

THE STC SYSTEM FOR THE CHIME 2018 CHALLENGE

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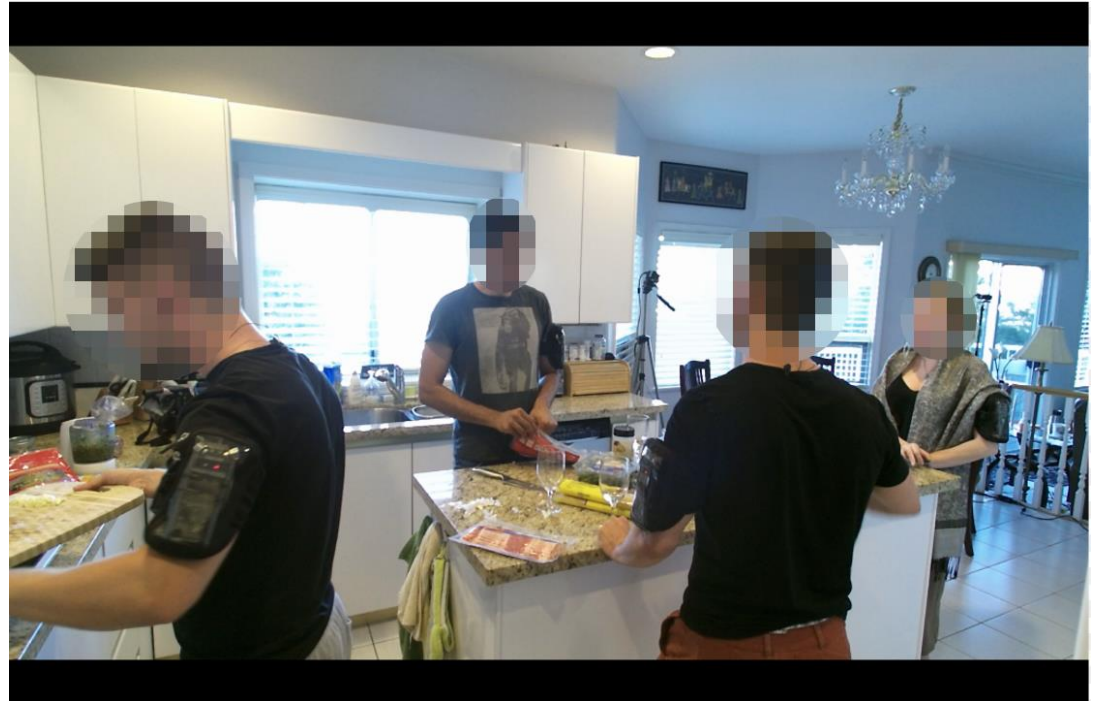
STC-INNOVATIONS

- 1** Top-3 in Babel OpenKWS, 1st NIST i-vector Machine Learning Challenge 2014, 2nd NIST LRE 2015, 2nd NIST SITW, 2015 2nd ANTISPOOF 2015, 2017 1st ANTISPOOF 2017
- 2** Multi-disciplinary team with expertise in general machine learning, speech recognition, NLU, bi-modal (voice+face) identification
- 3** Close partnership with ITMO University

- ▶ Introduction
- ▶ Unsuccess story
- ▶ Success story
- ▶ Conclusions
- ▶ Final results on eval and future work

Main challenges

- ▶ Conversational speech
- ▶ Noisy real-world environment
- ▶ Far-field conditions
- ▶ Great amount of overlapped speech



Beamforming and Enhancement: Unsuccess story

Unsuccess story

- ▶ MVDR + CGMM/Music/estnoise_g mask

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Unsuccess story

- ▶ MVDR + CGMM/Music/estnoiseeg mask
- ▶ DeepBeam [Qian, 2018] *





*<https://github.com/auspicious3000/deepbeam>

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*<https://github.com/fgnt/nn-gev>

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- ▶ Deep Clustering
- ▶ Permutation invariant training (PIT)

