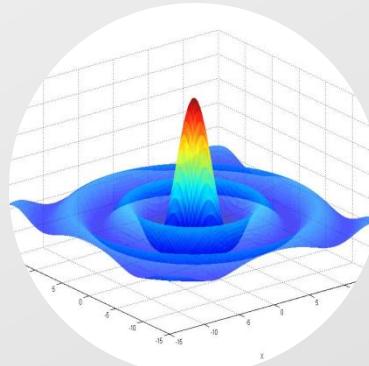


Project1 : Speech Synthesis And Perception With Envelope Cue



主讲老师：王小静
办公地点：慧园2栋411



Overview

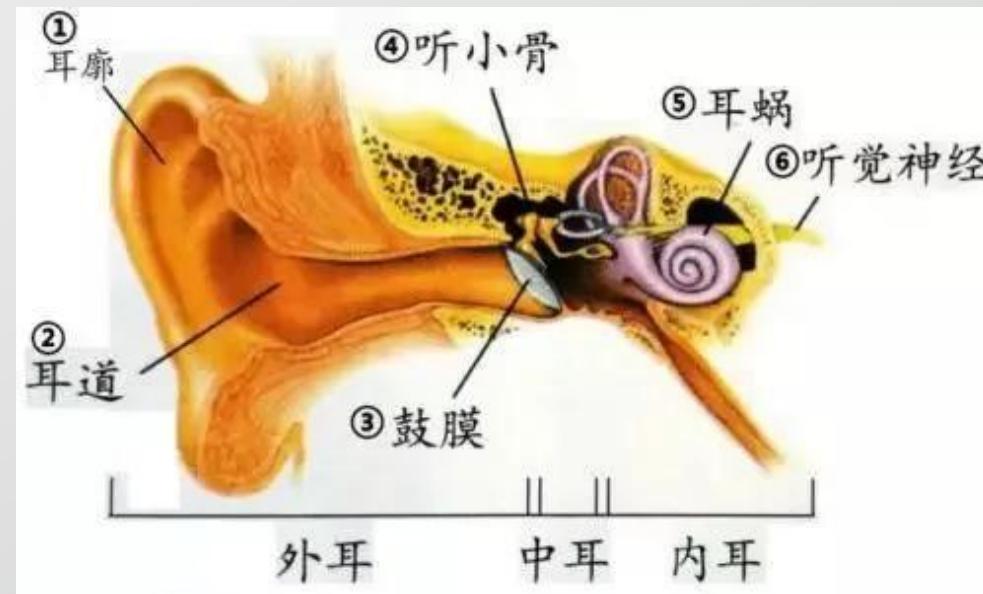
- In this tutorial, you will learn to synthesize a speech signal based on ***multi-band envelope cues***.
- In lab 5, you've learned:
 - how to design a low-pass/band-pass filter
 - how to extract envelope
 - how to generate a speech-spectrum shaped noise (SSN)
 - how to do energy normalization
 - how to read/save a ‘*.wav’ file

Background: The principle of hearing

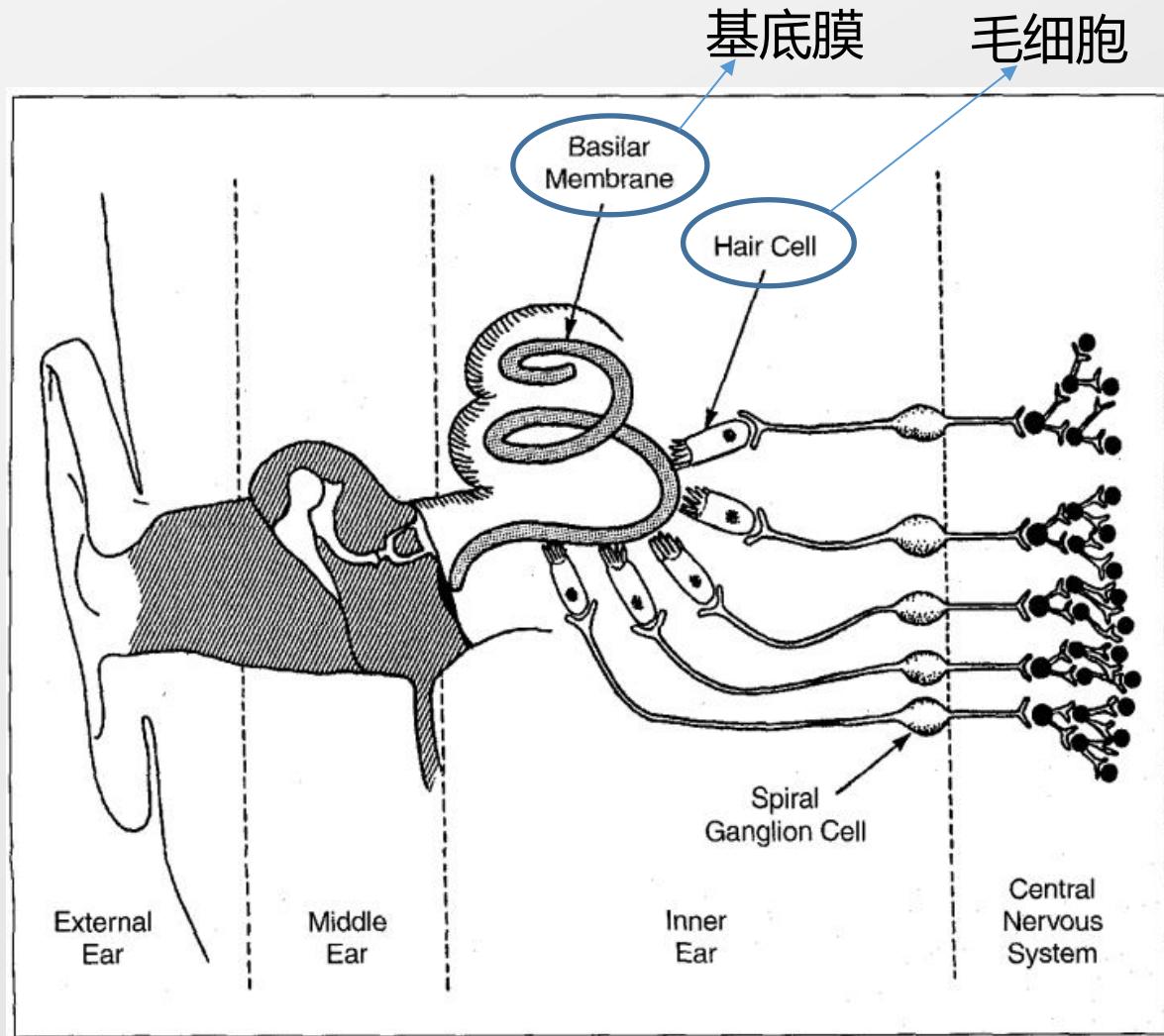
acoustic pressure waves → outer ear → eardrum(鼓膜)



brain ← auditory nerve ← cochlea(耳蜗) ← ossicles(听小骨)



