

Many-To-Many Audio Spectrogram Transformer: Transformer for Sound Event Localization and Detection

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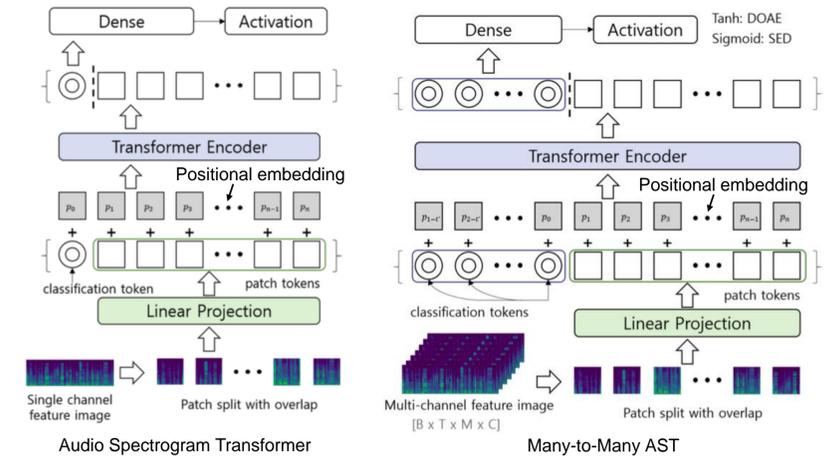
Abstract

Over the past few years, convolutional neural networks (CNNs) have been established as the core architecture for audio classification and detection. Recently, Transformers, which are pure attention-based architectures, have achieved excellent performance in various fields, showing that CNNs are not essential. In this paper, we investigate the reliance on CNNs for sound event localization and detection by introducing the Many-to-Many Audio Spectrogram Transformer (M2M-AST), a pure attention-based architecture. We adopt multiple classification tokens in the Transformer architecture to easily handle various output resolutions.

Proposed Method

Many-to-Many Audio Spectrogram Transformer (M2M-AST)

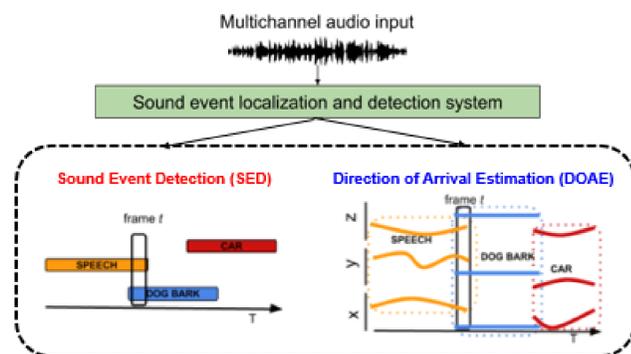
- M2M-AST focus on applying standard Transformer architecture for SELD
- Audio Spectrogram Transformer (AST) [3]
 - Patch embedding (token)** is extracted from small image patch through linear projection
 - Classification token** is an extra learnable embedding to perform classification
 - Positional embedding** is a learnable embedding to make spatial information between patches
- Difference compared to AST:
 - Multi-channel feature image** images are required to obtain spatial location information
 - Multiple classification tokens** are used to make a series of output rather than single output



DCASE Challenge Task 3

Sound Event Localization and Detection

- SELD recognizes the sound event and its direction simultaneously
- Input:
 - Directional microphone recordings from a tetrahedral array
 - First-order Ambisonic (FOA) recordings
- Output:
 - Active sound event
 - Onset/offset
 - Spatial location



System overview of sound event localization and detection

Result & Ablation Study

Feature and label configuration

| | Format | Feature | # Channels (C) | Label |
|------|------------------|--------------------------|----------------|----------------------------|
| SED | Microphone array | Logmel | 1 | Multi label binarization |
| DOAE | Ambisonic | Logmel, intensity vector | 7 | Cartesian coordinate (xyz) |

Model configuration

| | Task | Pre-trained model | Loss |
|----------|------|-------------------|-----------------|
| M2M-AST1 | SED | DeiT | BCE |
| M2M-AST2 | SED | M2M-AST1 | soft f-loss [1] |
| M2M-AST3 | DOAE | DeiT | MSE |
| M2M-AST4 | DOAE | M2M-AST3 | masked MSE |

