

Mixup Augmentation for Generalizable Speech Separation

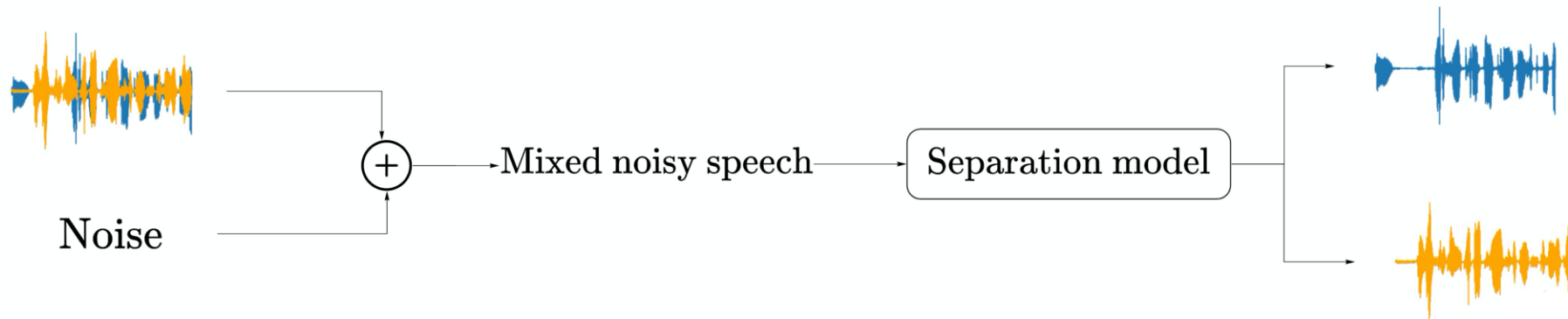
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Introduction

- Single channel speech separation in noisy environments



- Applications: Hearing aids, captioning & transcription (YouTube), human robotic interaction, automatic speech recognition

Motivation

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 - Improve generalization of separation models across datasets
 - Improve separation performance in unseen noisy conditions
 - Traditional regularization techniques, augmentations did not improve generalization
- Contributions
 - Extend Mixup augmentation and variations for time-domain speech separation
 - Proposed Data-only Mixup improves inter corpus separation performance

