

Acoustic scene and events recognition:

how similar is it to speech recognition and music genre/instrument recognition ?

G. Richard DCASE 2016

Thanks to my collaborators: S. Essid, R. Serizel, V. Bisot



Content

Some tasks in audio signal processing:

- What is scene recognition and sound event recognition ?
- What is speech recognition/speaker recognition/Music genre recognition,...?
- How similar are the different problems ?
- Are the tasks difficult for humans ?
- (Very) Brief historical overview of speech/audio processing
- Looking at recent trends for acoustic scenes recognition (DCASE2016)
- A recent and specific approach
- Discussion/Conclusion



Acoustic scene and sound event

Some example of acoustic scenes



Some example of sound events





Acoustic scene and sound event recognition

Acoustic scene recognition:

 « associating a semantic label to an audio stream that identifies the environment in which it has been produced »



 Related to CASA (*Computational* Auditory Scene Recognition) and SoundScape cognition (*psychoacoustics*)

D. Barchiesi, D. Giannoulis, D. Stowell and M. Plumbley, « Acoustic Scene Classification », IEEE Signal Processing Magazine [16], May 2015



Acoustic scene and sound event recognition

Sound event recognition

 "aims at transcribing an audio signal into a symbolic description of the corresponding sound events present in an auditory scene".



Applications of scene and events recognition

- Smart hearing aids (Context recognition for adaptive hearing-aids, Robot audion,..)
- Security (see for example the LASIE project)
- indexing,
- sound retrieval,
- predictive maintenance,
- bioacoustics,
- environment robust speech reco,
- ederly assistance



Use Case 3: The Missing Person: <u>http://www.lasie-project.eu/use-cases/</u>

14/09/2016



Is « Acoustic Scene/Event Recognition » just the same as

- Speech recognition ?
- Speaker recognition ?
- Music genre recognition ?
- Music instrument reccognition ?



. . .



From Speech to Text



« I am very happy to be here »

Input is an audio signal Output: sequence of words Associates an « acoustic recognition » model and a « language model

Acoustic model:

- Classification of an audio stream in 35 classes (« phonemes ») ... but many more if triphones are considered (even with *tied-states*)

- Class should be independent of the speaker and of pitch





Recognizing who speaks



« Tuomas Virtanen »

Input is an audio signal Output: name of a person No language model

Acoustic model:

- Classification of an audio stream in N classes (« speakers »)
- Class should be independent of the individual events (phonems) pronounced



What is Music genre recognition ?

From music to genre label



Input is an audio signal Output: Genre of the music No language model, but hierarchical model possible

Acoustic model:

- Classification of an audio stream in N classes (« genre »)
- Class should be (more or less) independant of the individual events (instruments, pitch, harmony, ...).



What is Music instrument recognition ?

From music to instrument labels



« Tenor saxophone, Bass, piano »

Input is an audio signal Output: name of the instrument playing concurrently No language model, but hierarchical model possible

Acoustic model:

- Classification of an audio stream in N classes (« instruments »)
- Multiple classes active concurrently
- Class should be (rather) independant of pitch.



Is « Acoustic Scene/Event Recognition » as difficult for humans as

- Speech recognition ?
- Speaker recognition ?
- Music genre recognition ?
- Music instrument recognition ?



. . .

Complexity of the tasks for humans

Speech recognition :

- 0.009% error rate for connected digits
- 2 % error rate for non sense sentences (1000 words vocabulary)
- Phoneme recognition (CVC or VCV) in noise: 25% error rate at -10db SNR

Speaker recognition

 About 1.3% of False Alarm and 3% Misses in a task « are the two speech signals from the same speaker ? »

R. Lippmann, Speech recognition by machines and humans, Speech Communication, Vol. 22, No 1, 1997
B. Meyer & al. "Phoneme confusions in human and automatic speech recognition", Interspeech 2007
W. Shen & al., "Assessing the speaker recognition performance of naive listeners using mechanical turk," in Proc. of ICASSP 2011

Complexity of the tasks for humans

Music Genre recognition

 55% accuracy (on average) for 19 musical genres including « Electronic&Dance", "Hip-Hop », « Folk » but also « easylistening », « vocals »

Music instrument recognition

 46% for isolated tones to 67 % accuracy for 10s phrases for 27 instruments

Sound scenes recognition

• 70% accuracy for 25 acoustic scenes

K. Seyerlehner, G. Widmer, P. Knees "Comparison of Human, Automatic and Collaborative Music Genre Classification and User Centric Evaluation of Genre Classification Systems", In Proc. of Workshop on Adaptive Multimedia Retreival (AMR-2010), 2010.

