IPTC-Amazon collaboration

Presentation (Nov 15th, 2023)



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IPTC-Amazon final event: agenda

- Innovations in Text-to-Speech Technology at Amazon
- Possibilities of collaboration at Amazon
- Projects developed in 2022-23
 - Sign language motion generation from high level sign characteristics
 - Speaker diarization with multimodal inputs
 - Pose and spatial movement as input for dynamic content search & generation
 - Entangling AI-audio synthesis models and multimodal representations
 - Zero-shot sonorizing of video sequences



Daniel Sáez-Trigueros

Innovations in Text-to-Speech Technology at Amazon

In this presentation, Daniel will introduce the Text-to-Speech team at Amazon and talk about some of the research projects they have published during 2023. The goal is to give an overview of the kind of problems that the team is facing and the innovative solutions that they are working on to tackle them.



Short CV

Daniel Sáez Trigueros received his Bachelor's degree in Telecommunication Technical Engineering from the University of Málaga in 2013, and a Ph.D. in Machine Learning from the University of Hertfordshire in 2019. His Ph.D was done in collaboration with IDscan Biometrics Ltd and focused on the development of novel face recognition techniques using machine learning.

In 2019, Daniel joined the Text-to-Speech Research team at Amazon as an Applied Scientist. Since then, he has further developed his expertise in deep learning and generative models, working on and leading different projects related to speech synthesis. In 2023, he became an advisor to master students from Technical University of Madrid as part of an IPTC-Amazon collaboration.



Roberto Barra Chicote

Possibilities of collaboration at Amazon

Internships at Amazon for students, researchers and professors



Short CV

Roberto Barra Chicote received his Master's degree in Telecommunication at ETSIT-UPM in 2005. His Ph.D was focused on speech synthesis: Contributions to the Analysis, Design and Evaluation of Strategies for Corpus-based Emotional Speech Synthesis. His PhD obtained several national awards.

From 2009, to 2015, Roberto was assistant professor in the Speech Technology Group at ETSIT-UPM, working in several projects including speech technologies.

In 2015, Roberto joined the Text-to-Speech Research team at Amazon, and now he is Principal Scientist in this team.

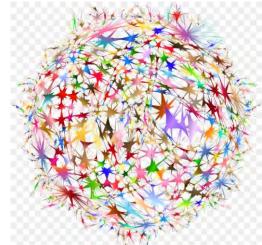


IPTC-Amazon collaboration

This collaboration has been focused on developing technologies to

extract and combine self-supervised representations for multimedia processing.

These technologies have a big potential in many areas such as content generation (audio, image, video, or sign language representation), classification, labelling or searching.





IPTC-Amazon: Team, IPTC students

- Juan Moreno Galiano (juan.moreno.galiano@alumnos.upm.es)
- María Villa Monedero (maria.villa.monedero@alumnos.upm.es)
- Laura Fernández Galindo (laura.fernandez.galindo@alumnos.upm.es)
- María Sánchez Ruiz (<u>maria.sanruiz@alumnos.upm.es</u>)
- Andrzej Daniel Dobrzycki (<u>daniel.dobrzycki@alumnos.upm.es</u>)



IPTC-Amazon: Students formation

- Introduction to **deep learning** strategies:
 - CNN, RNNs, Transformers, Adversarial and Contrastive learning, etc.
- **Tools** for deep learning:
 - CLIP or LIP and variants like WavCLIP, AudioCLIP or MotionCLIP. Also, video and audio processing tools (like OpenPose, MediaPipe, etc.).
- Framework for experimentation.
- Evaluation with several datasets.





IPTC-Amazon: Students supervision

- Every student has been supervised by
 - one researcher from IPTC and
 - another researcher from Amazon.
 - Meetings every week (aprox.).



 Joint meetings and sessions every 1.5 or 2 months to present the last achievements.



IPTC-Amazon: Results

- Web: provisional link (only direct access)
 - <u>https://iptc.upm.es/education/iptc-amazon-collaboration</u>
- **Prototypes and demonstrations** to show the main research achievements.
- Papers submissions to international conferences or journals.