Separation model architecture

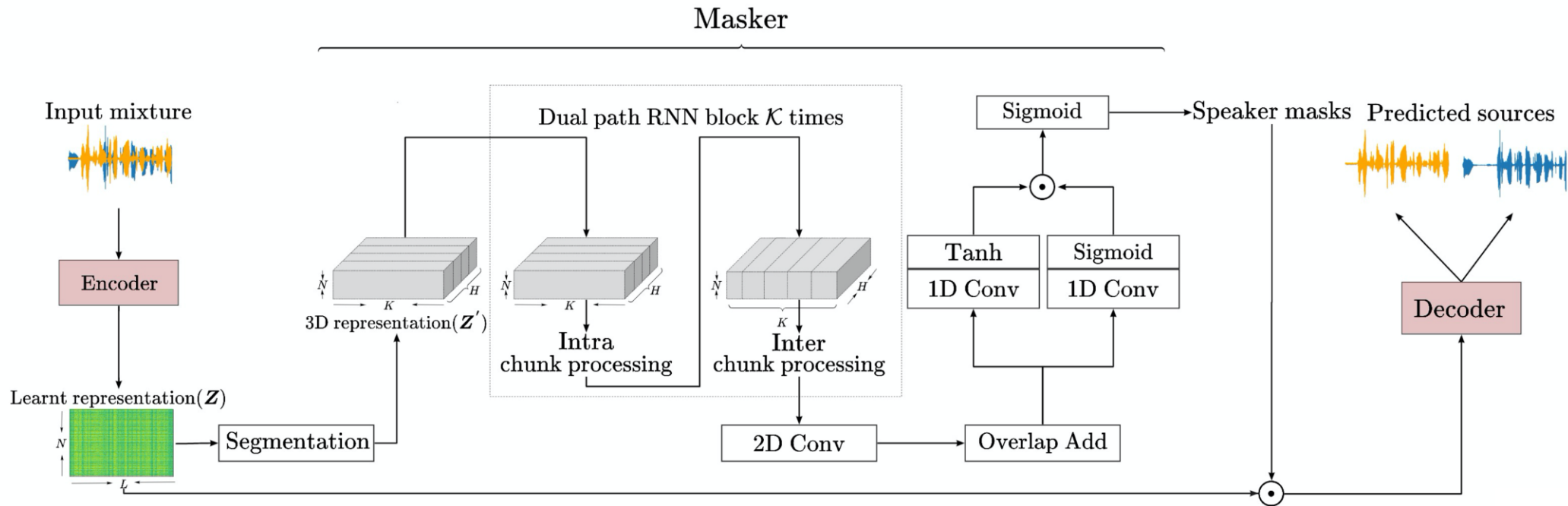
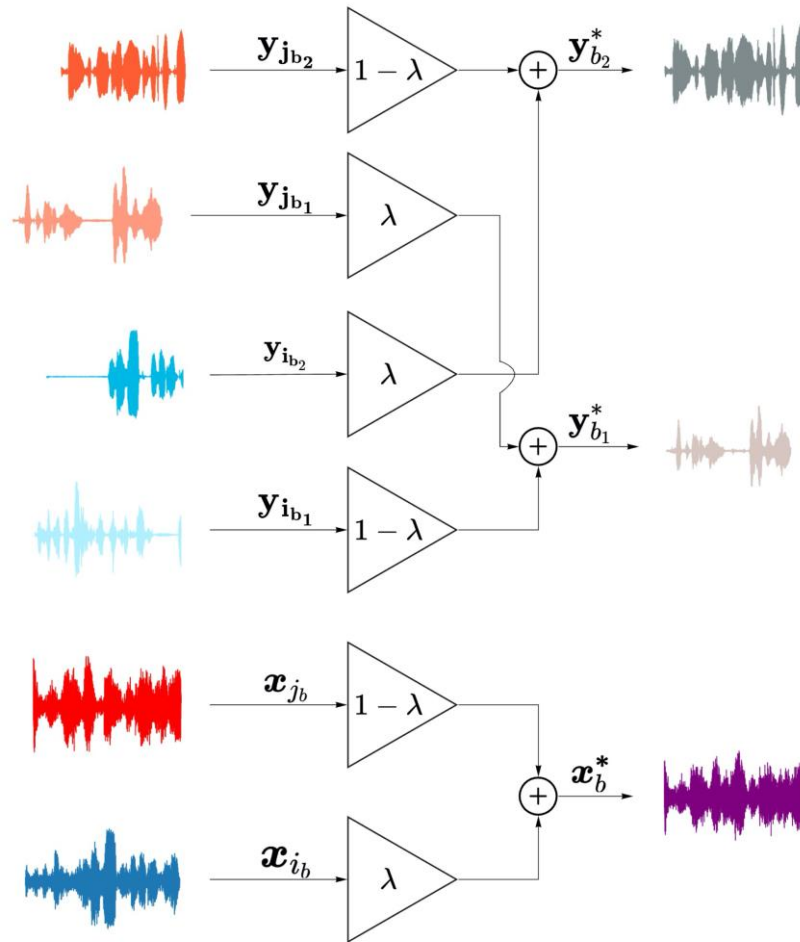


Fig. DPRNN [1] separation model

[1] Luo et al., Dual-path RNN: efficient long sequence modelling for time-domain single-channel speech Separation, in Proc. Interspeech, 2019.

