

DSP Lab. Week 1 Drawing sinusoidal waves

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❖ ANALOG/ELECTRONIC:

✓ Circuits: resistors, capacitors, op-amps



❖ DIGITAL/MICROPROCESSOR

√ Convert x(t) to numbers stored in memory





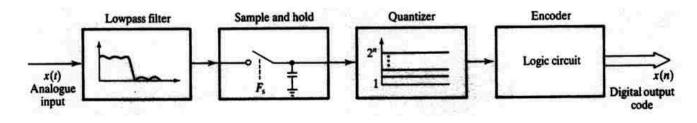


❖ A-to-D

√ Convert x(t) to numbers stored in memory

❖ D-to-A

- √ Convert y[n] back to a "continuous-time" signal, y(t)
- √ y[n] is called a "discrete-time" signal



- **❖ SAMPLING PROCESS**
 - √ Convert x(t) to numbers x[n]
 - √ "n" is an integer; x[n] is a sequence of values √ Think of "n" as the storage address in memory.
 - UNIFORM SAMPLING at t = nTs

$$\checkmark$$
 IDEAL: $x[n] = x(nTs)$

- **❖ SAMPLING RATE (fs)**
 - fs = 1/Ts
 - ✓ NUMBER of SAMPLES PER SECOND ■ Ts = 125 microsec \rightarrow fs = 8000 samples/sec
 - UNITS ARE HERTZ: 8000 Hz
- **❖ UNIFORM SAMPLING at** t = nTs = n/fs • IDEAL: x[n] = x(nTs) = x(n/fs)

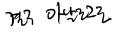
C-to-D

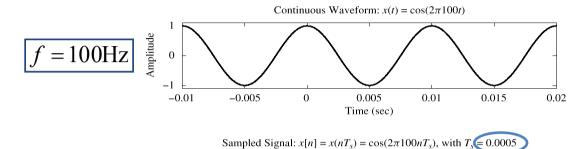
x(t)

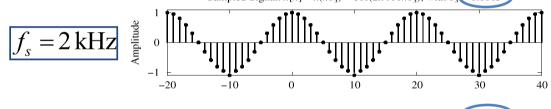
n (t= nts)

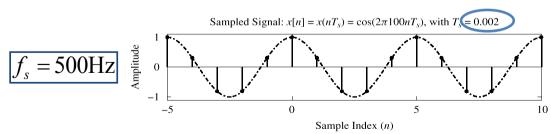
IT= NTo 1 (rad/s) W= 22 of (rad)

radion Laguary normalized "









 f_s : number of samples per a second

 $n = T/T_s$: number of samples per a period

- ❖ HOW OFTEN ?
 - DEPENDS on FREQUENCY of SINUSOID
 - ANSWERED by SHANNON/NYQUIST Theorem
 - ALSO DEPENDS on "RECONSTRUCTION"

Shannon Sampling Theorem

A continuous-time signal x(t) with frequencies no higher than f_{max} can be reconstructed exactly from its samples $x[n] = x(nT_s)$, if the samples are taken at a rate $f_s = 1/T_s$ that is greater than $2f_{\text{max}}$.

- ❖ x[n] is a SAMPLED SINUSOID A list of numbers stored in memory
- ❖ EXAMPLE: audio CD
- CD rate is 44,100 samples per second
 - 16-bit samples Stereo uses 2 channels
- **❖ Number of bytes for 1 minute is**
 - 2 X (16/8) X 60 X 44100 = 10.584 Mbytes

Syunghee University

$$x(t) = A\cos(\omega t + \varphi) \implies \lambda_{M^{2}} A\cos(\hat{\omega}n + \varphi)$$

$$x[n] = x(nT_{s}) = A\cos(\omega nT_{s} + \varphi)$$

$$x[n] = A\cos((\omega T_{s})n + \varphi)$$

$$x[n] = A\cos(\hat{\omega}n + \varphi)$$

$$\hat{\omega} = \omega T_{s} = \frac{\omega}{f_{s}}$$

DEFINE DIGITAL FREQUENCY
Called as Normalised Radian frequency

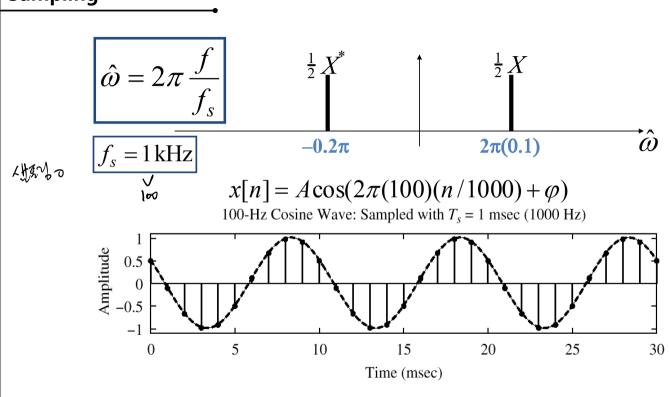
Kyunghee University

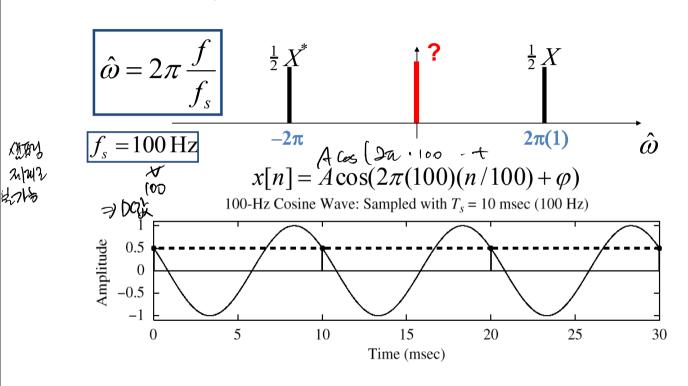
Sampling

- \spadesuit VARIES from 0 to 2π , as f varies from 0 to the sampling frequency
- ❖ UNITS are radians, not rad/sec
 - DIGITAL FREQUENCY is NORMALIZED

$$\hat{\omega} = \omega T_s = \frac{2\pi f}{f_s}$$

 f_s : number of samples per a second





- ❖ Spectrum of x[n] has more than one line for each complex exponential
 - Called ALIASING
 - MANY SPECTRAL LINES
- \clubsuit SPECTRUM is PERIODIC with period = 2 π
 - Because

$$A\cos(\hat{\omega}n + \varphi) = A\cos((\hat{\omega} + 2\pi)n + \varphi)$$

$$x(t) = A\cos(2\pi f t + \phi)$$

$$x(t) = A\cos(2\pi t) t + \psi$$

$$x[n] = x(nT_s) = A\cos(2\pi f nT_s + \phi)$$

$$y(t) = A\cos(2\pi(f + \ell f_s)t + \phi)$$

$$y[n] = y(nT_s) = A\cos(2\pi(f + \ell f_s)nT_s + \phi)$$

$$y[n] = y(nT_s) = A\cos(2\pi(f + \ell))$$

$$= A\cos((2\pi f T_s)n + (2\pi \ell f_s T_s)n + \phi)$$

$$= A\cos((2\pi fT_{\circ})n + 2\pi \ell n + \phi)$$

$$= A\cos((2\pi f T_s)n + 2\pi \ell n + \phi)$$

$$s((2\pi fT_s)n + 2\pi \ell n + \phi)$$

$$+\phi$$
)

$$(\hat{\omega} - A\cos(\hat{\omega}n + \phi) = x[n] \qquad \hat{\omega} = 2\pi f T_{\alpha} = 0$$

$$= A\cos(\hat{\omega}n + \phi) = x[n] \qquad \hat{\omega} = 2\pi f T_s = \frac{2\pi f}{f_s}$$

$y[n] = A\cos((2\pi fT_s)n + \phi) = A\cos(\hat{\omega}n + \phi) = x[n]$

Aliases of the frequency f with respect to the sampling frequency fs

$$ullet$$
 Other Frequencies give the same $\ \hat{\omega}$

$$f_{200}$$

$$(t) - \cos(400\pi t)$$

$$x_1(t) = \cos(400\pi t)$$
 sampled at $f_s = 1000$ Hz

$$-\cos(400\pi^{-n})$$

$$(2400\pi i)$$

$$x_2[n] = \cos(2400\pi \frac{n}{1000}) = \cos(2.4\pi n)$$

$$[n] = \cos(2A\pi n)$$

$$x_2[n] = \cos(2.4\pi n) = \cos(0.4\pi n + 2\pi n) = \cos(0.4\pi n)$$

$$\Rightarrow x_2[n] = x_1[n]$$

$$[n] - [n]$$

$$-\cos(0.4\pi n)$$

$$(2.4\pi n)$$

$$x_1[n] = \cos(400\pi \frac{n}{1000}) = \cos(0.4\pi n)$$

 $x_2(t) = \cos(2400\pi t)$ sampled at $f_s = 1000 \,\text{Hz}$

 $2400\pi - 400\pi = 2\pi(1000)$

$$w[n] = w(nT_s) = A\cos(2\pi(-f_0 + \ell f_s)nT_s - \phi)$$







$$=x[n$$

$$=x[n]$$

$$= A\cos(-2\pi f_0 n T_s + 2\pi \ell - \phi)$$
$$= A\cos(2\pi f_0 n T_s + \phi)$$

$$= A\cos(-2\pi f_0 n T_s + 2\pi \ell f_s T_s - \phi)$$

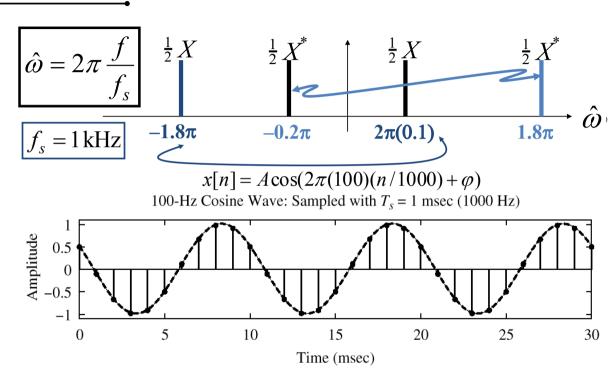
$$f_0 + \ell f_s$$

$$w(t) = A\cos(2\pi(-f_0 + \ell f_s)t - \phi)$$

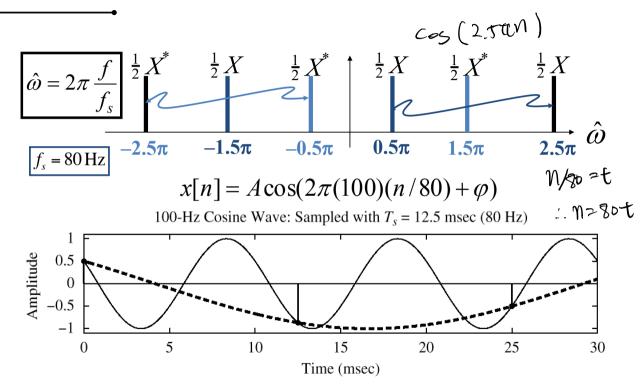
 \Rightarrow ADDING f_s or $2f_s$ or $-f_s$ to the FREQ of x(t) gives exactly the same x[n] • The samples, $x[n] = x(n/f_s)$ are EXACTLY THE SAME VALUES

 \Leftrightarrow GIVEN x[n], WE CAN'T DISTINGUISH f_0 FROM $(f_0 + f_s)$ or $(f_0 + 2f_s)$

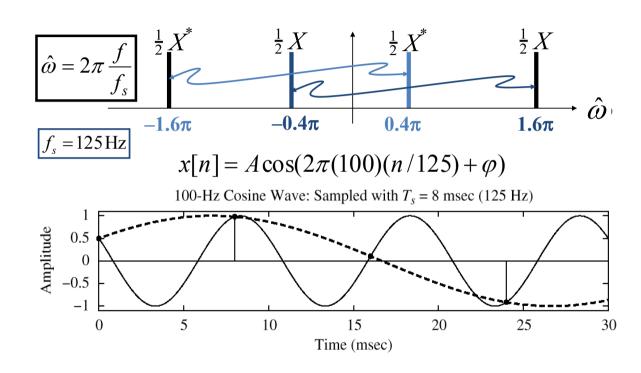
❖ The frequency are callled aliases of the frequency f with respect to the sampling frequency f_s











C-program (Sinusoidal wave) ex1

```
#include <iostream> // cout, cin
   #include 〈fstream〉 // ifstream, ofstream 입출력 파일 라이브러리
   using namespace std;
   #define PHI 3.141592
   int main(){
                                      // 출력 파일 선언
         ofstream outFile;
         outFile.open("data.txt", ios::out); // 출력 파일 data.txt 열기.
         float t, dt, f0;
         t = 0;
                                                                                                  main
         dt = 1./44000; // fs = 44000Hz smapling frequency
10.
                                                                                                  Function
11.
         f0 = 440;
                  // 440Hz signal
12.
         for(int i=0;i<400;i++,t+=dt)
Return
             outFile << t << " " << sin(2.*PHI*f0*t) << endl;
                                                                                                             ☑ 꺾은선형
                                                                                                             ᄪᅄ
14.
         outFile_close();
                                                                                                            更증패
15.
         return 0;
                                                                                                    분산형
16. }
               // 프로그램을 만드는 폴더에서 data.txt를 읽어서 excel로 그래프 그린다.
               // 두 개의 열을 drag한 후, "삽입→분산형"으로 그래프를 그린다.
```

C-program (Sinusoidal wave) ex2

```
x(t) = 2\cos\left(2\pi(50)t + \frac{\pi}{2}\right) + \cos(2\pi(150)t)
f_0 = 50Hz, T_0 = 0.02
f_s = 300Hz, T_s = \frac{1}{300}
```

```
#include <iostream> // cout, cin
#include <fstream> // ifstream. ofstream 입출력 파일 라이브러리
using namespace std:
#define PHI 3.141592
int main()
                              // 출력 파일 선언.
 ofstream outFile;
 outFile.open("data.txt", ios::out); // 출력 파일 data.txt 열기.
 float t = 0, fs = 300., dt = 1. / fs; // 시간, 샘플링 주파수, 샘플링 주기
                                    // 기본주파수. n개 주기 파형. 신호 샘플 개수
 int f0 = 50, n=3, smp cnt;
 smp cnt = (fs / f0)*n;
 for (int i = 0; i \le smp cnt; i++, t += dt)
   outFile << t << " " << 2*cos(2.*PHI*50*t+0.5*PHI) + cos(2.*PHI*150*t) << endl;
 outFile.close();
 return 0;
```

n주기 파형, $n \times T_0 = (신호 샘플 개수) \times T_s$



Week 1 assignment

ये छाम्ब यिष्णार रि

1. 어떤 주기 함수가 다음 식 $x(t) = 2 + 4\cos\left(30\pi t - \frac{1}{5}\pi\right) + 3\sin(40\pi t) + 4\cos\left(60\pi t - \frac{1}{3}\pi\right)$ 으로 주어졌다. 기본 주파수 f_0 , 샘플링 주파수 f_s 를 결정하여 3주기의 x(t)를 출력하라.

2. 변조(Modulation): 신호 $\mathbf{x}(t) = \cos(2\pi f_0 t) \sin(2\pi f_c t)$ 를 그려라. 이 그림은 base band frequency $\mathbf{f}_0 = 200$ Hz인 신호를 carrier frequency $\mathbf{f}_c = 1600$ Hz로 modulation한 신호이다.



Week 1 assignment

"KLAS에 제출할 때 다음 사항을 꼭 지켜주세요"

- 1. 파일명 : "Lab00_요일_대표자이름.zip" Ex) Lab01_목_홍길동.zip (압축 둘은 자유롭게 사용)
- 2. 제출 파일 (보고서와 프로그램을 압축해서 제출)
 - 보고서 파일 (hwp, word): 이름, 학번, 목적, 변수, 알고리즘(순서), 결과 분석, 느낀 점
 - 프로그램

