
Image Resampling

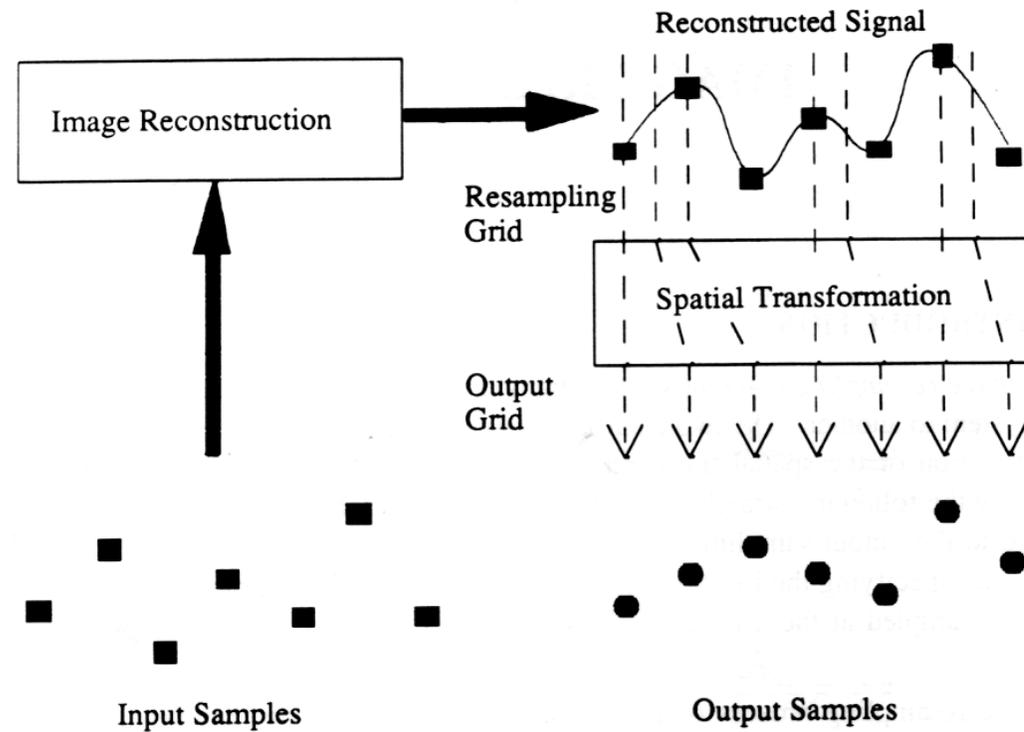
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Objectives

- In this lecture we review image resampling:
 - Ideal resampling
 - Mathematical formulation
 - Resampling filter
 - Tradeoffs between accuracy and complexity
 - Software implementation

