





# Augmenting Images for ASR and TTS through Single-loop and Dual-loop Multimodal Chain Framework

Johanes Effendi<sup>1,2</sup>, Andros Tjandra <sup>1</sup>, Sakriani Sakti<sup>1,2</sup>, Satoshi Nakamura<sup>1,2</sup>

<sup>1</sup>Nara Institute of Science and Technology, Japan

<sup>2</sup>RIKEN, Center for Advanced Intelligence Project (AIP), Japan {johanes.effendi.ix4, andros.tjandra.ai6, ssakti, s-nakamura}@is.naist.jp

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## Outline

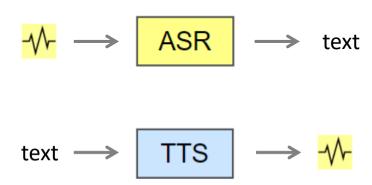


- Introduction
- Multimodal machine chain
- Experiment set-up
- Result and discussion
- Conclusion and future works

## Introduction



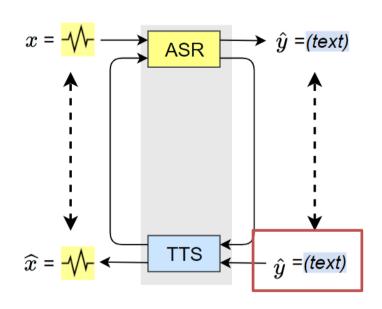
- Current state-of-the-art speech processing technology (i.e. ASR and TTS)
  - Rely on the availability of paired speech and transcription
- To improve: collect more data
- For some language, resources in such quantity are usually unavailable
- Some approaches to reduce data usage is needed



# Speech Chain (Tjandra et al., 2017)



- Enables the training of ASR and TTS to assist each other in semi-supervised learning
- Avoids the need of large amount of paired speech and text data
- But still need a large amount of unpaired speech and text data
- Speech and text is the source and target modality of ASR and TTS



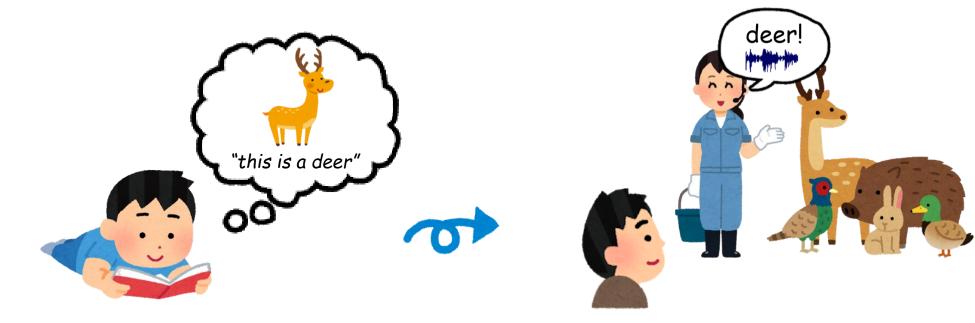
Can we improve ASR and TTS without speech or text data?



## How human perceive senses



- Human communication channels is not only auditory but also visual
- Multiple information sources are perceived together
- Able to learn even when no paired data are available (less supervision)



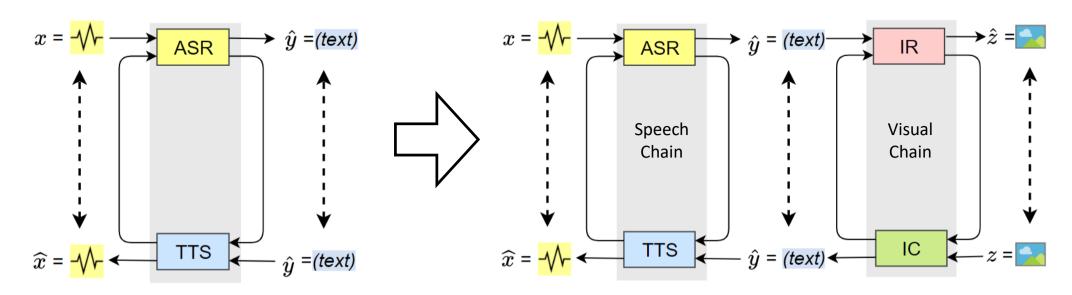
Learning by textual+visual

Learning by auditory+visual



# Multimodal Machine Chain (Effendi et al., 2019)





- Proposed to mimic overall human communication and accommodate visual modality
- Speech chain (ASR+TTS) and visual chain (IC+IR)
- Evaluated on single-speaker synthesized speech
- IR model: difficulty to handle unseen images