The function of cochlea



mechanical vibrations

vibrations in fluid

pressure variations

displacements of basilar membrane

bending of hair cells

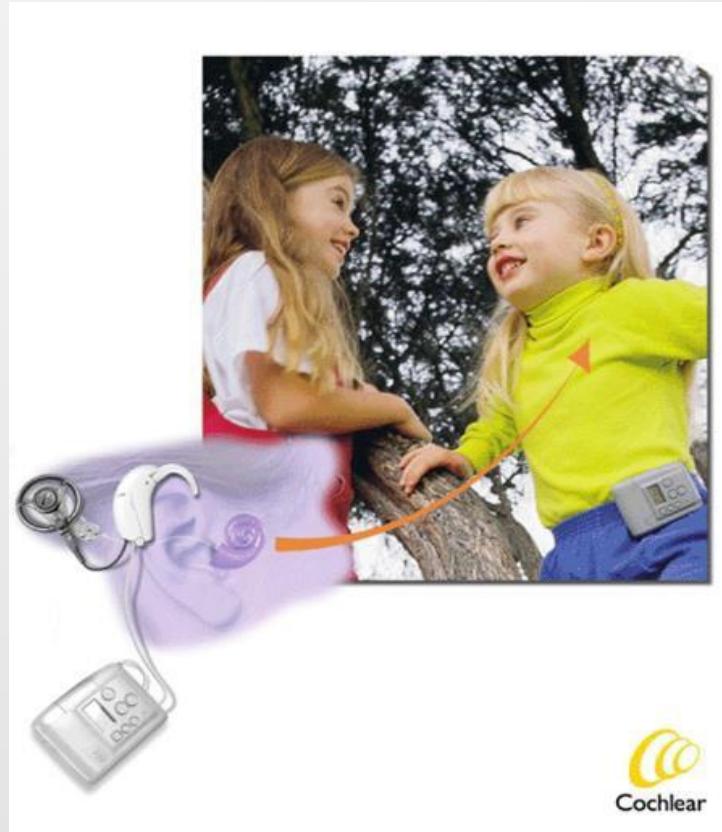
electrochemical substance

neurons fire



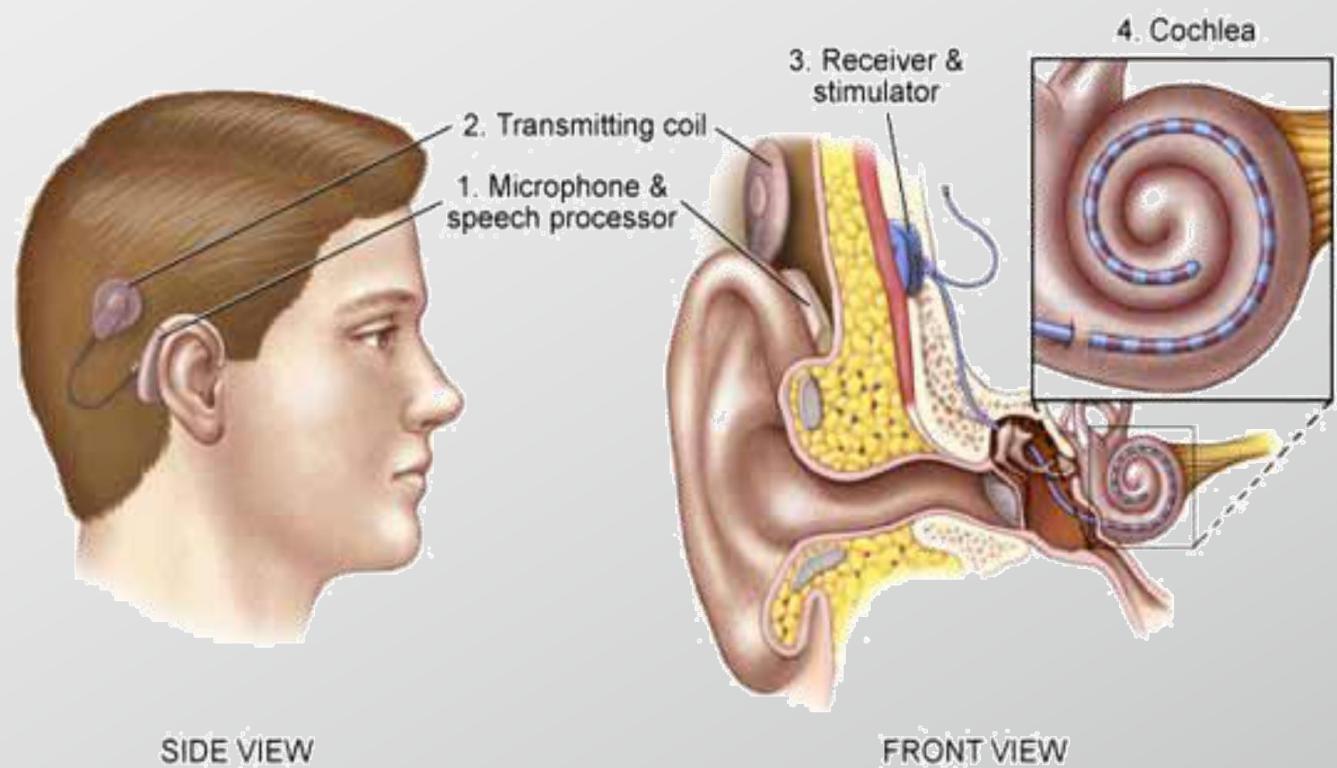
Deafness & cochlear implants

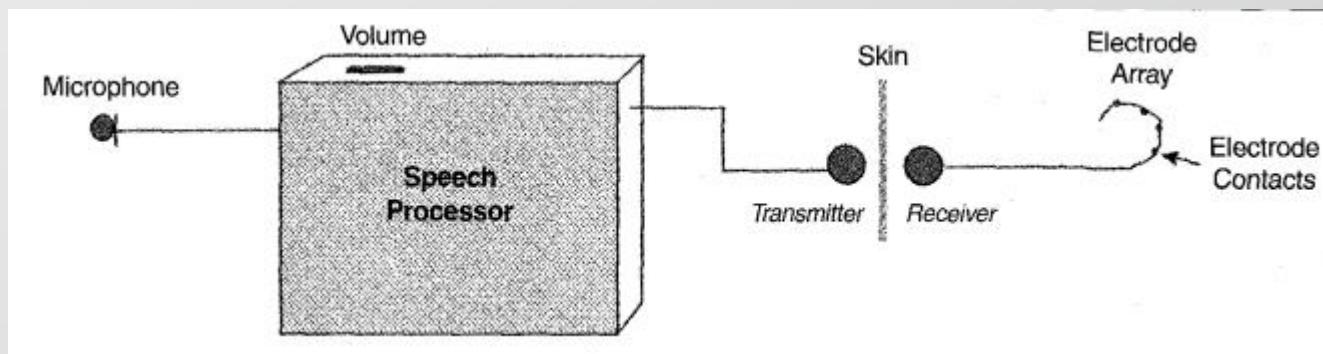
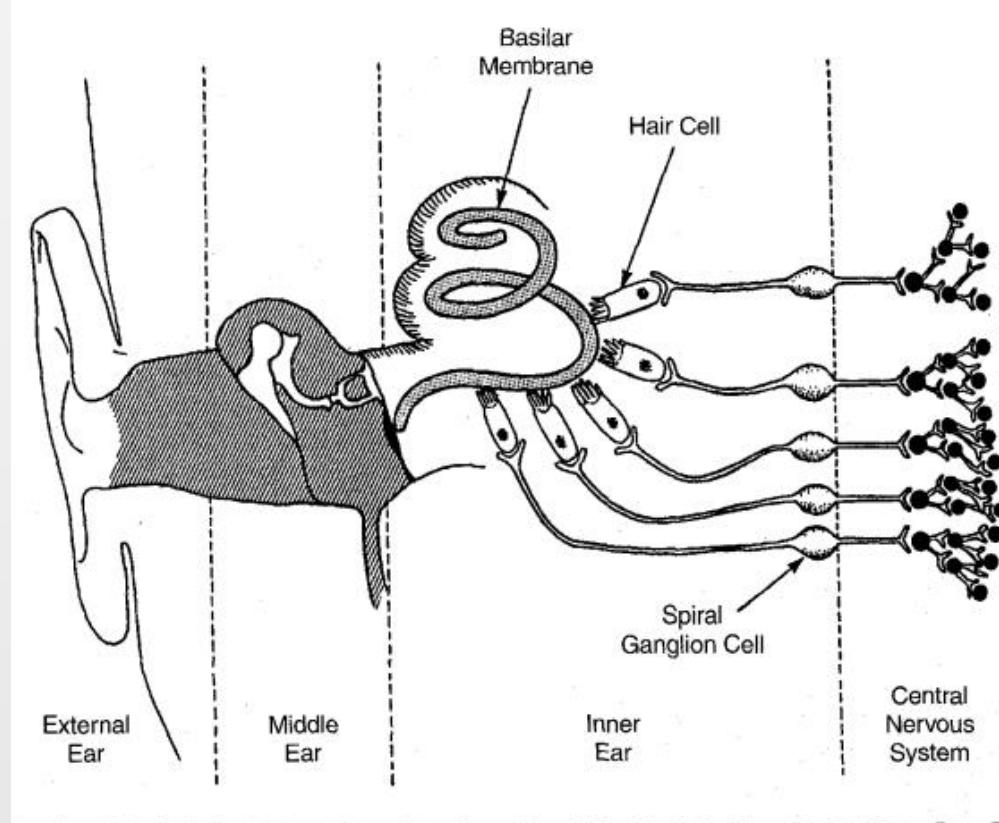
The most common cause of deafness is the loss of hair cells, rather than the loss of auditory neurons.

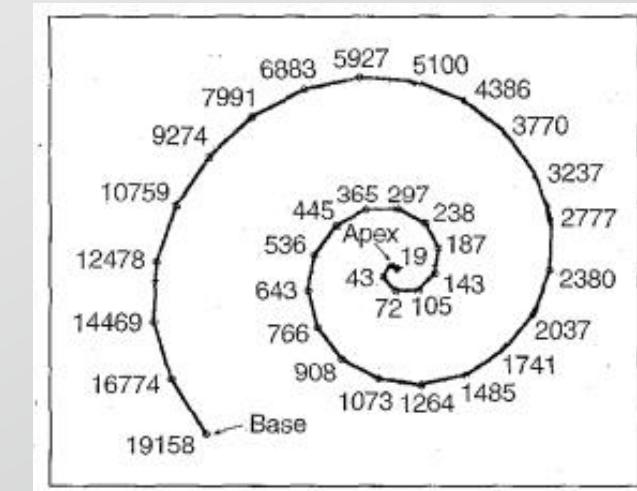
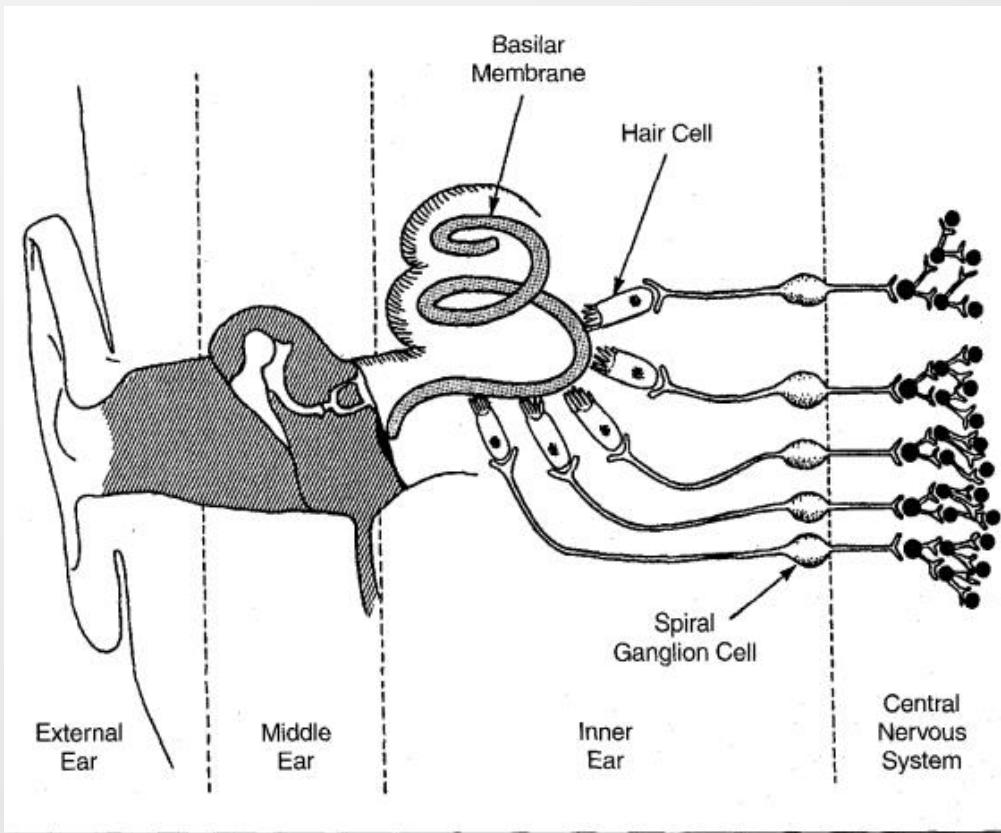


The principle of cochlear implants

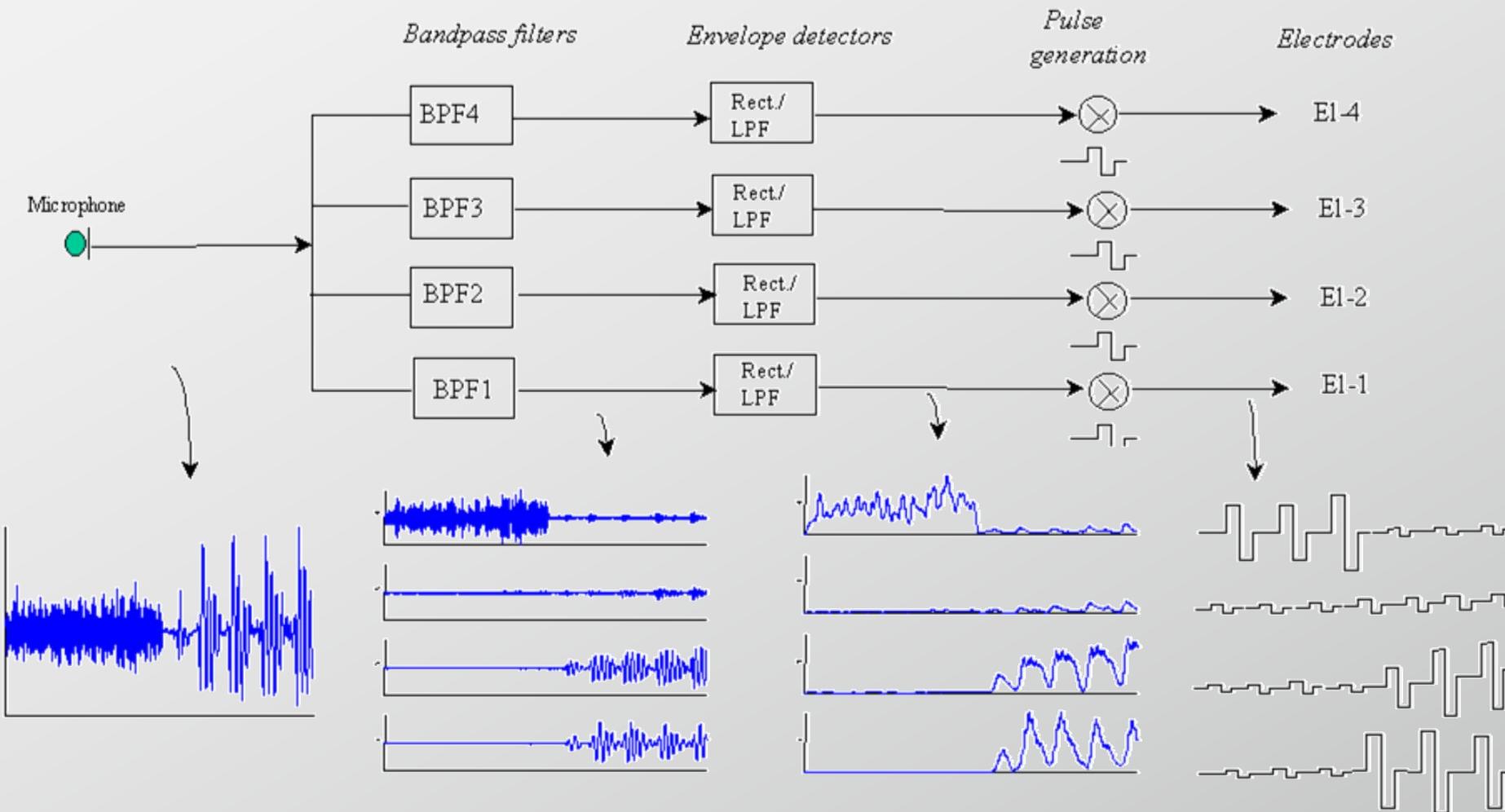
- The microphone collects signals, and the speech processor processes the collected signals.
- The wireless transmitter sends the processed signal to the inner side of the cochlea implants.
- Wireless receiver receives signals.
- Received signals stimulate the auditory nerve through electrodes to reconstruct signals.



