

Offre n°2022-05062

Post-Doctoral Research Visit F/M Robust and Generalizable Deep Learning-based Audio-visual Speech Enhancement

Le descriptif de l'offre ci-dessous est en Anglais

Type de contrat : CDD

Niveau de diplôme exigé : Thèse ou équivalent

Fonction : Post-Doctorant

Contexte et atouts du poste

This project is within the framework of collaboration between the [Multispeech team](#) and the [RobotLearn team](#) at Inria. The post-doctoral researcher will be working under the co-supervision of [Mostafa Sadeghi](#) (researcher, [Multispeech team](#)), [Xavier Alameda-Pineda](#) (researcher and team leader of [RobotLearn team](#)), and [Radu Horaud](#) (senior researcher, [RobotLearn team](#)).

Work environment: [Multispeech team](#), Inria Nancy, France.

Starting date & duration: October 2022 (flexible), for a duration of one year.

Mission confiée

Background: Audio-visual speech enhancement (AVSE) refers to the task of improving the intelligibility and quality of a noisy speech signal utilizing the complementary information of visual modality (lip movements of the speaker) [1], which could be very helpful in highly noisy environments. Recently, and due to the great success and progress of deep neural network (DNN) architectures, AVSE has been extensively revisited [1]. Existing DNN-based AVSE methods are categorized into *supervised* and *unsupervised* approaches. In the former category, a DNN is trained on a large audiovisual corpus, e.g., AVSpeech [2], with diverse enough noise instances, to directly map the noisy speech signal and the associated video frames of the speaker into a clean estimate of the target speech signal. The trained models are usually very complex and contain millions of parameters. The unsupervised methods [3] follow a statistical modeling-based approach combined with the expressive power of DNNs, which involves learning the prior distribution of clean speech using deep generative models, e.g., variational autoencoders (VAEs) [4], on clean corpora such as TCD-TIMIT [5], and estimating clean speech signal in a probabilistic way. As there is no training on noise, the models are much lighter than those of supervised methods. Furthermore, the unsupervised methods have potentially better generalization performance and robustness to visual noise thanks to their probabilistic nature [6-8]. Nevertheless, these methods are very recent and significantly less explored compared to the supervised approaches.

Principales activités

Project description: In this project, we plan to devise a robust and efficient AVSE framework by thoroughly investigating the coupling between the recently proposed deep learning architectures for speech enhancement, both supervised and unsupervised, benefiting from the best of both worlds, along with the state-of-the-art generative modeling approaches. This will include, e.g., the use of dynamical VAEs [9], temporal convolutional networks (TCNs) [10], and attention-based strategies [11,12]. The main objectives of this project are summarized as follows:

1. Developing a neural architecture that identifies reliable (either frontal or non-frontal) and unreliable (occluded, extreme poses, missing) lip images by providing a normalized score at the output;
2. Developing deep generative models that efficiently exploit the sequential nature of data;
3. Integrating the developed visual reliability analysis network within the deep generative model that accordingly decides whether to utilize the visual data or not. This will provide a flexible and robust audiovisual fusion and enhancement framework.

References:

- [1] D. Michelsanti, Z. H. Tan, S. X. Zhang, Y. Xu, M. Yu, D. Yu, and J. Jensen, "An overview of deep learning-based audio-visual speech enhancement and separation," *IEEE/ACM Transactions on Audio, Speech, and Language Processing*, vol. 29, 2021.
- [2] A. Ephrat, I. Mosseri, O. Lang, T. Dekel, K. Wilson, A. Hassidim, W.T. Freeman, M. Rubinstein, "Looking-to-Listen at the Cocktail Party: A Speaker-Independent Audio-Visual Model for Speech Separation," *SIGGRAPH* 2018.