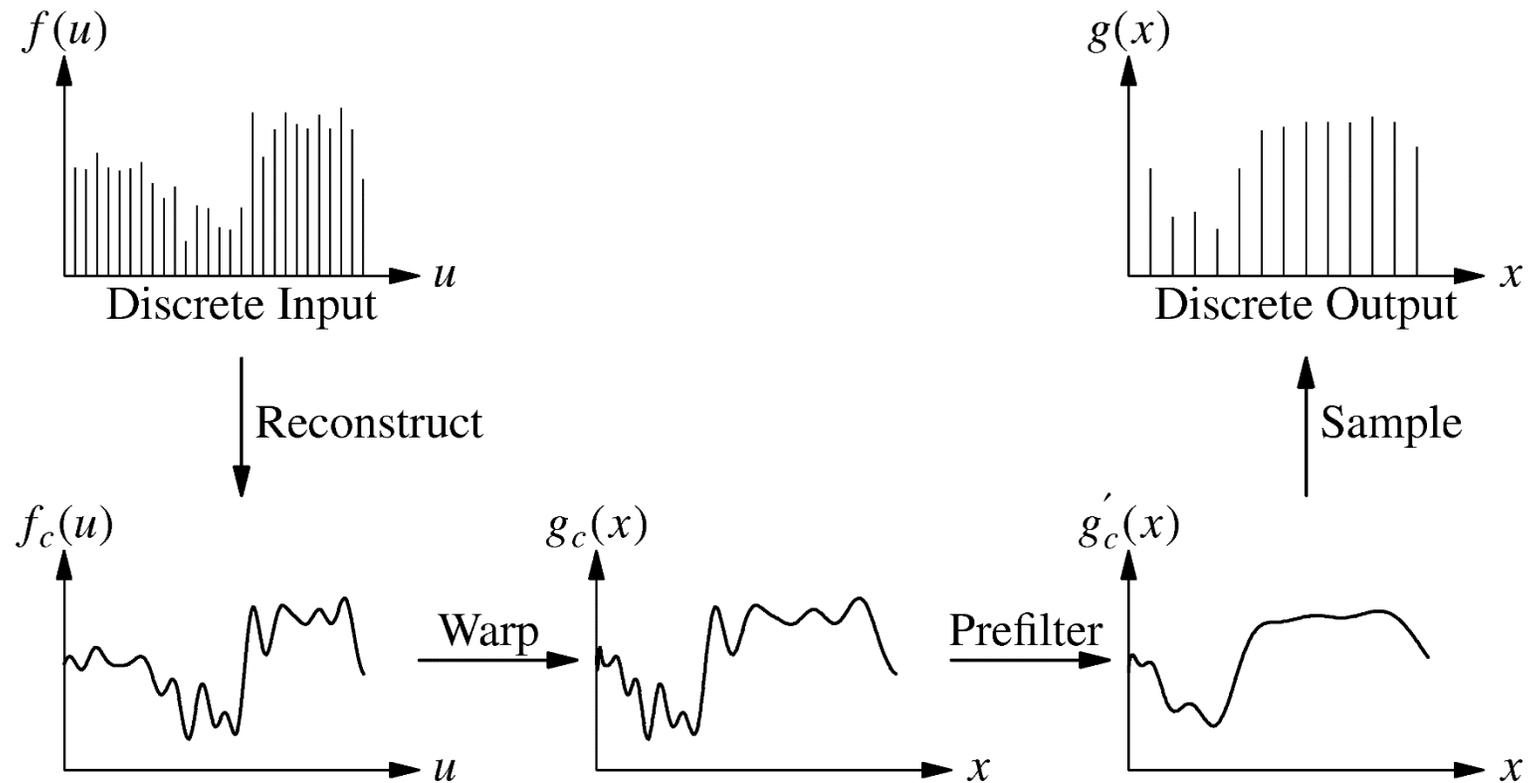
Definition

- Image Resampling: Process of transformation a sampled image from one coordinate system to another.



Ideal Resampling

- Ideal Resampling: reconstruction, warping, prefiltering, sampling



Mathematical Formulation (1)

Stages :

Math Definition :

Discrete Input

$$f(u), u \in z$$

Reconstruction Input

$$f_c = f(u) * r(u) = \sum_{k \in z} f(k)r(u - k)$$

Warped Signal

$$g_c(x) = f_c(m^{-1}(x))$$

Continuous Output

$$g'_c(x) = g_c(x) * h(x) = \int g_c(t)h(x - t)dt$$

Discrete Output

$$g(x) = g'_c(x)S(x)$$

Two filtering components : reconstruction and prefiltering (bandlimiting warped signal before sampling). Cascade them into one by working backwards from $g(x)$ to $f(u)$:

$$g(x) = g'_c(x) \quad \text{for } x \in z$$

$$g(x) = \int f_c(m^{-1}(t))h(x - t)dt = \int \left[\sum_{k \in z} f(k)r(m^{-1}(t) - k) \right] h(x - t)dt$$