Martin. (1999). "Sound-Source Recognition: A Theory and Computational Model". Ph.D. thesis, MIT V. Pelton & al., "Recognition of everyday auditory scenes : Potentials, latencies and cues, in Proc. AES, 2001





A (very) brief historical overview of

- Speech Recognition
- Music instrument/genre recognition
- Acoustic scenes/Event recognition



1952: Analog Digit Recognition, 1 speaker Features: ZCR in 2 bands <i>Davis, Biddulph, Balashek</i>	1962: Digital vowel Recognition, N speakers Taxonomy consonant/ vowel Features: Filterbank (40 filt.) <i>Schotlz, Bakis</i>	1980: MFCC Davis, Mermelstein 1980 - : HMM, GMM, Baker, Jelinek, Rabiner	,
1956: Analog 10 syllable recognition 1 speaker Features: Filterbank (10 filt.)	1971: Isolated word Recognition, Few speakers, DTW Features: Filterbank <i>Vintsjuk,</i>	1975-1985: Rule-based Expert systems 1000 words, few speakers Features: ManyFilterbank detection, Formant center fr energy, « frication » Decision trees, probabilistic <i>Woods, Zue, Lamel,</i>	2009 - : <i>Mel spectrogram</i> DNN <i>Hilton , Dahl</i> s, LPC, V/U requencies, labelling

1952: Analog Digit Recognition, 1 speaker	1962: Digital vowel Recognition, N speakers Taxonomy consonant/ vowel Features: Filterbank (40 filt.) <i>Schotlz, Bakis</i>	1980: MFCC Davis, Mermelstein	
Davis, Biddulph, Balashek		1980 - : HMM, GMM, Baker, Jelinek, Rabiner	,
1956: Analog 10 syllable recognition 1 speaker Features: Filterbank (10 filt.)	1971: Isolated word Recognition, Few speakers, DTW Features: Filterbank <i>Vintsjuk,</i>	1975-1985: Rule-based Expert systems 1000 words, few speakers Features: ManyFilterbank detection, Formant center fr energy, « frication » Decision trees, probabilistic <i>Woods, Zue, Lamel,</i>	2009 - : Mel spectrogram DNN Hilton , Dahl s, LPC, V/U requencies, labelling
17 14/09/2016 DC	ASE 2016	Gaël RICHARD	

1952: Analog Digit Recognition, 1 speaker Features: ZCR in 2 bands <i>Davis, Biddulph, Balashek</i>	1962: Digital vow Recognition, N spe Taxonomy consona Features: Filterban <i>Schotlz, Bakis</i>	el akers nt/ vowel < (40 filt.) 1980: MFCC <i>Davis, Mermelstei</i> 1980 - : HMM, <i>Baker, Jelinek, F</i>	n , GMM, Rabiner ,
			2009 - :
1956: Analog 10 syllable recognition 1 speaker			<i>Mel spectrogram</i> DNN <i>Hilton , Dahl…</i>
Features: Filterbank (10 filt.)	1971: Isolated w Recognition, Few speakers, DT Features: Filterbar <i>Vintsjuk,</i>	brd W k M M M M M M M M M M M M M	e-based eakers Iterbanks, LPC, V/U center frequencies, abilistic labelling
18 14/09/2016 DC	ASE 2016	Gaël RICHARD	TELECOM Paristech

1952: Analog Digit Recognition, 1 speaker Features: ZCR in 2 bands <i>Davis, Biddulph, Balashek</i>	1962: Digital vow Recognition, N spe Taxonomy consona Features: Filterban Schotlz, Bakis	el akers int/ vowel k (40 filt.) 1980 : MFCC <i>Davis, Mermelst</i> 1980 - : HMI <i>Baker, Jelinek,</i>	ein M, GMM, Rabiner ,
1956: Analog 10 syllable recognition 1 speaker Features: Filterbank (10 filt.)		1975-1985: Ru Expert systems	2009 - : <i>Mel spectrogram</i> DNN <i>Hilton , Dahl…</i> lle-based
	1971: Isolated w Recognition, Few speakers, DT Features: Filterbac <i>Vintsjuk,</i>	ord Features: Manyl detection, Forman energy, « frication bk Decision trees, pro <i>Woods, Zue, Lame</i>	peakers Filterbanks, LPC, V/U t center frequencies, » bbabilistic labelling e <i>l,</i>
19 14/09/2016 DC	ASE 2016	Gaël RICHARD	TELECOM Partistecta