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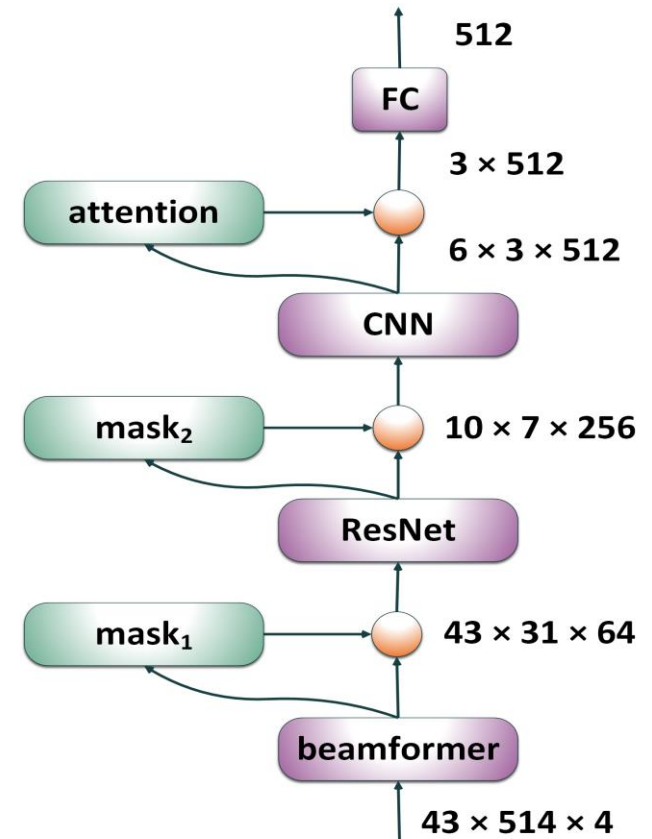


Success story

Multi-channel speaker-aware model training: embeddings

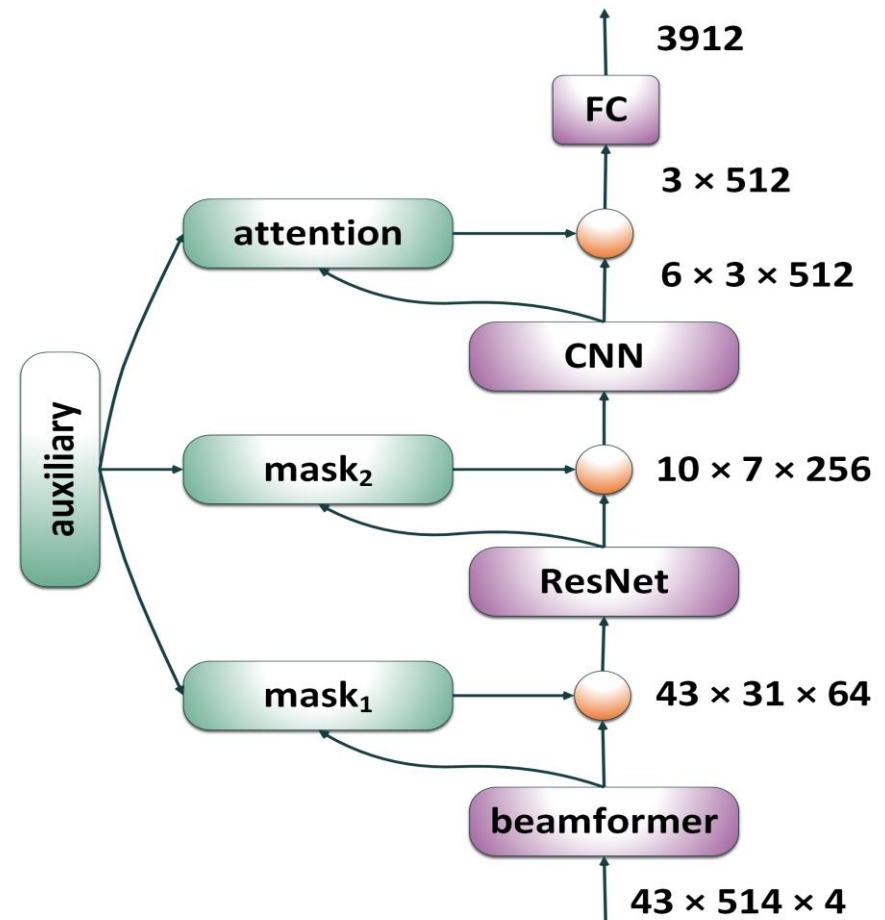
- ▶ embedding training by triplet ranking loss [Ye and Guo, 2018]