## Our proposed model



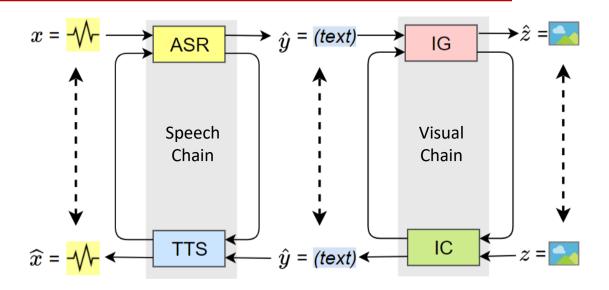
Components	Effendi et al. (2019)	This study
TTS	Single-speaker	Multi-speaker with one-shot speaker adaptation (Tjandra et al., 2018)
<b>Evaluated on</b>	Synthesized speech by Google TTS	Natural speech (Flickr8k Audio)
Image production	Image retrieval	Adversarial-based image generation
#loop	Dual-loop only	Single and dual-loop

- Image generation (IG) to handle unseen images
- Tested on multi-speaker natural speech dataset
- Multi-speaker TTS with embedding from DeepSpeaker (Li et al., 2017)
- One-shot speaker adaptation (Tjandra et al., 2018)



# [Proposed] MMC1 and MMC2





 $x = \sqrt[]{z}$   $z = \sqrt[]{z}$   $\lim_{x \to \infty} \hat{y} = (text)$   $\lim_{x \to \infty} \hat{y} = (text)$   $\lim_{x \to \infty} \hat{y} = (text)$ 

Dual-loop MMC1

Single-loop MMC2

- MMC1: dual-loop architecture with text as the bridge
- MMC2: alternative for application example on multi-source multimodal model
- Human brain process visual and auditory components of speech in a unified manner (Calvert, 2001)
  - Introduce sharing between ASR and IC -> ImgSp2Txt

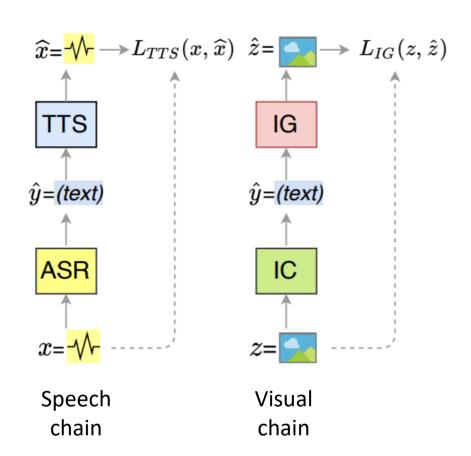


## MMC1 unrolled process:

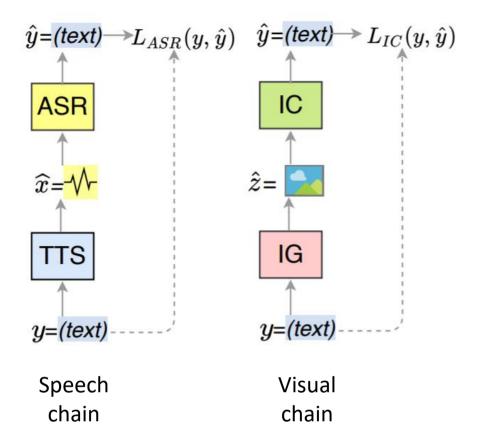
## The loop inside the speech chain and visual chain



