10.5120/18593-9835

{bibtex}pxc3899835.bib{/bibtex}

Abstract

Uma B. V.

In mobile devices, perceived speech signal degrades significantly in the presence of background noise as it reaches directly at the listener's ears. There is a need to improve the intelligibility and quality of the received speech signal in noisy environments by incorporating speech enhancement algorithms. This paper focuses on speech enhancement method including auditory masking properties of the human ear to improve the intelligibility and quality of the speech signal in the presence of near-end noise. Implemented by dynamically enhancing the speech signal when the near-end noise dominates. Intelligibility and quality of enhanced speech signal are measured using SII and PESQ. Experimental results show improvement in the intelligibility and quality of the enhanced speech signal with the proposed approach over the unprocessed speech signal. This particular approach is far more efficient in overcoming the degradation of speech signals in noisy environments.

Refer

ences

- Premananda B. S., and Uma B. V., "Speech Enhancement Algorithm to Reduce the Effect of Background Noise in Mobile Phones", International Journal of Wireless and

Mobile Networks (IJWMN), Vol. 5, No. 1, pp. 177 - 189, Feb. 2013.

- Premananda B. S., and Uma B. V., "Low Complexity Speech Enhancement Algorithm for Improved Perception in Mobile Devices", International Workshop on Wireless and Mobile Networks, WiMoNe-2012, Lecture Notes in Electrical Engineering, Vol. 131, Springer, pp. 699 707, Feb. 2013.
- Premananda B. S., Manoj and Uma B. V., "Near-End Perception Enhancement using Dominant Frequency Extraction", International Journal of Advanced engineering and Research Development (IJAERD), Vol. 1, No. 6, pp. 351-358, June 2014.
- Bastian Sauert and Peter Vary, " Near End Listening Enhancement Considering Thermal Limit of Mobile Phone Loudspeakers, " Proceedings of Elektronische Sprach Signal Verarbeitun (ESSV), Vol. 61, Germany, pp. 333–340, Sept. 2011.
- Bastian Sauert and Peter Vary, " Near End Listening Enhancement: Speech Intelligibility Improvement in Noisy Environments ", Proceedings of IEEE International Conference on Acoustics, Speech and Signal Processing, France, pp. 493 496, 2006.
- Bastian Sauert and Peter Vary, " Near-End Listening Enhancement Optimized with respect to Speech Intelligibility Index and Audio Power Limitations & quot;, Proceedings of European Signal Processing Conference, Aalborg, Denmark, pp. 1919 1923, August 2010.
- Taal C. H., Jensen J., and Leijon A., "On Optimal Linear Filtering of Speech for Near-End Listening Enhancement", IEEE Signal Processing Letters, Vol. 20, No. 3, pp. 225 228, March 2013.
- Taal C. H., and Jesper Jensen, "SII-based Speech Pre-processing for Intelligibility Improvement in Noise", Proceeding of Interspeech, Lyon, France, pp. 3582 3586, August 2013.
- Jeon Yu-young and Lee Sang-min, " A Speech Enhancement Algorithm to Reduce Noise and Compensate for Partial Masking Effect ", Journal of Central South University of Technology, Vol. 18, issue 4, pp. 1121 1127, August 2011.
- Gunawan T. S., and Ambikairajah E., "Speech Enhancement using Temporal Masking and Fractional Bark Gammatone Filters", Proceedings of the 10th Australian International Conference on Speech Science & Technology, Sydney, pp. 420 425, 2004.
- Gunawan T. S., Khalifa O. O., and Ambikairajah E., "Forward Masking Threshold Estimation using Neural Networks and its Application to Parallel Speech Enhancement", in International Conference on Computer and Communication Engineering, Vol. 11, No 1, pp. 15 26, 2010.
- Yi Hu and Philipos C. Loizou, "Incorporating a Psychoacoustic Model in Frequency Domain Speech Enhancement", IEEE signal processing letters, Vol. 11, No. 2, pp. 270 273, February 2004.
- Eberhard Zwicker and Hugo Fastl, Psychoacoustics, Facts and Models. New York: Springer, 2007.
- American National Standard. Methods for the Calculation of the Speech Intelligibility Index. ANSI S3. 5-1997, 1997.
 - "PESQ: An Introduction", Psytechnics Limited, White paper, September 2001.
- Rix A. W., Beerends J. G., Hollier M. P., and Hekstra A. P., "Perceptual Evaluation of Speech Quality (PESQ)-A New Method for Speech Quality Assessment of Telephone Networks and Codecs", IEEE International Conference on Acoustics, Speech, and Signal Processing, Vol. 2, pp. 749 752, May 2001.

- ITU-T P. 862, Perceptual evaluation of speech quality, and objective method for end-to-end speech quality assessment of narrowband telephone networks and speech codecs, ITU-T Recommendation P. 862, 2000.
- Ephraim Y., and Malah D., "Speech Enhancement using a Minimum Mean Square Error Short-Time Spectral Amplitude Estimator", IEEE Transaction on Acoustic, Speech, and Signal Processing, Vol. ASSP-32, pp. 1109 1121, December 1984.
- Malihe Hassani and Karami Mollaei M. R., "Speech Enhancement based on Spectral Subtraction in Wavelet Domain", Seventh IEEE International Colloquium on Signal Processing and its Applications, pp. 366 370, March 2011.

Index Terms

Computer Science

Signal Processing

Keywords

Gain Masking Near-end noise Speech enhancement Speech intelligibility Speech quality