1964 - : musical timbre perception <i>Clarke, Fletcher,</i> <i>Kendall</i>	2000 - : First use of MFCC for music modelling <i>Logan</i>	2004 - : Instrument recognition (polyphonic music) Multiple timbre features + GMM, SVM, Eggink, Essid,	2009 - : instrument recognition DNN, Hamel, Lee
1995 - : Music instrument recognitio on isolated notes <i>Kaminskyj, Martin,</i> <i>Peeters ,</i>	2001 - : Genre recognition Multiple musically motivated features GMM Tzanetakis,	5 + 2007 - : Instrumen recognition : exploiti source separation, dictionary learning NMF, Matching pursui Cont, Kitahara,Heittola Leveau, Gillet,	t ing t, a,



1964 - : musical timbre perception <i>Clarke, Fletcher, Kendall</i>	2000 - : First use of MFCC for music modelling <i>Logan</i>	2004 - : Instrument recognition (polyphonic music) Multiple timbre features + GMM, SVM, Eggink, Essid,	2009 - : instrument recognition DNN, Hamel, Lee
1995 - : Music instrument recognitic on isolated notes <i>Kaminskyj, Martin,</i> <i>Peeters ,</i>	2001 - : Genre recognition Multiple musically motivated features GMM Tzanetakis,	2007 - : Instrument recognition : exploiting source separation, dictionary learning NMF, Matching pursuin Cont, Kitahara,Heittola Leveau, Gillet,	t ing t, a,



1964 - : musical timbre perception <i>Clarke, Fletcher,</i> <i>Kendall</i>	2000 - : First use of MFCC for music modelling <i>Logan</i>	2004 - : Instrument recognition (polyphonic music) Multiple timbre features + GMM, SVM, Eggink, Essid,	2009 - : instrument recognition DNN, Hamel, Lee
1995 - : Music instrument recognitic on isolated notes <i>Kaminskyj, Martin,</i> <i>Peeters ,</i>	2001 - : Genre recognition Multiple musically motivated features GMM <i>Tzanetakis</i> ,	+ 2007 - : Instrumen recognition : exploit source separation, dictionary learning NMF, Matching pursui Cont, Kitahara,Heittol Leveau, Gillet,	t ing it, a,



1964 - : musical timbre perception <i>Clarke, Fletcher,</i> <i>Kendall</i>	2000 - : First use of MFCC for music modelling <i>Logan</i>	2004 - : Instrument recognition (polyphonic music) Multiple timbre features + GMM, SVM, Eggink, Essid,	2009 - : instrument recognition DNN, Hamel, Lee
1995 - : Music instrument recognitic on isolated notes <i>Kaminskyj, Martin,</i> <i>Peeters ,</i>	2001 - : Genre recognition Multiple musically motivated features GMM <i>Tzanetakis</i> ,	2007 - : Instrumen recognition : exploit source separation, dictionary learning NMF, Matching pursui Cont, Kitahara, Heittol Leveau, Gillet,	t ing t, a,



1964 - : musical timbre perception <i>Clarke, Fletcher,</i> <i>Kendall</i>	2000 - : First use of MFCC for music modelling <i>Logan</i>	2004 - : Instrument recognition (polyphonic music) Multiple timbre features + GMM, SVM, Eggink, Essid,	2009 - : instrument recognition DNN, <i>Hamel, Lee</i>
1995 - : Music instrument recognitic on isolated notes <i>Kaminskyj, Martin,</i> <i>Peeters ,</i>	2001 - : Genre recognition Multiple musically motivated feature GMM <i>Tzanetakis</i> ,	2007 - : Instrumen recognition : exploi source separation, dictionary learning NMF, Matching pursu Cont, Kitahara,Heitto Leveau, Gillet,	nt ting uit, bla,



