

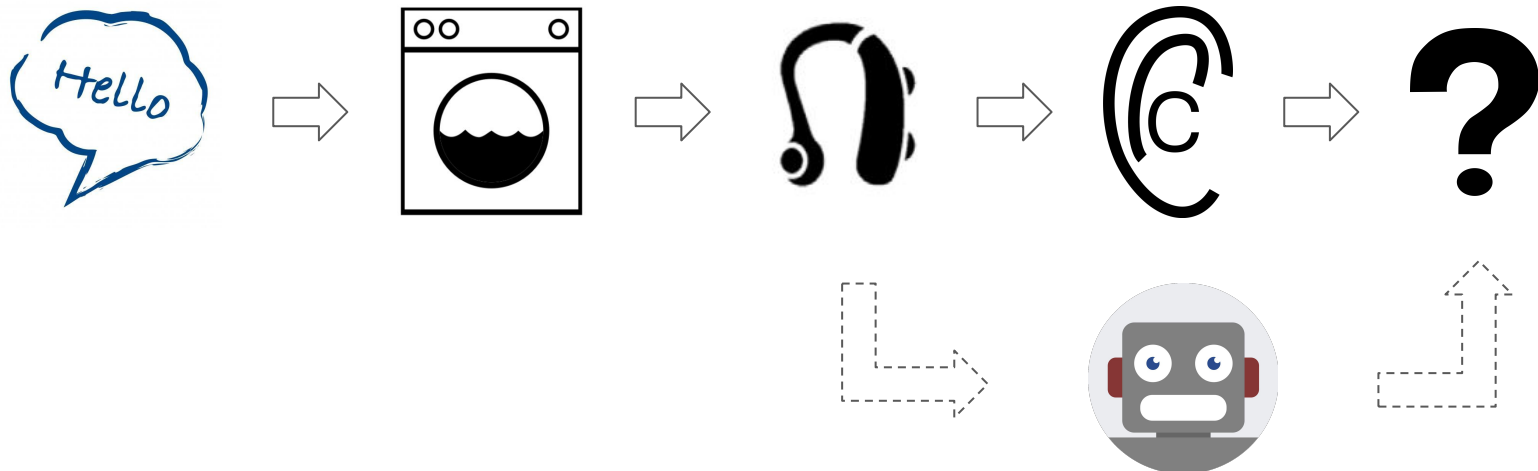


Sheffield Systems E029 & E032 for the First Round Clarity Prediction Challenge

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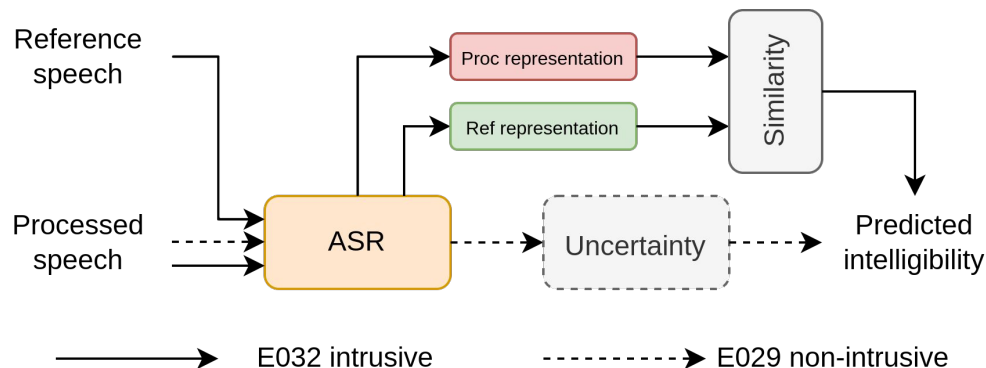
Background



How an automatic speech recogniser (ASR)
can be used to predict the intelligibility?

