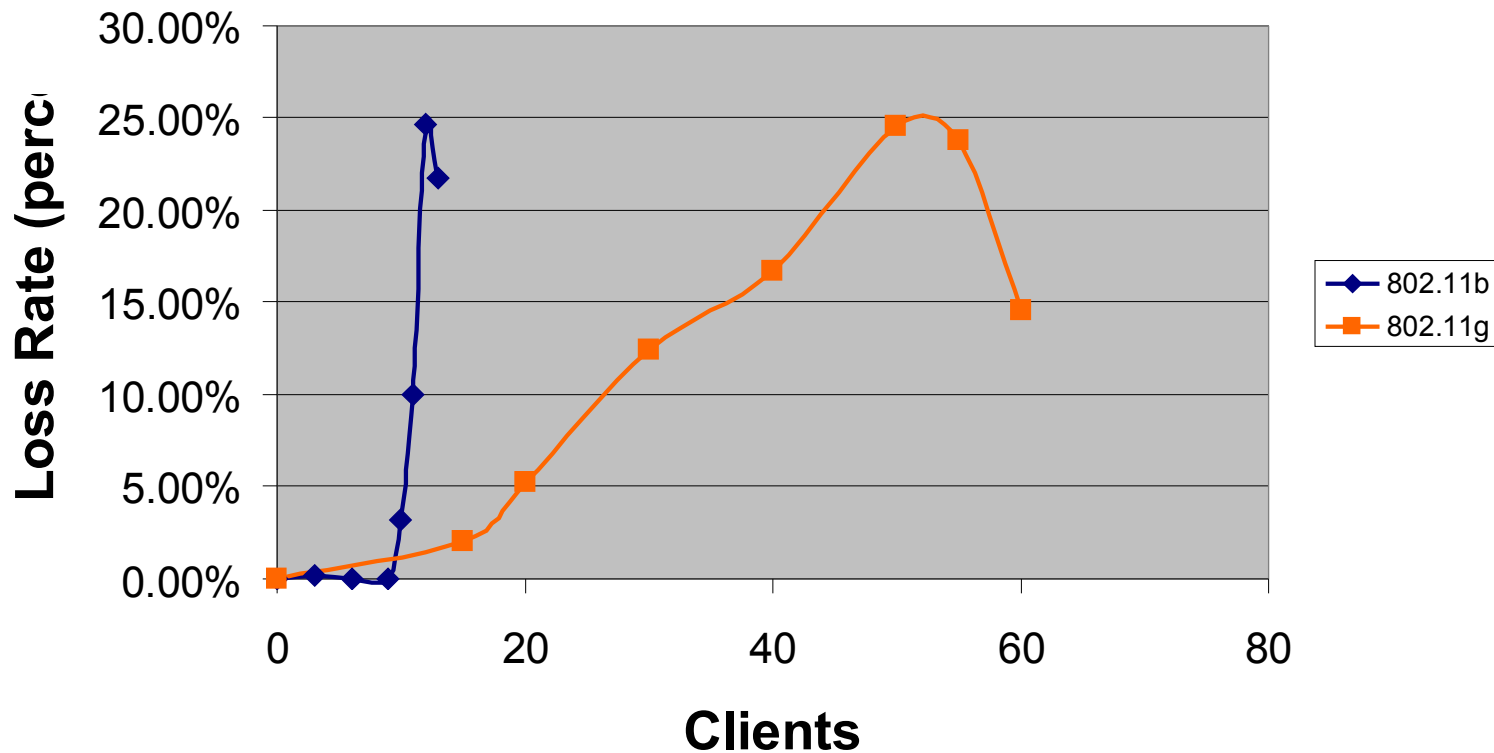
# Infrastructure (Cont.)

| <b>Calls Supported</b> | <b>802.11b</b> | <b>802.11g</b> |
|------------------------|----------------|----------------|
| <b>Theoretical Max</b> | 86             | 422            |
| <b>Analytical</b>      | 8              | 34             |
| <b>Measured</b>        | 11             | 30             |

- Supported calls is substantially lower than expected
- Small packets and packet overhead reduce maximum calls
- Values inflated due to limited number of computers in the testbed

