

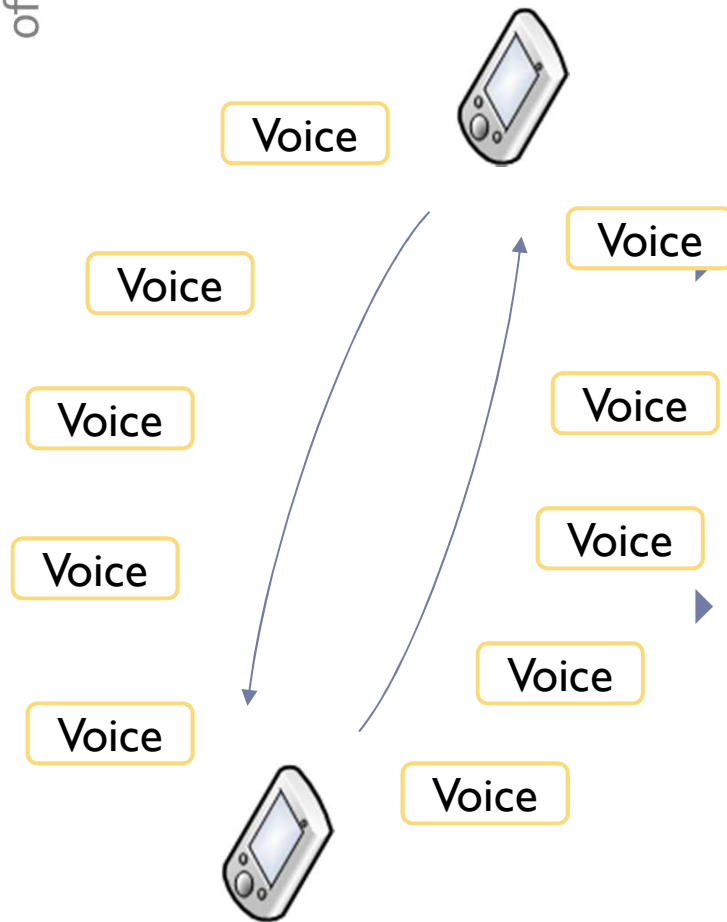
# Supporting VoIP in IEEE802.11 Distributed WLANs

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# VoIP Applications



## ▶ Constant Streaming Traffic

- ▶ Packetize interval usually 10-30 ms
- ▶ 8 – 160 bytes each packet
- ▶ RTP/UDP

## ▶ Time Sensitive

- ▶ Interactive voice service
- ▶ If round-trip delay is high (e.g. >300 ms), users may suffer in conversation

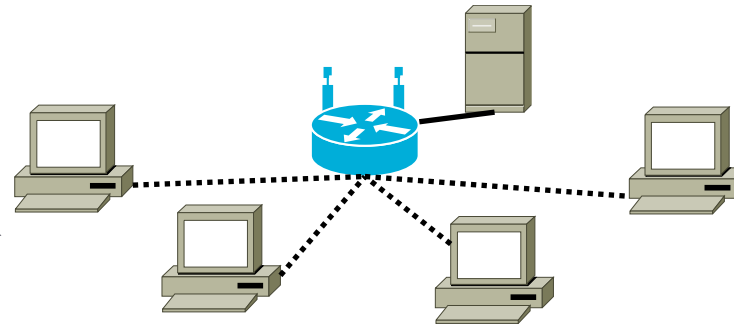
## ▶ Tolerate to Some Packet Losses

- ▶ Conversation can go on even when some packets are not delivered in time (e.g. 10%)
- ▶ Packet lost concealment techniques

# VoWLAN Scenarios

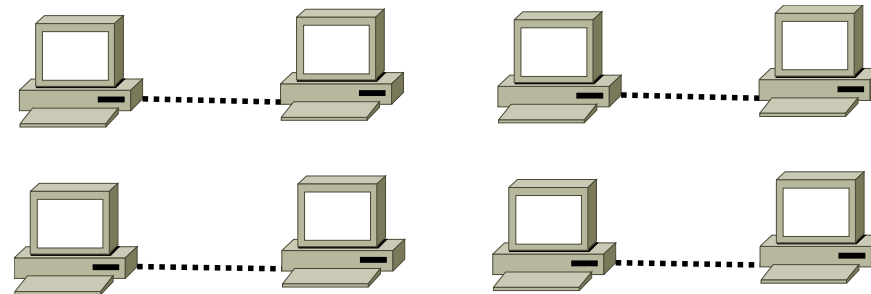
## Existing Research

- ▶ Infrastructure
- ▶ Unbalanced uplink/downlink traffic



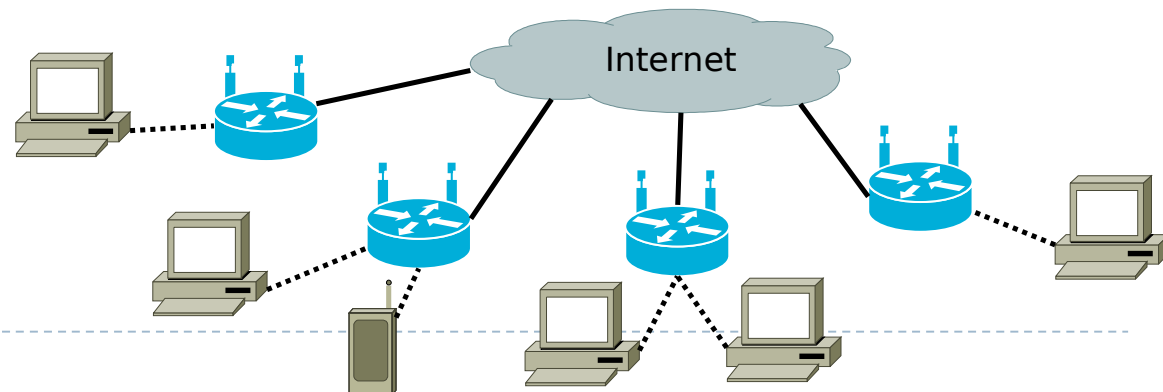
## My Research

- ▶ Fully distributed
- ▶ Contentions for transmission opportunity



## Real World

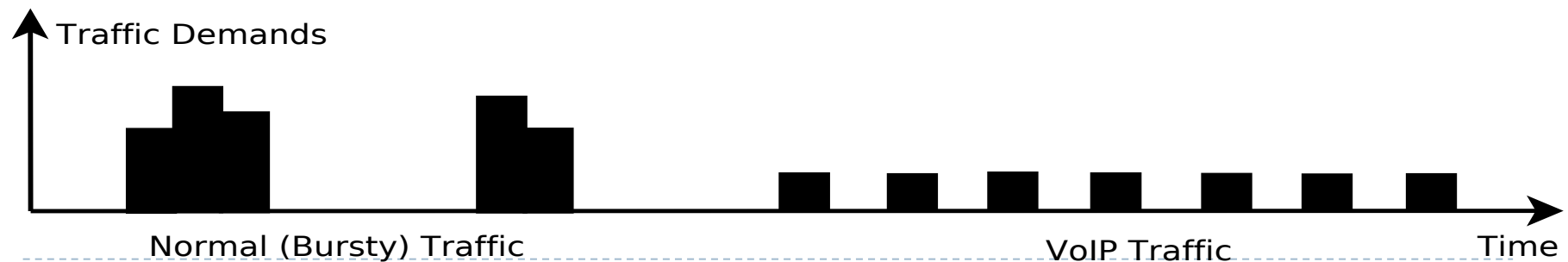
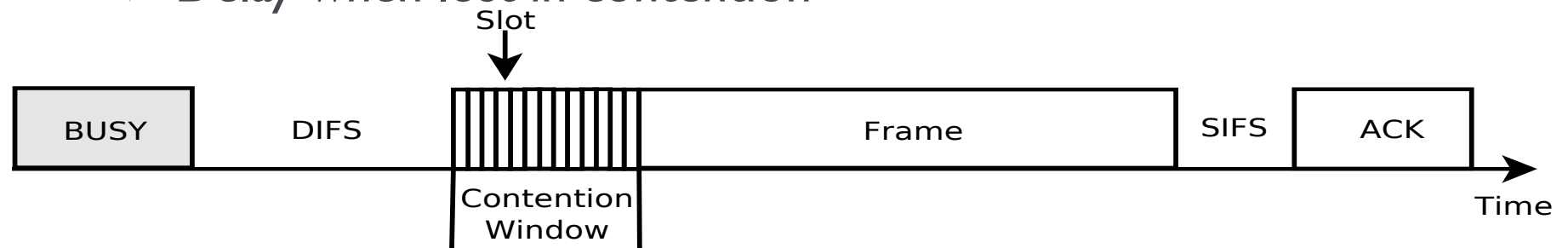
- ▶ Mixed