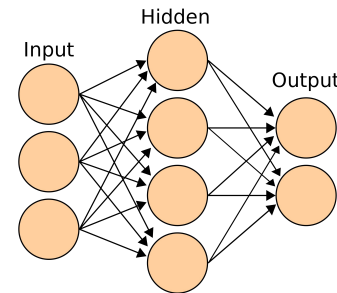
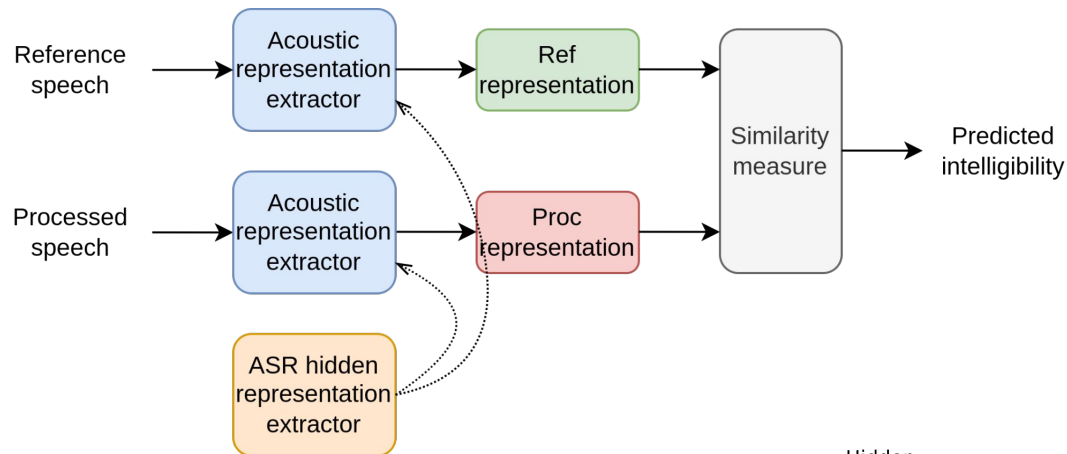
Overview

- ❖ E032 (intrusive): *Exploiting Hidden Representations from a DNN-based Speech Recogniser for Speech Intelligibility Prediction in Hearing-impaired Listeners*
- ❖ E029 (non-intrusive): *Unsupervised Uncertainty Measures of Automatic Speech Recognition for Non-intrusive Speech Intelligibility Prediction*



E032 overview

- ❑ Method
 - ❑ DNN-based ASR model
 - ❑ Hidden representations
 - ❑ Similarity computation
- ❑ Experimental setup
- ❑ Results
- ❑ Conclusions