Experimental results for dev

| | # Params | ER ₂₀ ^o | F ₂₀ ^o | LE _{CD} | LR _{CD} |
|-----------------------|----------|-------------------------------|------------------------------|------------------|------------------|
| CRNN (Baseline FOA) | 0.5M | 0.69 | 33.9 % | 24.1° | 43.9 % |
| CRNN (Baseline-Large) | 184M | 0.65 | 45.6 % | 22.6° | 55.0 % |
| CRNN [2] | 14M | 0.65 | 48.3 % | 22.0° | 62.6 % |
| M2M-AST1&3 | 172M | 0.55 | 62.6 % | 17.5° | 74.0 % |
| M2M-AST1&4 | 172M | 0.52 | 64.4 % | 16.0° | 74.0 % |
| M2M-AST2&3 | 172M | 0.52 | 64.0 % | 17.7° | 74.7 % |
| M2M-AST2&4 | 172M | 0.50 | 65.7 % | 16.3° | 74.7 % |

- All results are based on logmel energy and intensity vectors as input features
- The proposed pure transformer model outperforms the CRNN-based models listed in the table

Batch size and frame length

| # Batch | SED (F ₁ , LR _{CD}) | | | DOAE (LE _{CD}) | | |
|-----------|--|--------------|--------------|--------------------------|-------|--------------|
| | 1 sec | 2 sec | 3 sec (Used) | 1 sec | 2 sec | 3 sec (Used) |
| 24 (Used) | (68.3, 66.3) | (75.0, 73.2) | (74.0, 74.0) | 26.3° | 22.2° | 21.8° |
| 48 | (69.5, 70.9) | (75.7, 72.1) | (75.2, 73.6) | 27.9° | 23.1° | 23.0° |
| 96 | (70.7, 70.3) | (75.8, 68.7) | - | 27.0° | 24.4° | - |

Patch split with overlap

| | # Patches | SED (F ₁ , LR _{CD}) | DOAE (LE _{CD}) |
|------------------|-----------|--|--------------------------|
| No Overlap | 144 | (71.6, 60.2) | 27.3° |
| Overlap-2 | 189 | (73.8, 68.6) | 24.6° |
| Overlap-4 | 240 | (74.1, 70.6) | 24.1° |
| Overlap-6 (Used) | 348 | (74.0, 74.0) | 21.8° |
| Overlap-8 | 540 | (74.9, 72.5) | 21.0° |

Output resolution

| Output resolution | Output size (t') | SED (F ₁ , LR _{CD}) | DOAE (LE _{CD}) |
|-------------------|------------------|--|--------------------------|
| 25 ms | 120 | (75.3, 73.8) | 22.2° |
| 33.3 ms | 90 | (76.5, 75.1) | 22.1° |
| 50 ms | 60 | (74.4, 72.8) | 22.7° |
| 100 ms (Used) | 30 | (74.0, 74.0) | 21.8° |

Pre-training and loss function

| | Pre-trained model | Loss | SED (F ₁ , LR _{CD}) | DOAE (LE _{CD}) |
|-------------------------------|-------------------|-------------|--|--------------------------|
| No pre-train (SED) | - | BCE | (60.4, 54.5) | - |
| ImageNet pre-train (M2M-AST1) | DeiT | BCE | (74.0, 74.0) | - |
| SELD pre-train (M2M-AST2) | M2M-AST1 | soft f-loss | (75.8, 74.7) | - |
| No pre-train (DOAE) | - | MSE | - | 22.5 |
| ImageNet pre-train (M2M-AST3) | DeiT | MSE | - | 21.8 |
| SELD pre-train (M2M-AST4) | M2M-AST3 | masked MSE | - | 19.1 |

Conclusion

In this paper, we describe how to apply the standard Transformer architecture to SELD. As a consequence, we introduce M2M-AST, a pure Transformer model for SELD. Existing SELD networks have commonly used hybrid architectures that combine CNNs with RNNs or self-attention layers. We empirically show that M2M-AST can replace these hybrid networks in SELD, SED, and DOAE. The Experimental results represent the potential of a pure Transformer to lower the reliance on CNNs in SELD. Traditional neural networks use pooling layers to change the output shape. However, due to the pooling size of this pooling layer, the output resolution cannot be configured freely. On the other hand, M2M-AST has the advantage of being able to easily design to have a variety of output resolutions.

Reference

- [1] T. Tanaka et al., "F-measure based end-to-end optimization of neural network keyword detectors," in 2018 Asia-Pacific Signal and Information Processing Association Annual Summit and Conference (APSIPA ASC), 2018, pp. 1456-1461.
- [2] T. N. T. Nguyen et al., "Dcase 2021 task 3: Spectrotemporally-aligned features for polyphonic sound event localization and detection," DCASE2021 Challenge, Tech. Rep., November 2021.
- [3] Gong et al., AST: Audio Spectrogram Transformer, Interspeech 2021