

Lawrence R. Rabiner - Publication List

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4. Predicting Binaural Gain in Intelligibility and Release From Masking for Speech, H. Levitt and L. R. Rabiner, *Journ. Acoust. Soc. Amer.*, Vol. 42, No. 4, pp. 820-829, October 1967.
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17. An Example of Speech Processing by Computer, L. R. Rabiner and R. W. Schafer, *Digest Record of the 1969 Joint Conference on Mathematics and Computer Aids to Design*, Anaheim, CA, pp. 207-248, October 1969.
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