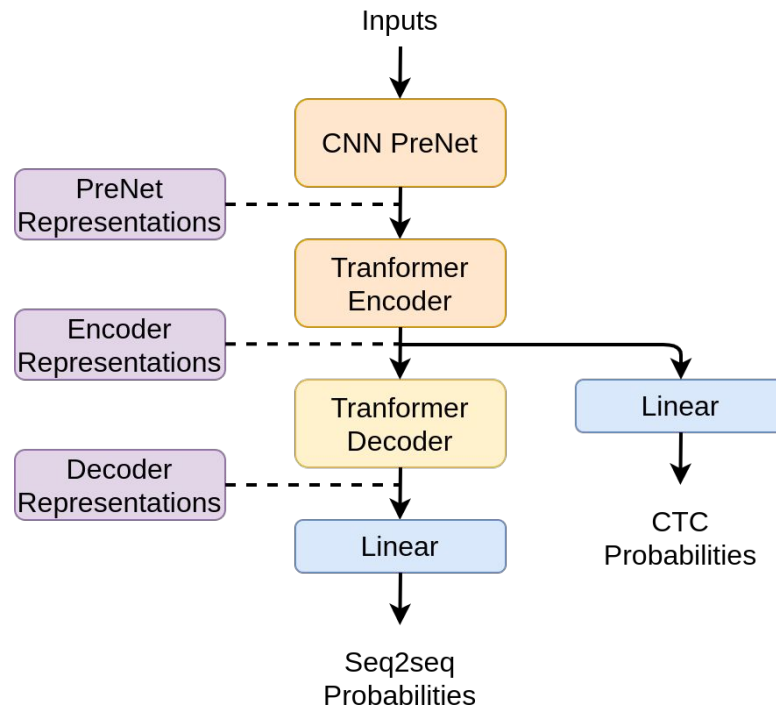


- Transformer-based end-to-end ASR model
- Hybrid training: Connectionist Temporal Classification + Attention-based Sequence-to-sequence
- SpeechBrain LibriSpeech ASR transformer recipe¹
- Fine-tuned from pretrained LibriSpeech (960h) model, i.e., strong knowledge on clean speech recognition

¹<https://github.com/speechbrain/speechbrain/tree/develop/recipes/LibriSpeech/ASR/transformer>

Hidden representations

- PreNet representations: low-level acoustic features
- Encoder representations: high-level acoustic features
- Decoder representations: features with language model knowledge



Similarity computation

Given two hidden vectors of the reference and processed speech:

$$h, \hat{h} \in \mathcal{R}^d$$

The similarities of the PreNet and Encoder representations:

$$\text{sim}(H^{bi}, \hat{H}^{bi}) = \frac{1}{T} \sum_{t=1}^T \max \left\{ \rho_t^{ll}, \rho_t^{lr}, \rho_t^{rl}, \rho_t^{rr} \right\}$$

The similarity of the Decoder representations, fast dynamic time warping is used:

$$\text{sim}(H_w, \hat{H}_w) = \frac{1}{T_w} \sum_{t=1}^{T_w} \cos(H_w(t), \hat{H}_w(t))$$

$$\text{sim}(H^{bi}, \hat{H}^{bi}) = \max \left\{ \text{sim}(H_w^l, \hat{H}_w^l), \text{sim}(H_w^l, \hat{H}_w^r), \text{sim}(H_w^r, \hat{H}_w^l), \text{sim}(H_w^r, \hat{H}_w^r) \right\}$$

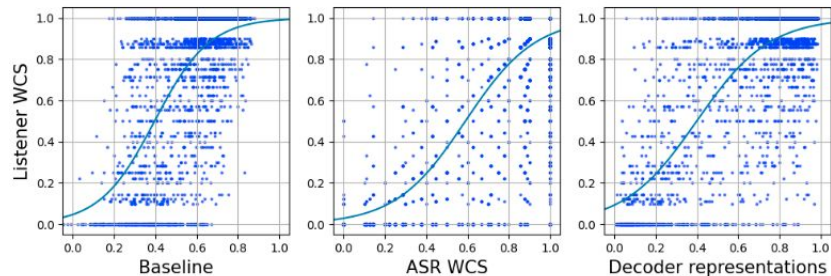
Experimental Setup

- Data split: 70% of CPC1_train_data is used as training set, 30% is used as dev set
- ASR training:
 - Cambridge MSBG hearing loss model as front-end to simulate hearing losses
 - Librispeech train-clean-100 + CEC1 training noise (CLS) for 10 epochs
 - CPC1 training set for 10 epochs
- Evaluation:
 - Root mean square error (RMS), normalised cross-correlation (NCC), Kendall's Tau coefficient (KT)
 - A logistic function fitting on the dev set
- Baselines:
 - MSBG + MBSTOI (CPC1 baseline)
 - ASR word correctness score (WCS)

