<u>Uji Kelayakan Jaringan Lokal Universitas Jenderal Soedirman Untuk Implementasi VoIP</u>

Title	Uji Kelayakan Jaringan Lokal Universitas Jenderal Soedirman Untuk Implementasi VoIP
Author Order	of
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Abstract	UNSOED has LAN connection which covers almost all of its units, with minimum bandwidth consupmtionat anaverage $\tilde{A}f$, \tilde{A} , \tilde{A} of 275,318 Kbpsfrom a total of 3 Mbpsso we may apply VoIP system in order to replace voice communication system whichis still $\tilde{A}f$, \tilde{A} , \tilde{A} using PSTN and PABX. Thereforewe need to know about jitter and packet loss of LAN, and also the codec needs to be determined $\tilde{A}f$, \tilde{A} , \tilde{A} due to voice quality expected and also bandwidth consumpted. Measurements carried out in two scenarios based on topologies used $\tilde{A}f$, $\tilde{A}f$, in UNSOED (star and ring), in the first scenario (star and ring) server placed on server farm meanwhile the second scenario (ring) $\tilde{A}f$, $\tilde{A}f$, $\tilde{A}f$, server placed on Pertanian node. Codec adjusted to the available bandwidth capacity and total of phone number that has to handled. $\tilde{A}f$, $\tilde{A}f$, $\tilde{A}f$. The result show that 0,759 ms jitter obtained from scenario 1 and 0,513ms from scenario 2, both of them included in good category $\tilde{A}f$, $\tilde{A}f$, $\tilde{A}f$, with 0-20 ms of range value, otherwise 0% of packetloss obtained from both scenarios these indicated that LANincluded in good $\tilde{A}f$, $\tilde{A}f$, $\tilde{A}f$ category. There is iLBCcodec with 30 ms packet size which is the mosteconomic on bandwidth cost among G.729, G.711, and iLBC $\tilde{A}f$, $\tilde{A}f$, $\tilde{A}f$, $\tilde{A}f$, because it cost1995,84Kbpsof total bandwidth usedforhandling 140 phone number. Beside that, it has good voice quality $\tilde{A}f$, $\tilde{A}f$, with 4,138996 MOS value. So if we use iLBC (30 ms) codec on the VoIPsystem, there will be still a lot of free bandwidth which can $\tilde{A}f$, $\tilde{A}f$, aused for other applications, it is about 695,873 Kbps(worst condition) and also get a good-voice quality-VoIP.
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