Success story



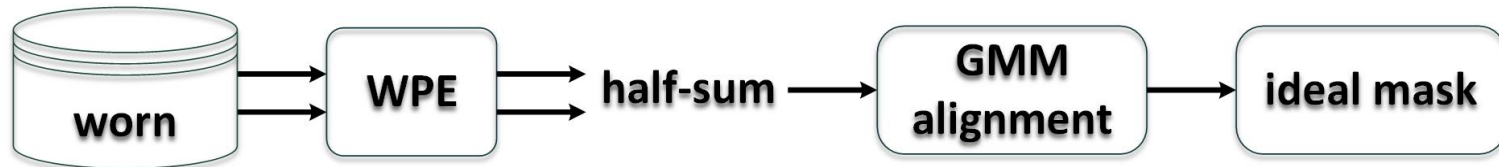
Multi-channel speaker-aware model training: final model

- ▶ auxiliary inputs [Zmolikova, 2018]
- ▶ residual attention network [Wang, 2017]
- ▶ speaker-adapted classifier *
- ▶ sum and average all embeddings for speaker in utterance



*<https://github.com/Microsoft/LightGBM>

Speaker adaptation by frame-level mask: training



	<sil>						word						<sil>				<noise>			<sil>							
P01(id 1)	0	0	0	0	0	0	1	1	1	1	1	1	0	0	0	0	1	1	1	0	0	0	0				
	<sil>		word		<sil>		<laught>						<sil>				word		<sil>								
P02(id 2)	0	0	2	2	2	0	0	0	2	2	2	2	2	2	0	0	0	0	0	0	2	2	0	0			
	<sil>		word				<sil>		word				<sil>				<spn>			<sil>							
P03(id 4)	0	0	0	4	4	4	4	0	0	4	4	4	4	0	0	0	0	0	0	4	4	4	0	0	0	0	0
	<sil>						<laught>						<sil>			word			<sil>			word					
P04(id 8)	0	0	0	0	0	0	0	0	8	8	8	8	0	0	0	8	8	8	0	0	0	0	8	8	8		
Ideal mask (general)	0	0	2	6	6	4	5	1	11	15	15	7	2	2	8	8	9	5	5	4	2	10	8	8			
Ideal targets (if P01)	1	1	0	0	0	0	1	1	1	0	0	1	0	0	0	0	1	1	1	0	0	0	0	0			

Speaker adaptation by frame-level mask: filtering

Original acoustic feats	$x_{t,1}$	$x_{t+1,1}$...					$x_{t+23,1}$
	\vdots	\vdots	\ddots					\vdots
	$x_{t,n}$	$x_{t+1,n}$...					$x_{t+23,n}$
Speaker mask	0.6	0.7	0.5	0.1	0.2	0.3	0.4	0.4
Filtered acoustic feats	$x_{t,1}$	$x_{t+1,1}$	$x_{t+2,1}$	Throw out			$x_{t+22,1}$	$x_{t+23,1}$
	\vdots	\vdots	\vdots				\vdots	\vdots
	$x_{t,n}$	$x_{t+1,n}$	$x_{t+2,n}$				$x_{t+22,n}$	$x_{t+23,n}$

* <https://github.com/speechpro/mixup> (for Kaldi)