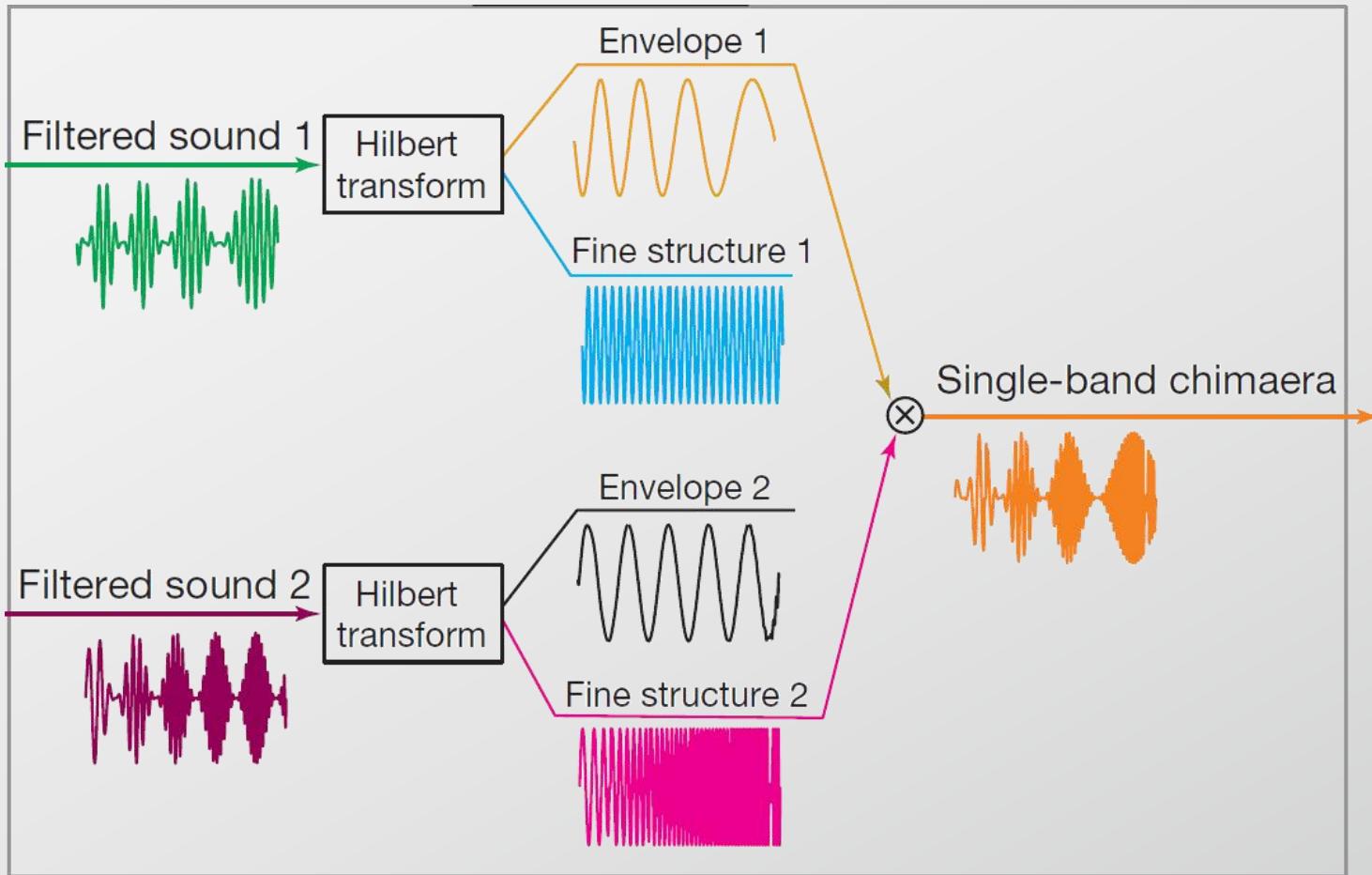




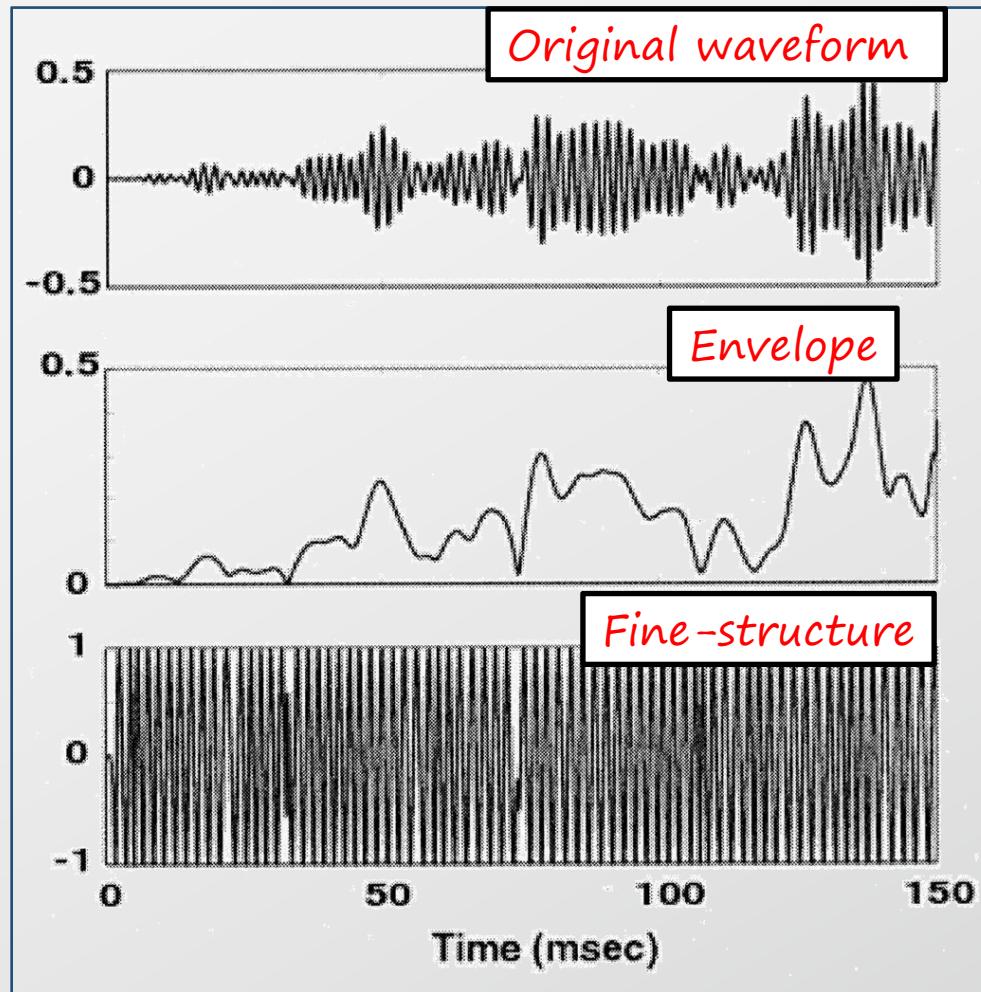
Speech processing in cochlear implants



Acoustic cues of speech signal



- Envelope and fine- structure
 - **Envelope:** amplitude modulation, low- frequency
 - **Fine-structure:** frequency modulation, high- frequency



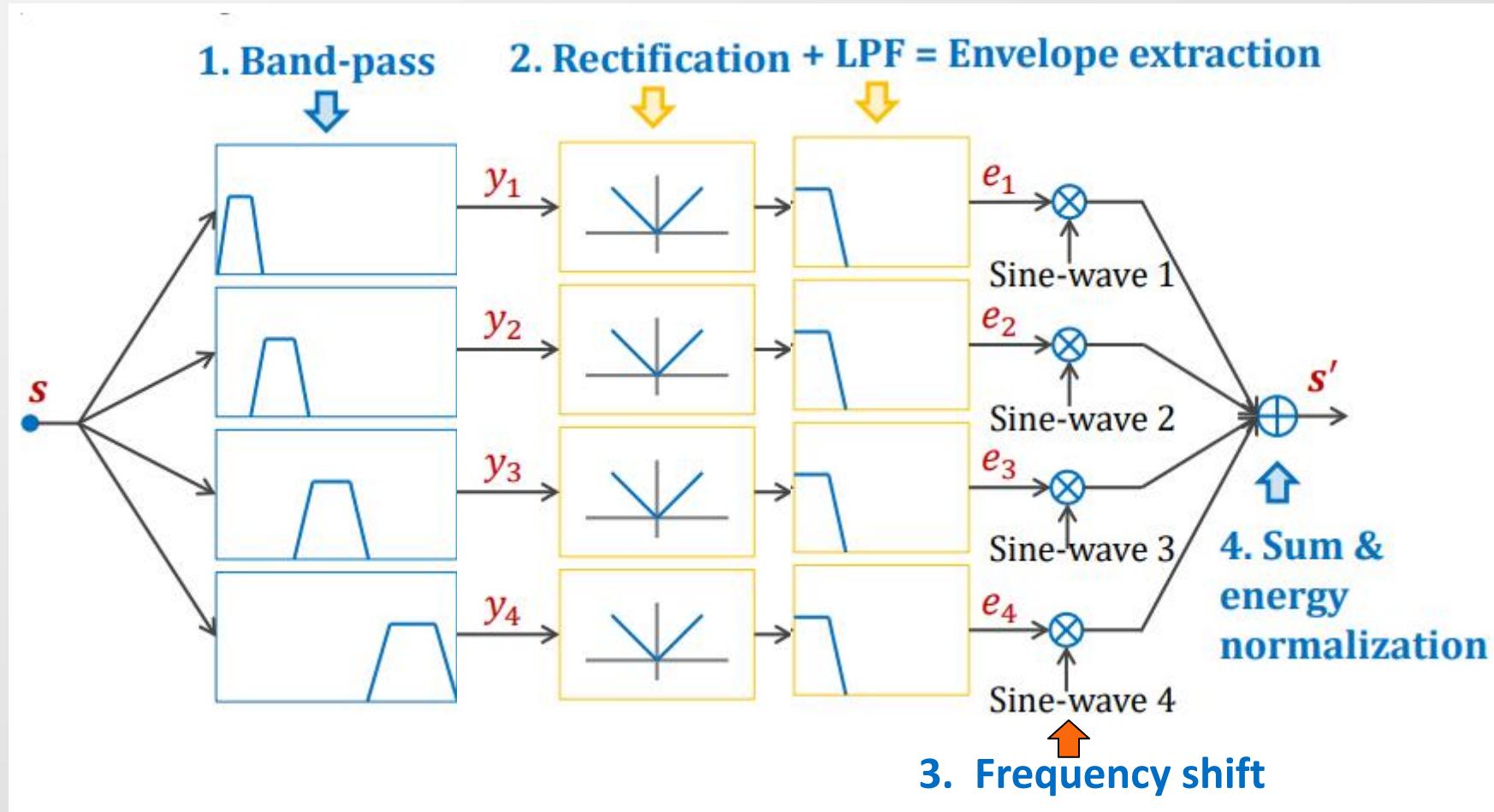
- **Envelope:**

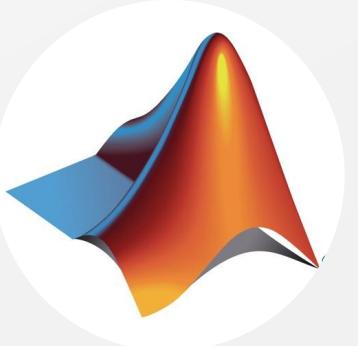
The envelope is most important for speech reception

- **Fine-structure:**

The fine structure is most important for pitch perception and sound localization.