1980 - : HMM, GMM in	1993 Computa (Audio stream se Use of auditory p Blackboard mod	tional ASA egregation) periphery model el ('IA)	recognition More specific sparsity, NMF <i>Chu & al, Cau</i>	methods exploiting , image features … <i>ichy & al,…</i>
speech/speaker recognition, <i>Baker, Jelinek,</i> <i>Rabiner ,</i>	M. Cook & al.	2003: Acoustic s recognition <i>MFCC+HMM+GN</i> <i>Eronen & al.</i>	scene 1M	2014 - : DNN for acoustic event recognition <i>Gencoglu & al</i>
1983,1990 Au Analysis (Perception/Psyc Scheffer, Bregma	uditory Sound rchology): nan,	1998 Acoustic scene recognition Use of HMM Clarksson &al.	2005: Event MFCC+ other Feature reduct GMM <i>Clavel & al.</i>	recognition feat. tion by PCA
	1997 Act 5 classes PLP + filte RNN or K- Sabwney	Dustic scenes recognition of sound r bank features, NN & al		



1980 - : HMM, GMM in	1993 Computational ASA (Audio stream segregation) Use of auditory periphery model Blackboard model ('IA)		recognition More specific methods exploiting sparsity, NMF, image features <i>Chu & al, Cauchy & al,</i>	
speech/speaker recognition, <i>Baker, Jelinek,</i> <i>Rabiner ,</i>	M. Cook & al.	2003: Acoustic s recognition <i>MFCC+HMM+GN</i> <i>Eronen & al.</i>	scene 1M	2014 - : DNN for acoustic event recognition <i>Gencoglu & al</i>
			0005	
1983,1990 A Analysis (Perception/Psy <i>Scheffer, Bregn</i>	vchology): <i>nan,</i>	1998 Acoustic scene recognition Use of HMM Clarksson &al.	2005: Event re MFCC+ other fe Feature reduction GMM <i>Clavel & al.</i>	ecognition eat. on by PCA
	1997 Acc 5 classes of PLP + filter RNN or K- Sahwney 8	oustic scenes recognition of sound r bank features, NN & al.		

14/09/2016



1980 - : HMM, GMM in	1993 Computa (Audio stream so Use of auditory Blackboard mod	ational ASA egregation) periphery model lel ('IA)	recognition More specific r sparsity, NMF, <i>Chu & al, Cau</i> d	nethods exploiting image features chy & al,
speech/speaker recognition, <i>Baker, Jelinek,</i> <i>Rabiner ,</i>	M. Cook & al.	2003: Acoustic s recognition <i>MFCC+HMM+GM</i> <i>Eronen & al.</i>	scene 1M	2014 - : DNN for acoustic event recognition <i>Gencoglu & al</i>
1983,1990 Au Analysis (Perception/Psyc Scheffer, Bregma	uditory Sound chology): an, …	1998 Acoustic scene recognition Use of HMM Clarksson &al.	2005: Event r MFCC+ other fe Feature reducti GMM <i>Clavel & al.</i>	ecognition eat. on by PCA
	1997 Ac 5 classes PLP + filte RNN or K	oustic scenes recognition of sound er bank features, -NN		

Sahwney & al.

14/09/2016



1980 - : HMM, GMM in	1993 Computational ASA (Audio stream segregation) Use of auditory periphery model Blackboard model ('IA)		recognition More specific sparsity, NMF <i>Chu & al, Ca</i>	recognition More specific methods exploiting sparsity, NMF, image features <i>Chu & al, Cauchy & al,</i>	
speech/speaker recognition, <i>Baker, Jelinek,</i> <i>Rabiner ,</i>	M. Cook & al.	2003: Acoustic recognition <i>MFCC+HMM</i> +G <i>Eronen</i> & al.	c scene SMM	2014 - : DNN for acoustic event recognition <i>Gencoglu & al</i>	
1983,1990 Au	uditory Sound	1998 Acoustic scene	2005: Event	recognition	
Analysis (Perception/Psyc Scheffer, Bregma	chology): an,	recognition Use of HMM Clarksson &al.	MFCC+ other Feature reduc GMM <i>Clavel & al.</i>	feat. ction by PCA	
	1997 A 5 classes PLP + filt	coustic scenes recognition s of sound ter bank features,			

RNN or K-NN Sahwney & al.