IPTC-Amazon: collaboration teams

- Sign language motion generation from high level sign characteristics
 - Student: María Villa Monedero
 - Advisors: Rubén San-Segundo, Manuel Gil-Martín, Andrzej Pomirski
- Speaker diarization with multimodal inputs
 - Student: Juan Moreno Galiano
 - Advisors: Alberto Belmonte, Ivan Valles
- Pose and spatial movement as input for dynamic content search & generation
 - Student: Andrzj Daniel Dobrzycki
 - Advisors: Ana Bernardos, Daniel Saez
- Entangling AI-audio synthesis models and multimodal representations
 - Student: Laura Fernández Galindo
 - Advisors: Julián David Arias Londoño, Juan Ignacio Gódino, Luis Hernández, Giulia Comini
- Zero-shot sonorizing of video sequences
 - Student: María Sánchez Ruiz
 - Advisors: Mateo Cámara, José Luis Blanco, Luis Hernández, Adam Gabrys



IPTC-Amazon: Projects

Projects:

- Sign language motion generation from high level sign characteristics
- Speaker diarization with multimodal inputs
- Pose and spatial movement as input for dynamic content search & generation
- Entangling AI-audio synthesis models and multimodal representations
- Zero-shot sonorizing of video sequences



Sign language motion generation from high level sign characteristics



Sign language motion generation and analysis

Introduction and Objectives

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Hand configurations



Landmarks



What have we done?



• Creation of a language sign dataset



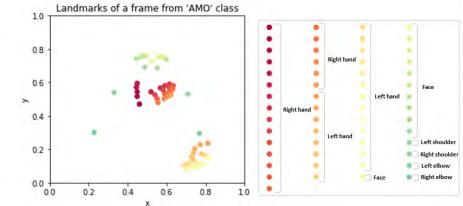
• Based on a Spanish to English translator transformer



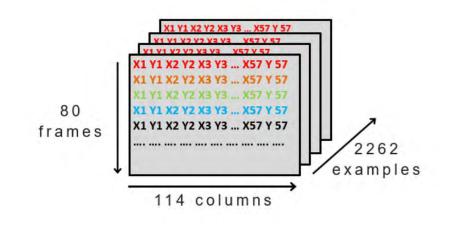
Dataset Creation

- 6.786 videos
- 24 frames per sign
- Three different points of view with three different avatars
- Mediapipe for extraction of coordinates.