Mixup [Medennikov, 2018] *

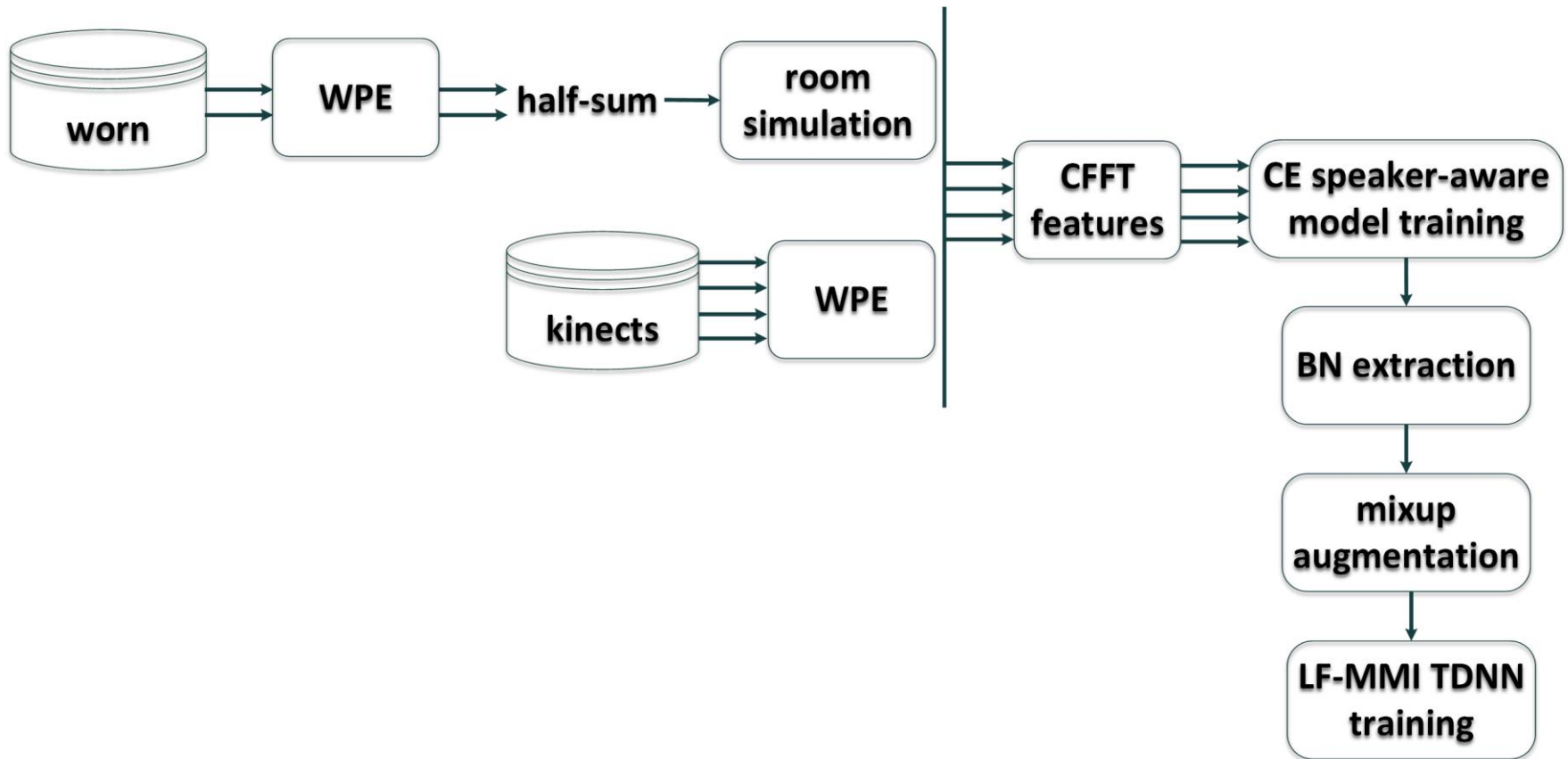
- ▶ virtual training examples by combining existing ones
- ▶ especially effective on mismatched test data