# Issues for VoIP in IEEE802.11 WLAN 1

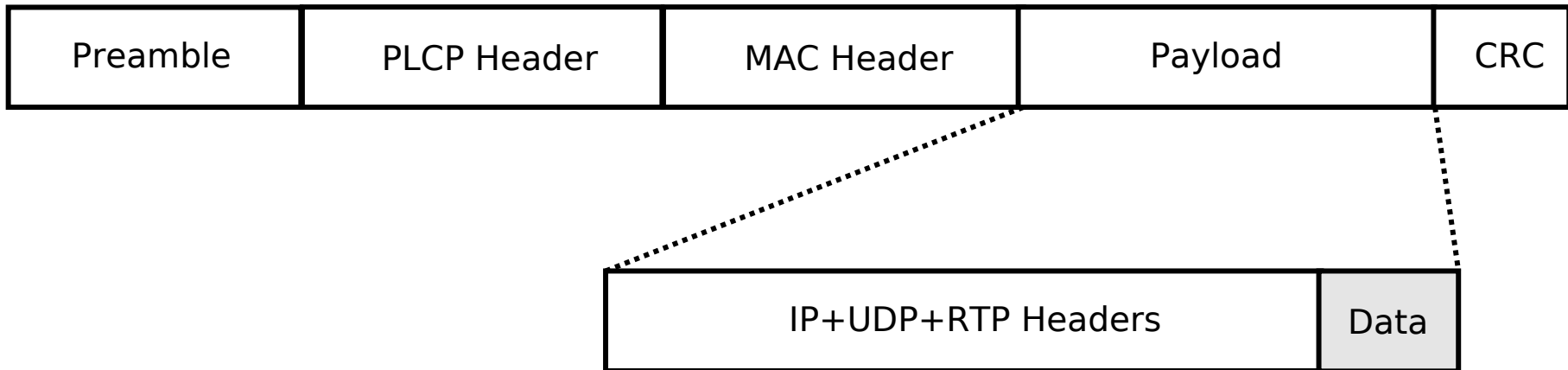
## ▶ Shared Wireless Channel

- ▶ Overhead by contending for transmission opportunity (CSMA/CA)
- ▶ More contentions for VoIP traffic
- ▶ Delay when lost in contention

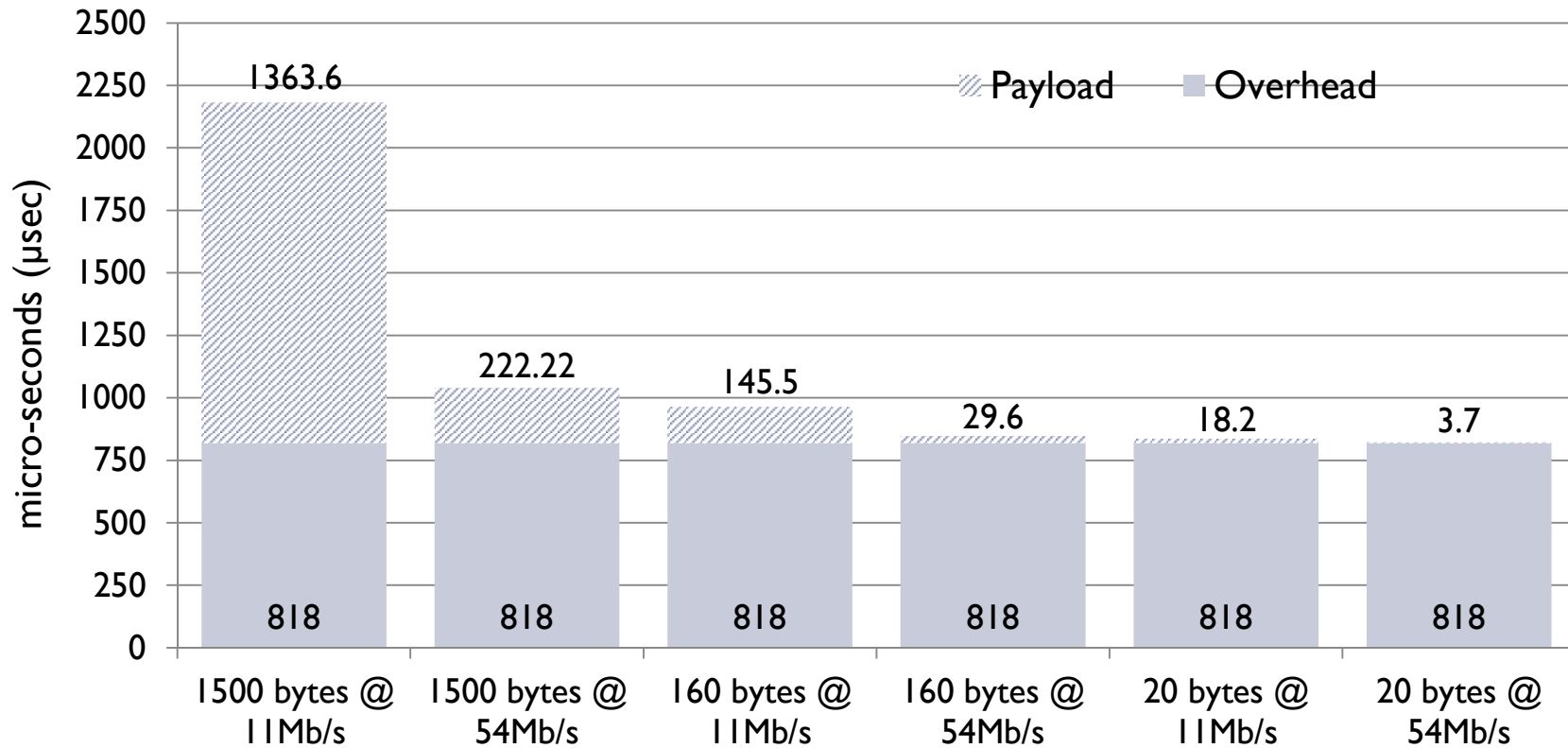


## Issues for VoIP in IEEE802.11 WLAN 2

- ▶ **Very Low Efficiency over Wireless Channel**
  - ▶ Control data overhead is too large
  - ▶ Payload is tiny for each voice packet

