

Daftar Pustaka

- [1] IETF, “rtcweb” .[Online]. Available: <https://datatracker.ietf.org/wg/rtcweb/charter/> Accessed March 29, 2019.
- [2] Atwah, R., Iqbal, R., Shirmohammadi, S., & Javadtalab, A. (2015). A Dynamic Alpha Congestion Controller for WebRTC. 2015 IEEE International Symposium on Multimedia (ISM). doi:10.1109/ism.2015.63
- [3] Haensge, K., & Maruschke, M. (2015). 2015 18th International Conference on Intelligence in Next Generation Networks. doi:10.1109/icin.2015.7073800.
- [4] Ayadi, I., Diaz, G., & Simoni, N. (2013). QoS-based network virtualization to future networks: An approach based on network constraints. 2013 Fourth International Conference on the Network of the Future (NoF). doi:10.1109/nof.2013.6724515.
- [5] Barayuga, V. J. D., & Yu, W. E. S. (2014). Study of Packet Level UDP Performance of NAT44, NAT64 and IPv6 Using Iperf in the Context of IPv6 Migration. 2014 International Conference on IT Convergence and Security (ICITCS). doi:10.1109/icitcs.2014.7021814.
- [6] García, B., Gallego, M., Gortázar, F., & Bertolino, A. (2018). Understanding and estimating quality of experience in WebRTC applications. *Computing*. doi:10.1007/s00607-018-0669-7
- [7] Suci, G., Stefanescu, S., Beceanu, C., & Ceaparu, M. (2020). WebRTC role in real-time communication and video conferencing. 2020 Global Internet of Things Summit (GIoTS). doi:10.1109/giots49054.2020.91196
- [8] S.Kuribayashi, “Optimal Joint Multiple Resource Allocation Method for Cloud Computing Environments,” *International Journal of Research and Reviews in Computer Science (IJRRCS)*, Vol. 2, No.1, pp.1-8, Feb. 2011.
- [9] S.Tsumura and S.Kuribayashi: “Simultaneous allocation of multiple resources for computer communications networks,” In *Proceeding of 12th Asia-Pacific Conference on Communications (APCC2006)*, 2F-4, Aug. 2006
- [10] M.Gusat, R.Birke and C.Minkenberg, “Delay-based cloud congestion control,” In *Proceeding of GLOBECOM’2009*, Nov. 2009.
- [11] B.Raghavan, K.Vishwanath, S.Ramabhadran, K.Yocum and A.C.Snoeren, “Cloud control with distributed rate limiting,” In *Proceeding of SIGCOMM’07*, Aug. 2007.
- [12] A. Sergiienko, Sergiienko, “WebRTC Blueprints, Develop your very own media applications and services using WebRTC,” in *Birmingham*, Packt Publishing, ISBN 978-1-78398-310-0, May 2014.A. Zeidan, A. Lehmann, and U. Trick, “WebRTC enabled multimedia conferencing and collaboration solution,” *WTC 2014*, in Berlin, Germany, Print ISBN 978-3-8007-3602-7, June 2014.
- [13] Pimentel, V., & Nickerson, B. G. (2012). Communicating and Displaying Real-Time Data with WebSocket. *IEEE Internet Computing*, 16(4), 45–53. doi:10.1109/mic.2012.64
- [14] Sredojević, B., Samardžija, D., & Posarac, D. (2015). WebRTC technology overview and signaling solution design and implementation. 2015 38th International Convention on Information and Communication Technology, Electronics and Microelectronics (MIPRO). doi:10.1109/mipro.2015.7160422
- [15] Tilkov, S., & Vinoski, S. (2010). Node.js: Using JavaScript to Build High-Performance Network Programs. *IEEE Internet Computing*, 14(6), 80–83. doi:10.1109/mic.2010.145
- [16] IPERF, “Networking with iperf” .[Online]. Available: <http://openmaniak.com/iperf.php>. Accessed December 29, 2013
- [17] Chodorek, A., Chodorek, R. R., & Wajda, K. (2019). An Analysis of Sender-driven WebRTC Congestion Control Coexisting with QoS Assurance Applied in IEEE 802.11 Wireless LAN. 2019 International Conference on Software, Telecommunications and Computer Networks (SoftCOM). doi:10.23919/softcom.2019.8903749
- [18] Chodorek, R. R., Chodorek, A., Rzym, G., & Wajda, K. (2017). A Comparison of QoS Parameters of WebRTC Videoconference with Conference Bridge Placed in Private and Public Cloud. 2017 IEEE 26th International Conference on Enabling Technologies: Infrastructure for Collaborative Enterprises
- [19] Alimudin, A., & Muhammad, A. F. (2018). Online Video Conference System Using WebRTC Technology for Distance Learning Support. 2018 International Electronics Symposium on Knowledge Creation and Intelligent Computing (IES-KCIC). doi:10.1109/kcic.2018.8628568
- [20] Varma, M., Yarnagula, H. K., & Tamarapalli, V. (2017). WebRTC-based peer assisted framework for HTTP live streaming. 2017 9th International Conference on Communication Systems and Networks (COMSNETS). doi:10.1109/comsnets.2017.7945420
- [21] H. Schulzrinne, S. Casner, R. Frederick, and V. Jacobson, “RTP: A transport protocol for real-time applications,” *RFC 3550*, 2003.

- [22] Zhang Yamei, & Cai Pengfei. (2010). Research on using UDP to traverse NAT under P2P network environment. 2010 3rd International Conference on Advanced Computer Theory and Engineering(ICACTE). doi:10.1109/icacte.2010.5579841