When the input is image or speech only data



When the input is text only data

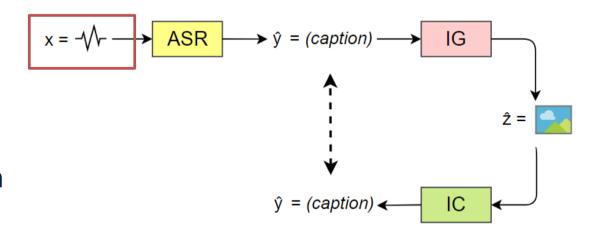


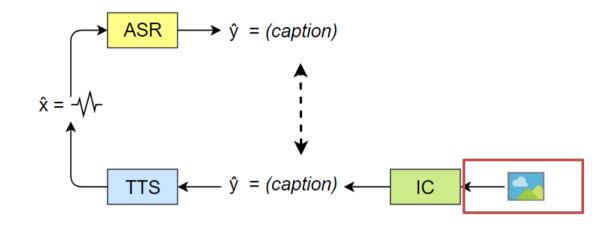
## MMC1 unrolled process:

#### Speech chain and visual chain collaboration

- Speech chain and visual chain collaborates through text modality
- The loss calculated from intermediate text
- Backpropagate the last element of the chain
- Simple filtering in text hypothesis

This is our main interest, to see if the imageonly data can help improve ASR

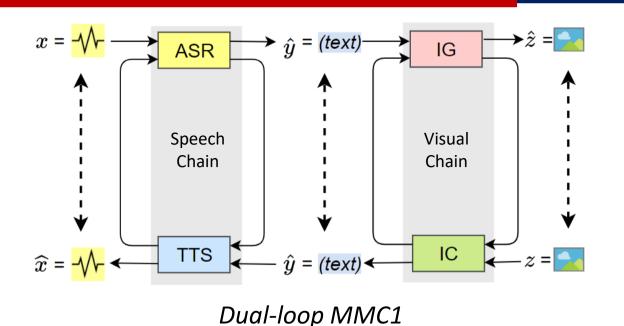


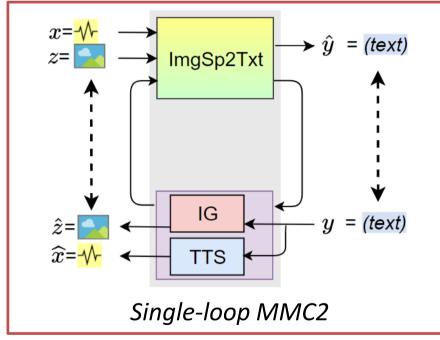




# [Proposed] MMC1 and MMC2







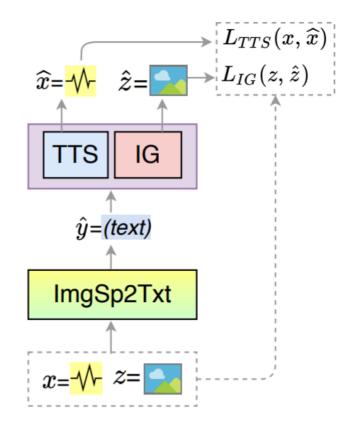
- MMC1: dual-loop architecture with text as the bridge
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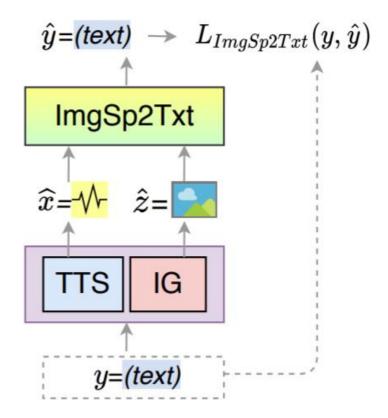
## MMC2 unrolled process



