Audio Classification - Model layer

Hugging Face model zoo options (good starting checkpoints)

Transformers on spectrograms (patch-based ViTs)

- AST (Audio Spectrogram Transformer) MIT/ast-finetuned-audioset-10-10 4593 Multi-label Audioset head; strong general audio tagging baseline.
- PaSST (Patchout Spectrogram Transformer) e.g.,
 kkoutini/passt_s_kd_ast10_10 Patchout regularization → efficient + robust on long clips.
- HTS-AT (Token-Semantic Transformer for Audio Tagging) —
 TencentGameMate/chinese-htsat (family) Strong music/environmental tagging;
 pairs well with CLAP text heads.
- Audio-MAE (Masked Autoencoding pretrain on spectrograms) e.g., MIT/ssastbase-patch400 (SSAST/AudioMAE family) Self-supervised pretraining → label-efficient fine-tuning.

Self-supervised waveform encoders (SSL)

- Wav2Vec2 / HuBERT / WavLM facebook/wav2vec2-base, superb/hubert-large-superb-er, microsoft/wavlm-base-plus Encode raw wave; add small classifier head for KWS/ASC/emotion.
- **BEATs** microsoft/BEATs / microsoft/BEATs—iter3 SSL optimized for non-speech acoustic events; strong ESC-50/ASC results.

CNN baselines (fast, small, deployable)

- PANNs (CNN14/ResNet22) qiuqiangkong/panns_cnn14 Mature CNN family; good speed/quality trade-off.
- YAMNet (MobileNet-V1) ports on HF (search "yamnet") Tiny, mobile-friendly; handy for real-time tagging.

Audio-text contrastive (zero-shot & tagging)

- CLAP (Contrastive Language-Audio Pretraining) laion/clap-htsat-unfused / laion/clap-htsat-fused Zero-shot label search via text prompts; also fine-tunes to tags.
- AudioCLIP community ports (search "audioclip") Image/audio/text shared space; useful for multimodal label spaces.

Task-specific heads

- Keyword spotting (Speech Commands) superb/wav2vec2-base-superb-ks
- Acoustic scenes dcase2020/task1a-baseline-* (various community ports)

Music tagging — M-A-P/MERT-v1-95M (music spectral transformer), mtg-jamendo-*
adapters

Pick by use-case: **Audioset-style multi-label** \rightarrow AST/PaSST/HTS-AT/BEATs; **KWS** \rightarrow Wav2Vec2/WavLM small; **On-device** \rightarrow YAMNet/PANNs; **Zero-shot** \rightarrow CLAP.

Architectural innovations (what actually moves the needle)

- Spectrogram ViTs (AST/PaSST/HTS-AT): Convert log-mels to patch tokens, use selfattention to model long contexts; patchout (PaSST) randomly drops patches for regularization & speed. Attention pooling over time/freq replaces fixed global pooling for better temporal localization.
- Self-supervised waveform encoders (Wav2Vec2/HuBERT/WavLM/BEATs): Pretrain
 on raw audio with contrastive/masked objectives → strong universal features;
 downstream = a small classification head. BEATs tailors SSL to non-speech acoustic
 semantics.
- Weak-label learning & MIL pooling (AudioSet/FSD50K): Use segment-level tokens
 with MIL/attention pooling to aggregate clip-level predictions; handles noisy/weak labels
 and long clips.
- Audio-text contrastive (CLAP/AudioCLIP): Align audio and natural-language labels → zero-shot classification; lets you "prompt" new classes without retraining.
- Learnable front-ends (LEAF, SincNet): Replace fixed mel filters with learnable filterbanks, improving robustness across domains/devices.
- Multi-scale & efficient designs: Dilated CNNs, temporal pooling pyramids, and patch subsampling (PaSST) keep latency + memory low for deployment.