# Loss Rates

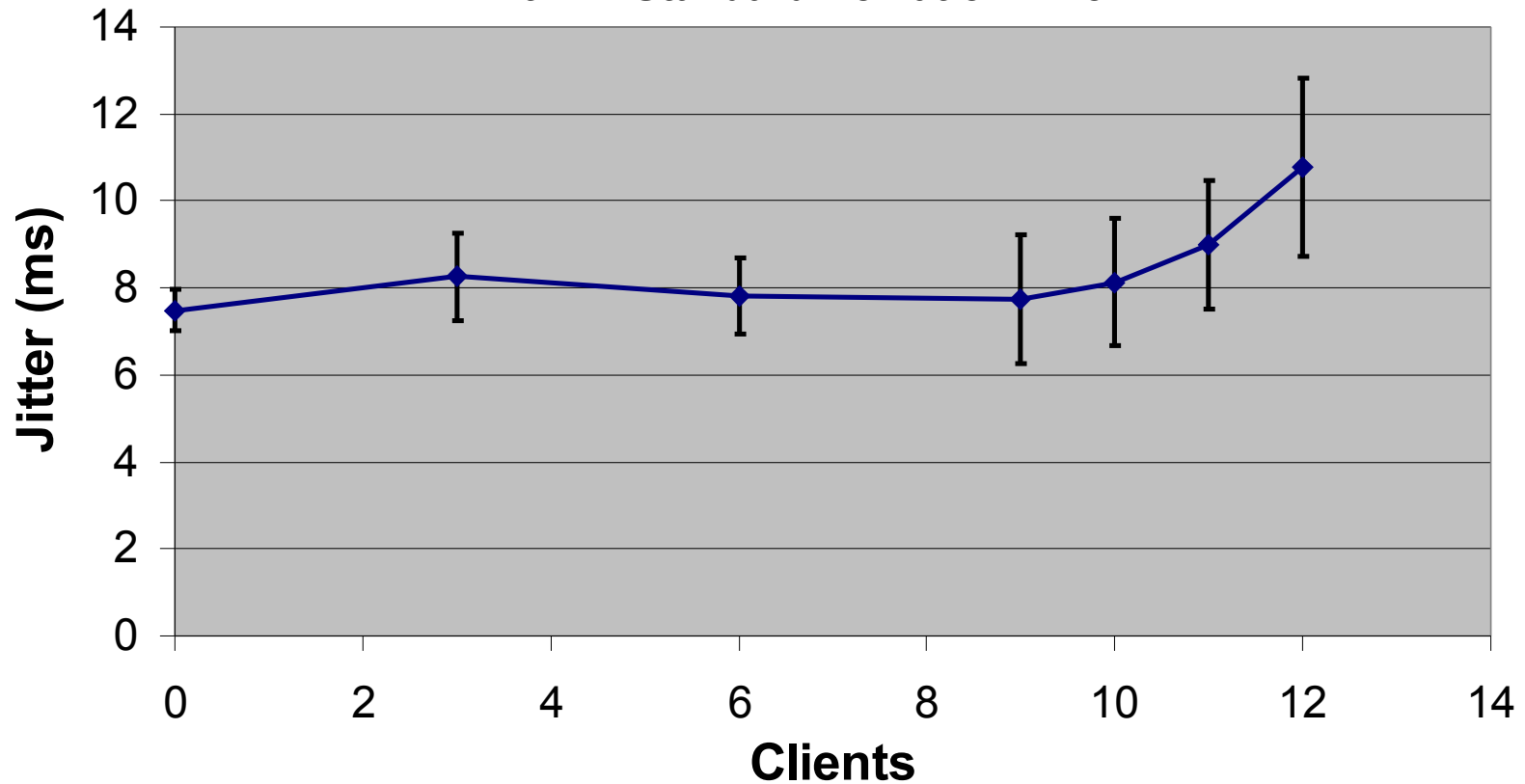
## 802.11b and 802.11g Infrastructure Losses