- [3] M. Sadeghi, S. Leglaive, X. Alameda-Pineda, L. Girin, and R. Horaud, "Audio-visual speech enhancement using conditional variational auto-encoders," *IEEE/ACM Transactions on Audio, Speech and Language Processing*, vol. 28, pp. 1788–1800, 2020.
- [4] D. P. Kingma and M. Welling, "Auto-encoding variational Bayes," in *International Conference on Learning Representations (ICLR)*, 2014.
- [5] N. Harte and E. Gillen, "TCD-TIMIT: An Audio-Visual Corpus of Continuous Speech," *IEEE Transactions on Multimedia*, vol.17, no.5, pp.603-615, May 2015.
- [6] M. Sadeghi and X. Alameda-Pineda, "Switching variational autoencoders for noise-agnostic audio-visual speech enhancement," in *ICASSP*, 2021.
- [7] Z. Kang, M. Sadeghi, R. Horaud, "Face Frontalization Based on Robustly Fitting a Deformable Shape Model to 3D Landmarks," in *International Conference on Computer Vision (ICCV) Workshops*, Montreal - Virtual, Canada, Oct. 2021, pp. 1–16.
- [8] Z. Kang, M. Sadeghi, R. Horaud, X. Alameda-Pineda, J. Donley, and A. Kumar, "The impact of removing head movements on audio-visual speech enhancement," in *ICASSP*, 2022, pp. 1–5.
- [9] L. Girin, S. Leglaive, X. Bie, J. Diard, T. Hueber, and X. Alameda-Pineda, "Dynamical variational autoencoders: A comprehensive review," *Foundations and Trends in Machine Learning*, vol. 15, no. 1-2, 2021.
- [10] C. Lea, R. Vidal, A. Reiter, and G. D. Hager. "Temporal convolutional networks: A unified approach to action segmentation." In *European Conference on Computer Vision*, pp. 47-54. Springer, Cham, 2016.
- [11] A. Vaswani, N. Shazeer, N. Parmar, J. Uszkoreit, L. Jones, A. N Gomez, L. Kaiser, and I. Polosukhin, "Attention is all you need," in *Advances in neural information processing systems*, 2017, pp. 5998–6008.
- [12] J. Jiang, G. G Xia, D. B Carlton, C. N Anderson, and R. H Miyakawa, "Transformer VAE: A hierarchical model for structure-aware and interpretable music representation learning," in *ICASSP*, 2020, pp. 516–520.

Compétences

Requirements & skills: The preferred profile is described below.

- Master's degree, or equivalent, in the field of speech/audio processing, computer vision, machine learning, or in a related field,
- Ability to work independently as well as in a team,
- Solid programming skills (Python, PyTorch), and deep learning knowledge,
- Good level of written and spoken English.

Avantages

- Subsidized meals
- Partial reimbursement of public transport costs
- Leave: 7 weeks of annual leave + 10 extra days off due to RTT (statutory reduction in working hours) + possibility of exceptional leave (sick children, moving home, etc.)
- Possibility of teleworking (after 6 months of employment) and flexible organization of working hours
- Professional equipment available (videoconferencing, loan of computer equipment, etc.)
- Social, cultural and sports events and activities
- Access to vocational training
- Social security coverage

Rémunération

Salary: 2653€ gross/month

Informations générales

- **Thème/Domaine :** Langue, parole et audio Statistiques (Big data) (BAP E)
- **Ville :** Villers lès Nancy
- **Centre Inria :** [Centre Inria de l'Université de Lorraine](#)
- **Date de prise de fonction souhaitée :** 2022-10-03
- **Durée de contrat :** 12 mois
- **Date limite pour postuler :** 2022-07-10

Contacts

- **Équipe Inria :** [MULTISPEECH](#)
- **Recruteur :**
Sadeghi Mostafa / mostafa.sadeghi@inria.fr

A propos d'Inria

Inria est l'institut national de recherche dédié aux sciences et technologies du numérique. Il emploie 2600 personnes. Ses 215 équipes-projets agiles, en général communes avec des partenaires académiques, impliquent plus de 3900 scientifiques pour relever les défis du numérique, souvent à l'interface d'autres disciplines. L'institut fait appel à de nombreux talents dans plus d'une quarantaine de métiers différents. 900 personnels d'appui à la recherche et à l'innovation contribuent à faire émerger et grandir des projets scientifiques ou entrepreneuriaux qui impactent le monde. Inria travaille avec de nombreuses entreprises et a accompagné la création de plus de 200 start-up. L'institut s'efforce ainsi de répondre aux enjeux de la transformation numérique de la science, de la société et de l'économie.

L'essentiel pour réussir

How to apply: Interested candidates are encouraged to contact Mostafa Sadeghi (mostafa.sadeghi@inria.fr), Xavier Alameda-Pineda (xavier.alameda-pineda@inria.fr), and Radu Horaud (radu.horaud@inria.fr), and upload the required documents (CV, transcripts, motivation letter, and recommendation letters) to the dedicated Inria platform.

Attention: Les candidatures doivent être déposées en ligne sur le site Inria. Le traitement des candidatures adressées par d'autres canaux n'est pas garanti.

Consignes pour postuler

Sécurité défense :

Ce poste est susceptible d'être affecté dans une zone à régime restrictif (ZRR), telle que définie dans le décret n°2011-1425 relatif à la protection du potentiel scientifique et technique de la nation (PPST). L'autorisation d'accès à une zone est délivrée par le chef d'établissement, après avis ministériel favorable, tel que défini dans l'arrêté du 03 juillet 2012, relatif à la PPST. Un avis ministériel défavorable pour un poste affecté dans une ZRR aurait pour conséquence l'annulation du recrutement.

Politique de recrutement :

Dans le cadre de sa politique diversité, tous les postes Inria sont accessibles aux personnes en situation de handicap.