- 3-Dimensional matrix
- 2.626 gestures
- 80 frames per gesture





Transformer Development and Evaluation

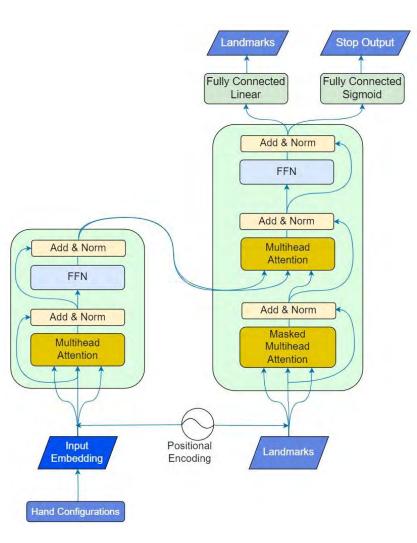
Transformer development



- Sign phonemes
- TokenandPositionEmbedding



- Landmarks
- PositionEmbedding
- Predicts landmarks



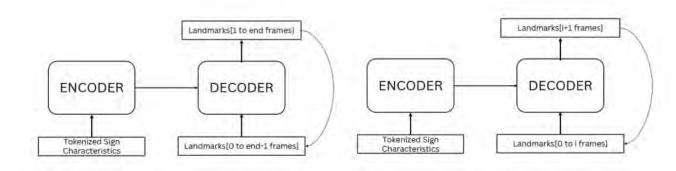


Transformer Development and Evaluation

Strategies and Evaluation

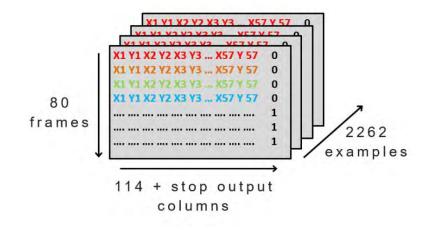


- Padding & Interpolation strategies
- Data augmentation
- Mean DTW





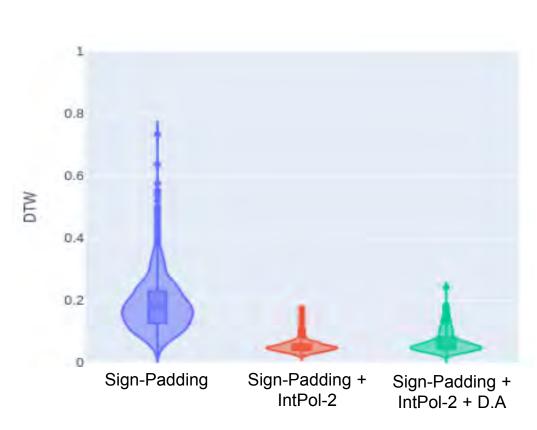
- Padding & Interpolation strategies
- Data augmentation
- ROC Curves and P-values

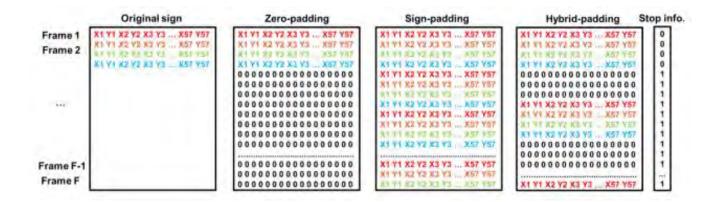




Landmarks Generation

Padding Strategies
Interpolation Strategies
Data Augmentation





	Sign-Padding	Sign-Padding + IntPol-2	Sign-Padding + IntPol2 + D.A
DTW mean	0.1880 ± 0.0857	0.0523 ± 0.0177	0.0677 ± 0.0340

