Impact of Data Rate for VoIP Quality over Direct Sequence 802.11b

Dr. Khalid Hamid Bilal Khartoum, Sudan dr.khalidbilal@hotmail.com

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Abstract

Wireless LAN connects two or more devices using Orthogonal Frequency Division Multiplexing (OFDM) or Direct Sequence Spread Spectrum (DSSS) modulation techniques to establish communication between devices within a limited range. This paper mainly aimed to study wireless local area network WLAN and analyze VoIP over WLAN and explore the evaluation of VoIP over WLAN in term of quality of service using OPNET simulator to analysis and evaluated network.

Keywords: WLAN, QoS, Data rate, Direct sequence.

1. Introduction

A wireless network is any kind of computer network that is connected wirelessly, meaning that the nodes are connected to each other or to the telecommunications network (Which connected them to the internet or backbone wired network) without the need of wires. Wireless networks use the electromagnetic waves (commonly radio waves) for carrying the signals and data between the nodes and it is implemented at the physical layer meant to replace the wires. [1] Some common types of wireless networks are as follows:

- Personal Area Network (PAN)
- Wireless Local Area Network (WLAN)
- Wireless Metropolitan Area Network (WMAN)

2. VoIP Protocol Stack

As its name implies, VoIP utilizes IP as its basic transport method. VoIP utilizes both the Transmission Control Protocol (TCP) and User Data gram Protocol (UDP) over IP.It is important to note that VoIP works with any protocols tack that supports IP .End users of VoIP can add enterprise VoIP systems to their existing infrastructure relatively quickly and easily. The widest VoIP protocols are **SIP Protocol:** As with HTTP, SIP messages can be broken into two major categories, including messages from clients to servers and messages from servers back to clients. And **RTP:** that supports user voice. Each RTP packet contains small sample of the voice conversation. The size of the packet and the size of the voice sample inside the packet will depend on the CODEC used. And **H.323:** is made up of several parts. Each part is responsible for specific tasks, such as call setup and phone registration. [2]

3. Quality of Service

Quality of Service (QoS) is the ability to measure the network's performance such that it delivers predictable results. Quality of Service helps to differentiate between the type of service required and the type of traffic and is a very important tool for VoIP services. The VoIP, quality means the ability to talk and listen clearly without any unwanted noise. The three major factors that affect the speech quality in VoIP are: delay, Jitter and MOS value. [3]

Delay: It is the amount of time it takes a signal to propagate through a copper wire or a Fibres optic and the Jitter is difference in the expected time of arrival and the actual time of arrival of the packet is called jitter.and the MOS value inNormally a observer gives each sentence a rating as follows: (1) bad; (2) poor; (3) fair; (4) good; (5) excellent by listening to the whole conversation. Many studies and research are found in performance of voice over internet protocol in wlan such as:

Author	Title of study	Туре	Place of publication	Date of
		of study		publicati on
L. Cai, Y. Xiao, X. Shen, L. Cai, and J. W. Mark	VoIP over WLAN: Voice paper capacity, admission control,		Int. J. Commun. Syst., vol. 19, no. 4, pp.	2006 [4]
	QoS, and MAC		491–508	
M. H. Miraz, S. A. Molvi, M. Ali,	Analysis of QoS of VoIP	paper	vol. I, pp. 2–7	2014[5]
M. A. Ganie, and A. H. Hussein	Traffic through WiFi- UMTS Networks			
N. M. Gambhir	Measurement of Speech	MSc		2009[6]
	Quality in VoIP over			
	Wireless LAN during			
	Handoff,			
A. Mahmood	Performance Evaluation of	MSc	Environments, no	2008[7]
	WLAN for Mutual			
	Interaction between Unicast			
	and Multicast			
	Communication Sessions			

All these conventional studies had evaluated and analysis of the performance of VoIP over Wirelesslan and conduct several of test cases in VoIP by constructing different simulation scenarios under software and concluded that VoWLAN is a promising but very challenging technology that needs more efforts to achieve potential success in the future. Some important open issues for further investigation: voice/data capacity analysis for effective admission control, voice performance analysis in QoS-enhanced 802.11e WLANs, admission control and optimal bandwidth aggregation in WLAN/cellular systems, network security and voice performance trade-off, etc.