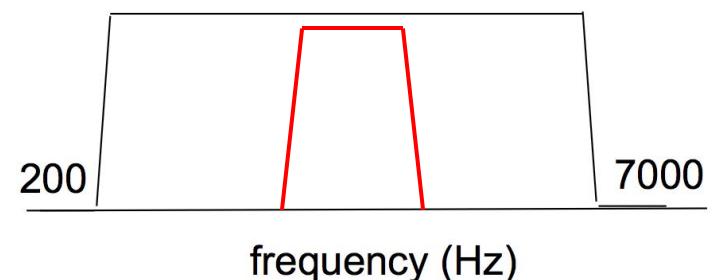
Speech synthesis with envelope cue: Tone-vocoder



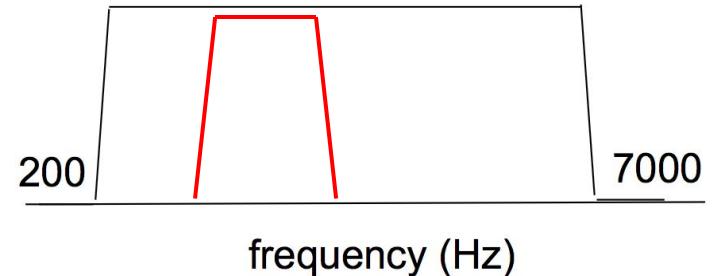


Example

- Set the number of bands to N.
- For the i^{th} band
 - 1) Design band-pass (e.g., 400-1000 Hz) filter at band i
`>> fs=16000; %sampling frequency
>> [b, a]=butter(4, [400 1000]/(fs/2)); %band-pass filter`
 - 2) Do band-pass filtering at band i
`>> y= filter(b, a, s); % s is speech signal, and y is the band-passed signal at band i`
 - 3) Do full-wave rectification, and low-pass filtering to get the envelope at band i
 - 4) Generate a sinewave, whose frequency equals to the center frequency of the i^{th} band-pass filter
 - 5) Multiply the envelope signal in 3) and sinewave in 4)
 - 6) Repeat for all N bands
 - 7) Sum up the outputs from all bands (denoting the summed outputs as s')
 - 8) Do energy normalization, i.e., letting the energy of s' equals to that of s
 - 9) Save the wavefile for signal s'



- Set the number of bands to N.
- For the i^{th} band
 - 1) Design band-pass (e.g., 400-1000 Hz) filter at band i
 $\gg \text{fs}=16000; \text{\%sampling frequency}$
 $\gg [\text{b}, \text{a}]=\text{butter}(4, [400 1000]/(\text{fs}/2)); \text{\%band-pass filter}$
 - 2) Do band-pass filtering at band i
 $\gg \text{y}=\text{filter}(\text{b}, \text{a}, \text{s}); \text{\% s is speech signal, and y is the band-passed signal at band } i$
 - 3) Do full-wave rectification, and low-pass filtering to get the envelope at band i
 - 4) Generate a sinewave, whose frequency equals to the center frequency of the i^{th} band-pass filter
 - 5) Multiply the envelope signal in 3) and sinewave in 4)
 - 6) Repeat for all N bands
 - 7) Sum up the outputs from all bands (denoting the summed outputs as s')
 - 8) Do energy normalization, i.e., letting the energy of s' equals to that of s
 - 9) Save the wavefile for signal s'



Question? How to divide pass-band from 200 Hz to 7000 Hz?

Determine the frequency range of the i^{th} band :

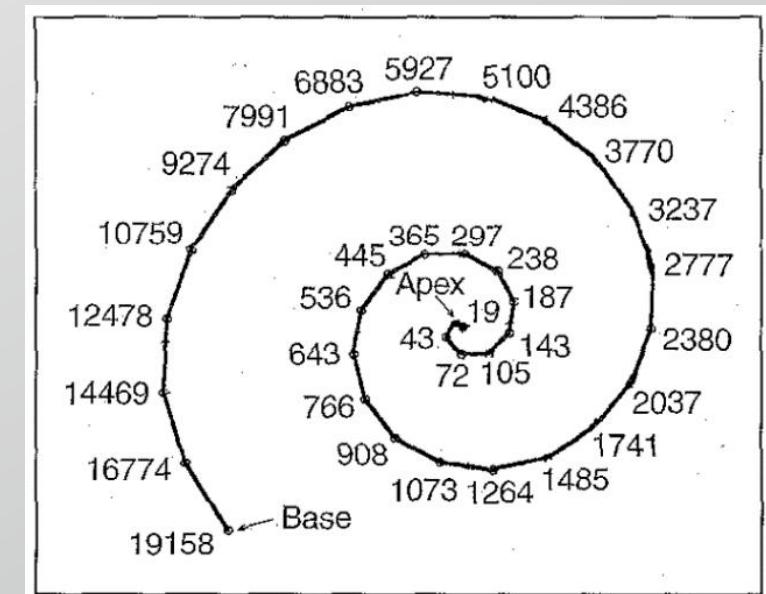
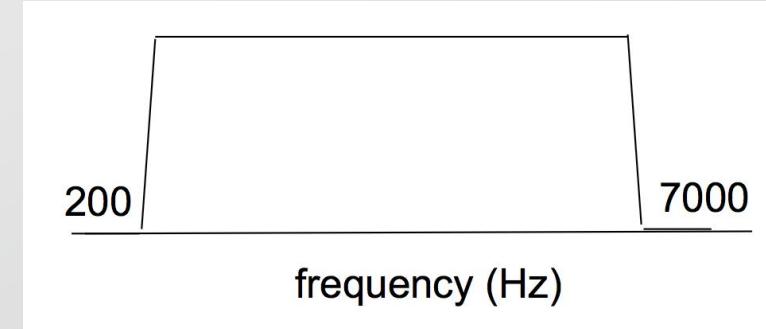
- Frequency-to-place mapping as:

$$f = 165.4 \times (10^{0.06 \cdot d} - 1)$$

where f is -3 dB cut-off frequency, and d is the distance (in millimeter) along the cochlea.

1) Equally divide the cochlea length.

2) Equal frequency interval division



Project tasks - 1

- Sentences for pro 1: ‘C_01_01.wav’ & ‘C_01_02.wav’
- Task 1
 - Set LPF cut-off frequency to 50 Hz.
 - Implement tone-vocoder by changing the number of bands to N=1, N=2, N=4, N=6, and N=8.
 - Save the wave files for these conditions, and describe how the number of bands affects the intelligibility (i.e., how many words can be understood) of synthesized sentence.

Project tasks -2

- **Task 2**
 - Set the number of bands $N=4$.
 - Implement tone-vocoder by changing the LPF cut-off frequency to 20 Hz, 50 Hz, 100 Hz, and 400 Hz.
 - Describe how the LPF cut-off frequency affects the intelligibility of synthesized sentence.

Project tasks -3

- **Task 3**

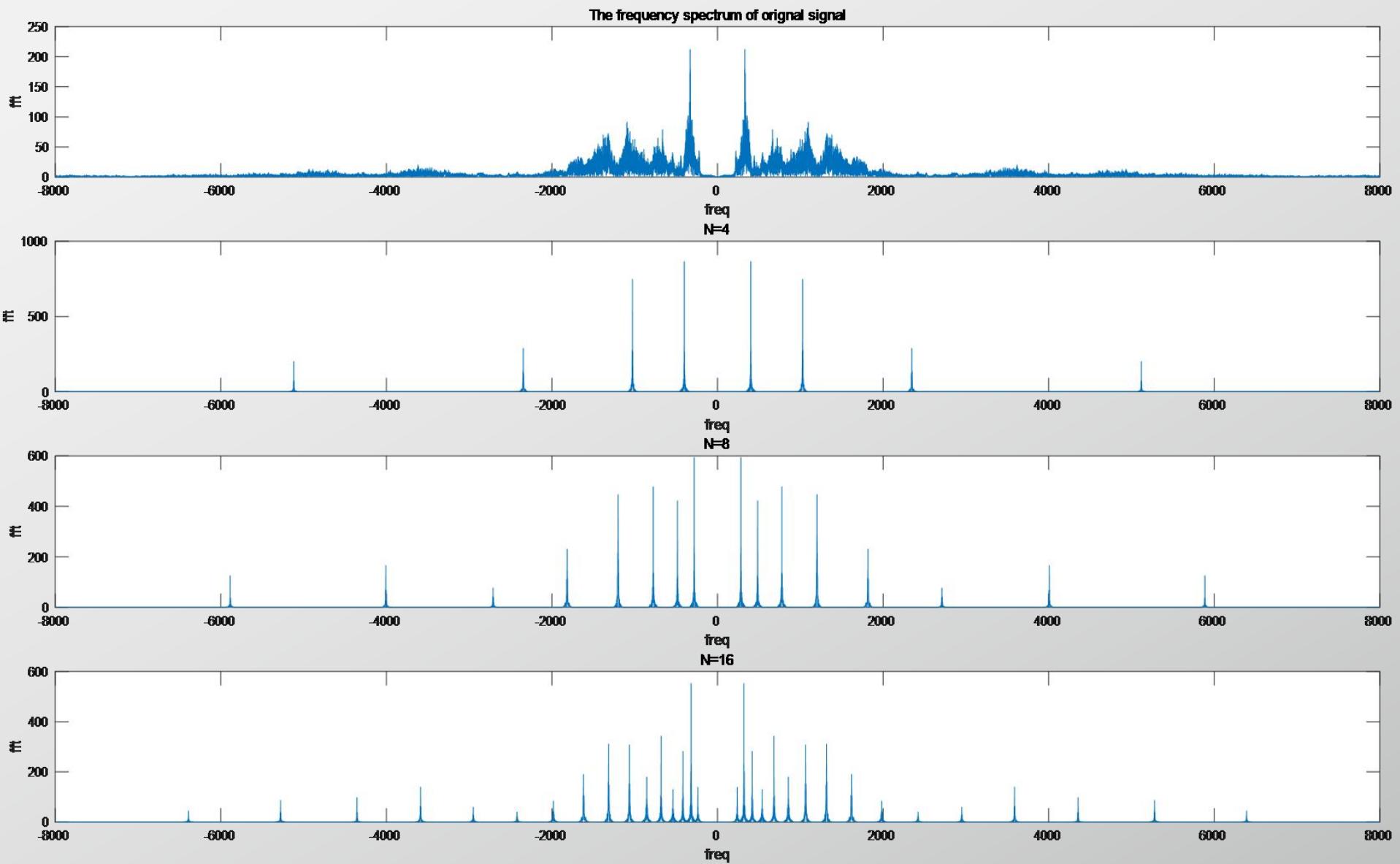
- Generate a noisy signal (summing clean sentence and SSN) at SNR -5 dB.
- Set LPF cut-off frequency to 50 Hz.
- Implement tone-vocoder by changing the number of bands to $N=2$, $N=4$, $N=6$, $N=8$, and $N=16$.
- Describe how the number of bands affects the intelligibility of synthesized sentence, and compare findings with those obtained in task 1.

Project tasks -4

- **Task 4**

- Generate a noisy signal (summing clean sentence and SSN) at SNR -5 dB.
- Set the number of bands to N=6.
- Implement tone-vocoder by changing the LPF cut-off frequency to 20 Hz, 50 Hz, 100 Hz, and 400 Hz.
- Describe how the LPF cut-off frequency affects the intelligibility of synthesized sentence.

Results



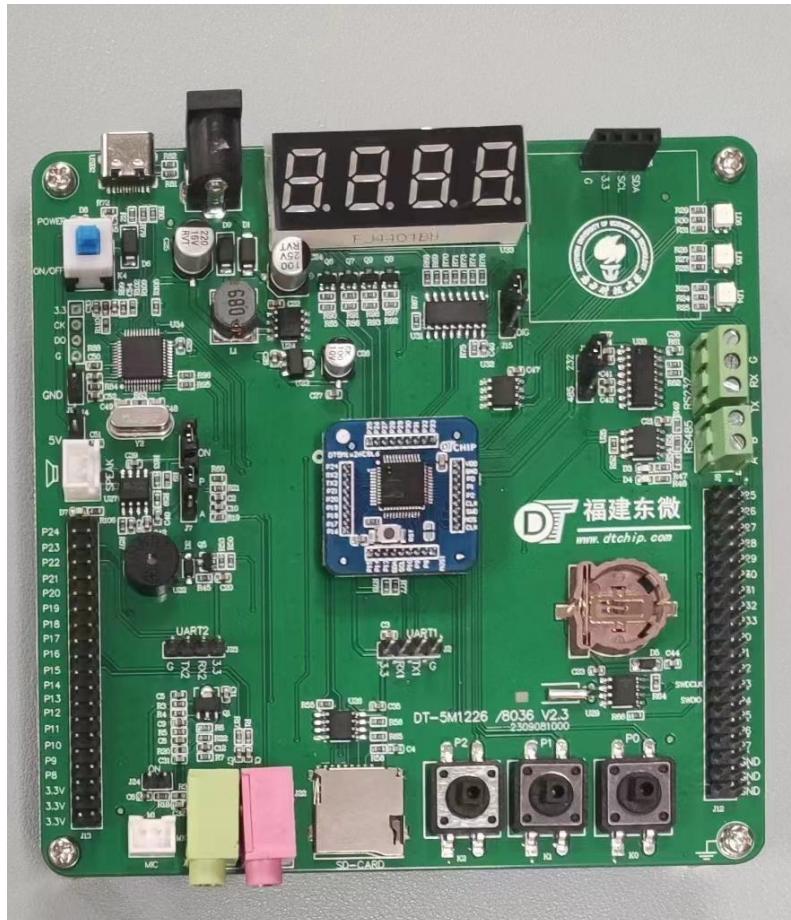
硬件. DT5M1x26学习板简介—12.7培训会

实验目标： 基于DT5M1x26芯片实现合成后的音频播放。

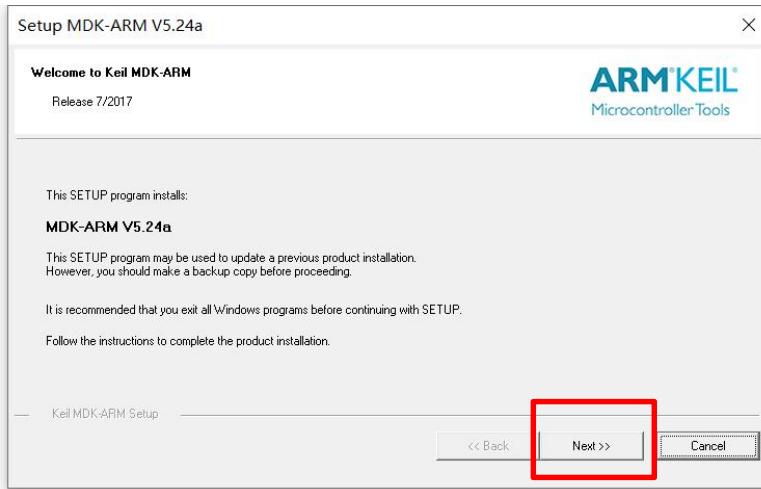
硬件介绍： DT5M1x26学习板是基于ARM Cortex-M0的架构的低功耗MCU板。板载Daplink通过Type-C口连接学习板和电脑，实现程序下载，将耳机或音响的音频线插入PCM OUT口，可以实现语音播放。