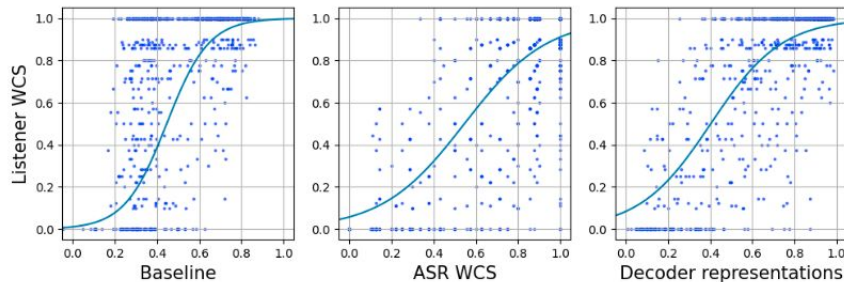


E032 Overall results

	RMSE ↓	NCC ↑	KT ↑
Closed-set			
Baseline	0.285	0.621	0.398
ASR WCS	0.250	0.729	0.523
PreNet representations	0.347	0.299	0.182
Encoder representations	0.237	0.758	0.487
Decoder representations	0.231	0.773	0.498
Open-set			
Baseline	0.365	0.529	0.391
ASR WCS	0.250	0.723	0.534
PreNet representations	0.356	0.254	0.136
Encoder representations	0.241	0.751	0.534
Decoder representations	0.235	0.763	0.530



(a) *Closed-set*



(b) *Open-set*

Further analysis on the CPC1 closed-set:

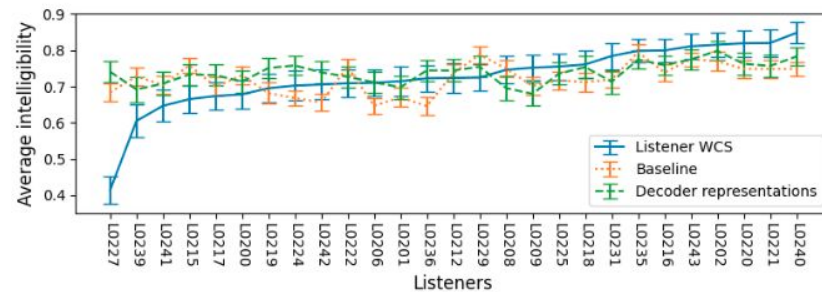
- Different training data
- With or without using the MSBG hearing loss model

MSBG	Training data	RMSE ↓	NCC ↑	KT ↑
with	LS	0.264	0.692	0.449
	LS+CLS	0.243	0.746	0.464
	LS+CPC1	0.233	0.768	0.503
	LS+CLS+CPC1	0.231	0.773	0.498
w/o	LS+CLS+CPC1	0.234	0.767	0.476

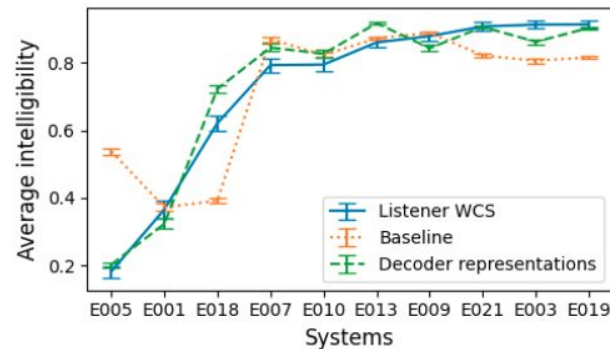
Further analysis on the CPC1 closed-set:

- Listener- and system-wise results

	RMSE ↓	NCC ↑	KT ↑
Listener-wise			
Baseline	0.078	0.414	0.311
Decoder representations	0.078	0.419	0.407
System-wise			
Baseline	0.147	0.798	0.244
Decoder representations	0.048	0.982	0.644



