

ANALYSIS OF EFFECT OF ZFONE SECURITY ON VIDEO CALL SERVICE IN WIRELESS LOCAL AREA NETWORK

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Abstract—Along with the development of WLAN (Wireless Local Area Network) network technology, many services that used cable networks began to migrate to wireless networks. VoIP (Voice over Internet Protocol) is one service implemented in wireless local area networks. However, VoIP that uses wireless technology as a data stream media Video Call service has a high risk of tapping pictures. To avoid tapping pictures, you can add a security system to the service, one of which is to use Zfone security. With him adding a security system that will influence the work of Video Call services on the quality of service. The author uses the General Network Design Process (GNDP) study method. After testing security by intercepting images, Zfone affects the security of video calls on WLAN by changing conversations between client images to black. Although Zfone secured from eavesdropping on video calls, Zfone also impacted the deterioration of the quality of Video Call services. The results of quality of service before using the Zfone security system with the parameters of delay, jitter, and packet loss are as follows; the delay time value is 68.93ms, the jitter time value is 2.04ms and the packet loss parameter before using Zfone security is 0%. The results of quality of service after using the Zfone security system with the parameters of delay, jitter, and packet loss are as follows; the value of delay time is 80.12ms and the value of jitter time is 4.36ms on packet loss parameters after using Zfone security which is 0%.

Keywords—VoIP, Zfone, Network Security, Quality Of Service, GNDP.

I. INTRODUCTION

The development of computer networks at this time which is rapidly making a big influence on the network technology itself, the current network technology that allows one to communicate with each other long-distance is like a necessity that can not be separated. One of the technologies used today is VoIP. Voice over Internet Protocol (VoIP) is a technology that allows long-distance voice conversations through internet media. In addition to conducting long-distance conversations over the internet, VoIP has a service that can carry out chat messages and even make video calls.

VoIP can also be implemented on a Local Area Network (LAN) or abbreviated as VoIP LAN. VoIP LAN networks are usually implemented on cable networks and combined with Public Switch Telephone Network (PSTN) networks. The development of network technology, the existence of

VoIP LAN that uses cable began to move using wireless network technology, namely wireless. The use of wireless networks as a medium for the flow of data makes VoIP services can be used in several communication technology tools such as personal computers, laptops, and smartphones.

The use of VoIP technology that is implemented on WLAN networks with video call services found in VoIP facilities is an excellent solution for conducting conversations at affordable costs compared to the VoIP services associated with PSTN. The use of VoIP technology with video call services is very beneficial for the user, but communication that is so efficient and affordable costs in terms of security is less paid attention. Therefore, when making a video call service that is connected to a WLAN, it is still very vulnerable in terms of security because the network works by spreading radio frequency signals, allowing unauthorized parties to enter the network and intercept the ongoing communication. This needs to be addressed because it is a very important matter concerning user privacy.

Based on this, what can be done is to add a security system to the WLAN network. By using Zfone software that uses the Zimmermann Real-Time Transport Protocol (ZRTP) protocol as its executor. The use of Zfone is an alternative method for security by encrypting the data traffic of a WLAN. By implementing an additional security system using Zfone software, the data packet must go through several security stages before being sent and arrive at its destination.

II. RESEARCH METHOD

In this study the author uses the General Network Design Process (GNDP) method with 6 stages as follows:

1. Assess needs and costs

The initial stages of the General Network Design Process (GNDP) there are two main things as follows:

a. Assessing Needs

This stage explains what is needed by the user by doing this research. The elements of availability that must be achieved in this research are video call

service security and Quality of Service WLAN network system.

- b. This stage is the stage for the budget required in this research both from hardware (hardware) and software (software)

2. Choosing topology and technology

This stage is the stage for the selection of topology and technology used in this research. In this study using the star topology. The wireless network in this study uses infrastructure mode, presented in Figure 1.

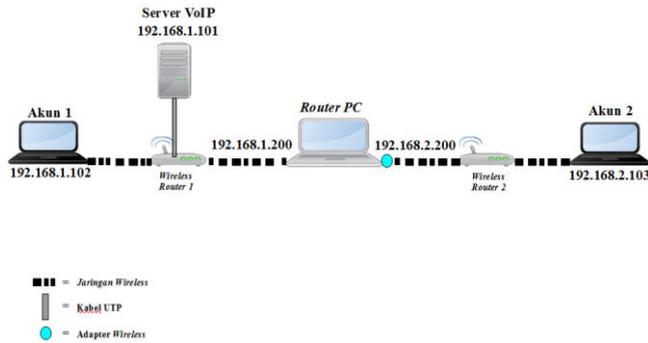


Fig 1. VoIP Network Topology Using WLAN

3. Workload model

This stage is determined by what it charges to the network researchers have created that. In this study, they charge two systems the VoIP system using WLAN and VoIP using WLAN.

