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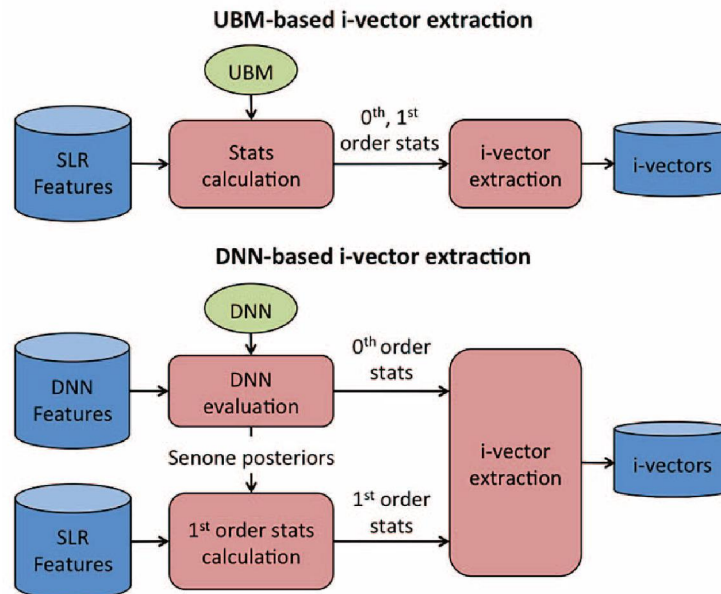
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Flow diagram of the GMM-based and the DNN-based i-vector approaches.
For more, please see “Study of Senone-Based Deep Neural Network Approaches for Spoken Language Recognition,”
by L. Ferrer, *et al.*, p. 105.

For the January 2016 issue, see p. 1 for Table of Contents
For the February 2016 issue, see p. 211 for Table of Contents



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