实验过程：

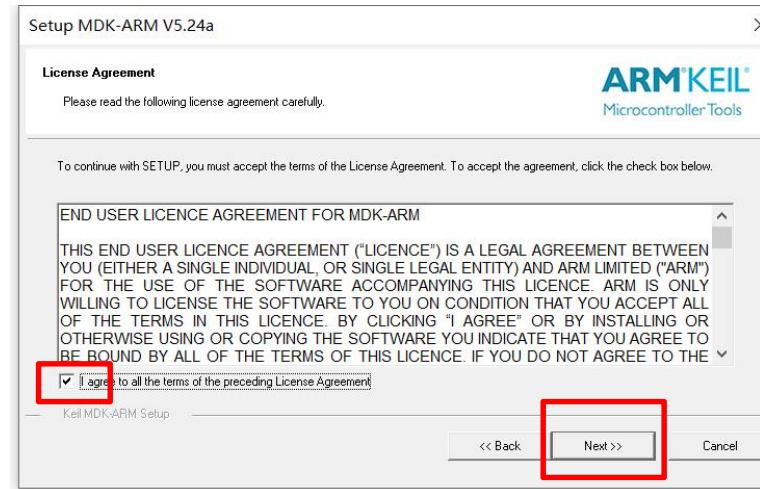
- 1、MATLAB进行语音合成信号处理
- 2、将合成后的音频数据导入嵌入式开发板DT5M1x26中，并利用其音频输出端口将语音信号播放出来。



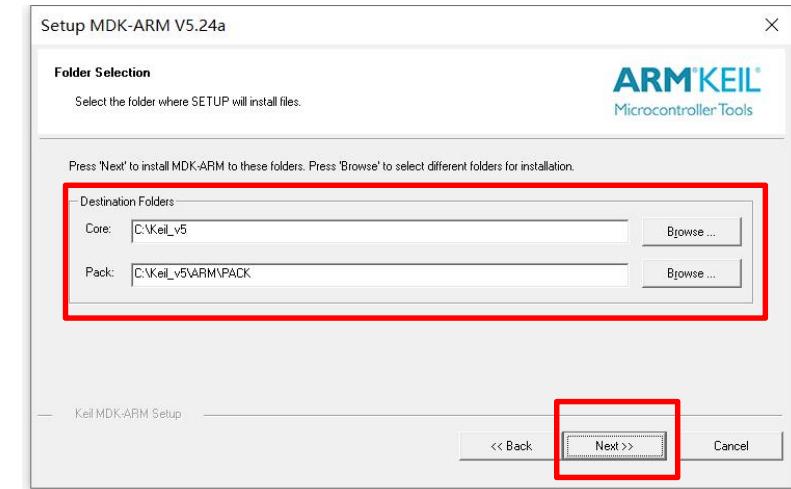
Preparation1: ARM KEIL 安装步骤-1



确认MDK-ARM

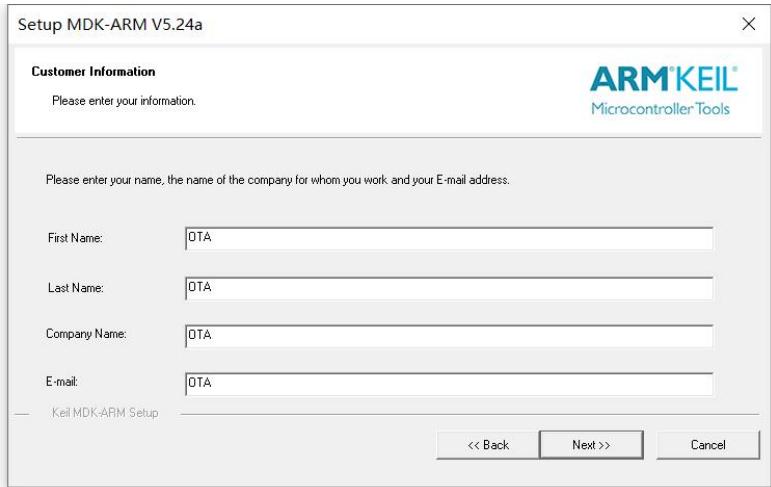


确认License

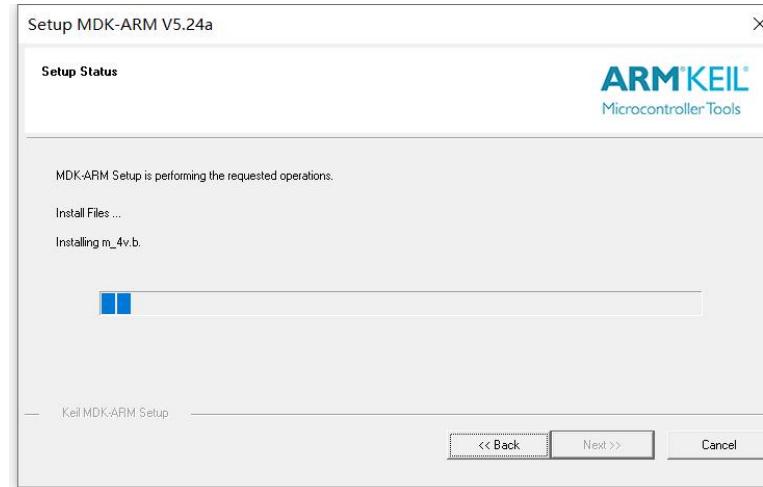


确认安装路径

ARM KEIL 安装步骤-2



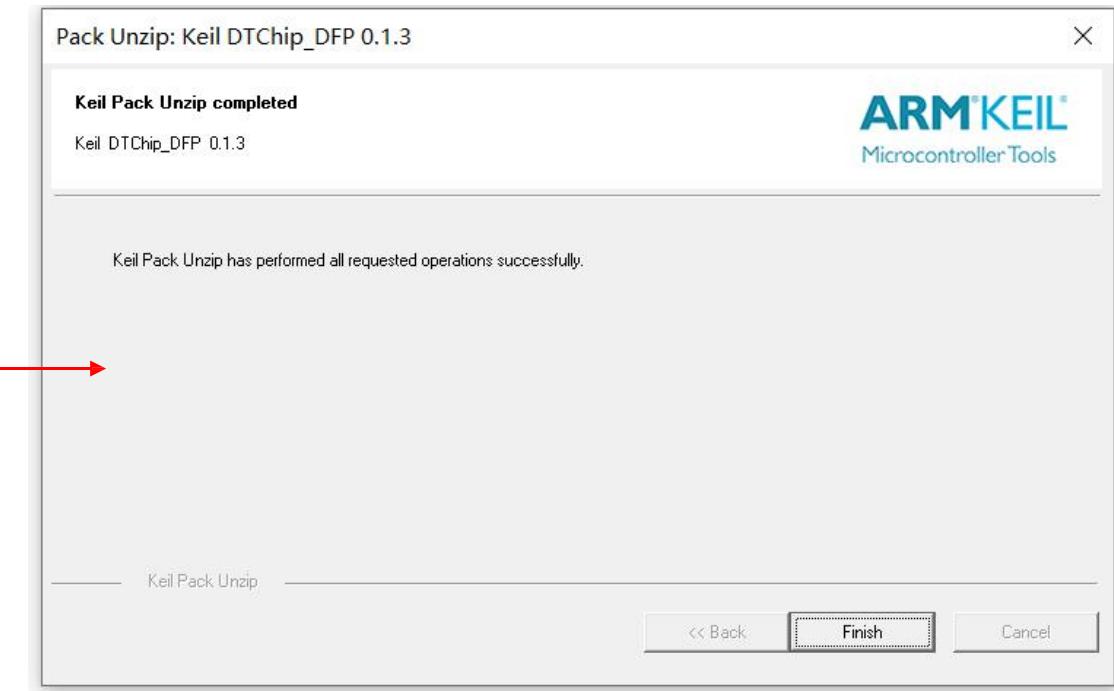
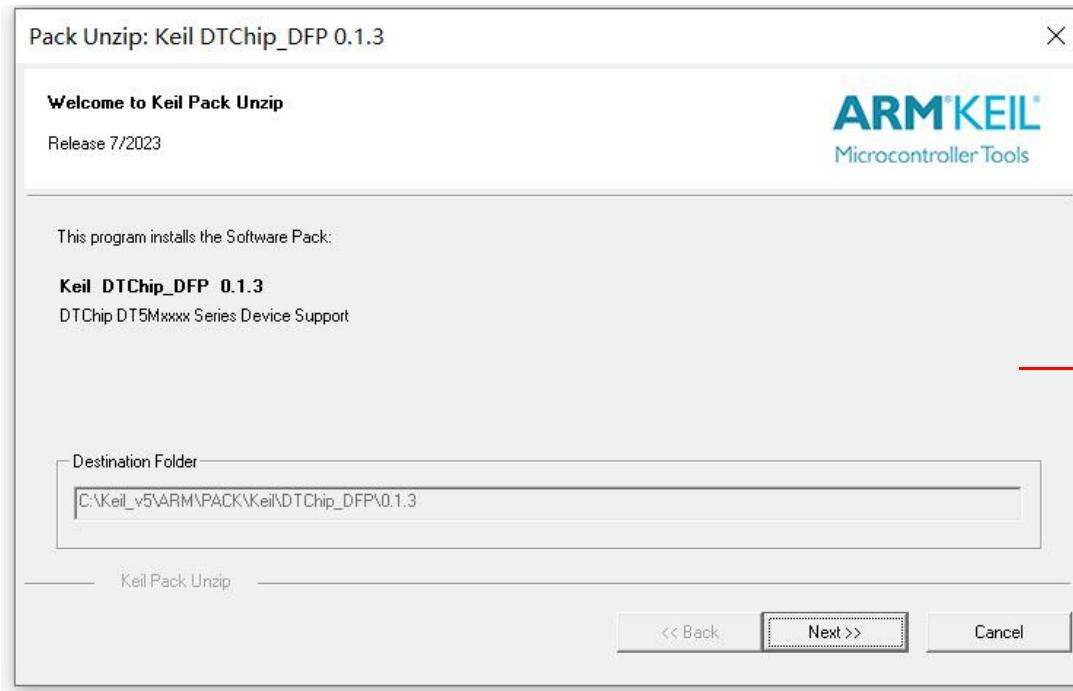
确认用户名



