

Daniel STOLLER

Ph.D. Candidate

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PERSONAL DATA

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RESEARCH INTERESTS

I aim to develop **machine learning** models that can **generalise from small amounts of training data**. My research specifically focuses on **multi-task learning**, **semi-supervised learning** using **generative adversarial networks**, and **transfer learning**. I apply these techniques to tasks in **music information retrieval**, **computer vision** and **NLP**.

SELECTED PROJECTS

Wave-U-Net for end-to-end audio source separation

Models for audio source separation usually operate on the magnitude spectrum, which ignores phase information and makes performance dependant on hyper-parameters for the spectral front-end. We propose the Wave-U-Net, an **end-to-end convolutional neural network** based on the U-Net that **directly operates on raw waveforms**, and show that **it can outperform a state-of-the-art spectrogram-based approach**.

Paper: “Wave-U-Net: A Multi-Scale Neural Network for End-to-End Audio Source Separation”

Code (Python, Tensorflow): www.github.com/f90/Wave-U-Net

Causal U-Nets for efficient sequence modelling

Convolutional networks developed for sequence modelling such as Wavenet achieve good performance, but are prohibitively slow. We exploit that **many important features are only slowly changing** and propose the Seq-U-Net that computes features at multiple time-scales. Our model achieves **comparable performance in text and audio generation** while using **significantly less memory and computation time**.

Paper: “Seq-U-Net: A One-Dimensional Causal U-Net for Efficient Sequence Modelling”

Code (Python, Pytorch): <https://github.com/f90/Seq-U-Net>

Adversarial semi-supervised audio source separation

State-of-the-art approaches for audio source separation use supervised deep learning, which requires multi-track datasets for training. We developed an **unsupervised training technique** based on **generative adversarial networks** that makes of more easily available solo recordings of sources and unpaired audio mixtures and can be added to previous approaches for improved separation performance.

Paper: “Adversarial semi-supervised audio source separation applied to singing voice extraction”

Code (Python, Pytorch): www.github.com/f90/AdversarialAudioSeparation

Semi-supervised training of generative adversarial networks (GANs)

While GANs can generate as well as translate images and other complex data with high quality, they need large amounts of training data. We show how GANs can be elegantly adapted to also leverage training samples that are partially incomplete or unlabelled. The resulting “FactorGAN” outperforms the standard GAN on image generation as well as image segmentation, especially if only few complete samples are given.

Paper: “Training GANs from Incomplete Observations using Factorised Discriminators”

Code (Python, Pytorch): <https://github.com/f90/FactorGAN>

PUBLICATIONS

- ArXiv Preprint Seq-U-Net: A One-Dimensional Causal U-Net for Efficient Sequence Modelling
Daniel Stoller, Mi Tian, Sebastian Ewert, Simon Dixon
- ICLR 2020 Training Generative Adversarial Networks from Incomplete Observations using Factorised Discriminators
Daniel Stoller, Sebastian Ewert, Simon Dixon
- ICASSP 2019 Conference paper End-to-end Lyrics Alignment for Polyphonic Music Using An Audio-to-Character Recognition Model
Daniel Stoller, Simon Durand, Sebastian Ewert
- ICLR 2019 Workshop paper GAN-based Generation and Automatic Selection of Explanations for Neural Networks
Saumitra Mishra, Daniel Stoller, Emmanouil Benetos, Bob L. Sturm, Simon Dixon
- Interspeech 2019 Conference paper Ensemble Models for Spoofing Detection in Automatic Speaker Verification
Bhusan Chettri, Daniel Stoller, Veronica Morfi, Marco A. Martínez Ramírez, Emmanouil Benetos, Bob L. Sturm
- EvoMUSART 2019 Conference paper Evolutionary Multi-objective Training Set Selection of Data Instances and Augmentations for Vocal Detection
Igor Vatolkin, Daniel Stoller
- ISMIR 2018 Conference paper Wave-U-Net: A Multi-Scale Neural Network for End-to-End Source Separation
Daniel Stoller, Sebastian Ewert, Simon Dixon
- LVA/ICA 2018 Conference paper Jointly Detecting and Separating Singing Voice: A Multi-Task Approach
Daniel Stoller, Sebastian Ewert, Simon Dixon
- MLSP 2018 Workshop paper Detection of Cut-Points for Automatic Music Rearrangement
Daniel Stoller, Vincent Akkermans, Simon Dixon
- ICASSP 2018 Conference paper Adversarial Semi-Supervised Audio Source Separation applied to Singing Voice Extraction
Daniel Stoller, Sebastian Ewert, Simon Dixon
- JNMR 2018 Journal article Intuitive and Efficient Computer Aided Music Rearrangement with Optimised Processing of Audio Transitions
Daniel Stoller, Igor Vatolkin, Heinrich Müller
- ISMIR 2016 Conference paper Analysis and Classification of Phonation Modes in Singing
Daniel Stoller, Simon Dixon
- ECDA 2013 Conference paper Impact of Frame Size and Instrumentation on Chroma-based Automatic Chord Recognition
Daniel Stoller, Matthias Mauch, Igor Vatolkin, Claus Weihs
Received “Best Paper Award”