Mixup



x_{i_b}, x_{j_b} : Two distinct mixtures from the mini-batch
 $y_{i_{b_1}}, y_{i_{b_2}}$ and $y_{j_{b_1}}, y_{j_{b_2}}$: ground-truth speech

x_b^* : Augmented mixture
 $y_{b_1}^*$ and $y_{b_2}^*$: Augmented ground truth speech

$$\lambda = \text{beta}(\alpha, \beta)$$

Mixup variants

- Complete Mixup: Augment training data using Mixup for all epochs

$$\mathcal{L}_{\text{CP}} = \mathcal{L}_{\text{augment}}, \quad 0 < e < E_{\text{max}}$$

E_{max} : Maximum number of epochs for training

- Partial Mixup: Regular training in initial epochs followed by Mixup Augmentation in subsequent epochs

$$\mathcal{L}_{\text{PA}} = \begin{cases} \mathcal{L}_{\text{regular}}, & 0 < e \leq E_{\text{early}} \\ \mathcal{L}_{\text{regular}}, & (E_{\text{early}} < e < E_{\text{max}}) \wedge (e|Q \neq 0) \\ \mathcal{L}_{\text{augment}}, & (E_{\text{early}} < e < E_{\text{max}}) \wedge (e|Q = 0) \end{cases}$$

E_{max} : Maximum number of epochs for training

E_{early} : Number of epochs until which Augmentation is applied for

Mixup variants

- Pretrained Mixup: Fine tune a pretrained model using Mixup Augmentation

$$\mathcal{L}_{PT} = \begin{cases} \mathcal{L}_{\text{regular}}, & 0 < e \leq E_{\text{max}} \\ \mathcal{L}_{\text{augment}}, & E_{\text{max}} < e < E_{pt} \end{cases}$$

E_{pt} : Maximum number of epochs pre-trained model is finetuned for

- Data only Mixup
 - Apply Mixup on mixtures only
 - Keep ground truth as most dominant sources in Mixup augmented mixture

$$\begin{cases} \mathbf{x}_b^\circ = \lambda \mathbf{x}_{i_b} + (1 - \lambda) \mathbf{x}_{j_b} \\ \mathbf{Y}_b^\circ = \mathbf{Y}_{i_b} \end{cases}$$

$$\mathcal{L}_{DO} = \mathcal{L}(\mathbf{Y}^\circ, \hat{\mathbf{Y}}^\circ), \quad 0 < e < E_{\text{max}}$$

Experiments & datasets

Dataset	Split	Hours	Speakers	Noise corpus
LibriMix[2]	train-100	58	251	WHAM[5]
LibriMix[2]	test	11	40	WHAM[5]
VCTK[3]	test	9	109	WHAM[5]
TIMIT[4]	test	10	630	Env noise corpus[6]

- Intra corpus - Train on LibriMix (train-100) & test on LibriMix (test)
- Inter corpus – Train on LibriMix (train-100) & test on TIMIT (test) and VCTK (test)

[2] Cosentino et al., LibriMix: An Open-Source Dataset for Generalizable Speech Separation, arXiv preprint arXiv:2005.11262

[3] C. Veaux et al., Superseded-CSTR VCTK corpus: English multispeaker corpus for cstr voice cloning toolkit, 2016

[4] J. S. Garofolo, TIMIT acoustic phonetic continuous speech corpus, Linguistic data consortium, 1993

[5] Wichern et al., WHAM!: Extending speech separation to noisy environments, in Proc. Interspeech, 2019

[6] Xu et al. A regression approach to speech enhancement based on deep neural networks, IEEE/ACM Trans. Audio, Speech, Lang. Process., 2014