4. Simulation

The network model was built and configured using OPNET simulator, and the simulation was run in two scenarios by using direct sequence 802.11b technology by using different data rate. The network model is shown in fig (1):

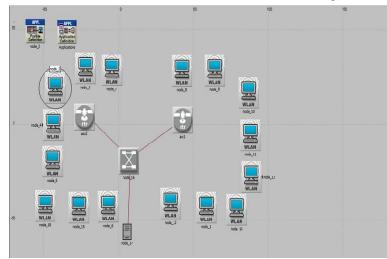


Fig (1): The Simulation Network Model

Table (1): Simulation Environments

Numbers of nodes	16 nodes		
Network scale	Office		
Specify size	100*100 m2		
Technology	802.11b direct sequence		
Data rate	(11, 5.5) Mbps		
Link model	100BaseT full duplex		
Application	Voice over IP call (PCM Quality)		
Duration of simulation	120 second		

5. Results

After running the simulation we get the flowing results for different scenario as showing below:

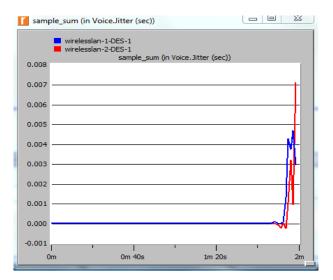


Fig (2): Voice Jitter

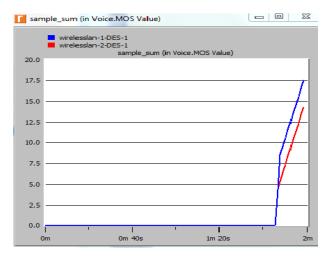


Fig (3): MOS Value

🚺 sample_	sum (in Vo	ice.Packet	Delay Vari	ation)		
_	wirelesslan wirelesslan					
0.070 -	sa	mple_sum (in Voice.Pa	cket Delay ∖	/ariation)	
0.065 -						
0.060 -						
0.055 -						
0.050 -						
0.045 -						
0.040 -						
0.035 -						
0.030 -						
0.025 -						
0.020 -						
0.015 -						
0.010 -						
0.005 -						
0.000	0m 20s	0m 40s	1m Os	1 m 20 s	1m 40s	200.00
Om Os	Um 20S	om 40s	1m Us	1m 20s	1m 40s	2m 0s

Fig (4): Voice Packet Delay

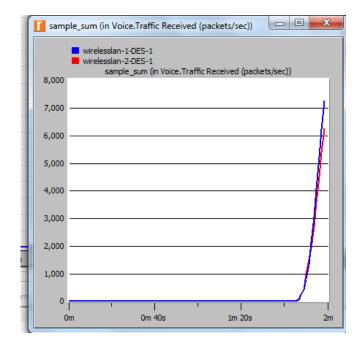


Fig (5): Voice Traffic Received

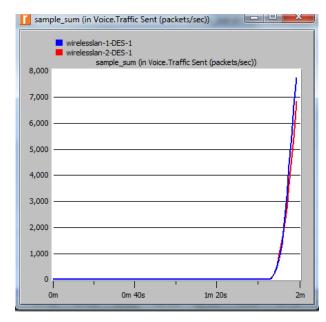


Fig (6): Voice Traffic Sent

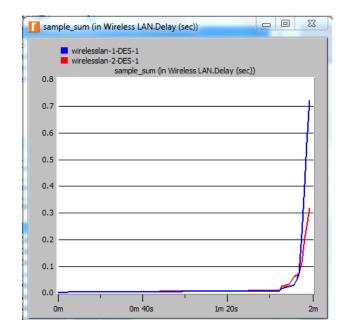


Fig (7): Wireless LAN Delay

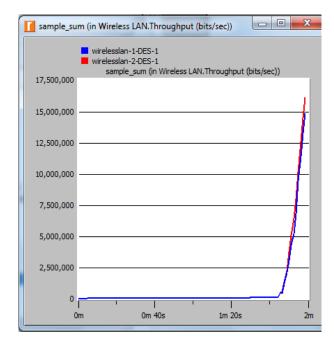


Fig (8): Wireless LAN Throughput

6. Results Discussion

From all previous results, when used data rate11Mpbs and no. of node 16 and technology 802.11b direct sequence in 120 sec and Data rate5.5Mpbs and no of node 16 and technology 802.11b direct sequence in 120 sec.

The result of QoS VS simulation time is:

- If the data rate increases the jitter decreases.
- If the data rate increases the MOS increases.
- If the data rate increases the packet delay increases.
- If the data rate increases the traffic receive increases.
- If the data rate increases the traffic sent increases.
- If the data rate increases the delay increases.
- If the data rate increases the throughput decreases.

7. Conclusion

From the previous scenarios and subsequent of analysis, we conclude that the possible performance quality of VoIP associated with the datarate. If MOS, traffic sent, traffic received and jitter are taken to be the most prioritized QoS factors, we must used high value of datarate over other datarate.

Reference

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