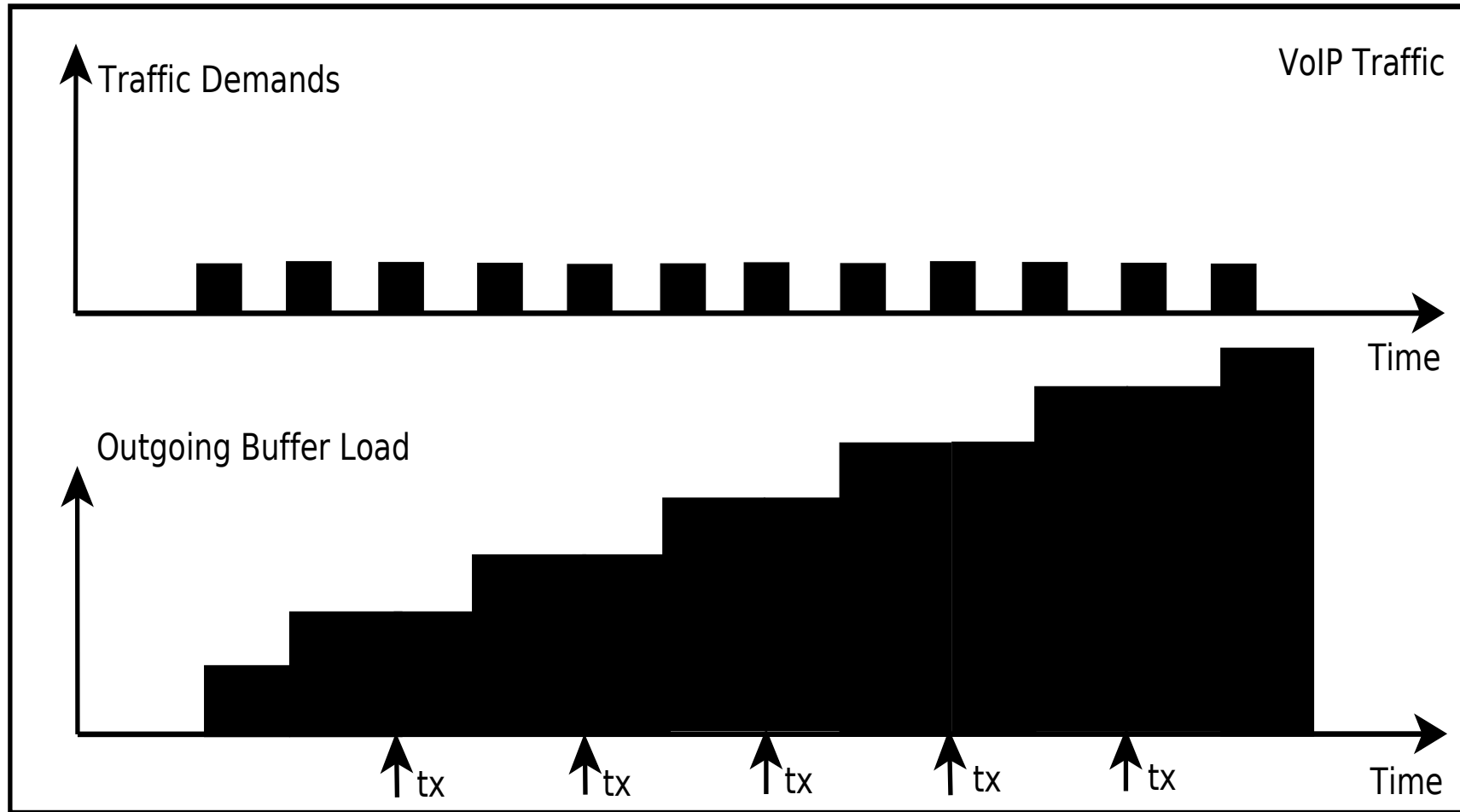


# Estimations



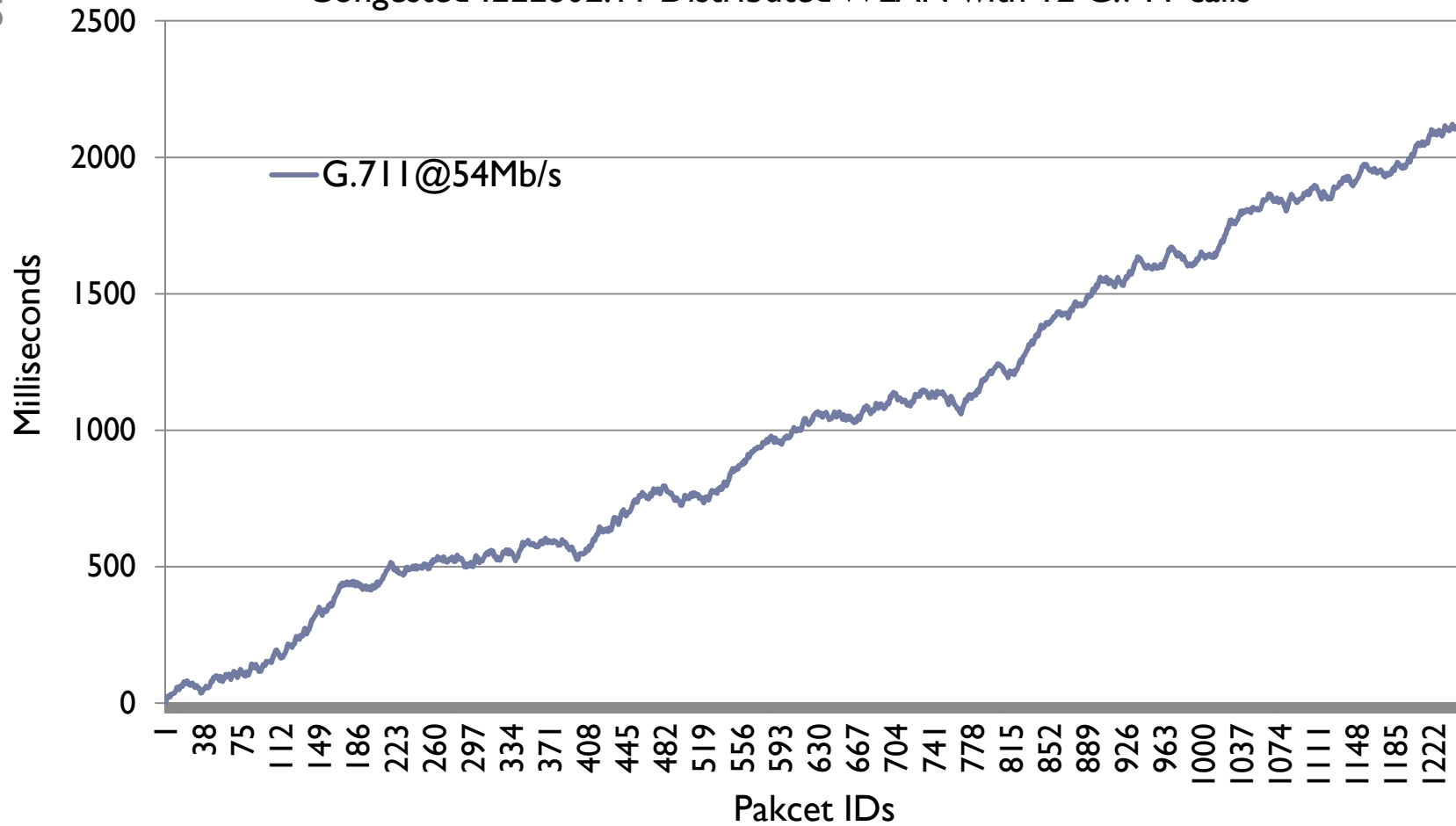
802.11b/g	(G.711)160 bytes @ 11Mb/s	(G.711)160 bytes @ 54Mb/s	(G.729)20 bytes @ 11Mb/s	(G.729)20 bytes @ 54Mb/s
VoIP Call Capacity	10.37883	11.79802	11.95886	12.16989