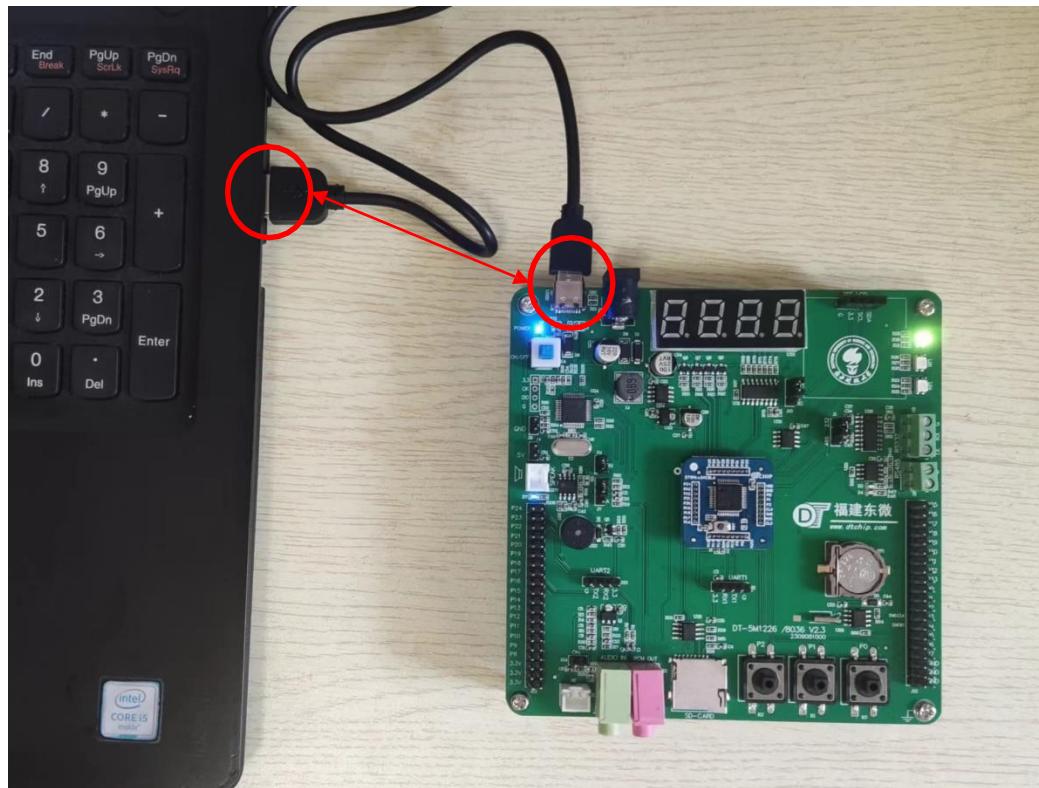
安装进程

确认完成

安装keil.DTChip_DFP.0.1.3.pack芯片支持包

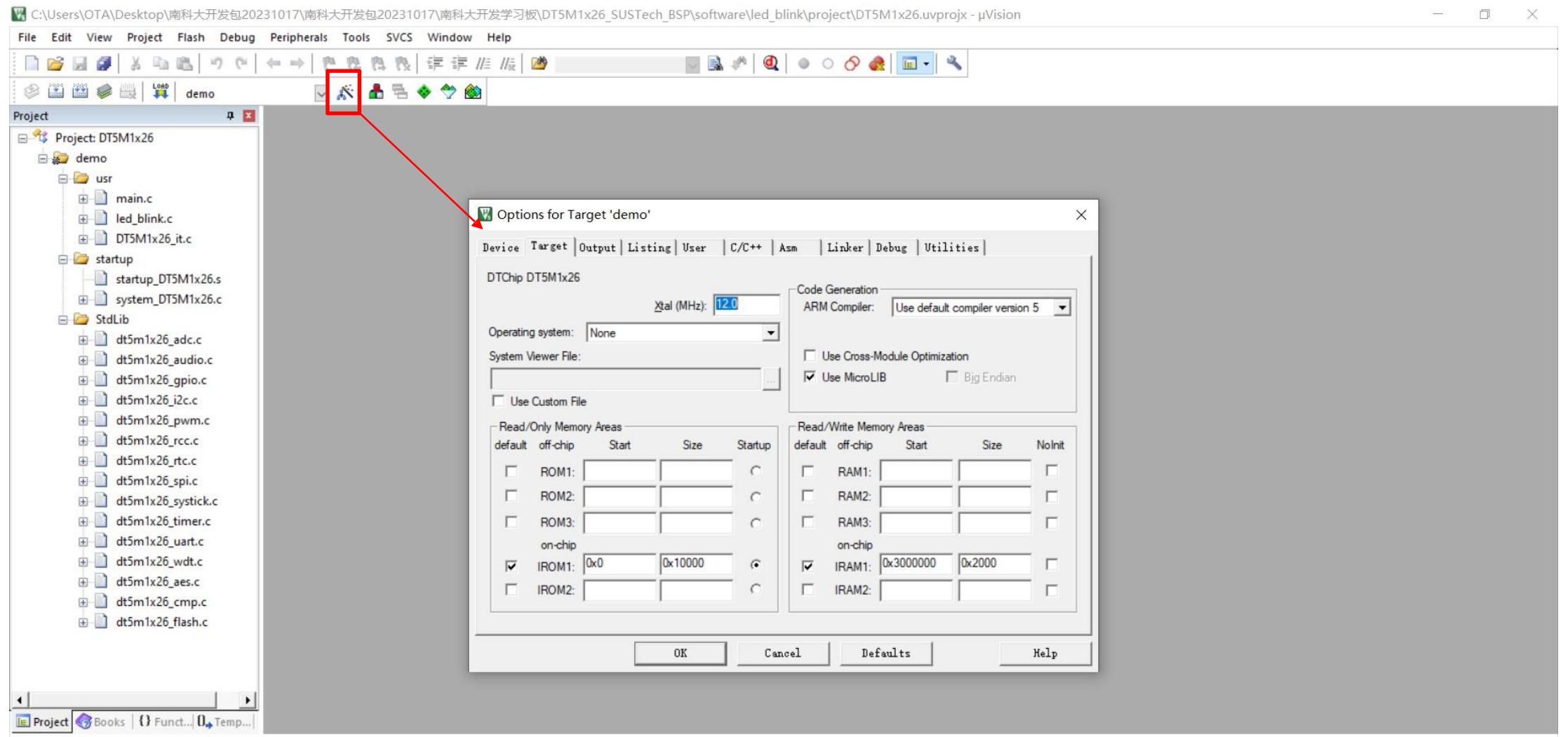


连接DT开发板和电脑，运行led_blink程序

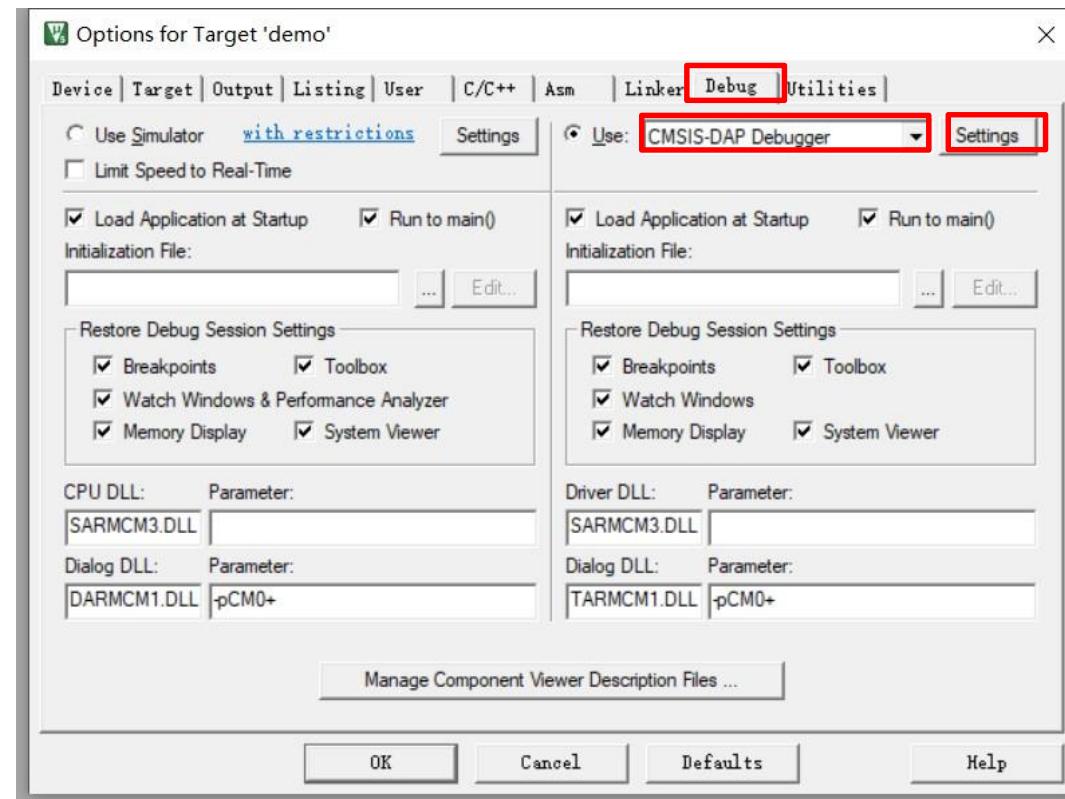


南科大开发包20231017 > 南科大开发学习板 > DT5M1x26_SUSTech_BSP > software > led_blink > project			
名称	修改日期	类型	大小
Listings	2023/10/16 16:45	文件夹	
Objects	2023/10/17 15:53	文件夹	
b2f.exe	2023/8/31 16:29	应用程序	10 KB
bios.hex	2023/8/31 16:29	HEX 文件	9 KB
build_hex.bat	2023/8/31 16:29	Windows 批处理...	1 KB
code_rom	2023/8/31 16:29	文件	5 KB
code_rom.hex	2023/8/31 16:29	HEX 文件	9 KB
DT5M1x26.bin	2023/10/17 15:53	binimage	1 KB
DT5M1x26.uvguixAdministrator	2023/10/17 16:01	ADMINISTRATO...	91 KB
DT5M1x26.uvguix.HIAPAD	2023/8/31 16:29	HIAPAD 文件	101 KB
DT5M1x26.uvoptx	2023/8/31 16:37	UVOPTX 文件	14 KB
DT5M1x26.uvprojx	2023/10/17 11:45	junction5 Project	19 KB
EventRecorderStub.scvd	2023/8/31 16:29	SCVD 文件	1 KB
flash.ini	2023/8/31 16:29	配置设置	1 KB
JLinkLog.txt	2023/8/31 16:29	文本文档	4 KB
JLinkSettings.ini	2023/8/31 16:29	配置设置	1 KB