4. Simulates behavior under expected load

This stage simulates a network system that has been determined at the workload model stage. At this stage, I also measure the Quality of Service against the VoIP system using WLAN.

5. Do a sensitivity test

At this stage, it implements the security system on a VoIP system using WLAN testing. The security used on this system is data encryption; the authors use Zfone software. Zfone uses the ZRTP protocol to encrypt data on the transport protocol used by the VoIP system using WLAN.

6. Process design as needed

At this stage, it results from the analysis carried out in the previous stages. The stage also redesigned if needed to get optimal results.

III. TESTING RESULTS

1. Call flow process

At this stage, we explain it about the process of a video call from the beginning of the video call until the end of the video call. Following result from testing the call flow process on a VoIPWLAN system without security and a VoIPWLAN system with Security:

a. VoIP without security

Following results from the process flow of the VoIPWLAN system without security presented in Figure 2.

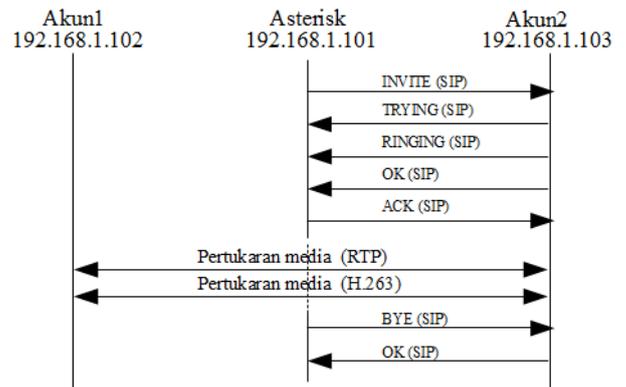
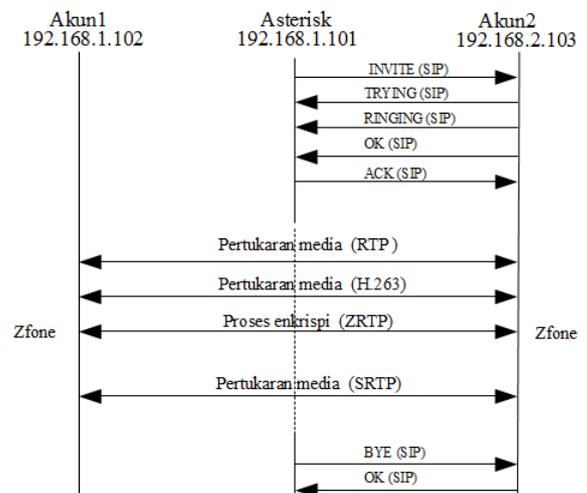


Fig 2. illustrates the process of initiating a video call flow

b. VoIPWLAN with security

The following result from the call flow process on the VoIPWLAN + ZRTP system presented in Figure 3.



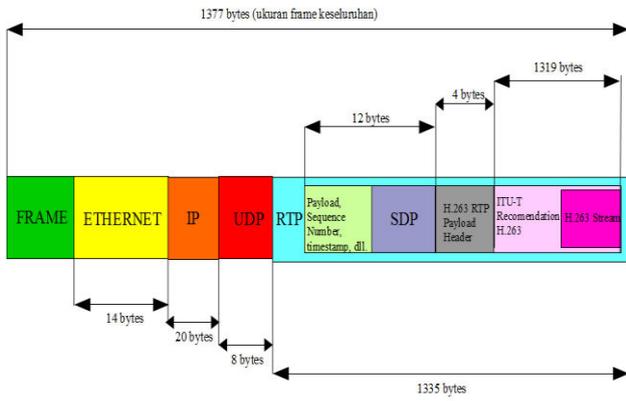
2. Transport protocol type

Based on the transport protocol used by the VoIPWLAN system before and after using security.

a. VoIPWLAN without security

The following results of testing the transport protocol VoIPWLAN system without security, are presented in Figure 4 and we present the frame shape that carries the transport protocol in Figure 5.