# Buffer Accumulation



# Delay Accumulation

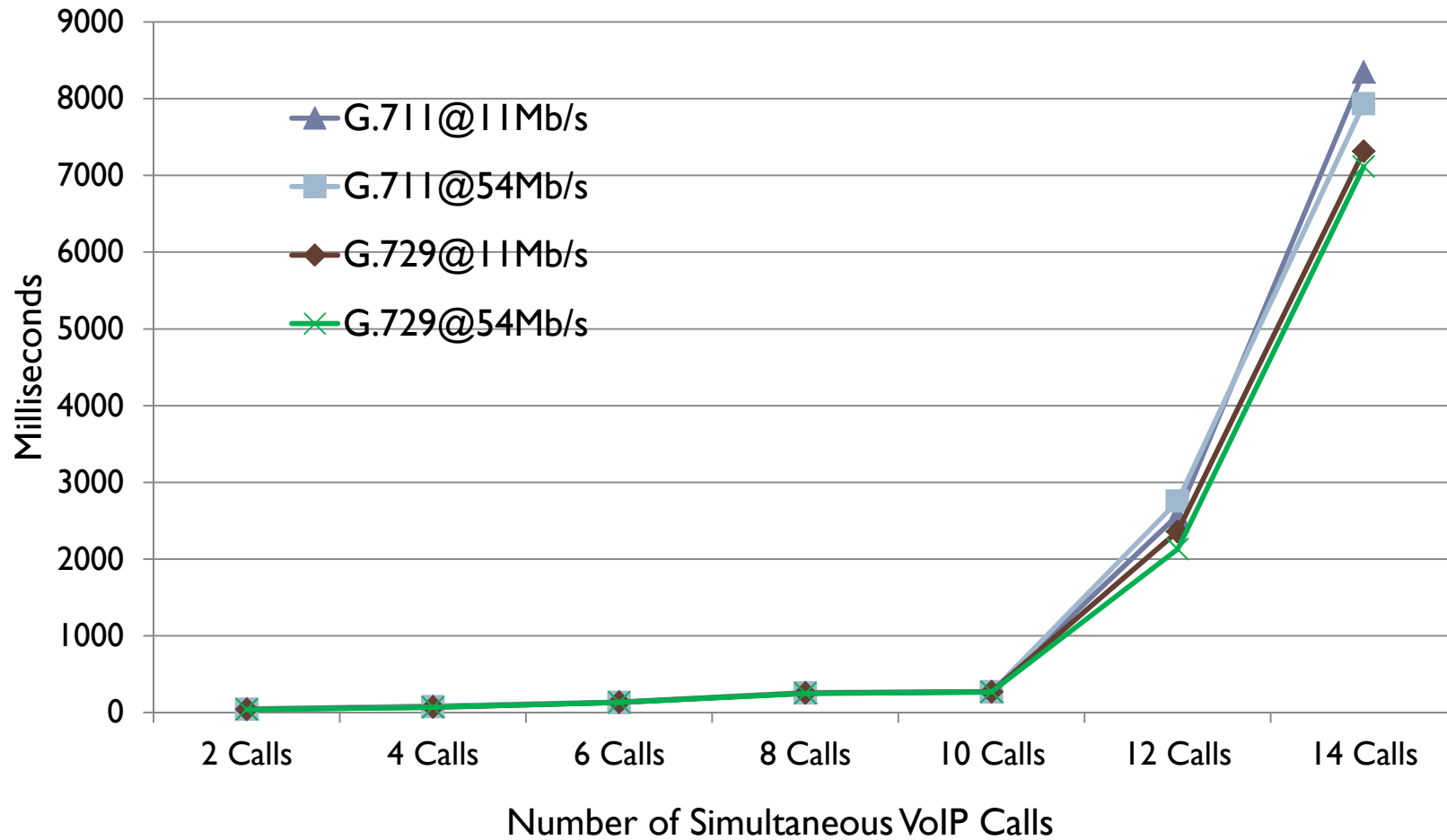
Delay Accumulation Observed from Round-Trip Time Measured in a Real Congested IEEE802.11 Distributed WLAN with 12 G.711 calls





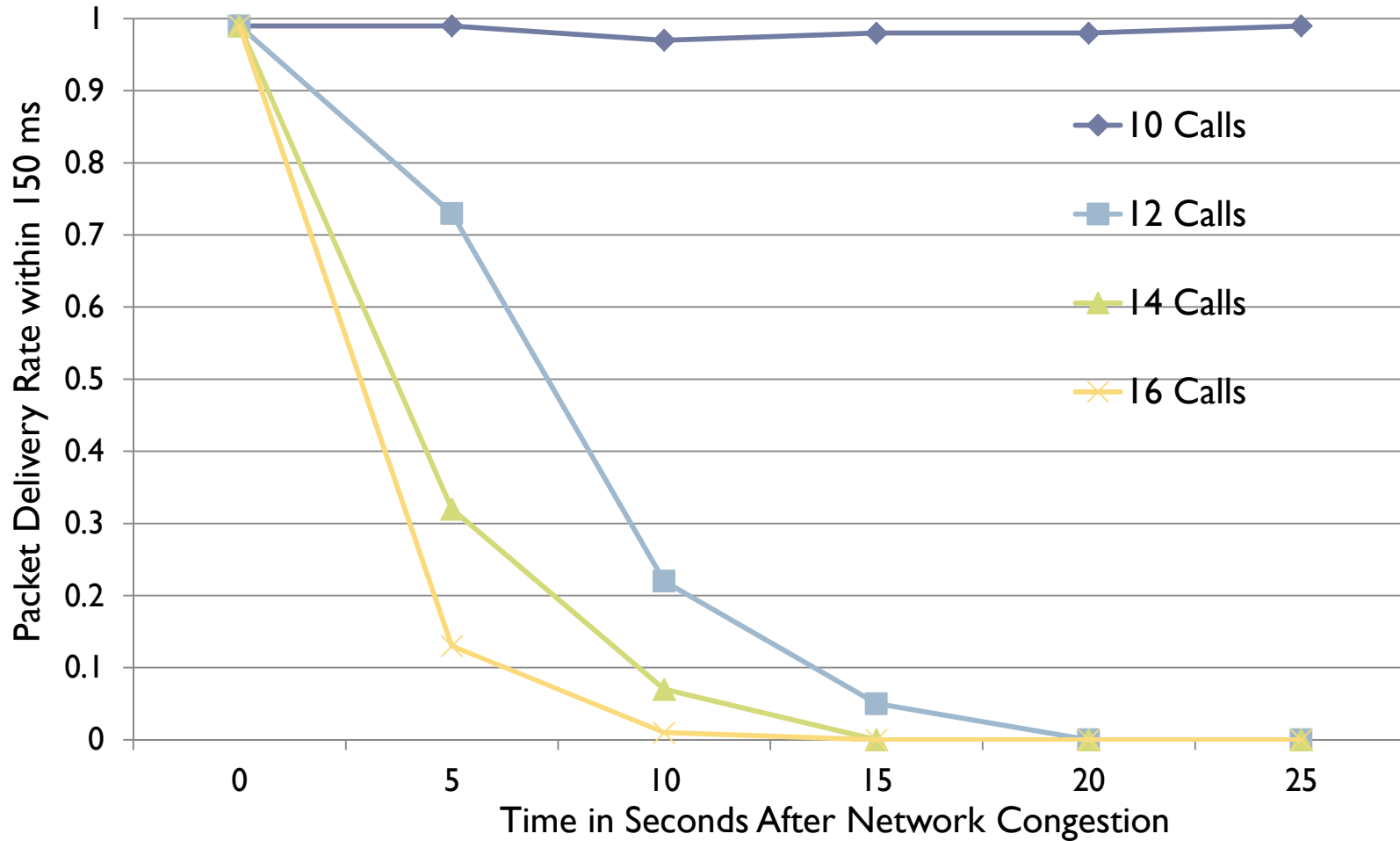
# Round Trip Time in Real 802.11 WLAN

Average RTT Measured by VoIP-RTT-Emulator in Real IEEE802.11b/g  
Distributed WLAN



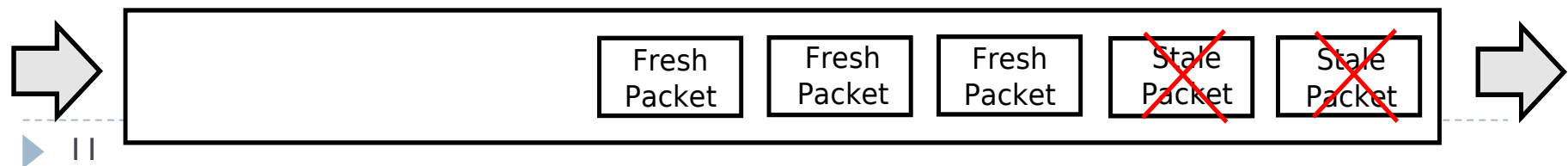
# Simulated Congestion Test

Normalised Packet Delivery Rate within 150 ms Delay Budget after Network Congestion Happens in Simulated IEEE802.11 Distributed WLAN