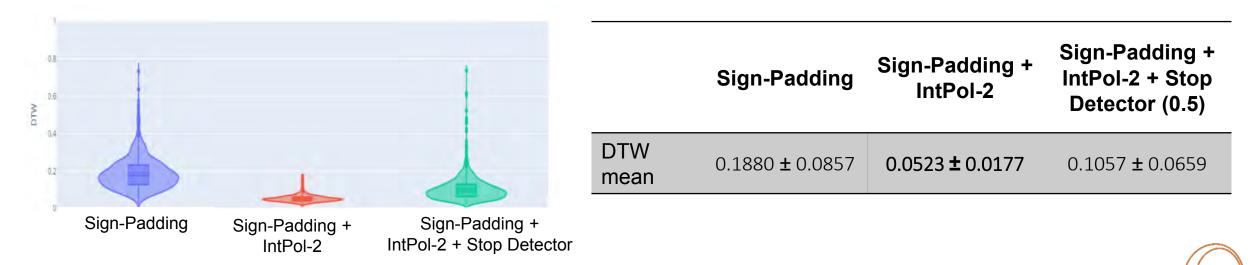


Stop Detector and Final System

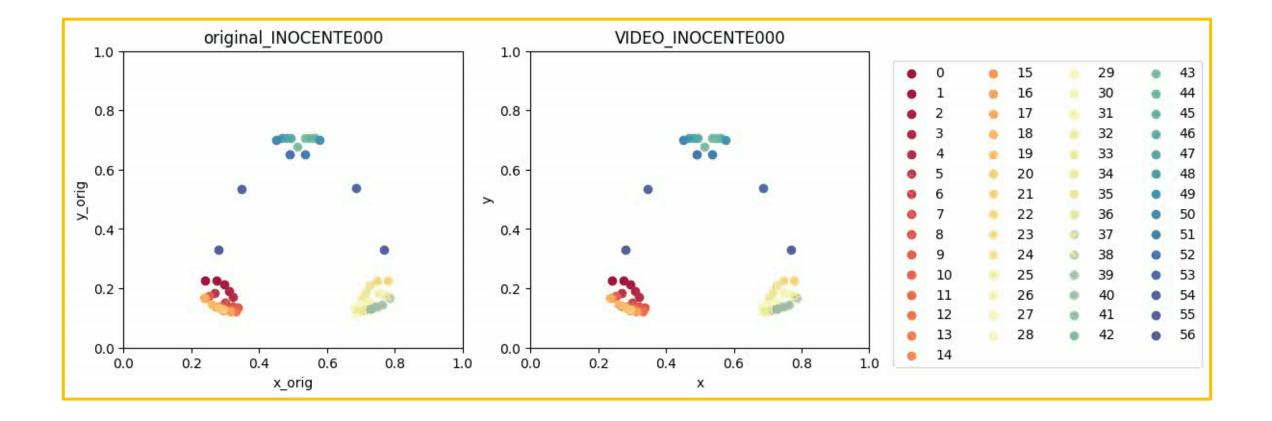
STOP Detector

	Hybrid-Padding	Hybrid-Padding + IntPol-2	Hybrid-Padding + IntPol2 + D.A
AUC	0.94025	0.94379	0.97653
P-Value	0.000097		<0.000001

🔅 Final System



Results





Journal Papers

M. Villa-Monedero, M. Gil-Martín, D. Sáez-Trigueros, A. Pomirski, and R. San-Segundo, "Sign

Language Dataset for Automatic Motion Generation", Journal of Imaging, In review, 2023.

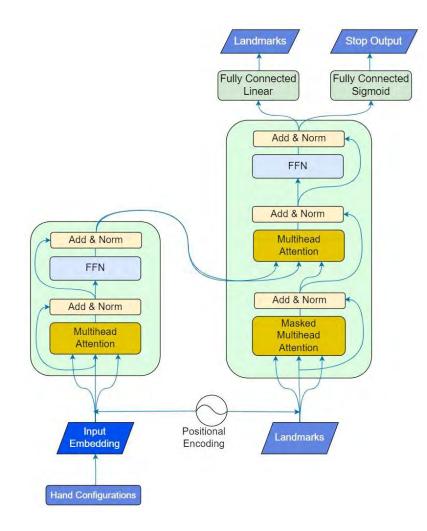
M. Gil-Martín, M. Villa-Monedero, A. Pomirski, D. Sáez-Trigueros, and R. San-Segundo, "Sign

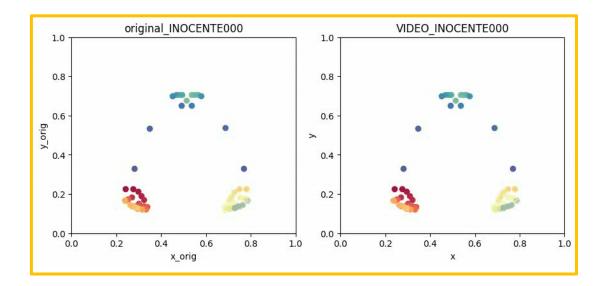
Language Motion Generation from Sign Characteristics", Sensors, In review, 2023.



Sign language motion generation and analysis

Conclusions and Greetings







Speaker diarization with multimodal inputs



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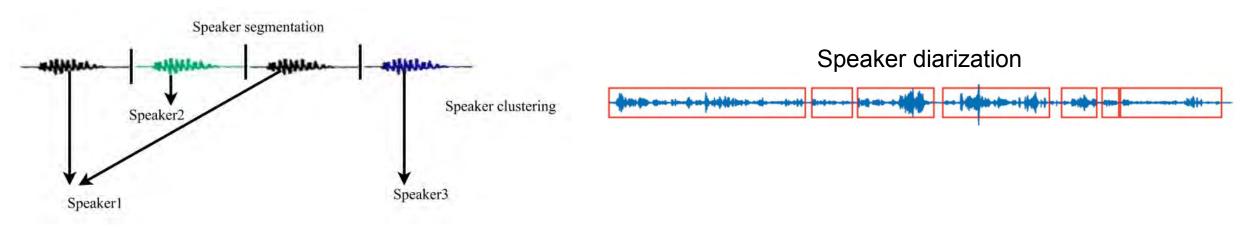
- Speaker diarization Traditional vs Multimodal
- Our case Tools and AI: Contrastive learning

