

# CONVOLUTION

# Mathematik

N-1

$$a[n] * b[n] = \text{output}[k] = \sum_{n=0}^{N-1} a[n] \times b[k-n]$$

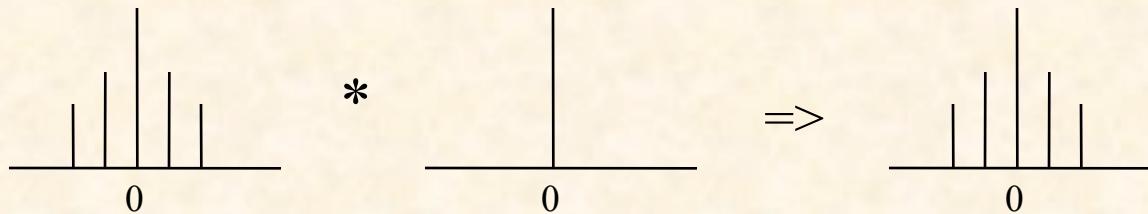
$$\text{length(output)} = \text{length}(a) + \text{length}(b) - 1$$

## Direct Convolution

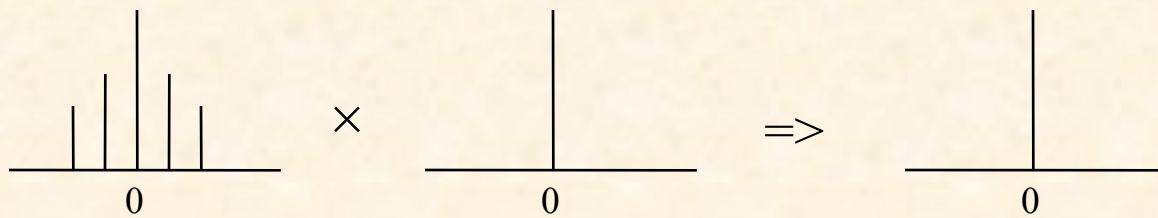
$a = [1 \ 0.75 \ 0.5 \ 1]$  Impulse Response  $b = [0.1 \ 1 \ 0 \ 0.5]$

c[1]	1*0.1	0.75*0.1	0.5*0.1	1*0.1			
		+	+	+			
c[2]		1*1	0.75*1	0.5*1	1*1		
			+	+	+		
c[3]			1*0	0.75*0	0.5*0	1*0	
				+	+	+	
c[4]				1*0.5	0.75*0.5	0.5*0.5	1*0.5
	=	=	=	=	=	=	=
result	0.1	1.075	0.8	1.1	1.375	1.25	0.5

Convolution:



Multiplikation:



*Convolution ist Multiplikation im Frequenzbereich!*

## Fast Convolution:

Produkt von zwei N großen DFT's ist gleich der Convolution von zwei N großen Signalen:

⇒ Convolution durch zero-padding, FFT's der Eingangssignale, Multiplikation und anschließender IFFT.

Sig. a à zero-pad à FFT



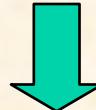
× à IFFT à result ( $a^*b$ )

Sig. b à zero-pad à FFT

Convolution  $\longleftrightarrow$  digitale Filter

FIR-Filter:

$$y[n] = (x_1 \times a[n]) \pm (x_2 \times a[n-1]) \pm \dots (x_i \times a[n-j])$$



$$y[n] = \sum_{m=0}^{N-1} x[m] \times a[n-m]$$

è Jeder FIR-Filter kann als Convolution bezeichnet werden.

# Convolution in C-Sound

## Opcode „convolve“:

ar1[,...[,ar4]]] convolve ain,ifilcod [, ichannel]

- Fast Convolution
- C-Sound Utility „cveral“ konvertiert Soundfile in .cv Datei  
à -b (*begin*), -d (*duration*), -c (*channel*), -s (*sample rate*)

## Opcode „dconv“

ar dconv asig, isize, irfn

- direct Convolution
- Keine Voranalyse nötig

## Faltungshall:

```
instr 1  
iamp = 0.2  
  
ain1 diskin "pattern2.wav", 1  
a1 convolve ain1, "smallchamber.cv",1  
  
out a1*iamp  
  
endin
```

```
instr 1  
iamp = 0.2  
  
ain1 diskin "pattern2.wav", 1  
a1 convolve ain1, „bigchurch.cv”,1  
  
out a1*iamp  
  
endin
```

instr 2

iamp = .005

ain1 diskin "pattern2.wav", 1  
a1 convolve ain1, "pattern2.cv" ; erste 2 sek.

imix = p4

adel delay imix\*ain1, 2

out a1\*iamp+adel

endin

;Score:

i2 0 17 .1

i2 + . .9

i2 + . .3

e

```
instr 3
```

```
iamp = .008
```

```
itable = 1
```

```
isize = ftlen(itable)
```

```
ain diskin "sheila.aiff", 1
```

```
a1 dconv ain, isize , itable
```

```
out a1*iamp
```

```
endin
```

```
;score
```

```
f1 0 65536 1 "bell.aiff" 0 0 1
```

```
i3 0 10
```

```
e
```