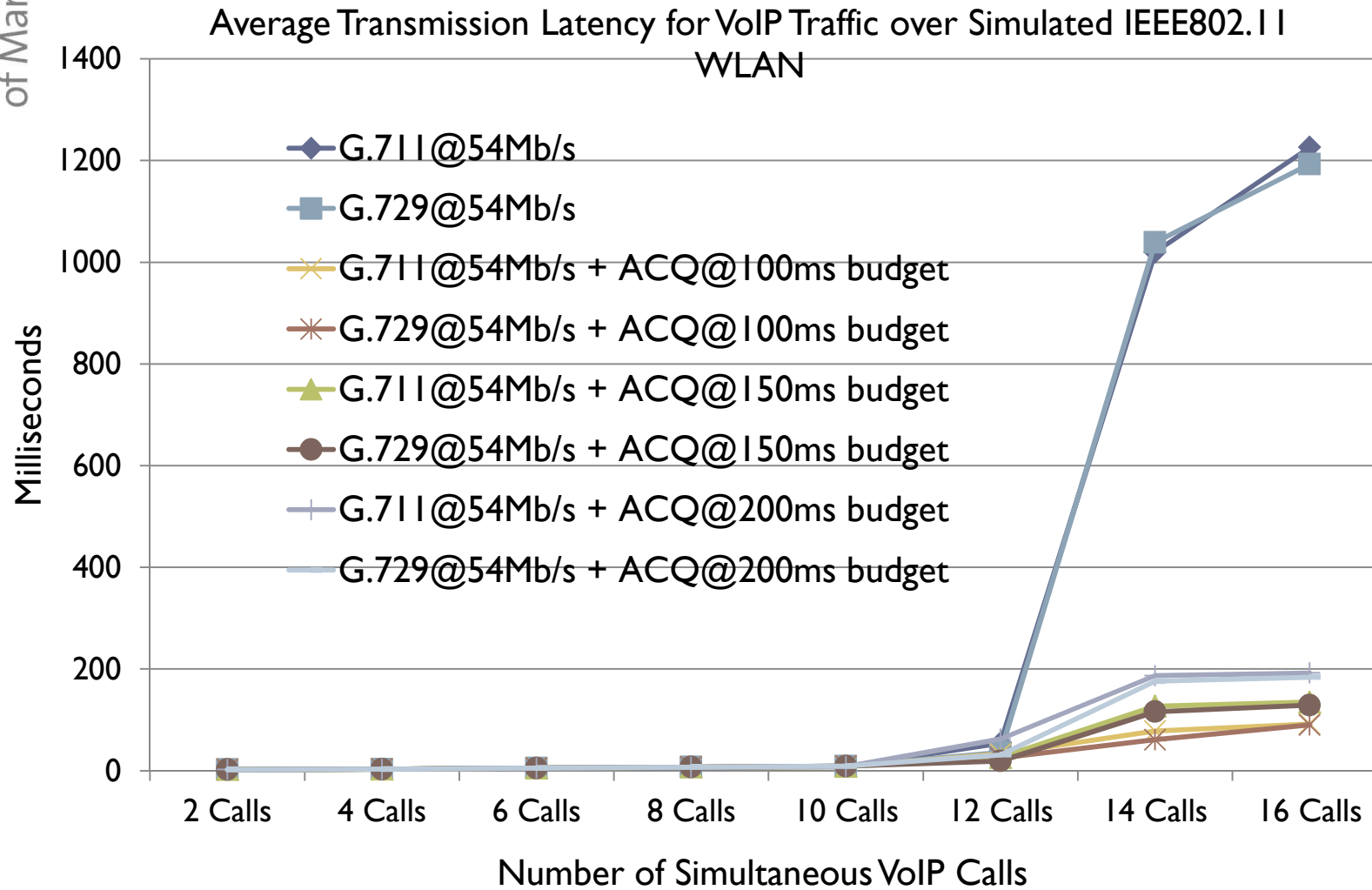


## Auto Cleaning Queue (ACQ)

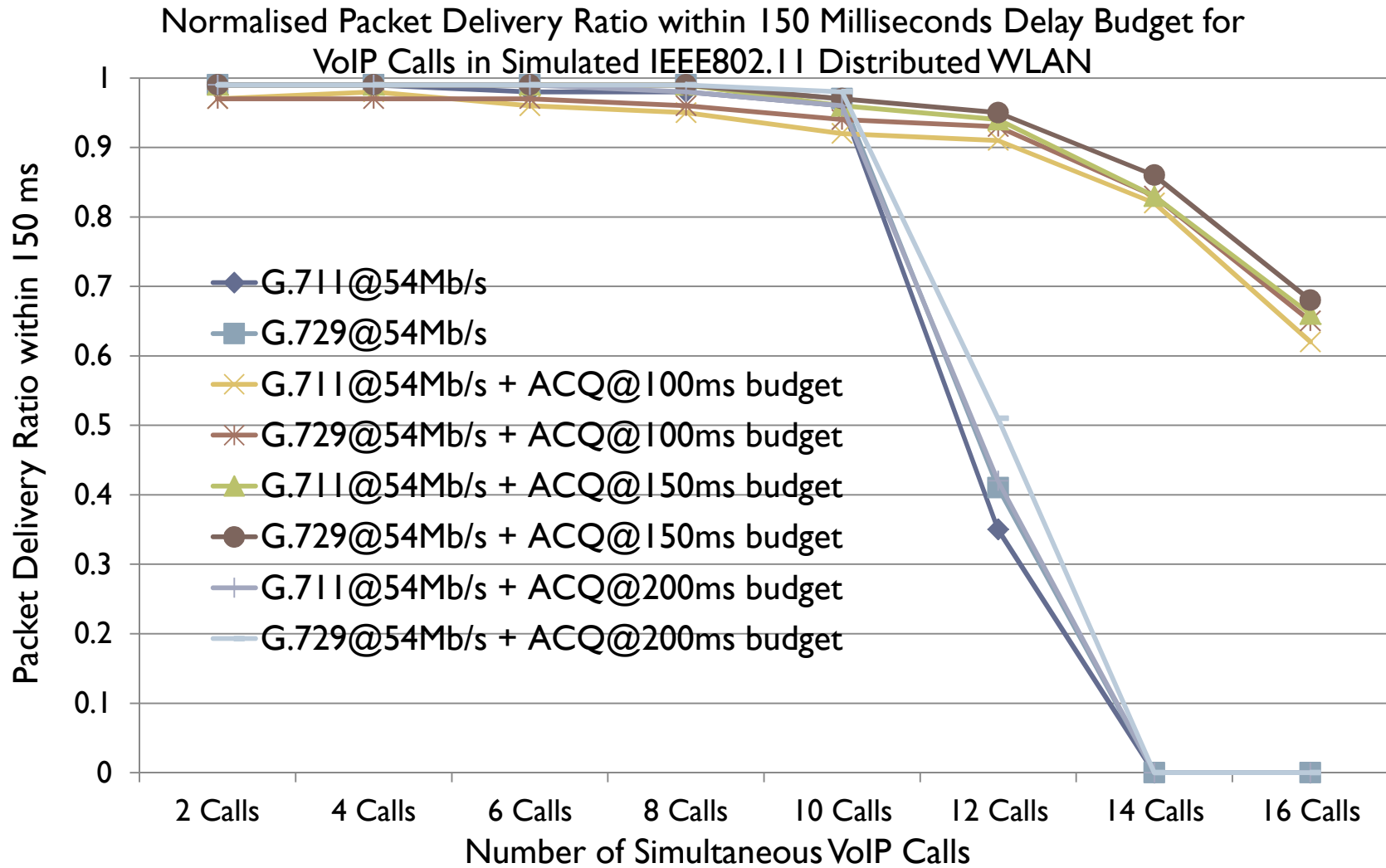
- ▶ **Stale Voice Packets in Outgoing Buffer**
  - ▶ Not possible being delivered in time
  - ▶ Receiver end already played a pseudo sample
  - ▶ Wasted transmission opportunity
- ▶ **Related Work**
  - ▶ Dynamic buffer size
  - ▶ Drop front/random drop (mainly for TCP)
- ▶ **ACQ**
  - ▶ Actively dropping voice packets older than  $T_{MAX}$
  - ▶ Giving transmission opportunity to 'fresh' voice packets
  - ▶ Compatible with IEEE802.11e Voice (AC\_VO) queue