✤ Dataset

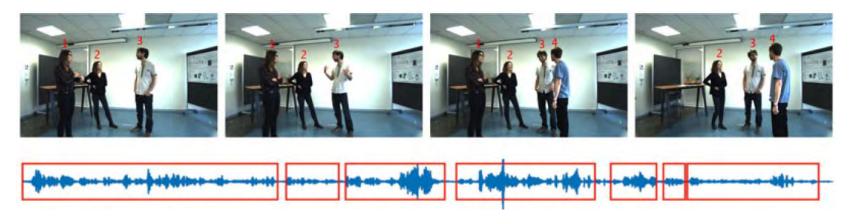
- Training and testing process
- Experiments
- ✤ Results
- Future improvements

Speaker diarization – Traditional vs Multimodal

✤ Answers to the question of "who spoke when"



Multimodal speaker diarization





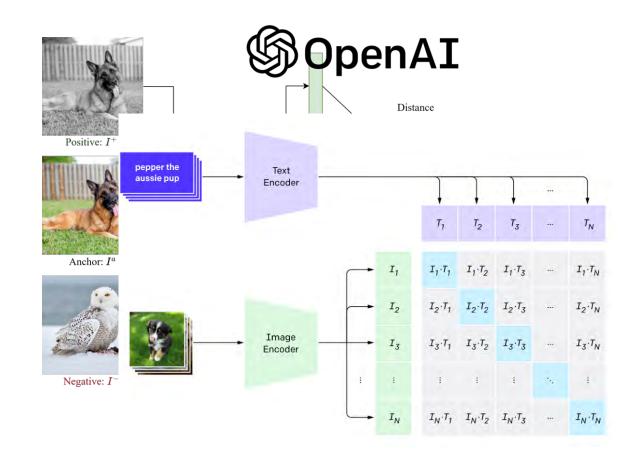
Our case – Tools and AI: Contrastive learning

We decided to focus on a concrete use case to ease the task

A video conference



Minimize the distance between similar pairs and maximize the distance of dissimilar pairs.



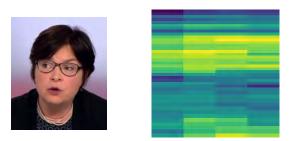


Dataset: AVSpeech



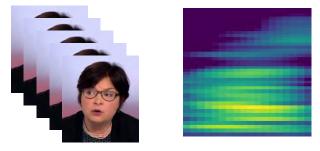
More than 200k videos of duration around 5-10 seconds

