



# Insights into End-to-End Learning Scheme for Language Identification

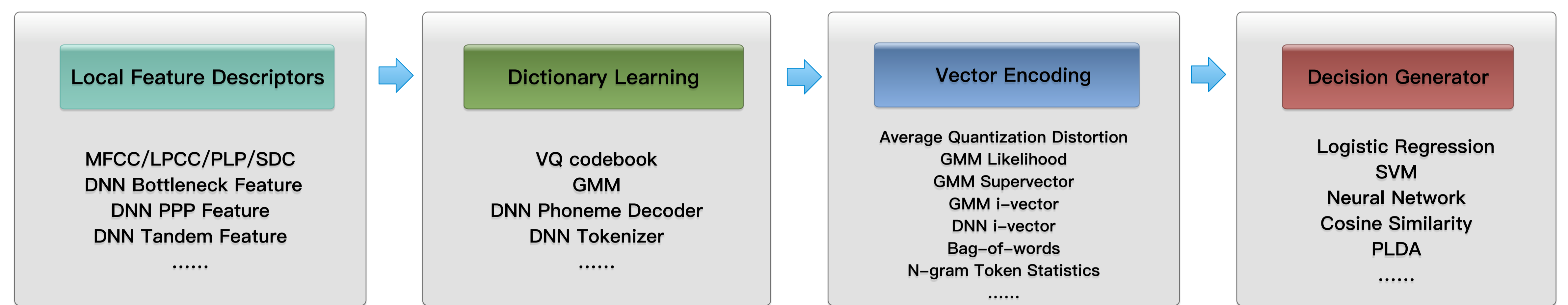
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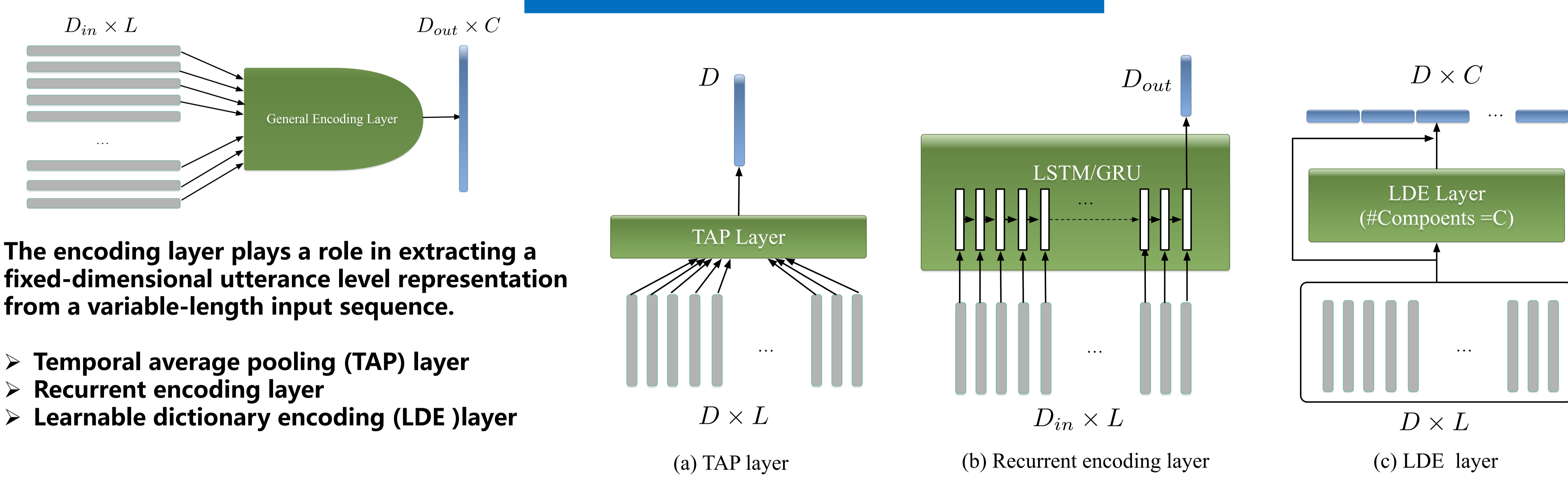
## Introduction