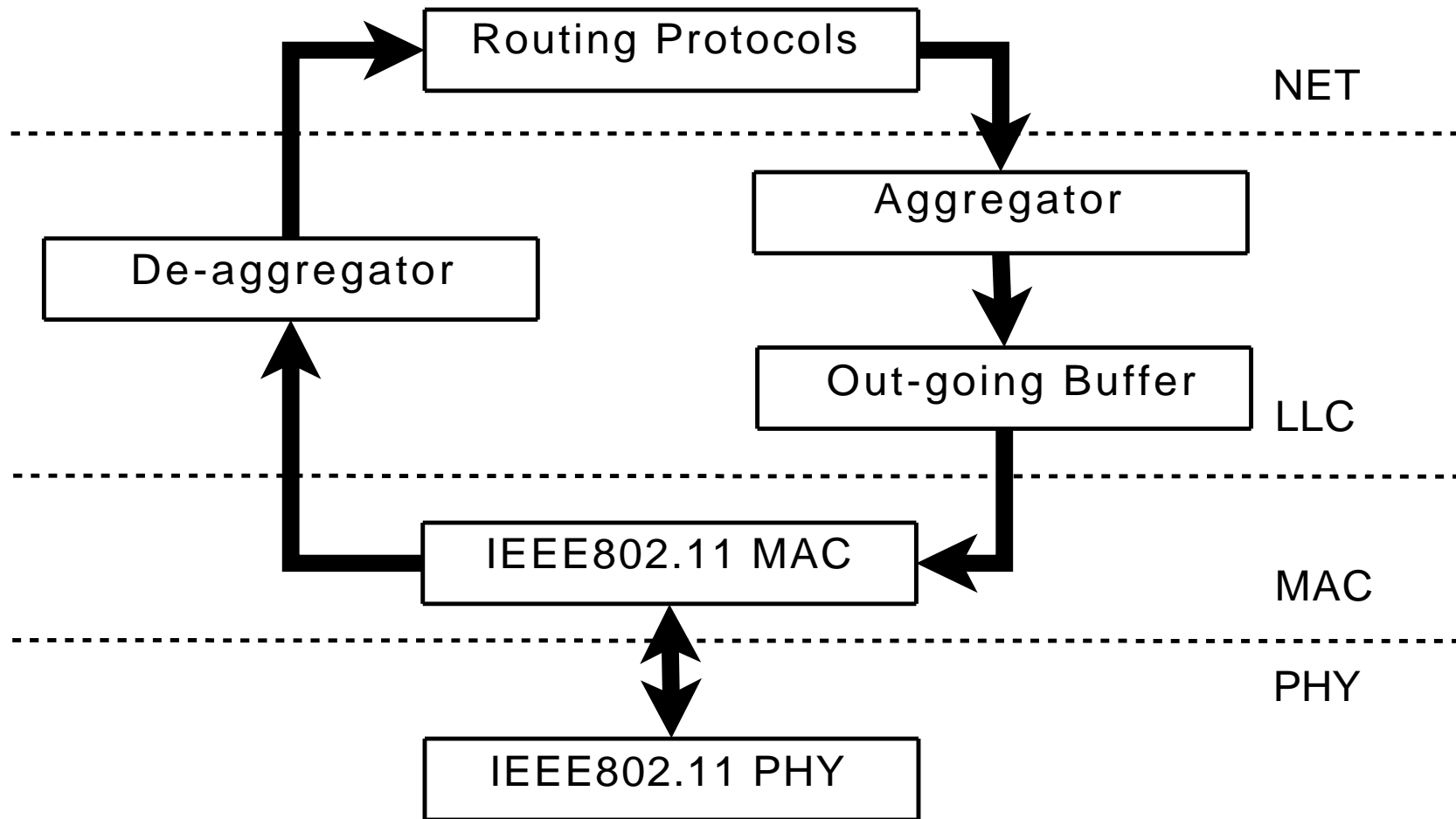
# ACQ Evaluation 1



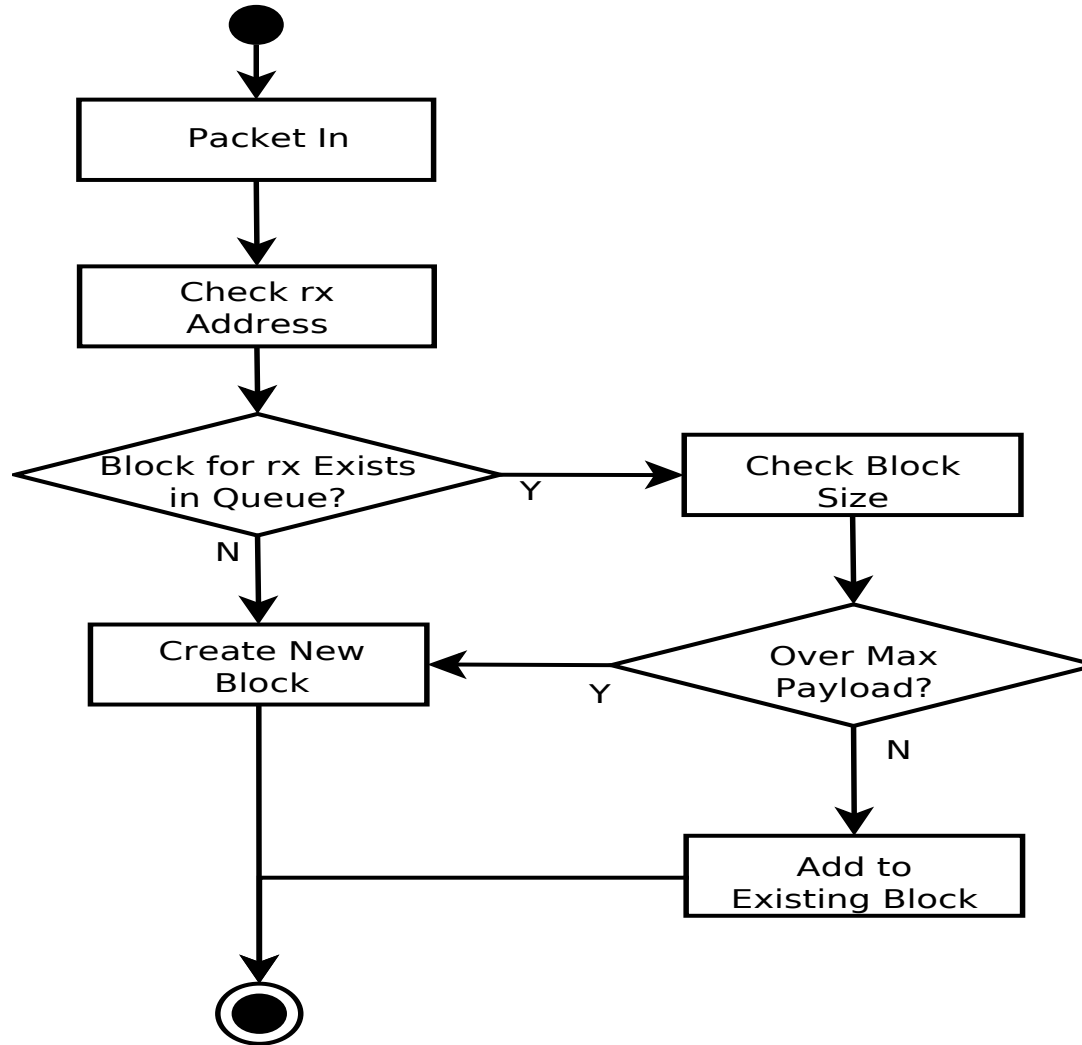