1980 - : HMM, GMM in	1993 Computational ASA (Audio stream segregation) Use of auditory periphery model Blackboard model ('IA)		recognition More specific methods exploiting sparsity, NMF, image features <i>Chu & al, Cauchy & al,</i>	
speech/speaker recognition, <i>Baker, Jelinek,</i> <i>Rabiner ,</i>	M. Cook & al.	2003: Acoustic recognition <i>MFCC+HMM+G</i> <i>Eronen</i> & al.	; scene MM	2014 - : DNN for acoustic event recognition <i>Gencoglu & al,</i>
1983,1990 Au Analysis (Perception/Psyc Scheffer, Bregma	uditory Sound hology): an, …	1998 Acoustic scene recognition Use of HMM Clarksson &al.	2005: Event MFCC+ other Feature reduct GMM <i>Clavel & al.</i>	recognition feat. tion by PCA
	1997 Acc 5 classes o PLP + filter RNN or K-	oustic scenes recognition of sound r bank features, NN		

Sahwney & al.

14/09/2016



And in 2016

• The example of Acoustic Scene recognition (DCASE2106)



The (partial) figure in 2016 (from DCASE 2016 – Acoustic Scene Detection)

Accuracy (Eval) जिप्ते	Input	Features	Classifier
	•	▼	
89.7 %	mono+binaural	MFCC+spectrograms	fusion
88.7 %	binaural	MFCC	-vector
87.7 %	mono	spectrogram	NMF
87.2 %	mono	various	fusion
86.4 %	mono	various	fusion
86.4 %	binaural	MFCC	l-vector
86.2 %	mono	mel energy	CNN
85.9 %	mono	MFCC distribution	SVM
85.6 %	mono	MFCC	DNN-GMM
85.4 %	mono	unsupervised	CNN ensemble
84.6 %	mono	mel energy	CNN
84.1 %	mono	various	ensemble
84.1 %	mono	spectrogram	CNN-RNN
83.8 %	mono	MFCC+mel energy	fusion
83.6 %	mono	MFCC+mel energy	fusion
83.3 %	mono	CQT	CNN
83.3 %	mono	spectrogram	CNN
83.3 %	mono	label tree embedding	CNN
83.1 %	mono	MFCC+mel energy	fusion
83.1 %	mono	MFCC	KNN



The (partial) figure in 2016 (from DCASE 2016 – Acoustic Scene Detection)

Some observations:

- Few systems exploit spatial information
- ... even though it is one of the important ideas of CASA...
- It seems that spatial information helps (as in speech recognition but has probably more potential here)

Inp	ut	
		•
	mono+binaural	
	binaural	
	mono	
	mono	
	mono	
	binaural	
	mono	
	TELECOM	



The (partial) figure in 2016 (from DCASE 2016 – Acoustic Scene Detection)

Some observations:

- MFCC are still very popular which seems surprising since an audio scene is not a speech signal :
 - 11 of the top 20 systems use MFCC

	MFCOspectrograms
	MFCC
	spectrogram
	various
	various
	MFCC
	mel energy
	MFCC istribution
	MFCC
	unsupervised
	mel energy
<	various
	spectrogram
	MFCC+nel energy
	MFCC+nel energy
	CQT
	spectrogram
	label tree embedding
	MFCC+ nel energy
	MECC

Are MFCC appropriate for acoustic scene/event recognition ?

- Pitch range is much wider in audio signal than in speech
- For high pitches the deconvolution property of MFCCs does not hold anymore (e.g. MFCC become pitch dependent)
- Their global characterization prevents MFCCs to describe localised time-frequency information and in that sense they fail to model well-known masking properties of the ear.
- MFCC are not highly correlated with the perceptual dimensions of "polyphonic timbre" in music signals despite their widespread use as predictors of perceived similarity of timbre.
- Sometimes MFCC are used exactly as for 8kHz sample speech (e.g. 13 coefficients) ...

Their use in general audio signal processing is therefore not well justified



G. Richard, S. Sundaram, S. Narayanan "An overview on Perceptually Motivated Audio Indexing and Classification", Proceedings of the IEEE, 2013.

A. Mesaros and T. Virtanen, "Automatic recognition of lyrics in singing," EURASIP Journal on Audio, Speech, and Music

B. Processing, vol. 2010, no. 1, p. 546047, 2010.

V. Alluri and P. Toiviainen, "Exploring perceptual and acoustical correlates of polyphonic timbre," Music Perception, vol. 27, no. 3, pp. 223–241, 2010

What are MFCC ?