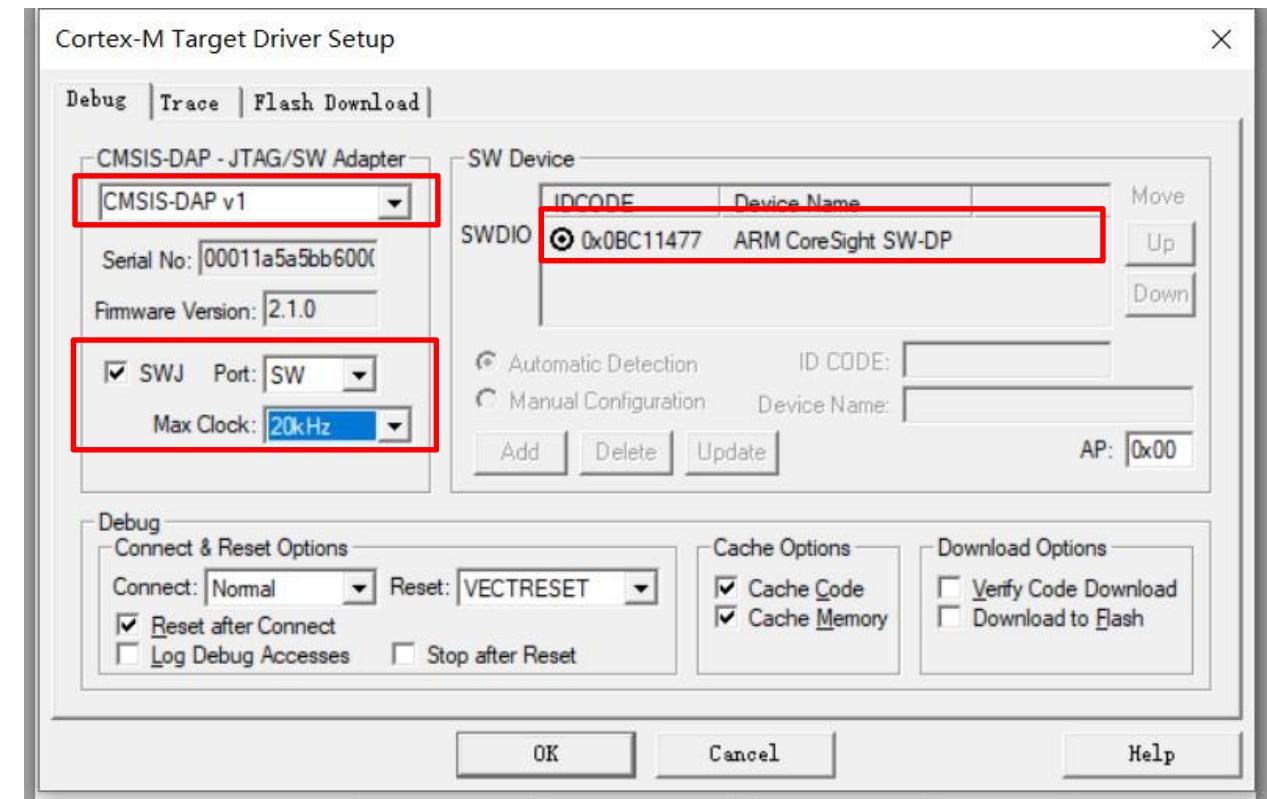
Options for Target



安装keil.DTChip_DFP.0.1.3.pack芯片支持包

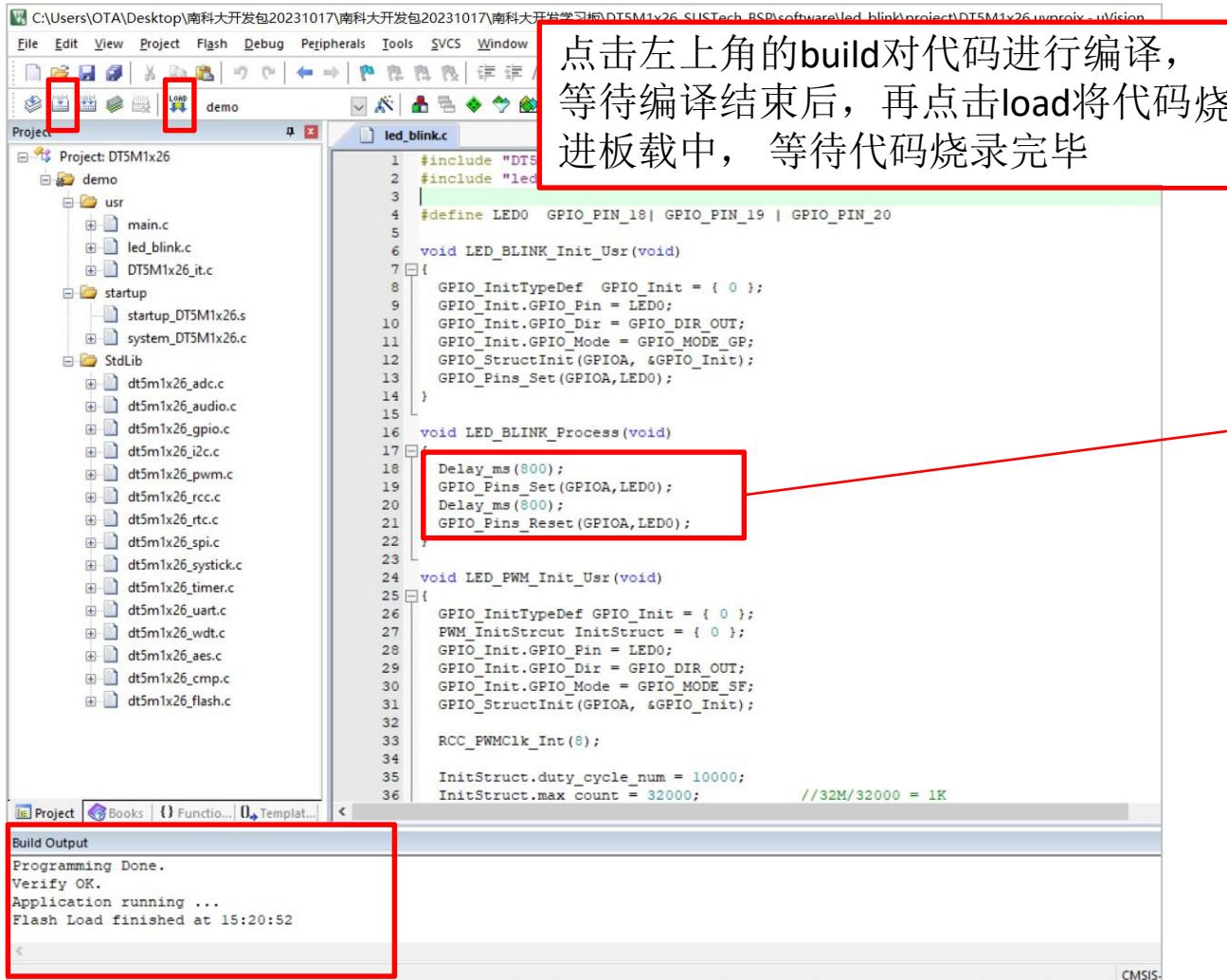


选择CMSIS-DAP Debugger选项，再点击setting进入下一步设置



在SW Device中，若出现设备则说明开发板与电脑连接成功

编译—> 下载



点击左上角的build对代码进行编译，等待编译结束后，再点击load将代码烧录进板载中，等待代码烧录完毕

```
#include "DT5M1x26.h"
#include "led.h"

#define LED0 GPIO_PIN_18|GPIO_PIN_19|GPIO_PIN_20

void LED_BLINK_Init(void)
{
    GPIO_InitTypeDef GPIO_Init = { 0 };
    GPIO_Init.GPIO_Pin = LED0;
    GPIO_Init.GPIO_Dir = GPIO_DIR_OUT;
    GPIO_Init.GPIO_Mode = GPIO_MODE_GP;
    GPIO_StructInit(GPIOA, &GPIO_Init);
    GPIO_Pins_Set(GPIOA, LED0);
}

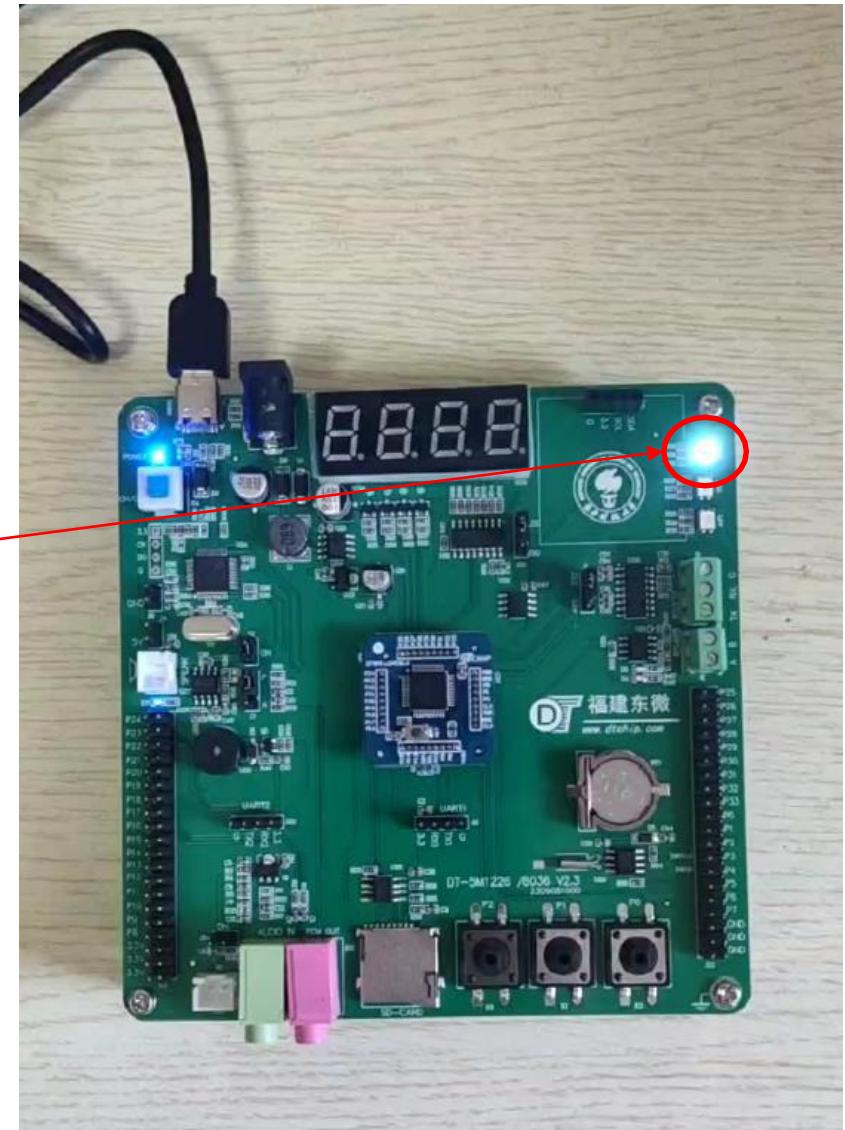
void LED_BLINK_Process(void)
{
    Delay_ms(800);
    GPIO_Pins_Set(GPIOA, LED0);
    Delay_ms(800);
    GPIO_Pins_Reset(GPIOA, LED0);
}

void LED_PWM_Init(void)
{
    GPIO_InitTypeDef GPIO_Init = { 0 };
    PWM_InitStruct InitStruct = { 0 };
    GPIO_Init.GPIO_Pin = LED0;
    GPIO_Init.GPIO_Dir = GPIO_DIR_OUT;
    GPIO_Init.GPIO_Mode = GPIO_MODE_SF;
    GPIO_StructInit(GPIOA, &GPIO_Init);

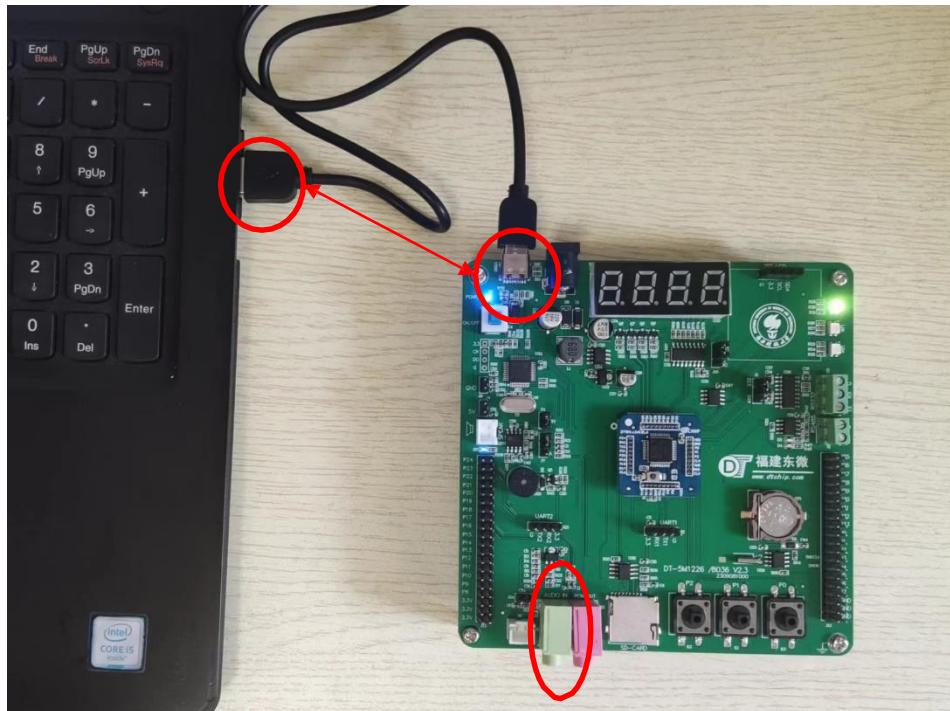
    RCC_PWMClk_Int();
    InitStruct.duty_cycle_num = 10000;
    InitStruct.max_count = 32000; //32M/32000 = 1K
}
```

Build Output

```
Programming Done.
Verify OK.
Application running ...
Flash Load finished at 15:20:52
```