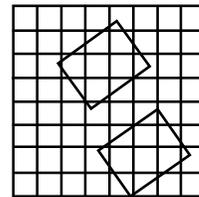
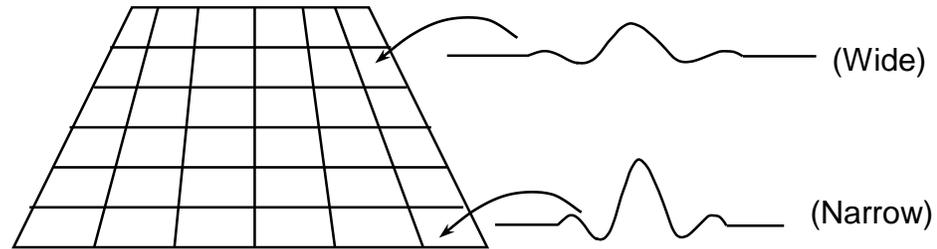
$$g(x) = \sum_{k \in z} f(k)\rho(x, k) \quad \text{where } \rho(x, k) = \int r(m^{-1}(t) - k)h(x - t)dt$$

Mathematical Formulation (2)

$$\rho(x, k) = \int r(m^{-1}(t) - k)h(x - t)dt$$

↑
Selects filter response
used to index filter
coefficients

← Spatially varying resampling
filter expressed in terms
of output space



Mathematical Formulation (3)

- We can express $\rho(x,k)$ in terms of input space:

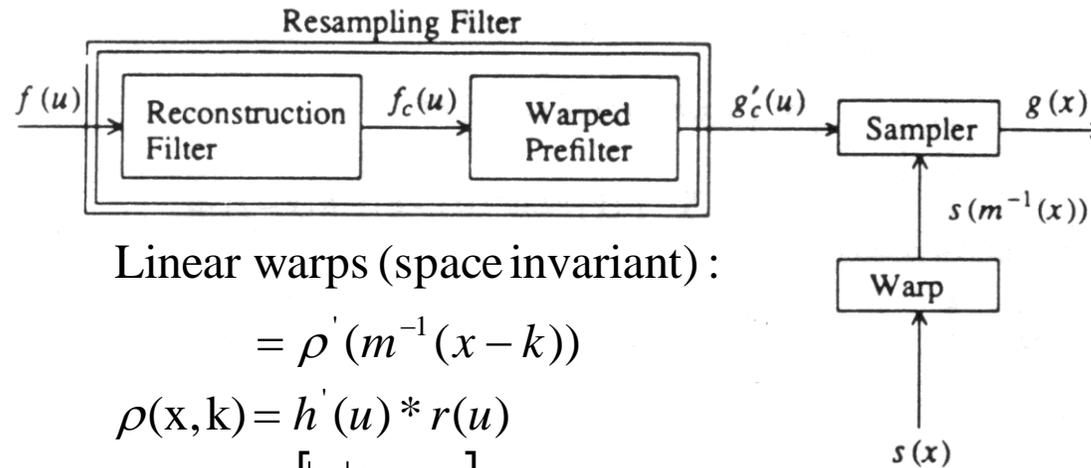
Let $t = m(u)$, we have

$$\rho(x, k) = \int r(u - k)h(x - m(u)) \left| \frac{\partial m}{\partial u} \right| dt$$

where $\left| \frac{\partial m}{\partial u} \right|$ is the determinant of Jacobian matrix :

$$1\text{D} : \left| \frac{\partial m}{\partial u} \right| = \frac{dm}{du} \quad 2\text{D} : \left| \frac{\partial m}{\partial u} \right| = \begin{vmatrix} x_u & x_v \\ y_u & y_v \end{vmatrix} \text{ where } x_u = \frac{\partial x}{\partial u}$$

Resampling Filter



Linear warps (space invariant) :

$$\begin{aligned}
 &= \rho'(m^{-1}(x - k)) \\
 \rho(x, k) &= h'(u) * r(u) \\
 &= [|J| h(uJ)] * r(u)
 \end{aligned}$$

$\rho_{mag}(x, k) = r(u)$: Shape of $r(u)$ remains the same, independently of $m(u)$, (independently of scale factor)

$\rho_{min}(x, k) = |J|h(uJ)$: Shape of prefilter is based on desired frequency response characteristic (performance in passband and stopband). Unlike $r(u)$, though, the prefilter must be scaled proportional to the minification factor (broader and shorter for more minification)

Fourier Transform Pairs

- The shape of ρ_{min} is a direct consequence of the reciprocal relation between the spatial and frequency domains.