- « Mel-Frequency Cepstral Coefficients »
- The most widely spread speech features (before 2012...)





What do the MFCC model ?

Interest

Speech source-filter production model (Fant 1960)

$$s(t) = g(t) * h(t)$$

 \checkmark The model in spectral domain

$$S(\omega) = G(\omega)H(\omega)$$

✓ Cepstre (real): a sum of two terms

 $c(\tau) = FFT^{-1}\log|S(\omega)| = FFT^{-1}\log|G(\omega)| + FFT^{-1}\log|H(\omega)|$

✓ Source contribution is removed by selecting the first few cepstral coefficients



MFCC capture "global" spectral envelope

Fourier transform of the cepstrum (first 45 coefficients)



It seems that MFCC's capacity to capture "global" spectral envelope properties is the main reason of their success in audio classification tasks.



DCASE 2016

The (partial) figure in 2016 (from DCASE 2016...)

Some observations:

- All but 4 systems use Neural Networks
- But the best systems without fusion do not use Neural networks
- Other recent ideas:
 - Use of i-vectors (from speaker recognition)
 - Exploit decomposition techniques (NMF)

Classifier
fusion
l-vector
NMF
fusion
fusion
l-vector
CNN
SVM
DNN-GMM
CNN ensemble
CNN
ensemble
CNN-RNN
fusion
fusion
CNN
CNN
CNN
fusion
kNN



A (very) recent system for Acoustic Scene recognition proposed in DCASE2016

An alternative approach to DNN



V. Bisot, R. Serizel, S.Essid and G. Richard, "Supervised NMF for Acoustic Scene Classification, techn rep. DCASE2016 challenge, 2016.

V. Bisot, R. Serizel, S.Essid and G. Richard, Feature Learning with Matrix Factorization Applied to Acoustic Scene Classification, submitted to special issue of IEEE Trans. On ASLP, 2016 *Available at: https://hal.archives-ouvertes.fr/hal-01362864*



Some hypotheses

Hypotheses

- An acoustic scene is characterised by the nature and occurrence of specific events
 - A car horn is mostly present in streets
- Most of the events have specific time-frequency content

Objective : to find a mean to capture event occurrencies and time-frequency content for acoustic scene recognition



An Acoustic Scene recognition system

Aim to decompose audio scene spectrograms in events using matrix factorization

- Learn a dictionary of audio event
- Use as features the projections on the learned dictionary

Additional possibility:

- Jointly learn the dictionary and the classifier
- Take into account the multi-class aspect of the problem

V. Bisot, R. Serizel, S.Essid and G. Richard, "Supervised NMF for Acoustic Scene Classification, techn rep. DCASE2016 challenge, 2016.
V. Bisot, R. Serizel, S.Essid and G. Richard, Feature Learning with Matrix Factorization Applied to Acoustic Scene Classification, submitted to special issue of IEEE Trans. On ASLP, 2016



Matrix factorization for feature learning



- V is the data Matrix
- **W** is the learned « dictionary » Matrix
- H is the « activation » matrix and the learned features



D. D. Lee and H. S. Seung, "Learning the parts of objects by non-negative matrix factorization," Nature, vol. 401, no. 6755, pp. 788–791, 1999.





Input feature for each recording

• The average of each h_k

$$\begin{array}{c} \mathbf{H}_{(K \times N)} \\ \hline \circ \bullet \bullet \bullet \bullet \circ \circ \\ \circ \circ \circ \bullet \bullet \bullet \circ \end{array} \begin{array}{c} \mathbf{h}_1 \\ \mathbf{h}_2 \end{array} \longrightarrow \overline{\mathbf{h}}_i = \frac{1}{M} \sum_m h_i(m) \end{array}$$

Classifier

Multinomial Linear Logistic Regression



Multinomial Linear Logistic Regression

Classifier cost to be minimized:

$$\ell_s(y, \mathbf{h}) = -\log(P(y = c | \mathbf{h}))$$

With

$$\begin{cases} \mathbf{P}(y=c|\mathbf{h}) = \frac{e^{(b_c + \mathbf{a}_c^T \mathbf{h})}}{1 + \sum_{j=1}^{C-1} e^{(b_j + \mathbf{a}_j^T \mathbf{h})}} ; c = 1, ..., C - 1\\ \\ \mathbf{P}(y=C|\mathbf{h}) = \frac{1}{1 + \sum_{j=1}^{C-1} e^{(b_j + \mathbf{a}_j^T \mathbf{h})}} \end{cases}$$

- $\mathbf{a}_c \in \mathbb{R}^K$ are the classifier weights
- y is one of the possible label



46

14/09/2016







TELECON Paristed

谬

What can be improved ?