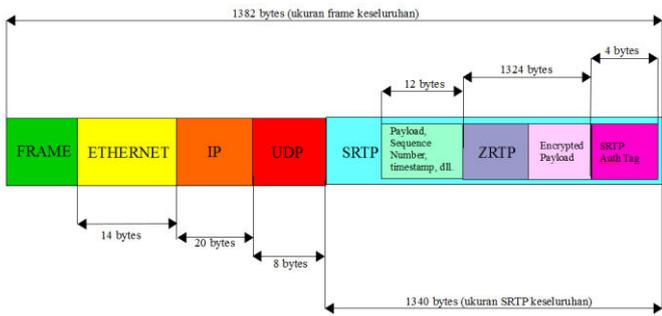
Time	Source	Destination	Protocol	Length	Info
56.24.3248050	192.168.0.102	192.168.0.103	SIP	1088	Request: INVITE sip:100208192.168.0.103
57.21.8300620	192.168.0.103	192.168.0.102	SIP	595	Status: 401 Unauthorized
58.21.9105700	192.168.0.102	192.168.0.103	SIP	310	Request: ACK sip:100208192.168.0.103
59.21.8973100	192.168.0.102	192.168.0.103	SIP	1259	Request: INVITE sip:100208192.168.0.103
60.21.8422950	192.168.0.102	192.168.0.103	SIP	538	Status: 100 Trying
63.21.9790700	192.168.0.102	192.168.0.103	SIP	538	Status: 100 Trying
67.24.1596700	192.168.0.102	192.168.0.103	SIP	1150	Status: 200 OK
69.24.4149110	192.168.0.102	192.168.0.103	SIP	464	Request: ACK sip:100208192.168.0.103:5060
70.24.41616300	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=0, TTime=379020745, Mark
71.24.4180640	192.168.0.103	192.168.0.102	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=4032, TTime=377848040, Mark
72.24.4484050	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=1, TTime=379020905
73.24.4483910	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=2, TTime=377848200
74.24.4487800	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=2, TTime=379020705
75.24.4789840	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=4034, TTime=377848300
76.24.4987210	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=1, TTime=379021215
77.24.4981490	192.168.0.102	192.168.0.103	RTP	1377	PT=ITU-T H.263, SSRC=0x4F18E178, Seq=29864, TTime=MODE A H263 payload (PSC) P-frame
78.24.4981500	192.168.0.102	192.168.0.103	H.263	1471	PT=ITU-T H.263, SSRC=0x4F18E178, Seq=29865, TTime=MODE A H263 payload (GRC)
79.24.4981610	192.168.0.102	192.168.0.103	H.263	703	PT=ITU-T H.263, SSRC=0x4F18E178, Seq=29866, TTime=, H.263 payload (GRC)
80.24.5018710	192.168.0.102	192.168.0.103	H.263	1447	PT=ITU-T H.263, SSRC=0x18156683, Seq=0, TTime=234091890, H.263 payload (PSC) I-frame
81.24.5020960	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x18156683, Seq=4035, TTime=377848520
82.24.5020820	192.168.0.102	192.168.0.103	H.263	1139	PT=ITU-T H.263, SSRC=0x18156683, Seq=1, TTime=234091890, H.263 payload (GRC)
83.24.5033980	192.168.0.102	192.168.0.103	H.263	1359	PT=ITU-T H.263, SSRC=0x18156683, Seq=2, TTime=234091890, H.263 payload (GRC)
84.24.5040700	192.168.0.102	192.168.0.103	H.263	800	PT=ITU-T H.263, SSRC=0x18156683, Seq=1, TTime=234091890, H.263 payload (GRC)
85.24.5031020	192.168.0.102	192.168.0.103	RTP	214	PT=ITU-T G.711 PCMU, SSRC=0x1AC34C3D, Seq=1, TTime=379021405



b. VoIPWLAN with security

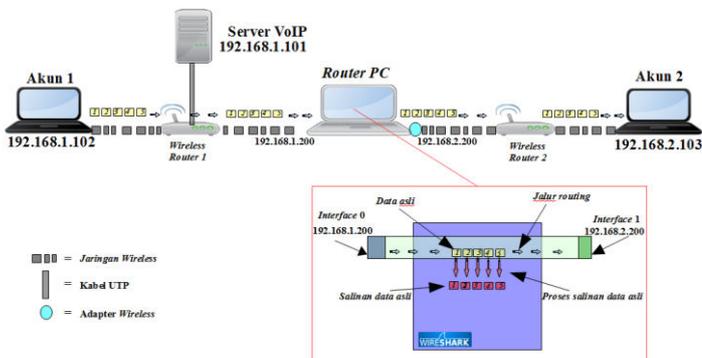
The following results of testing the transport protocol, VoIPWLAN system with security, are presented in Figure 7 and I present the frame shape that carries the transport protocol in Figure 8.