Generation of new training data

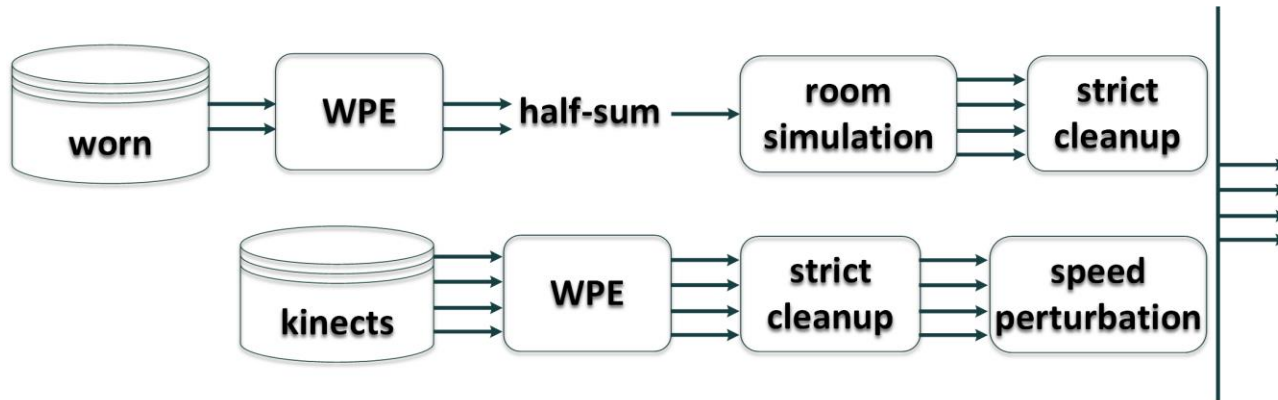
$$\tilde{x} = \lambda x_i + (1 - \lambda)x_j$$