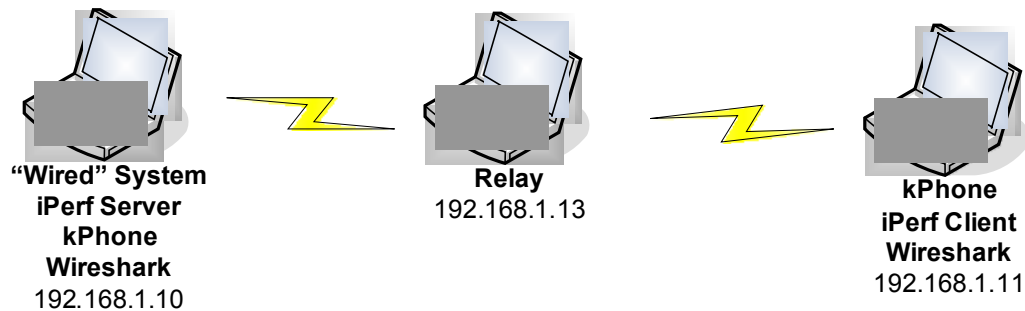
# Jitter

## 802.11b Infrastructure Jitter

with +/- Standard Deviation Error



# Linear Multi-hop



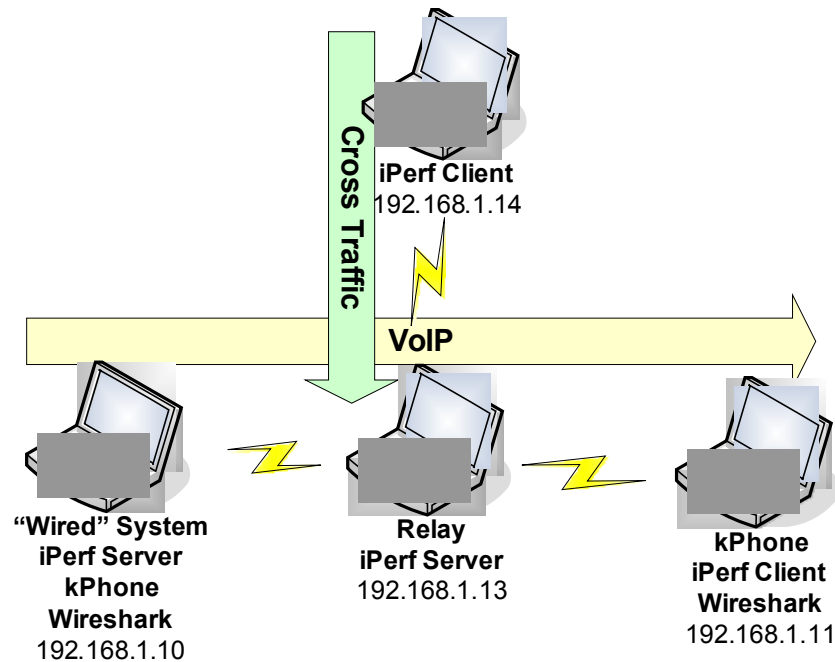
- Adhoc setup with Linux system acting as relay
- Increased contention
- Wireshark recording on endpoints (NOT in the middle!)

# Linear Multi-hop (Cont.)

|                 | 802.11b | 802.11g |
|-----------------|---------|---------|
| Calls Supported | 3       | 9       |

- Multiple hops reduces maximum
- Higher contention
- Excessive delay (2-5 seconds mean delay in test cases)

# Cross Traffic T-Mode



- Increased contention
- Start of a grid

# T-Mode (Cont.)

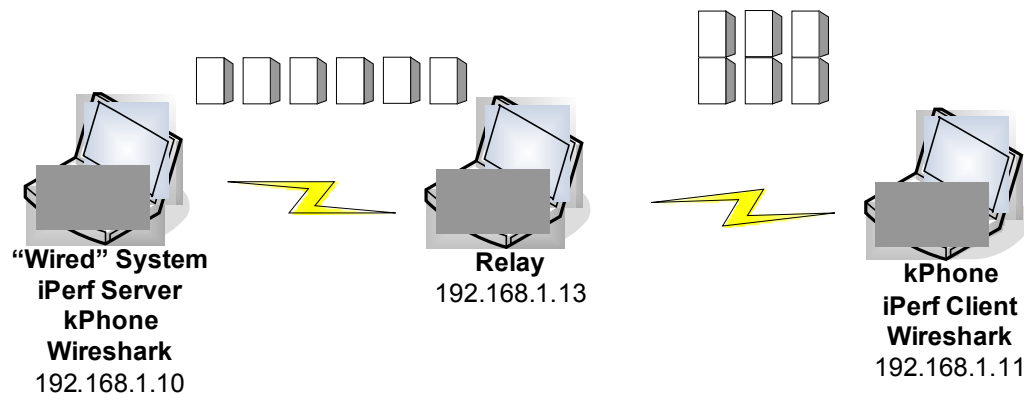
|                        | <b>802.11b</b> | <b>802.11g</b> |
|------------------------|----------------|----------------|
| <b>Calls Supported</b> | 5              | -              |

- Computers become overloaded in relay



# Improvements

- Turn RTS/CTS on
- VoIP packet size increase
- Packet Aggregation



# Improvements (Cont.)

|                            | <b>RTS/CTS</b> | <b>Double<br/>Size</b> | <b>Aggregate</b> |
|----------------------------|----------------|------------------------|------------------|
| <b>Calls<br/>Supported</b> | 3              | 5                      | 3-5              |

- RTS/CTS introduced extremely large delay but no improvements
- Some improvement in increased packet size
- Aggregation indicated small improvement
- Small improvements hard to measure in environment

# Conclusions

- WLAN characteristics result in large delay and low VoIP call capacity
- Ad-hoc networks more affected by delay, infrastructure more affected by loss
- Improvements can be made by reducing the number of packets contending at any given time



# Thanks for your attention

- Feel free to ask questions!



# All Losses

## 802.11 Losses