No.	Time	Source	Destination	Protocol	Length	Info
169	14.331039	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU
170	14.331040	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
171	14.331041	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU
172	14.331042	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
173	14.331043	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU
174	14.331044	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
175	14.331045	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU
176	14.411024	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU
177	14.411025	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
178	14.413716	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU
179	14.421235	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
180	14.469298	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
181	14.483563	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
182	14.483563	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
183	14.502752	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
184	14.502752	192.168.2.103	192.168.1.102	SRTP	218	PT=ITU-T G.711 PCMU
185	14.518012	192.168.1.102	192.168.2.103	SRTP	218	PT=ITU-T G.711 PCMU



3. Video tapping results

Tapping is done by stealing data packets that become transport protocols on VoIP systems using WLAN. The transport protocol carries data that contains sound and image media that are generated from conversations conducted by the client. To do this, the writer uses Wireshark and PC router as the executor of tapping, as shown in Figure 9.



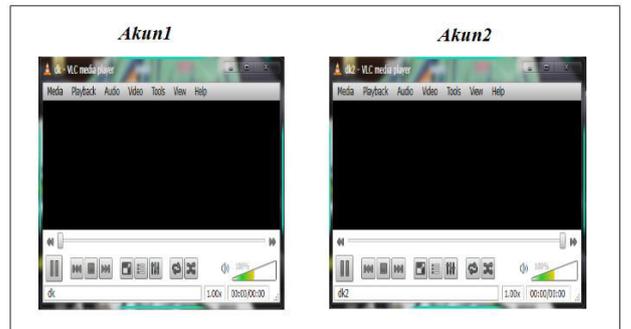
a. VoIPWLAN without security

Following result from tapping video calls on a VoIPWLAN system without security, presented in Figure 10.



b. VoIPWLAN with ZRTP

The following result from tapping video calls on a VoIPWLAN system with security, presented in Figure 11.



4. Quality of Service Results

I carry quality of Service testing out five times, testing is done by making calls to account1 to account2. The parameters used are delay, jitter and packet loss. They carry the test out on a VoIPWLAN system without security and a VoIPWLAN system with security[1]

a. Delay

The following is the delay value in the VoIPWLAN system without security and with security, presented in table 1.

No	Testing	VoIPWLAN	VoIPWLAN+Zfone
1	Ke-1	72,59	78,77
2	Ke-2	74,07	81,48
3	Ke-3	100,7	83,64
4	Ke-4	71,73	73,41
5	Ke-5	25,58	83,29
Average		68,93	80,12

b. Jitter

Following is the value of jitter on a VoIPWLAN system without security and with security, presented in table 2.

No	Testing	VoIPWLAN	VoIPWLAN+Zfone
1	Ke-1	1,75	2,50
2	Ke-2	2,43	1,81
3	Ke-3	2,58	8,07
4	Ke-4	2,44	5,74
5	Ke-5	0,99	3,67
Average		2,04	4,36

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c. Packet Loss

Following is the value of packet loss on a VoIPWLAN system without security and with security, presented in table 3.

No	Testing	VoIPWLAN	VoIPWLAN+Zfone
1	Ke-1	0	0
2	Ke-2	0	0
3	Ke-3	0	0
4	Ke-4	0	0
5	Ke-5	0	0
Average		0	0

CONCLUSIONS

Based on the research that has been done, the following conclusions can be drawn:

- Zfone's security system affects securing video calls on WLAN. When video tapping is done, the ZRTP protocol turns the image of conversations between clients black.
- The results of the quality of service before using the Zfone security system with the parameters of delay, jitter, and packet loss are as follows:
 - The value of the delay time is 68.93ms
 - The jitter time value is 2.04ms
 - The packet loss parameter before using Zfone security is 0%.
- The results of quality of service after using the Zfone security system with the parameters delay, jitter and packet loss are as follows:
 - The value of the delay time is 80.12ms
 - The jitter time value is 4.36ms
 - The packet loss parameter after using Zfone security is 0%..

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