EDUCATION

CURRENT	Ph.D. Candidate
SEP. 2015	Queen Mary University of London Topic: “Machine listening with limited annotations” First supervisor: Prof. Simon Dixon Second supervisor: Dr. Emmanouil Benetos
AUG. 2015	Master of Computer Science, Technical University of Dortmund Thesis: “Constraint-based rearrangement of music” Final mark: 1.0 with distinction (<i>equivalent to 1</i>) Advisor: Prof. Dr. Heinrich MÜLLER
FEB. 2013	Bachelor of Computer Science, Technical University of Dortmund Secondary subject: Economics Thesis: “Automatische Segmentierung dentaler CT-Daten” Final mark: 1.8 (<i>equivalent to 2.1</i>) Advisor: Prof. Dr. Heinrich MÜLLER

WORK EXPERIENCE

SEP. 2018	Research Intern
JUNE 2018	Spotify, London <i>Large scale deep learning for music understanding</i> Developed novel deep neural networks for lyrics recognition and alignment Greatly advanced state-of-the-art in terms of accuracy Work published in ICASSP 2019 conference paper Patent filed for the training procedure
JAN. 2017	Senior Teaching Assistant
SEP. 2016	Queen Mary University of London <i>Module ”Procedural programming”</i>
SEP. 2016	Research Intern
APRIL 2016	HeresyAI, London <i>Finding cut points in music using deep learning</i> Research project as part of the PhD Programme Developed an automatic system for identifying cut points in music enabling automatic music remixing in the app Mashtraxx
DEC. 2015	Teaching Assistant
SEP. 2015	Queen Mary University of London <i>Module ”Computer Systems and Networks”</i>
AUG. 2015	Student Assistant
MAY 2011	Chair for Algorithm Engineering, TU Dortmund <i>Research in music information retrieval</i>

SCHOLARLY REVIEWING ACTIVITIES

- Regular reviewer for the ISMIR conference (2015, 2018, 2019)
- IEEE Journal of Selected Topics in Signal Processing
- PeerJ Computer Science Journal

PATENTS

Systems and Methods for Aligning Lyrics using a Neural Network (Patent filed)

TECHNICAL SKILLS

FRAMEWORKS AND LIBRARIES: Tensorflow, Pytorch, Docker, Singularity, JUCE, openCV
PROGRAMMING LANGUAGES: Python, Java, C++, R, MATLAB, C#, C, SQL
SOFTWARE DEVELOPMENT: UML modelling, efficient algorithms and data structures,
IT project management

LANGUAGES

English (Fluent), German (Native), Chinese & Latin (Basic Knowledge)

INTERESTS AND ACTIVITIES

- Music: Drums and vocals in a band, keyboard and guitar
- Technology: Video recording & editing
- Sports: Badminton, Biking