230622 - DSAP - Digital Speech and Audio Processing

Coordinating unit: 230 - ETSETB - Barcelona School of Telecommunications Engineering
Teaching unit: 739 - TSC - Department of Signal Theory and Communications
Academic year: 2017
Degree: MASTER'S DEGREE IN INFORMATION AND COMMUNICATION TECHNOLOGIES (Syllabus 2009). (Teaching unit Optional)
DEGREE IN TELECOMMUNICATIONS ENGINEERING (Syllabus 1992). (Teaching unit Optional)
MASTER'S DEGREE IN TELECOMMUNICATIONS ENGINEERING (Syllabus 2013). (Teaching unit Optional)
ECTS credits: 5
Teaching languages: English

Teaching staff

Coordinator: Climent Nadeu
Others: Antonio Bonafonte Javier Hernando

Opening hours

Timetable: Tuesday and Thursday from 10:00 to 13:00

Prior skills

Signal Processing

Requirements

Signal processing

Degree competences to which the subject contributes

Specific:
1. Ability to apply information theory methods, adaptive modulation and channel coding, as well as advanced techniques of digital signal processing to communication and audiovisual systems.

Transversal:
2. TEAMWORK: Being able to work in an interdisciplinary team, whether as a member or as a leader, with the aim of contributing to projects pragmatically and responsibly and making commitments in view of the resources that are available.
3. EFFECTIVE USE OF INFORMATION RESOURCES: Managing the acquisition, structuring, analysis and display of data and information in the chosen area of specialisation and critically assessing the results obtained.
4. FOREIGN LANGUAGE: Achieving a level of spoken and written proficiency in a foreign language, preferably English, that meets the needs of the profession and the labour market.

Teaching methodology

- Lectures (50%)
- Application classes (with Matlab or similar) (50%)
- Team work: project, presentation
- Individual work: preparation and completion (out classroom) of application activities

Learning objectives of the subject
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Understanding and being competent on a relevant set of concepts and techniques in the field of digital audio processing, and their application to problems arising from real applications. Especially, speech and music signals and applications will be considered.

Learning results:
Ability to digitally process, in an application-oriented context, audio and speech signals, in order to analyze, model, extract information from, clean, modify, and generate/synthesize them.

### Study load

<table>
<thead>
<tr>
<th>Total learning time: 125h</th>
<th>Hours large group: 39h</th>
<th>31.20%</th>
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<tbody>
<tr>
<td>Hours medium group:</td>
<td>0h</td>
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<tr>
<td>Hours small group:</td>
<td>0h</td>
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<td>Guided activities:</td>
<td>0h</td>
<td>0.00%</td>
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<tr>
<td>Self study:</td>
<td>86h</td>
<td>68.80%</td>
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# Content

<table>
<thead>
<tr>
<th>Section</th>
<th>Learning time: 12h</th>
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<tbody>
<tr>
<td><strong>1. Introduction</strong></td>
<td>Theory classes: 6h</td>
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<tr>
<td></td>
<td>Self study : 6h</td>
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<tr>
<td><strong>Description:</strong></td>
<td></td>
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<tr>
<td>Course presentation</td>
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<tr>
<td>Audio diversity</td>
<td></td>
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<tr>
<td>Characteristics of speech and music. Production model</td>
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<tr>
<td>Hearing and auditory modeling</td>
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<tr>
<td>The short-time Fourier transform</td>
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| **2. Short-term analysis-synthesis of (cuasi)periodic signals**       | Theory classes: 6h |
|                                                                      | Self study : 6h    |
| **Description:**                                                     |                    |
| Filter-bank analysis/synthesis. The phase vocoder                    |                    |
| Filter-bank and spectrogram                                          |                    |
| Time-scale and pitch modification                                    |                    |
| QMF filters. MP3 coding.                                             |                    |

| **3. Modeling and representation of speech signals**                 | Theory classes: 6h |
|                                                                      | Self study : 6h    |
| **Description:**                                                     |                    |
| Production-based all-pole modeling                                   |                    |
| Pitch determination for speech and music                             |                    |
| LPC-based coding used in mobile telephony                            |                    |
### 4. Enhancement of speech and audio signals

**Description:**
- Cancellation: echo, interference
- Denoising: spectral subtraction, Wiener-based filtering, wavelets
- Blind source separation: ICA, CASA, NMF

**Learning time:** 12h
- Theory classes: 6h
- Self study: 6h

### 5. Multi-microphone audio processing

**Description:**
- Room acoustics
- Array beamforming
- Acoustic source localization and tracking

**Specific objectives:**

**Learning time:** 12h
- Theory classes: 6h
- Self study: 6h

### 6. Recognition and detection of audio and speech

**Description:**
- 6. Recognition and detection of audio and speech
- Pattern-matching approaches
- Audio activity detection
- Application to speech and speaker recognition

**Learning time:** 12h
- Theory classes: 6h
- Self study: 6h
Projects realization and presentation

<table>
<thead>
<tr>
<th>Learning time: 54h</th>
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<tbody>
<tr>
<td>Theory classes: 3h</td>
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<tr>
<td>Self study: 51h</td>
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**Description:**
Design, implementation and test of an audio processing system for a specific application. Oral presentation of 1) Project proposal, and 2) Project realization.

Qualification system

Attendance/participation in class (10%)  
Tests (30%)  
Project (50%)  
Presentation (10%)

Bibliography

**Basic:**

**Complementary:**

**Others resources:**
- Lecture slides
- Practical work statements and programs

**Audiovisual material**
- Slides
  - Slides used in lectures

**Computer material**
- Codi programes
  - Software codes in Matlab or similar