实验内容：使用DT5M1x26播放音频

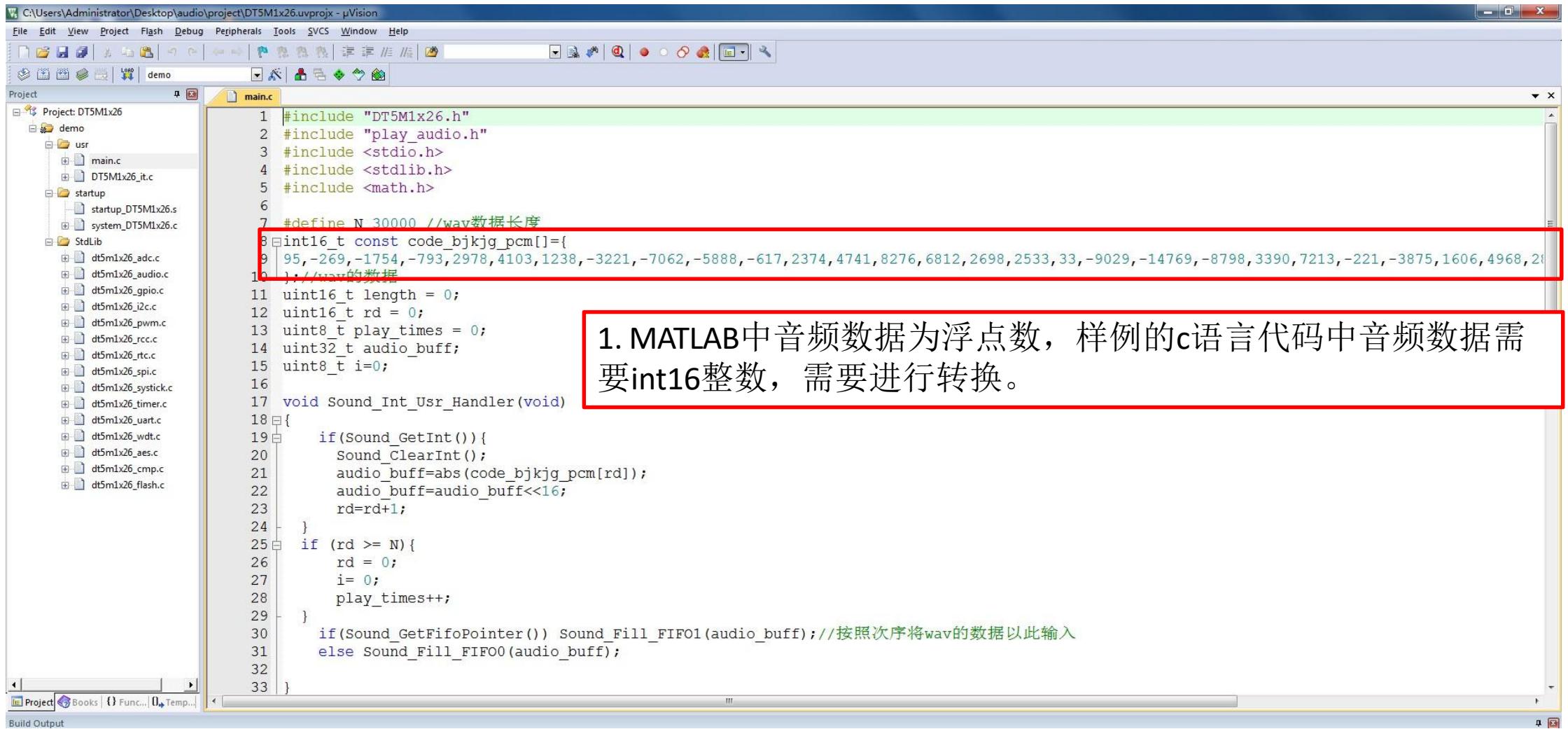


Listings	2023/9/24 23:53	文件夹
Objects	2023/9/24 23:53	文件夹
b2f.exe	2023/8/8 14:01	应用程序 10 KB
bios.hex	2023/8/8 14:01	HEX 文件 10 KB
build_hex.bat	2023/8/8 14:01	Windows 批处理... 1 KB
code_rom	2023/8/8 14:01	文件 5 KB
code_rom.hex	2023/8/8 14:01	HEX 文件 10 KB
DT5M1x26.bin	2023/9/24 23:49	BIN 文件 62 KB
DT5M1x26.uvguix.Administrator	2023/8/9 15:46	ADMINISTRATO... 180 KB
DT5M1x26.uvguix.HIAPAD	2023/8/8 14:01	HIAPAD 文件 100 KB
DT5M1x26.uvguix.toLanzen	2023/9/24 23:53	TOLANZEN 文件 96 KB
DT5M1x26.uvoptx	2023/9/24 23:53	UVOPTX 文件 14 KB
DT5M1x26.uvprojx	2023/9/24 23:53	Microvision Project 18 KB
EventRecorderStub.scvd	2023/8/8 14:01	SCVD 文件 1 KB
flash.ini	2023/8/8 14:01	配置设置 1 KB
fliter.c	2023/9/8 20:59	C 文件 2 KB
JLinkLog.txt	2023/8/9 11:50	文本文档 286 KB
JLinkSettings.ini	2023/8/8 14:01	配置设置 1 KB

注意事项：

1. MATLAB中音频数据为浮点数，样例的c语言代码中音频数据需要int16整数，需要进行转换。
- 2.由于板载Flash的内存有限，每次在代码中只能储存一定量的数据，请勿因存储过多而内存满。
- 3.样例中使用的wav音频在长度和采样率上与需要播放的音频不同，需要修改部分数据。

step1: 数据转换



The screenshot shows the µVision IDE interface with a project named "DT5M1x26" open. The "main.c" file is the active code editor. The code is written in C and includes headers for "DT5M1x26.h", "play_audio.h", "stdio.h", "stdlib.h", and "math.h". It defines a constant `N` representing the wav data length. A specific section of the code, which contains a large array of integers representing audio samples, is highlighted with a red rectangle. This array is declared as `int16_t const code_bjkjg_pcm[]`. Below this highlighted code, a callout box with a red border contains the following text:

1. MATLAB中音频数据为浮点数，样例的c语言代码中音频数据需要int16整数，需要进行转换。

```
#include "DT5M1x26.h"
#include "play_audio.h"
#include <stdio.h>
#include <stdlib.h>
#include <math.h>

#define N 30000 //wav数据长度

int16_t const code_bjkjg_pcm[] = {
    95,-269,-1754,-793,2978,4103,1238,-3221,-7062,-5888,-617,2374,4741,8276,6812,2698,2533,33,-9029,-14769,-8798,3390,7213,-221,-3875,1606,4968,28
}; //wav的数据

uint16_t length = 0;
uint16_t rd = 0;
uint8_t play_times = 0;
uint32_t audio_buff;
uint8_t i=0;

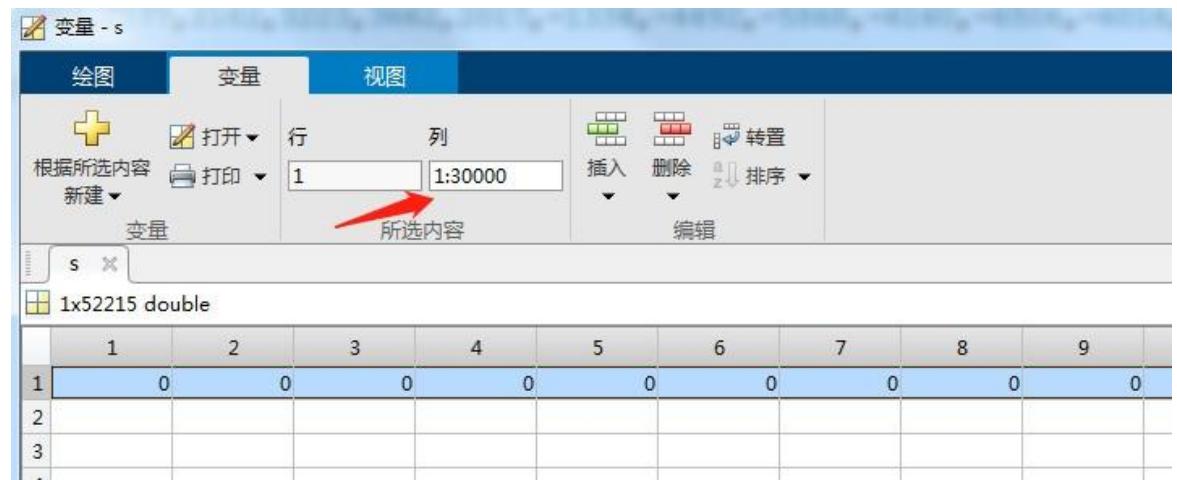
void Sound_Int_Usr_Handler(void)
{
    if(Sound_GetInt()){
        Sound_ClearInt();
        audio_buff=abs(code_bjkjg_pcm[rd]);
        audio_buff=audio_buff<<16;
        rd=rd+1;
    }
    if (rd >= N){
        rd = 0;
        i= 0;
        play_times++;
    }
    if(Sound_GetFifoPointer()) Sound_Fill_FIFO1(audio_buff); //按照次序将wav的数据以此输入
    else Sound_Fill_FIFO0(audio_buff);
}
```

step2: main.c函数中修改数据

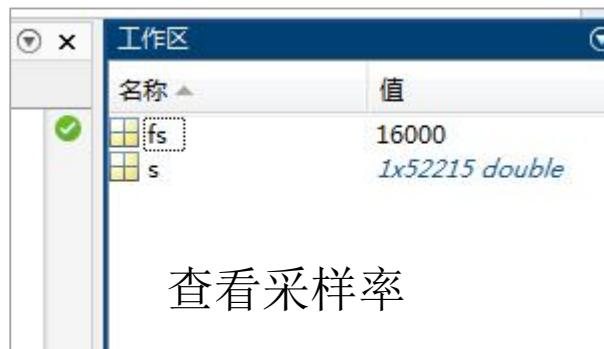
```
testaudio.m * +  
1 [s,fs]=audioread("C_01_01.wav");  
2 s=s';  
3 s=round(s*32768);  
4 %round是四舍五入取整数，在取滤波后值时有效
```

注意事项：

1. 设置行为1，列为1: 30000 (C_01_02为1:29636)
 2. Ctrl+c复制数组文件，再Ctrl+v粘贴到命令行窗口
 3. ctrl+a全选数组文件，并粘贴至main.c对应位置并进行修改



step3 设置采样率—>编译—>运行



在工作区中找到采样率fs，并在main.c代码中修改Sound_Init.sample_rate的赋值。

```
sound_init.unsigne_en = 0;
Sound_Init.sound_format = SOUND_FORMAT_PCM; //wav音频使用PCM采样
Sound_Init.sample_rate = SOUND_SAMPLE_RATE_8K; //wav音频的采样率，对应fs
//对应SOUND_SAMPLE_RATE_8K, SOUND_SAMPLE_RATE_16K, SOUND_SAMPLE_RATE_44.1K
Sound_Init.pcm_frist_value = 0;
Sound_StructInit(&Sound_Init);
```



Organization

- Each group consists of **three or four** students.
- Each group need present **one Lab project** (submit reports for both projects):
- The presentation date is Dec. 14th or 15th for project 1, and Jan. 4th or 5th for Project 2.
- Each presentation is 15 minutes (including Q & A)
 - All team members need to contribute to the presentation.
 - Presentation in English or Chinese.

The presentation should...

- Introduce
 - team
 - objective of the project
 - background review (search more additional information)
 - methodology
- Discuss
 - what you have learned from this study?
 - problems during this project and your solution
 - investigation beyond project tasks
 - critical thinking
- Present
 - relevant data, figure, etc.
 - the results for project tasks (e.g., with demo, Figure, etc.)
 - interpretation of project findings
- Appendix (if any)
- Team effort (e.g., individual contribution)
- Reference
- Q & A (answer questions raised from audience)

Questions



saveas--Save figure to specific file format

Syntax

```
saveas(fig,filename)  
saveas(fig,filename,formattype)
```

Description

`saveas(fig,filename)` saves the figure or Simulink® block diagram specified by `fig` to file `filename`. Specify the file name as a character vector that includes a file extension, for example, '`myplot.jpg`'. The file extension defines the file format. If you do not specify an extension, then `saveas` saves the figure to a FIG-file. To save the current figure, specify `fig` as `gcf`.

`saveas(fig,filename,formattype)` creates the file using the specified file format, `formattype`. If you do not specify a file extension in the file name, for example, '`myplot`', then the standard extension corresponding to the specified format automatically appends to the file name. If you specify a file extension, it does not have to match the format. `saveas` uses `formattype` for the format, but saves the file with the specified extension. Thus, the file extension might not match the actual format used.

print--Print figure or save to specific file format
savefig--Save figure and contents to FIG-file