Ablation of scalars for Mixup

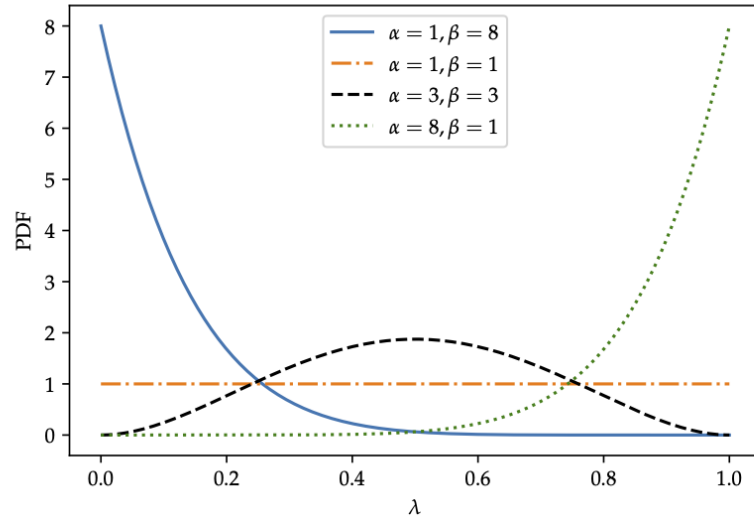


Fig. Probability density function of beta distribution with different α and β . $\lambda = \text{beta}(\alpha, \beta)$

$\alpha \backslash \beta$	1	3	8
1	11.69	11.70	11.51
3	11.64	11.70	11.64
8	11.97	11.70	11.57

(a) Complete Mixup

$\alpha \backslash \beta$	1	3	8
1	11.17	7.60	5.89
3	11.55	11.31	7.12
8	12.00	11.51	11.26

(b) Data Only Mixup

Table. SI-SNR_i (dB) of data-augmented DPRNN for various values of α and β

Intra corpus results

Augmentation type	Augmentation variation	SI-SNRi
None	-	12.00
SpecAugment[7]	Frequency masking	11.63
	Time masking	12.04
	T-F masking	12.05
Mixup	Complete	11.97
	Data-only	12.00
	Partial	11.50
	Pre-trained	12.00

- None of the augmented models significantly outperform non-augmented model

Table: Model trained and tested on LibriMix dataset

[7] Park et al., SpecAugment: A simple data augmentation method for automatic speech recognition, in Proc. Interspeech, 2019

Inter corpus results (1)

SNR	UAUG	SpecAugment			Mixup			
		TM	FM	T-F	PA	PT	CP	DO
-5	4.95	5.09	4.99	4.53	4.86	5.19	4.95	5.61
0	5.41	5.76	5.94	4.84	5.38	5.69	5.85	6.60
5	6.52	6.62	6.59	6.10	6.34	6.95	6.87	8.48
10	8.24	8.32	8.18	8.18	8.39	8.81	8.84	10.25
15	9.80	10.22	9.85	9.82	10.21	10.64	10.33	11.42
20	10.93	11.24	11.08	11.30	10.92	11.84	10.94	11.97
Avg	7.64	7.87	7.77	7.46	7.68	8.13	7.96	9.06

- Data-only Mixup improves separation performance on TIMIT dataset
- Noise types & speakers in TIMIT are different from LibriMix

Table: Model trained on LibriMix and tested on TIMIT dataset

Inter corpus results (2)

Augmentation type	Augmentation variation	SI-SNRi
None	-	11.07
SpecAugment[7]	Frequency masking	10.79
	Time masking	11.09
	T-F	11.04
Mixup	Complete	11.11
	Data-only	11.43
	Partial	10.93
	Pre-trained	11.06

- Data-only Mixup slightly improves separation performance on VCTK dataset
- Speakers in VCTK are different from LibriMix
- Noise samples in VCTK dataset is the same as LibriMix dataset

Table: Model trained on LibriMix and tested on VCTK dataset

[7] Park et al., SpecAugment: A simple data augmentation method for automatic speech recognition, in Proc. Interspeech, 2019

Conclusion & future work

- Data-only Mixup augmentation improves cross-corpus performance for speech separation model
- Data augmentation approach doesn't incur additional in network parameters
- Future work – Finding optimal augmentation combinations using learnt augmentation strategies