\leftrightarrow denotes Fourier Transform pair

$$H(u) = \int h(u) e^{-i2\pi fu} du$$

$$H(m(u)) = \int h(m(u)) e^{-i2\pi fu} du$$

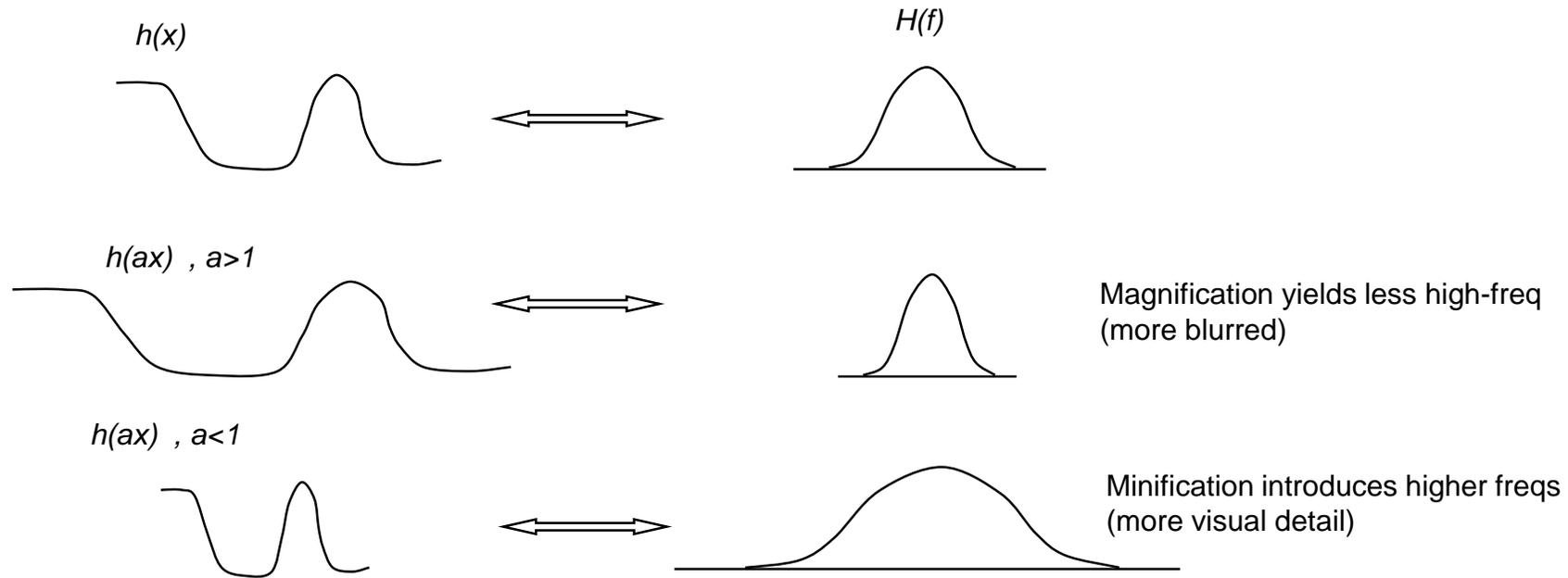
$$\text{Let } x = au = m(u) \text{ and } dx = \left| \frac{\partial m}{\partial u} \right| du$$