25 fps videos



31050 frames and 31050 40ms mel-spectrograms

5 fps videos

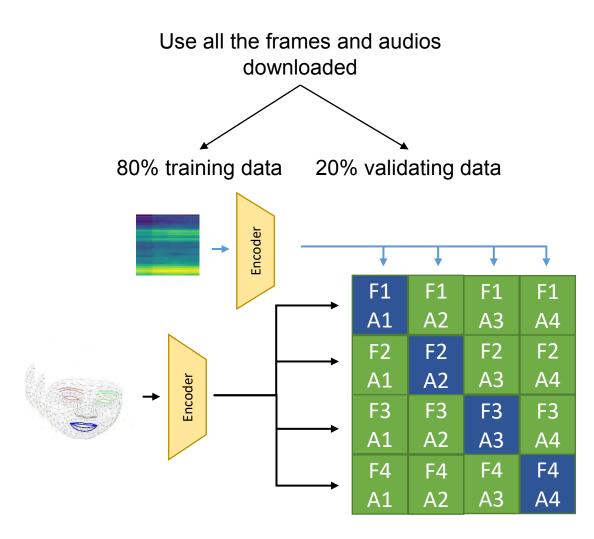


6210 5-frames sets and 6210 200ms mel-spectrograms



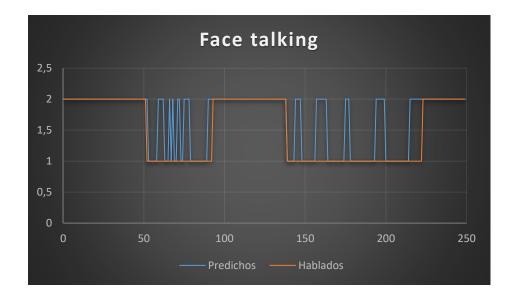
Training and testing the network

Training



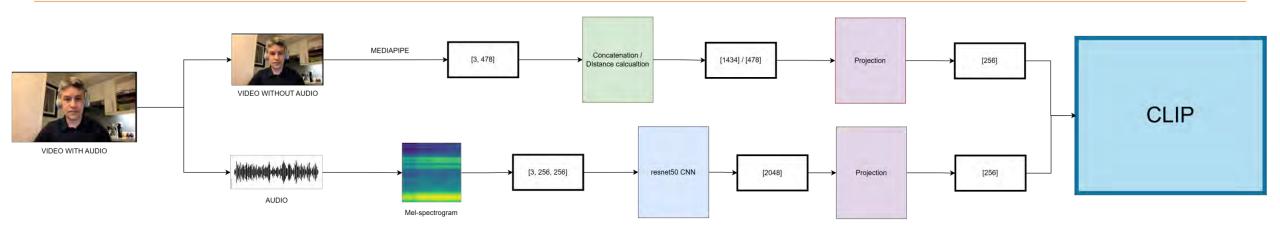
Testing

Manually label a videoconferece video and obtain the network prediction





Experiments



1. Inputs: Face cropped and 40 ms Mel-spectrogram image. Both CLIP branches with ResNet50.

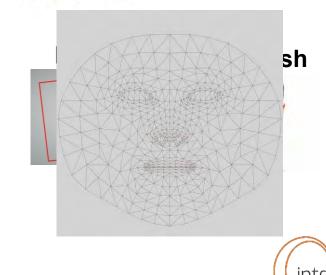
2. Inputs: Face image after MediaPipe keypoints extraction and 40 ms Mel-spectrogram image. Both CLIP branches with ResNet50.

3. Inputs: MediaPipe face keypoints (as vector) and 40 ms Melspectrogram image. DNN and ResNet50 in each CLIP Branch.

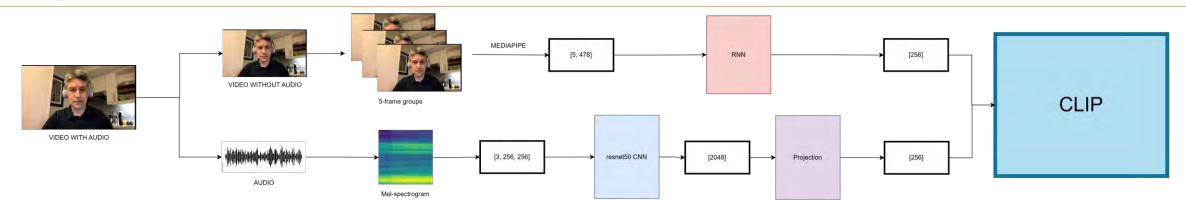
4. Inputs: MediaPipe face keypoints (as vector) normalized distance to a central point (nose) and 40 ms Mel-spectrogram image. DNN and ResNet50 in each CLIP Branch.

MediaPipe

Pose



Experiments and results



5. Inputs: 5 consecutive frames MediaPipe face Keypoints normalized distances and 200 ms Mel-Spectrogram image. RNN and ResNet50 in each CLIP branch.

Experiments	Accuracy	Comments
1 - Face and 40ms Spectrogram	<20%	It is not possible to generalize the problem to real world applications, added to the bad performance of the network
2 - Keypoints image and 40ms Spectrogram	<25%	Similar problem as in experiment 1
3 – Keypoints vector and 40ms Spectrogram	43.2%	The direct use of the keypoints values with a simpler network (DNN instead CNN) is more efficient to learn
4 – Keypoints distances and 40ms Spectrogram	82%	The distances provide information about the location and movement of different parts in the face
5 – 5 frames Keypoints distances and 200ms Spectrogram	76%	Not totally finished and reduced dataset. The RNN can learn movements in the keypoints to match better with a longer spectrogram

Future lines

- Increase the number of data to train the network using the network proposed in experiment 5. This is the main next step, as it will lead to much better results, as with the current data the testing results are "following" the real speaking moments.
- Explore the use of Transformers instead RNNs.
- ✤ Use filters to reduce the spurious misspredictions. Moving averaging, etc.
- Change the audio encoding. Some experiments with wav2vec network, but obtaining worse results than with the CNN. However, there are more encoding networks for the audio which can be explored and evaluated.