Exploit more sophisticated and task-adapted NMF

- Sparse NMF: towards more interpretable decomposition
- Convolutive NMF: to exploit 2D dictionnary elements
- ...

Jointly learn the dictionnary for feature extraction and the classifier

• For example : Task driven Dictionnary Learning



J. Mairal, F. Bach, and J. Ponce, "Task-driven dictionary learning," IEEE Transactions on Pattern Analysis and Machine Intelligence, vol. 34, no. 4, pp. 791–804, 2012.



Task driven Dictionnary Learning (TDL)

Supervised dictionary learning

Aim of TDL: jointly learn a good dictionary and the classifier along with activation sparsity constraints

Classify optimal projections on the dictionary

Solving the following problem:

$$\mathbf{h}_{i}^{\star}(\mathbf{v}_{i}, \mathbf{W}) = \min_{\mathbf{h} \in \mathbb{R}^{K}} \frac{1}{2} \|\mathbf{v}_{i} - \mathbf{W}\mathbf{h}\|_{2}^{2} + \lambda_{1} \|\mathbf{h}\|_{1} + \frac{\lambda_{2}}{2} \|\mathbf{h}\|_{2}^{2}$$



Adapted algorithm

Adaptation to our task

- Classifying averaged projections $\overline{\mathbf{h}}_i = \frac{1}{M} \sum_m h_i(m)$
- Exploit a Multinomial Linear Logistic Regression classifier (as before)
- Force non negativity for activations (e.g. projections)

V. Bisot, R. Serizel, S.Essid and G. Richard, "Supervised NMF for Acoustic Scene Classification, techn rep. DCASE2016 challenge, 2016.
V. Bisot, R. Serizel, S.Essid and G. Richard, Feature Learning with Matrix Factorization Applied to Acoustic Scene Classification, submitted to special issue of IEEE Trans. On ASLP, 2016 *Available at: https://hal.archives-ouvertes.fr/hal-01362864*





- This approach is efficient for Acoustic scene classification
 - Ranked 3rd in DCASE2016 challenge without exploiting DNN (but a little bit of fusion).
 - Is better than our DNN approach using the same datamatrix for the DCASE2016 development dataset
 - But less good (but not statistically significant) than DNN on LITIS dataset which is larger



Discussion / Wrap up

Acoustic Scene Recognition and Audio event recognition is a more recent field than speech recognition, speaker recognition, MIR, ...

The problems are « similar »

- The input signal is an audio signal
- The problem is to classify the input signal in different classes

... but also different

- The classes are very different and always well defined
- The audio signal is a complex mixtures of overlapping individual sounds which may be never observed in isolation or quiet environment
- Cannot really use a « Language » model, but taxonomy is possible
- The number of classes may differ very significantly...



Discussion / Wrap up

The influence of Speech domain is natural

- Due to the proximity of the different problems,
- Due to the fact that the speech community is much larger and has a stronger past history
- Due to the fact that speech models are trained on much larger and varied datasets
- Speech recognition is a complex audio signal classification problem.
- it is then natural to find in Acoustic Scene and Event Recognition the solutions proposed for speech/speaker
 - MFCC, i-vectors, GMM, HMM,and now DNNs
 - And DNNs do work in scene/event recognition



Discussion / Wrap up

- But the problem is also different and calls for task designed and adapted methods
 - Adapted to the specificities of the problem
 - Adapted to the scarcity of training (*annotated*) data
 - Adapted to the fact that individual classes (especially events) may be only observed in mixtures
 - Potential of novel paths is shown in the DCASE2016 results





- Yes, we are right in looking what the speech processing community is doing
- ... but we should adapt their findings to our problem
- and It is worth looking other domains...
- and it is worth developping new methods which are not a direct application of speech methods
- There may be a life besides DNNs especially for Acoustic Scene and Event recognition