(a) Listener-wise



(b) System-wise

E032 conclusions

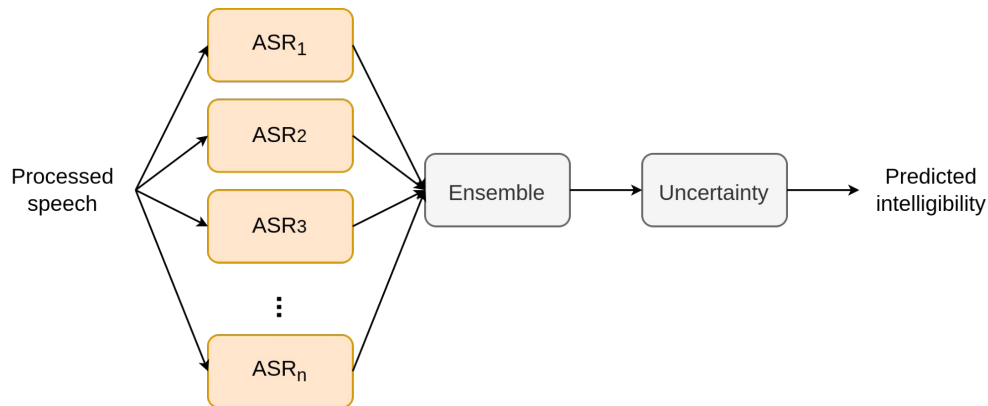
- Representations of a state-of-the-art ASR could be better at intelligibility prediction than acoustic representations of MBSTOI.
- Language knowledge matters, as Decoder > Encoder representations.
- ASR recognition results (WCS) might not be the best intelligibility predictor.
- ASR training data matters, meanwhile, ASR trained only by LS can still make good prediction.
- MSBG hearing loss simulation can improve performance.
- If considering only monotonicity, no listener intelligibility label is needed.

Questions on E032?

E029 overview

Method

- ASR model (same as E032)
 - Unsupervised sequence-level uncertainty estimation**
- ## Experimental setup (same as E032)
- Results
 - Conclusions



- Why uncertainty?
 - Intelligibility can be characterised as the probability of correct word recognition by human, meanwhile, uncertainty of ASR is also associated with the probability of ASR making correct predictions.
 - Uncertainty avoids cases like correct guess.
- Why unsupervised?
 - No listener intelligibility labels are needed.
- Why sequence-level?
 - No alignment is needed.
 - Contextual information could matter.

Given a sequence of input acoustic features and the corresponding transcript targets (BPE tokens):

$$\{x_1, \dots, x_N\} = \mathbf{x}, \quad \{y_1, \dots, y_L\} = \mathbf{y}$$

The ASR posterior can be expressed as:

$$P(y_l | \mathbf{y}_{<l}, \mathbf{x}; \boldsymbol{\theta}^{(m)}) = \lambda P_{CTC}(y_l | \mathbf{y}_{<l}, \mathbf{x}; \boldsymbol{\theta}^{(m)}) + (1 - \lambda) P_{seq2seq}(y_l | \mathbf{y}_{<l}, \mathbf{x}; \boldsymbol{\theta}^{(m)})$$

The sequence-level confidence $\mathcal{C}_S$ is computed as:

$$\mathcal{C}_S = \exp \left[\frac{1}{L} \ln \sum_{l=1}^L \max \frac{1}{M} \sum_{m=1}^M P(y_l | \mathbf{y}_{<l}, \mathbf{x}; \boldsymbol{\theta}^{(m)}) \right]$$

The sequence-level entropy $\mathcal{H}_S$ can be approximated with top samples in a beam-search candidates:

$$\mathcal{H}_S = - \sum_{b=1}^B \frac{\pi_b}{L^{(b)}} \ln P(\mathbf{y}^{(b)} | \mathbf{x}, \boldsymbol{\theta})$$

Where:

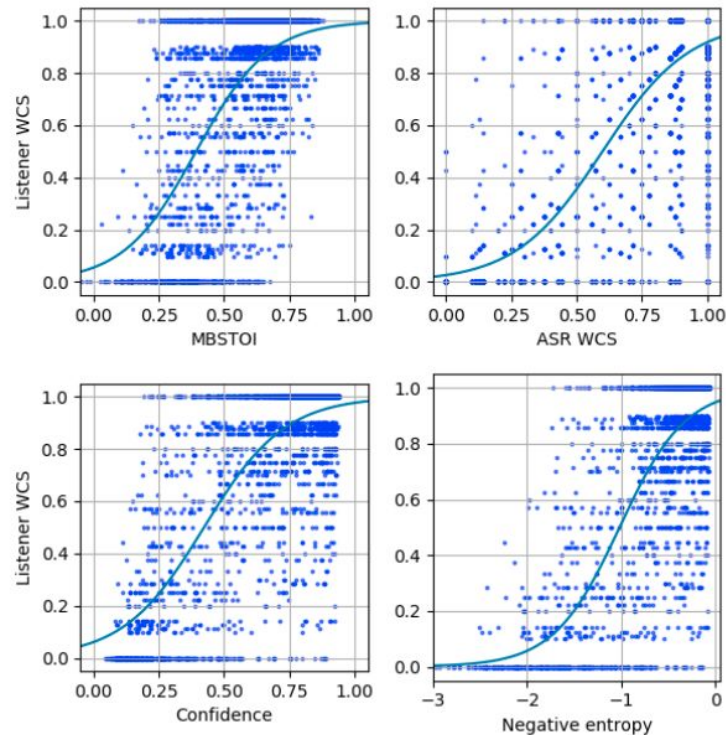
$$\pi_b = \frac{\exp \frac{1}{T} \ln P(\mathbf{y}^{(b)} | \mathbf{x}, \boldsymbol{\theta})}{\sum_k^B \exp \frac{1}{T} \ln P(\mathbf{y}^{(k)} | \mathbf{x}, \boldsymbol{\theta})}$$

$$\ln P(\mathbf{y}^{(b)} | \mathbf{x}, \boldsymbol{\theta}) = \sum_{l^{(b)}=1}^{L^{(b)}} \ln \frac{1}{M} \sum_{m=1}^M P(y_l^{(b)} | \mathbf{y}_{<l}^{(b)}, \mathbf{x}; \boldsymbol{\theta}^{(m)})$$



E029 Overall results

	WER	Measure	RMSE ↓	NCC ↑	KT ↑
<i>Closed-set</i>					
CPC1 Baseline	-	-	0.285	0.621	0.398
Proposed without MSBG	25.17	$\mathcal{C}_S$	0.241	0.751	0.472
		$-\mathcal{H}_S$	0.239	0.754	0.477
		ASR WCS	0.249	0.730	0.525
Proposed with MSBG	30.33	$\mathcal{C}_S$	0.234	0.767	0.497
		$-\mathcal{H}_S$	0.233	0.768	0.499
		ASR WCS	0.249	0.731	0.526
<i>Open-set</i>					
CPC1 Baseline	-	-	0.365	0.529	0.391
Proposed with MSBG	30.93	$\mathcal{C}_S$	0.248	0.729	0.512
		$-\mathcal{H}_S$	0.246	0.734	0.512
		ASR WCS	0.253	0.717	0.530



E029 conclusions

- The uncertainty estimated by an ensemble of powerful ASR models is naturally well correlated to speech intelligibility.
- MSBG hearing loss simulation can improve performance.
- The closed-set performance of the non-intrusive E029 is quite close to the intrusive E032.
- The intrusive E032 generalise better as its performance is better in the open-set.

System	RMSE ↓	NCC ↑	KT ↑
Closed-set			
E032(intrusive)	0.231	0.773	0.498
E029(non-intrusive)	0.233	0.768	0.499
Open-set			
E032(intrusive)	0.235	0.763	0.530
E029(non-intrusive)	0.246	0.734	0.512

Thank you for your attention!

The intrusive system E032 has been open-sourced in the Clarity Challenge github repository [recipes/cpc1/e032_sheffield](https://github.com/recipes/cpc1/e032_sheffield), please check ^_^:

