

一、簡介(Introduction)

利用麥克風陣列來做 Beamforming 偵測聲源所在方位，
再利用 Chebyshev 來壓抑雜訊，最後聲源方位找出。

二、理論(Theoretical)

無限遠處的 source 所發出訊號被兩個麥克風接收，

相對於原點的 phasor 和：

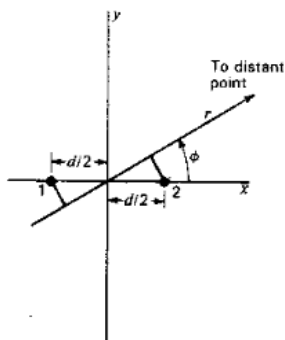
$$\Psi = d_r \cos \phi$$

$$d_r = \frac{2\pi d}{\lambda}$$

$$E(\phi) = E_0 e^{-j\psi/2} + E_0 e^{+j\psi/2}$$

$$= 2E_0 \frac{e^{-j\psi/2} + e^{+j\psi/2}}{2}$$

$$= 2E_0 \cos \frac{\psi}{2} = 2E_0 \cos \left(\frac{d_r}{2} \cos \phi \right)$$



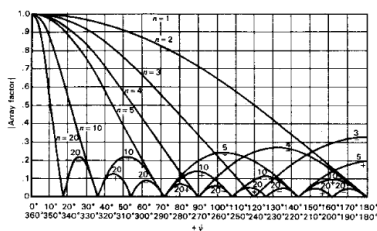
N 個麥克風的情形：

$$E(\phi) = e^{-j(N-1)\psi/2} + \dots + e^{-j\psi/2} + e^{+j\psi/2} +$$

$$e^{+j(N-1)\psi/2}$$

$$= e^{-j\frac{N-1}{2}\psi} \frac{1 - e^{-jN\psi}}{1 - e^{-j\psi}}$$

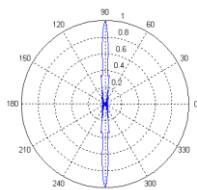
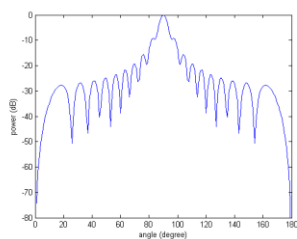
$$= \frac{\sin(N\psi/2)}{\sin(\psi/2)}$$



三、實驗(Experimental)

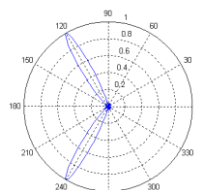
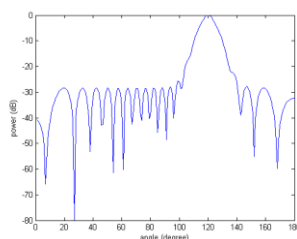
利用 Matlab 模擬 Beamforming 和 Chebyshev

$$N = 20, d = 0.5, r = 100, \lambda = 2$$

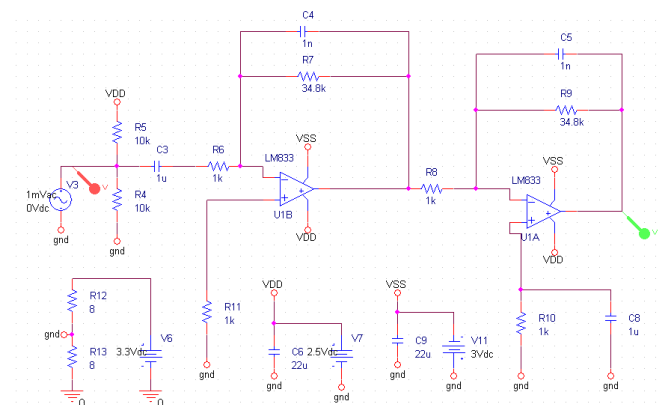


改變接收角度 θ ：

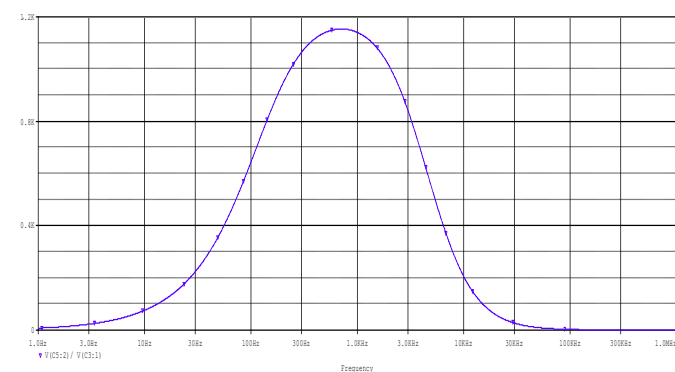
$$N = 20, d = 0.5, r = 100, \lambda = 2, R = 30, \theta = 30$$



模擬前端電路



Frequency response



四、參考文獻(Reference)

Beamforming 參考資料：

https://www.engr.wisc.edu/ece/faculty/vanveen_barry/A_SSP_Mag_88.pdf

<http://www.nari.ee.ethz.ch/~studerc/papers/10ISCASacm.pdf>

所用 IC 資料來源：

1. OP (LM833):

<http://www.national.com/ds/LM/LM833.pdf>

2. ADC (AD7828):

http://www.analog.com/static/imported-files/data_sheets/AD7824_7828.pdf