$$\tilde{y} = \lambda y_i + (1 - \lambda)y_j$$

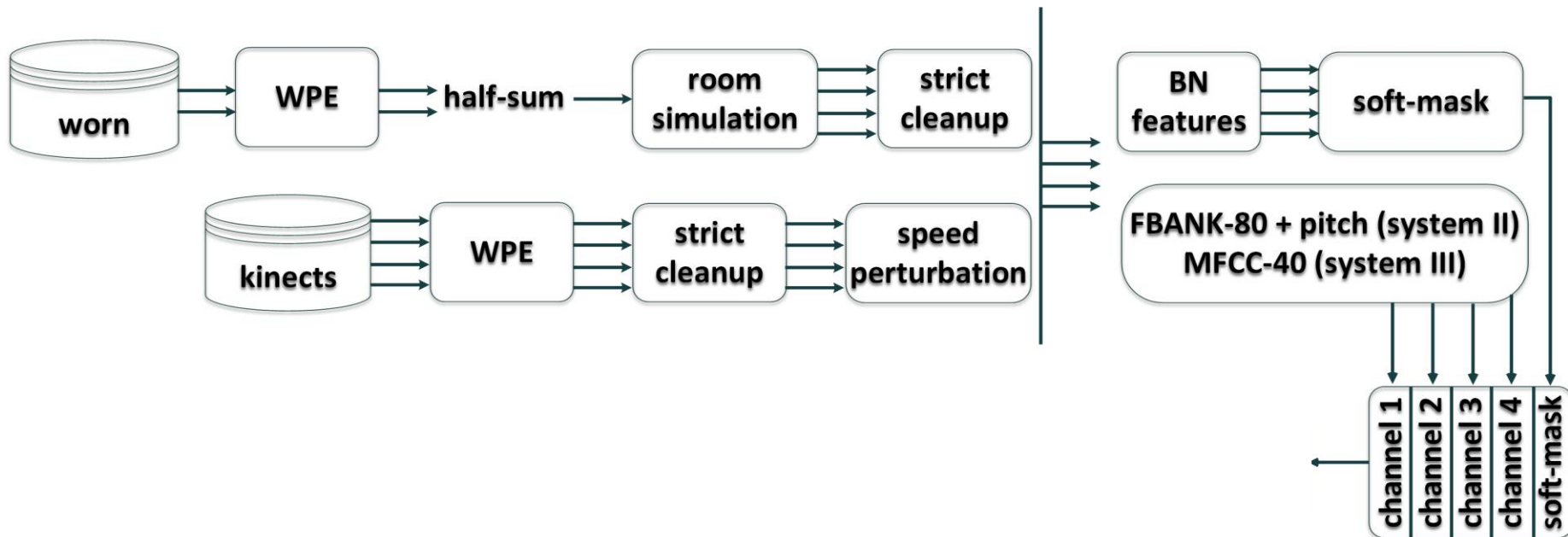
System I



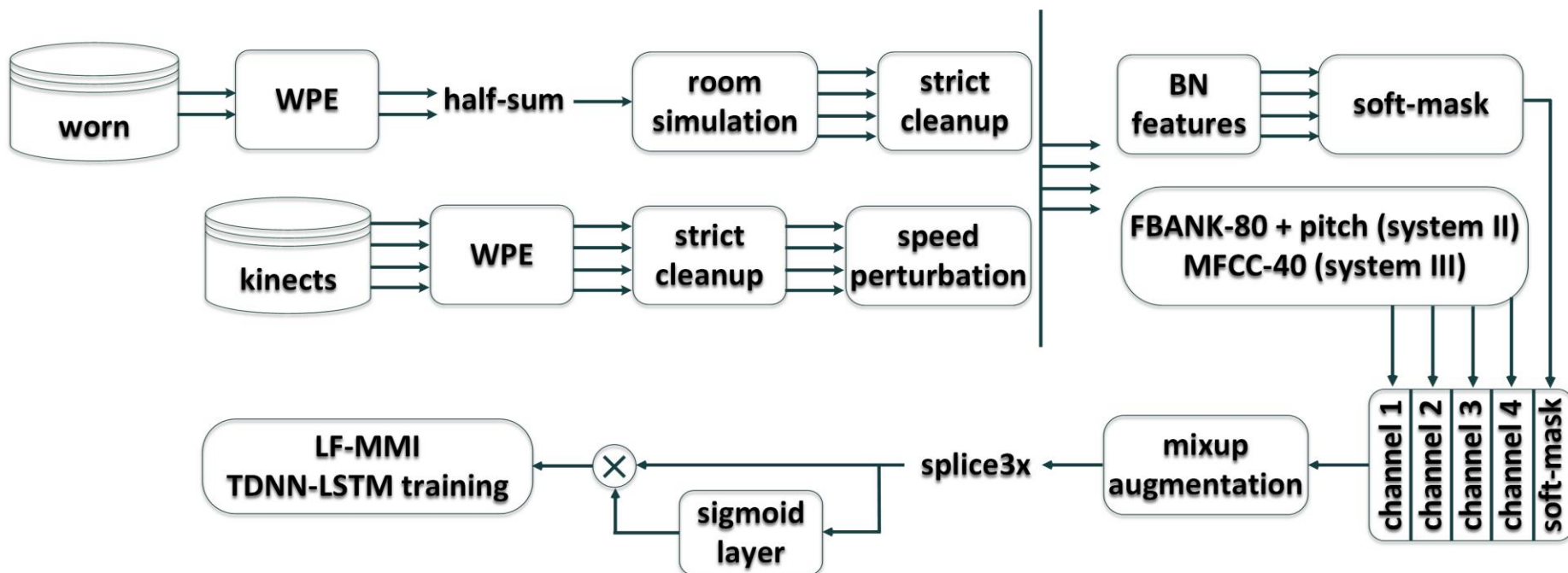
System II and III



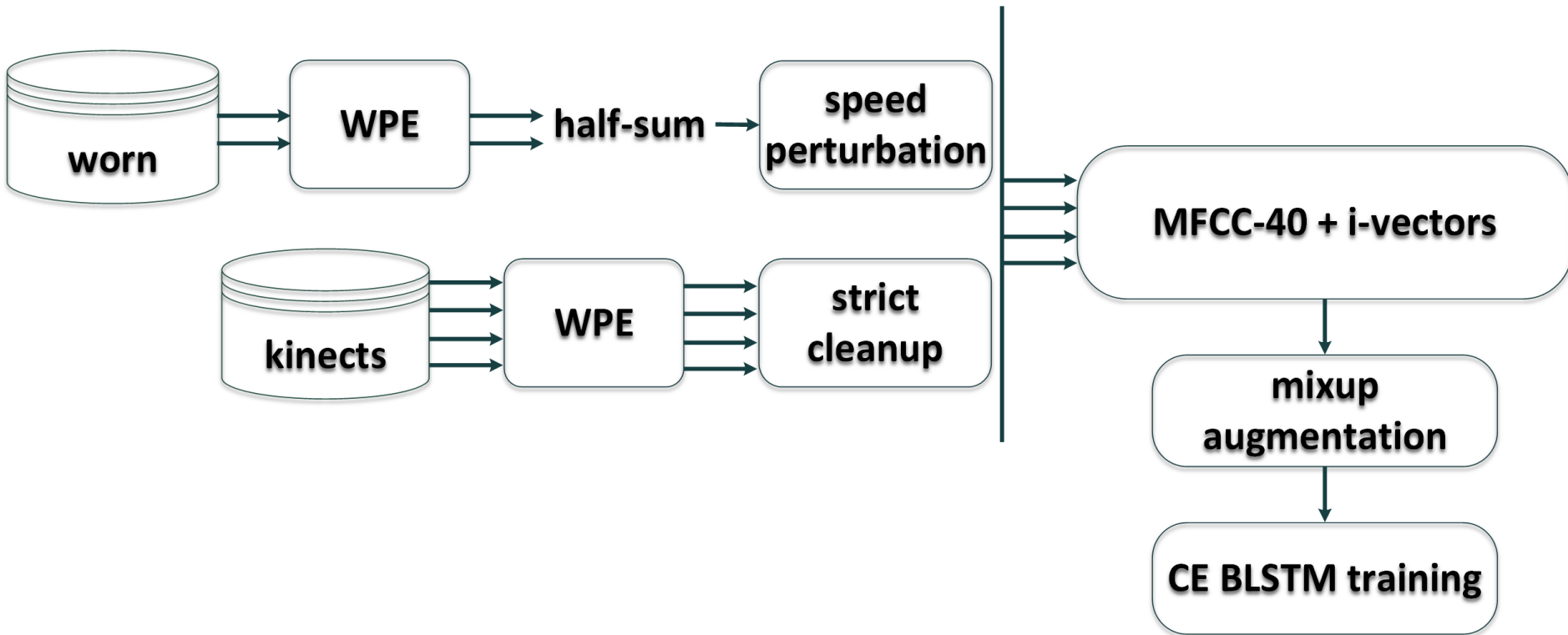
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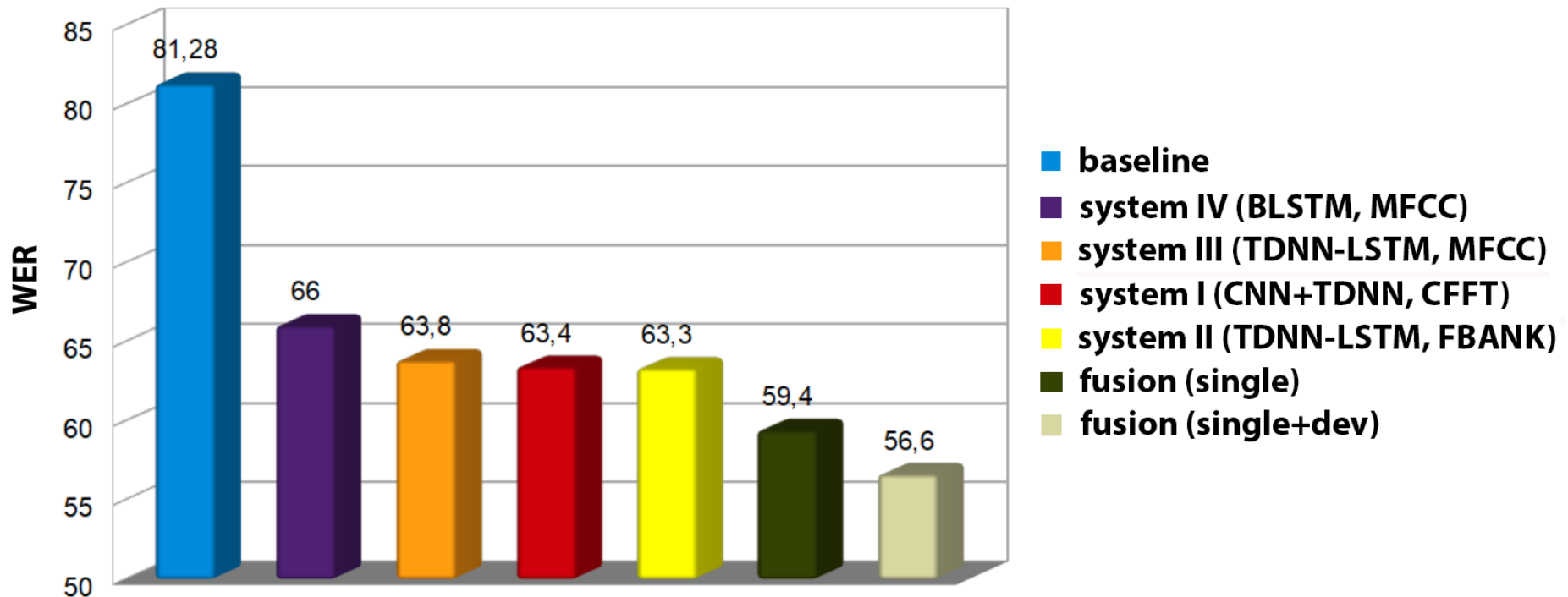
System IV



Decoding and models combination

- ▶ Decoding: application of softmax temperature to a prior distribution
- ▶ Fusion: posterior-level combination or two types of lattice-level combination

Fusion



WER (%) for the final system per session and location

Track	Session	Kitchen	Dining	Living	Overall
Single	S02	67.7	59.7	55.5	59.4
	S09	58.0	59.8	54.9	
Single+Dev	S02	65.5	56.2	52.4	56.6
	S09	55.7	56.8	51.9	
Multiple	S02	65.8	57.9	55.1	58.1
	S09	55.5	57.3	55.4	
Multiple+Dev	S02	62.1	52.2	50.2	53.5
	S09	51.2	51.6	51.4	

Summary

Track	Features	Adaptation	Model	Loss	WER
Single	CFFT	Auxiliary	CNN+TDNN	CE, LF-MMI	63.4
	FBANK	soft-mask	TDNN-LSTM	LF-MMI	63.3
	MFCC	soft-mask	TDNN-LSTM	LF-MMI	63.8
	MFCC	ivec+mask	BLSTM	CE	66.0
	Fusion (4 systems)				59.4
Single+Dev	Fusion (4 systems)				56.6
Multiple	Fusion (4 systems)*				58.1
Multiple+Dev	Fusion (4 systems)*				53.5

Conclusions

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- ▶ Both speaker separation and speaker adaptation are extremely important
- ▶ Data augmentation and normalization are reasonably effective for this type of data
- ▶ Fusion always gives a good performance improvement

Final results on eval and future work

Baseline	Our result	abs, %	rel, %
73.3	55.5	-17.8	-24.3

- ▶ Joint training of all components (front-end and back-end)
- ▶ Diarization for unsegmented real-world data

Contributions of applied methods

Method	Abs WER improvement, %
Array synchronization	0.9
Room simulator	1.6
Alignment transfer (worn half-sum → kinect)	1.3
Speaker adaptation (gating/throw out)	7/5
Speaker adaptation (i-vector)	2.4
Speaker adaptation (auxiliary)	4.1
Multi-channel model	2.2
Strict cleanup	1.3
WPE	1.4
Mixup	1.1
Speed Perturbation	0.9
Backstitch training	0.5
Fusion	3.9

THANK YOU

ABOUT THE COMPANY

STC-Innovations is a leader in the multimodal biometric market. STC-Innovations develops multimodal biometric solutions based on person-identifying technologies via voice, face and other noncontact biometric features.

STC-Innovations is a spin-off company of the Speech Technologies Center, leading global provider of innovative systems in high-quality recording, audio and video processing and analysis, speech synthesis and recognition, and real-time, high-accuracy voice and facial biometrics solutions with over 20 years of research, development and implementation experience in Russia and internationally.

STC is ISO-9001: 2008 certified.

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