When the input is image and/or speech only data



When the input is text only data



## Models used and its evaluation metrics



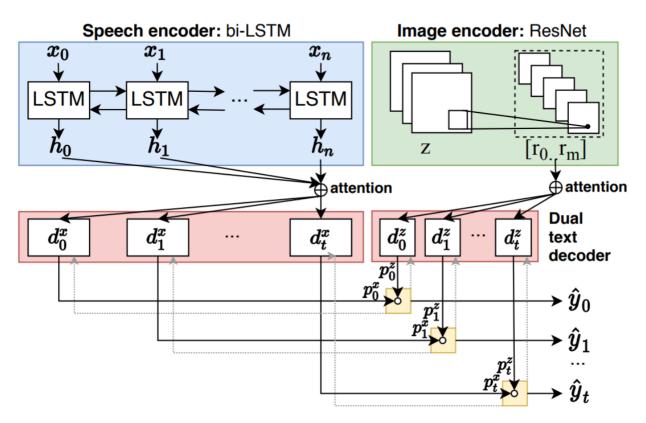
- ASR: Listen, Attend, and Spell (Chan+, 2016)
  - LSTM encoder-decoder ASR model
  - CER
- TTS: Tacotron (Wang+, 2017)
  - encoder-decoder speech synthesis model
  - using speaker embedding from DeepSpeaker (Li+, 2017) with size of 64
  - L2-norm<sup>2</sup>
- IC: Show, Attend, and Tell (Xu+, 2015)
  - modified to process 128x128
  - BLEU
- IG: AttnGAN (Xu+, 2017)
  - multistep image generation using adversarial loss
  - generate only until 128x128 image instead of 256x256
  - Inception Score



## Multimodal Chain Components



- ImgSp2Txt: average of  $p_t^x$  and  $p_t^x$  output layer probability
- When only image or speech are available, the decoder uses only the corresponding output layer.
- Trained in character-level granularity to match the best practice of ASR





## **Experiment Set-up**



- Flickr8k + audio (Rashtchian et al, 2010; Harwath and Glass, 2015)
  - 8000 photos of everyday activities and events
  - 5 captions per image, 8920 words vocabulary
  - Crowdsourced natural speech, 183 speakers, 64 hours
- We used the predefined train, dev, test subset
- But the target is to see how the proposed method perform in a single modality dataset
- So we make these data partition

Туре	Speech	Text	Image	# Image
Multimodal Paired	0	0	0	800
Multimodal Unpaired	Δ	Δ	Δ	1500
Speech only	Δ	X	X	1850
Image only	X	Х	Δ	1850

o: available paired

 $\Delta$ : available but unpaired

x: unavailable

# Result - Comparing MMC1 and MMC2



Training	Data Type	#Image	ASR (CER) ↓	IC (BLEU4) 个	TTS (L2² Norm) ↓	IG (Inception) 个
MMC 1 Dual-loop (Semi-supervised)	Multimodal (P)	800	36.35	12.75	0.77	5.90
	+ Multimodal (U)	1500	15.10	13.22	0.59	8.29
	+ Sp only (U)	1850	12.37	13.28	0.56	9.12
	+ Img only (U)	1850	12.06	13.29	0.56	9.11
Topline MMC1 (Supervised)	Multimodal (P)	6000	5.76	19.91	0.50	9.66
MMC 2 Single-loop (Semi-supervised)	Multimodal (P)	800	26.67	32.23	0.77	5.90
	+ Multimodal (U)	1500	14.88	55.15	0.65	10.12
	+ Sp only (U)	1850	13.81	58.03	0.62	10.65
	+ Img only (U)	1850	12.32	59.66	0.61	9.95
Topline MMC2 (Supervised)	Multimodal (P)	6000	5.16	79.88	0.50	9.66

ASR improvement even without speech and text data



# Discussion – Comparing MMC1 and MMC2



- ASR improvement even when using image dataset
- Sharing between ASR and IC in MMC2 yields better ASR in low-data scenario
- MMC2 ends up with 12.32 CER, on par with MMC1 12.06 CER

- Best score of MMC1 12.06 CER = 17.84 WER
- Comparable with Sun et al. (2016) = 13.81 WER
  - Fully supervised
  - Lattice rescoring algorithm
  - ASR implemented using non end-to-end method, with image to help decoding



## **Conclusion and Future Works**



- Improvements from previous multimodal machine chain:
  - adversarial-based image generation model
  - one-shot speaker adaptation
  - tested on multispeaker natural speech dataset
- Alternative single-loop multimodal chain
- Results shows that both multimodal chains:
  - enables improvement of speech processing components using an image-only dataset
  - by collaborating with image processing components
  - within multimodal machine chain architecture
- Future work: investigate various approaches of component combination

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