## ACQ Evaluation 2



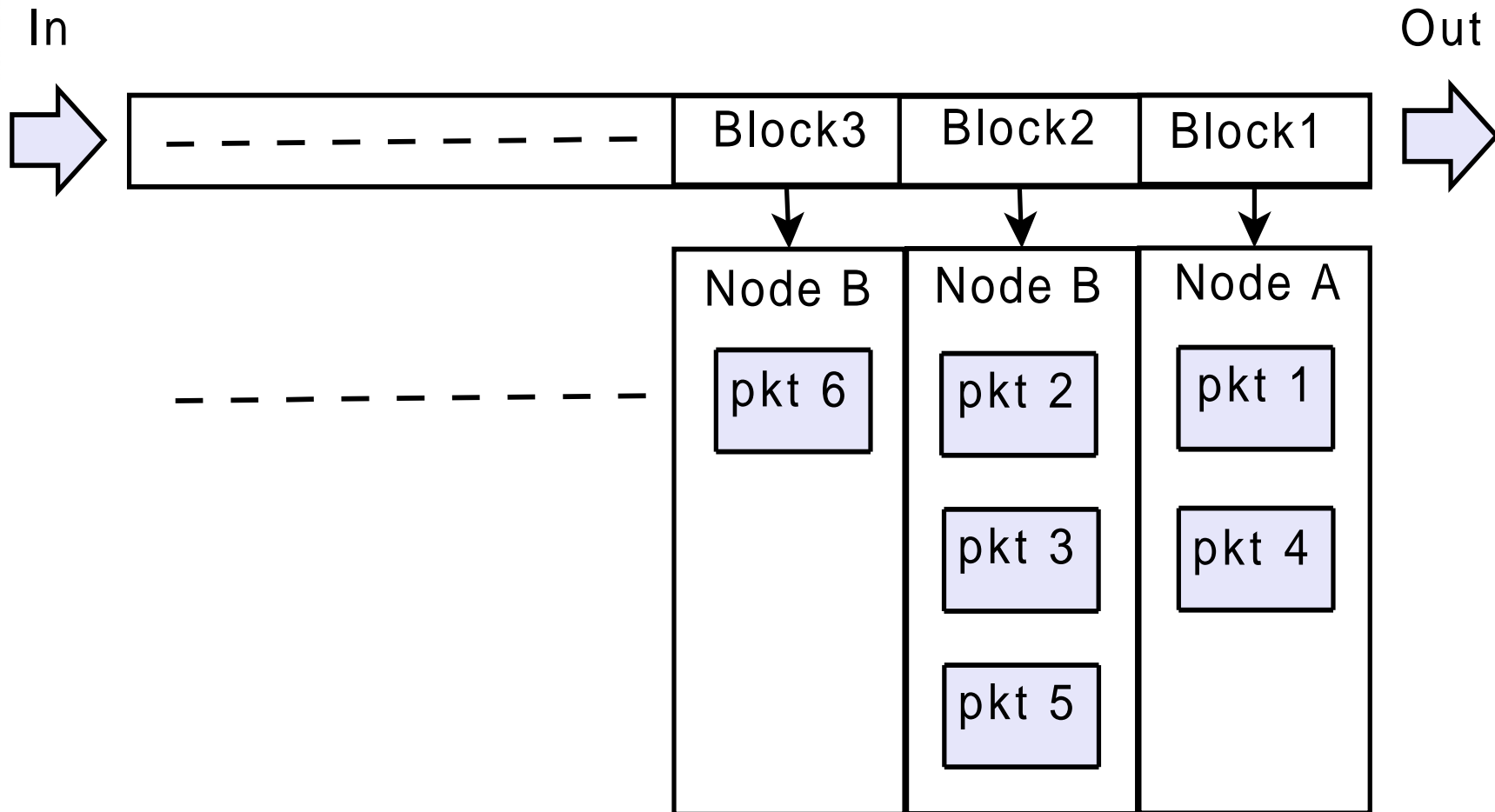
# Small Packet Aggregation for Wireless Networks