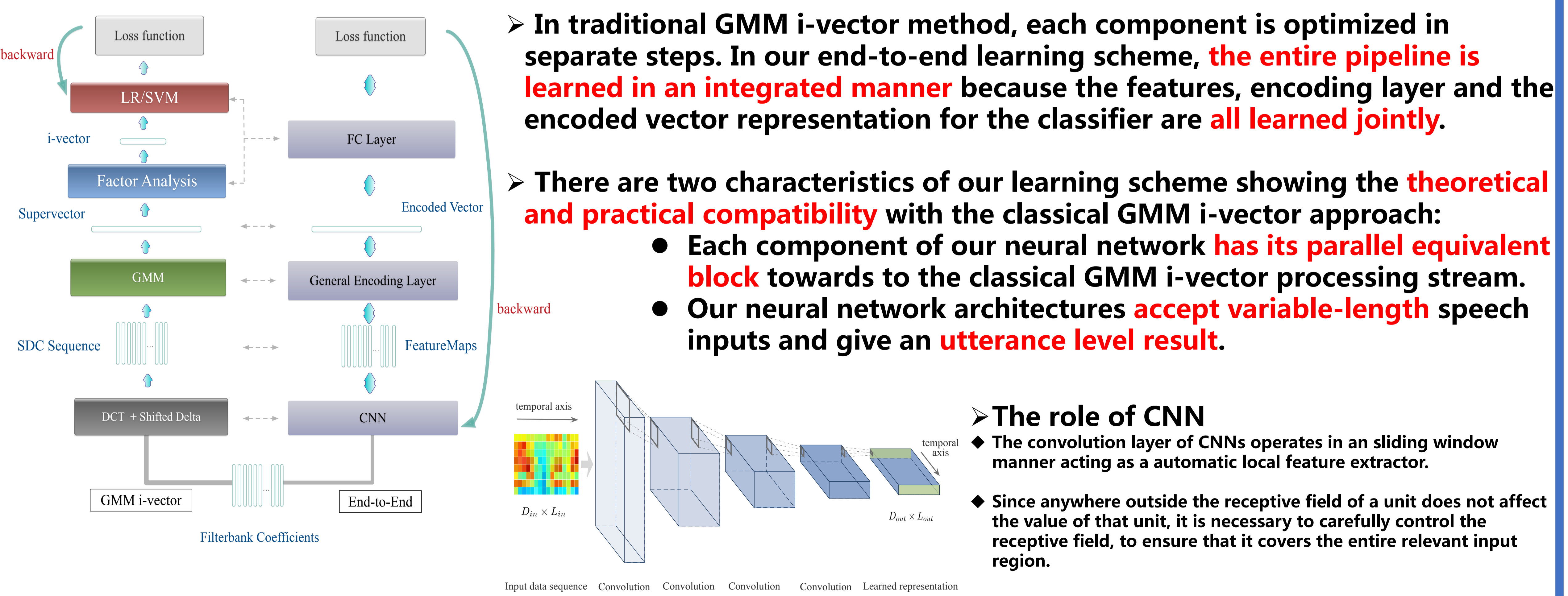
Four main steps in the conventional processing pipeline

The GMM i-vector based approaches are comprised of a series hand-crafted or ad-hoc algorithmic components, and they show strong generalization ability and robustness when data and computational resource are limited.

## General Encoding Layer



## End-to-End Learning Scheme



## Experimental Results and Discussion

**Table 1.** Performance on the 2007 NIST LRE closed-set task

System ID	System Description	$C_{avg}(\%)/EER(\%)$		
		3s	10s	30s
1	GMM i-vector	20.46/17.71	8.29/7.00	3.02/2.27
2	DNN i-vector	14.64/12.04	6.20/3.74	2.60/1.29
3	DNN PPP Feature	<b>8.00/6.90</b>	<b>2.20/1.43</b>	<b>0.61/0.32</b>
4	DNN Tandem Feature	9.85/7.96	3.16/1.95	0.97/0.51
5	DNN Phonotactic[22]	18.59/12.79	6.28/4.21	1.34/0.79
6	RNN D&C[22]	22.67/15.57	9.45/6.81	3.28/3.25
7	LSTM-Attention[21]	-/14.72	-/-	-/-
8	<b>CNN-TAP</b>	9.98/11.28	3.24/5.76	1.73/3.96
9	<b>CNN-GRU</b>	11.31/10.74	5.49/6.40	-/-
10	<b>CNN-LSTM</b>	10.17/9.80	4.66/4.26	-/-
11	<b>CNN-LDE</b>	<b>8.25/7.75</b>	<b>2.61/2.31</b>	<b>1.13/0.96</b>

- For ID2 to ID5, additional speech data with transcription and an extra DNN phoneme decoder is required, while our end-to-end systems only rely on the acoustic level feature of LID data.
- For each training step, an integer  $L$  within [200,1000] interval is randomly generated, and each data in the mini-batch is cropped or extended to  $L$  frames. In testing stage, all the 3s, 10s, and 30s duration data is tested on the same model. Because the duration length is arbitrary, we feed the testing speech utterance to the trained neural network one by one.
- It's very interesting that although recurrent layer introduces much more parameters comparing with TAP, it results in a slightly degraded performance. Specially, when the full 30s duration utterance is fed into our CNN-GRU/CNN-LSTM neural network trained within 1000 frames (10s), it suffers from "the curse of sentence length". The performance drops sharply and almost equals to random guess.

- Although recurrent layer can deal with variable-length inputs theoretically, it might be not suitable for the testing task with wide duration range and particularly with duration that are much longer than those used for training.
- The success of TAP and LDE layer inspires us that it might be more necessary to get utterance level representation describing the context-independent feature distribution rather than the temporal structure.