# SPAWN Packet En-queue Process



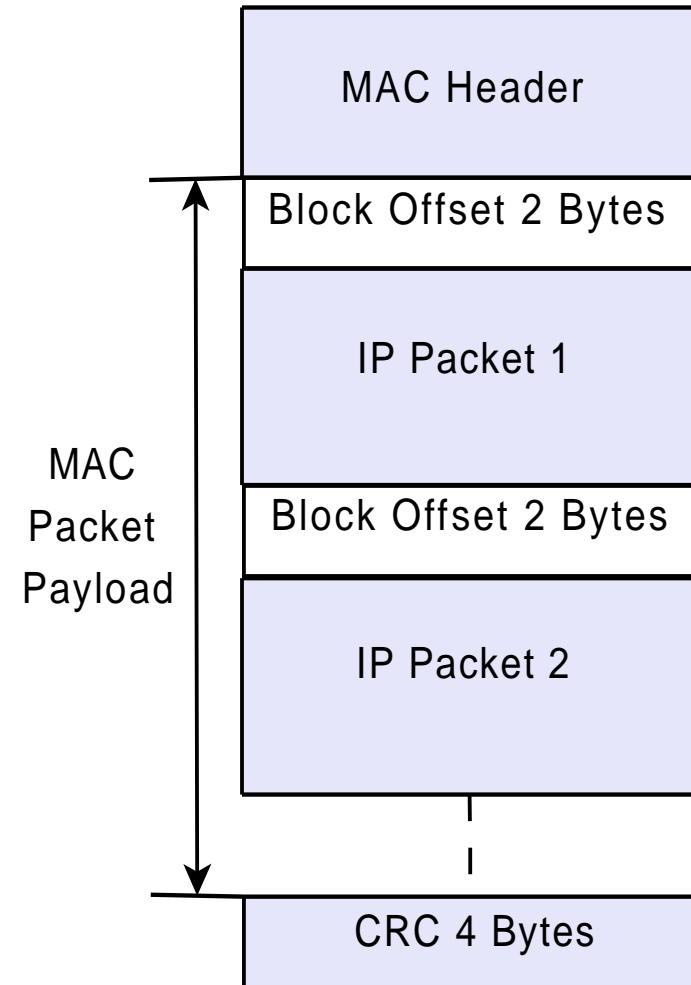
# SPAWN Outgoing Buffer Structure





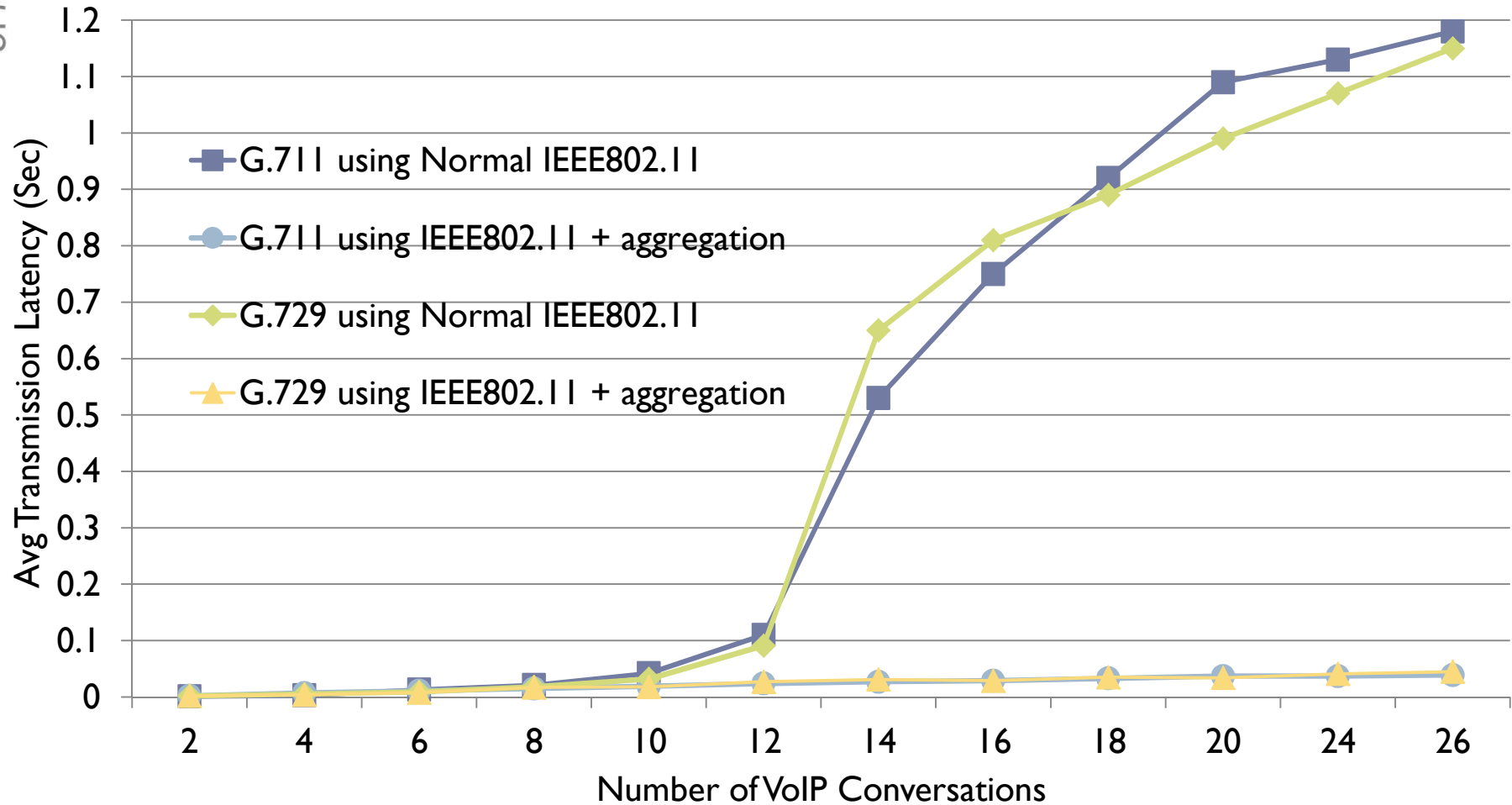
# SPAWN Block Structure

- ▶ **Compatible with IEEE802.11**
  - ▶ No modification on frame header and structure
  - ▶ No extra operations required below LLC layer
  - ▶ IEEE802.11n aggregation (A-MPDU) resulting in new MAC protocol
  
- ▶ **Flexible for Efficiency**
  - ▶ Variable number of packets
  - ▶ Not forced to reach the maximum payload size

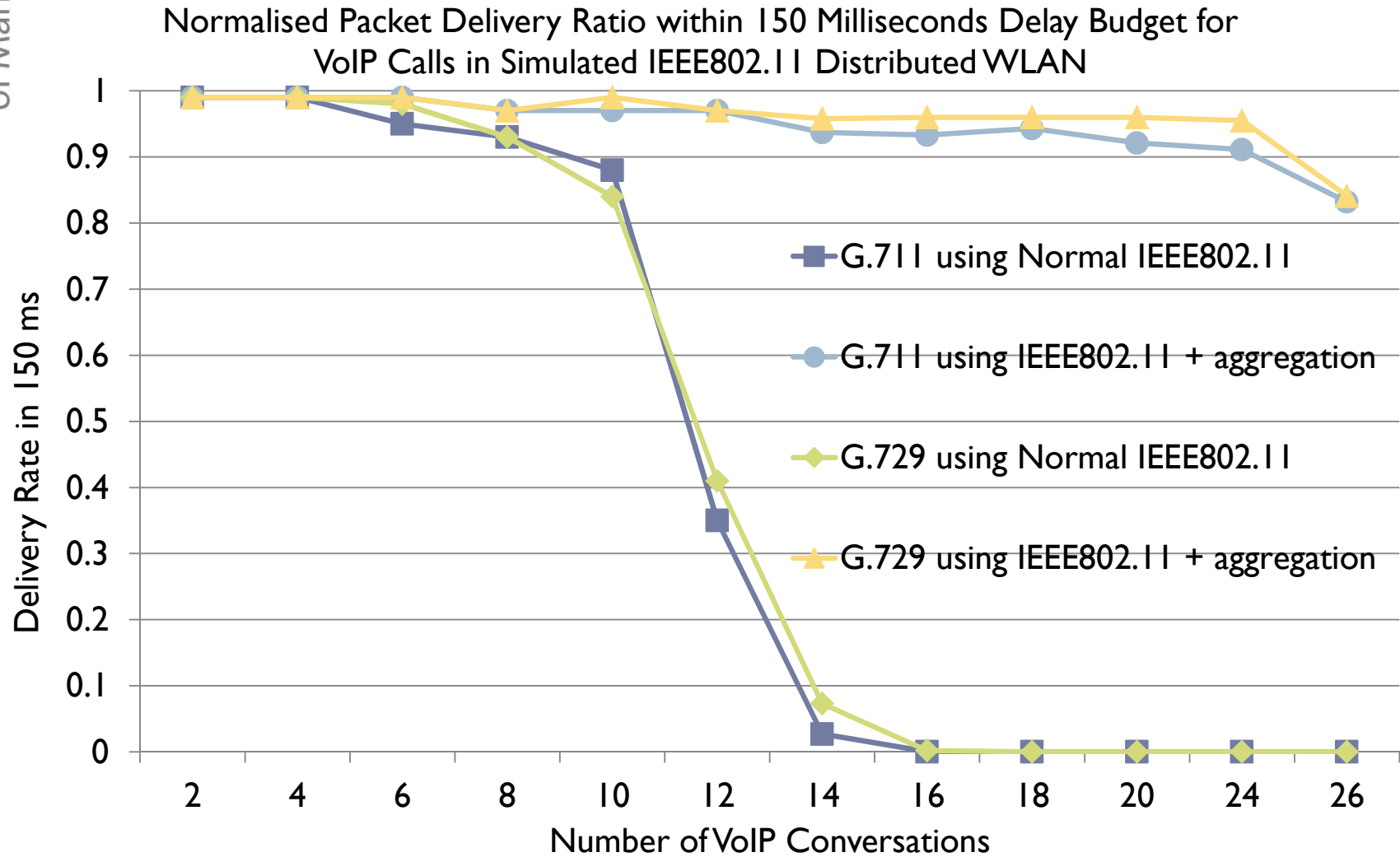


# SPAWN Evaluation 1

Average Transmission Latency for VoIP Traffic over Simulated IEEE802.11 WLAN



# SPAWN Evaluation 2



# Conclusions

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- ▶ **Outcomes**
  - ▶ Similar behaviour for VoIP over distributed and infrastructure WLANs
  - ▶ Delay accumulation after congestion
  - ▶ ACQ resolves the delay accumulation
  - ▶ SPAWN effectively increases the network efficiency
  - ▶ Compatible with IEEE802.11a/b/g MAC and IEEE802.11e
  
- ▶ **Future Work**
  - ▶ ACQ and SPAWN in real devices (seeking partners)
  - ▶ Dynamic  $T_{MAX}$  for ACQ
  - ▶ Multicasting SPAWN
  - ▶ Multi hops, multi flows, multi traffic types, etc.

Thank you!  
&  
Questions?

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