$$H(m(u)) = \int h(x) e^{-i2\pi f m^{-1}(x)} \frac{dx}{\left| \frac{\partial m}{\partial u} \right|} \text{ where } m^{-1}(x) = \frac{x}{a}; \quad \left| \frac{\partial m}{\partial u} \right| = |J| = a$$

$$h(au) \leftrightarrow \frac{1}{a} \int h(x) e^{-i2\pi \frac{fx}{a}} dx$$

$$h(au) \leftrightarrow \frac{1}{a} H\left(\frac{f}{a}\right)$$

Reciprocal Relationship



Intuition: $f = 1 / T$

Consequences: narrow filters in spatial domain (desirable) yield wide frequency spectrums (undesirable). Tradeoff between accuracy and complexity.

Software (1)

```
resample1D(IN, OUT, INlen, OUTlen, filtertype, offset)
unsigned char *IN, *OUT;
int INlen, OUTlen, filtertype, offset;
{
    int i;
    int left, right;    // kernel extent in input
    int pixel;         // input pixel value
    double u, x;       // input (u) , output (x)
    double scale;      // resampling scale factor
    double (*filter)(); // pointer to filter fct
    double fwidth;     // filter width (support)
    double fscale;     // filter amplitude scale
    double weight;     // kernel weight
    double acc;        // convolution accumulator

    scale = (double) OUTlen / INlen;
```

Software (2)

```
switch(filtertype) {
case 0: filter = boxFilter; // box filter (nearest nbr)
        fwidth = .5;
        break;
case 1: filter = triFilter; // triangle filter (lin intrp)
        fwidth = 1;
        break;
case 2: filter = cubicConv; // cubic convolution filter
        fwidth = 2;
        break;
case 3: filter = lanczos3; // Lanczos3 windowed sinc fct
        fwidth = 3;
        break;
case 4: filter = hann4; // Hann windowed sinc function fwidth = 4;
        // 8-point kernel
        break;
}
```

Software (3)

```
if(scale < 1.0) { // minification: h(x) -> h(x*scale)*scale
    fwidth = fwidth / scale; // broaden filter
    fscale = scale; // lower amplitude

    /* roundoff fwidth to int to avoid intensity modulation */
    if(filtertype == 0) {
        fwidth = CEILING(fwidth);
        fscale = 1.0 / (2*fwidth);
    }
} else fscale = 1.0;

// project each output pixel to input, center kernel, and convolve
for(x=0; x<OUTlen; x++) {
    /* map output x to input u: inverse mapping */
    u = x / scale;

    /* left and right extent of kernel centered at u */
    if(u - fwidth < 0) {
        left = FLOOR (u - fwidth);
    else left = CEILING(u - fwidth);
    right = FLOOR(u + fwidth);
}
```

Software (4)

```
/* reset acc for collecting convolution products */
acc = 0;

/* weigh input pixels around u with kernel */
for(i=left; i <= right; i++) {
    pixel = IN[ CLAMP(i, 0, INlen-1)*offset];
    weight = (*filter)((u - i) * fscale);
    acc += (pixel * weight);
}

/* assign weighted accumulator to OUT */
OUT[x*offset] = acc * fscale;
}
}
```

Software (5)

```
double boxFilter(double t)
{
    if((t > -.5) && (t <= .5)) return(1.0);
    return(0.0);
}
```

```
double triFilter(double t)
{
    if(t < 0) t = -t;
    if(t < 1.0) return(1.0 - t);
    return(0.0);
}
```

Software (6)

```
double cubicConv(double t)
{
    double A, t2, t3;

    if(t < 0) t = -t;
    t2 = t * t;
    t3 = t2 * t;

    A = -1.0; // user-specified free parameter
    if(t < 1.0) return((A+2)*t3 - (A+3)*t2 + 1);
    if(t < 2.0) return(A*(t3 - 5*t2 + 8*t - 4));
    return(0.0);
}
```

Software (7)

```
double sinc(double t)
{
    t *= PI;
    if(t != 0) return(sin(t) / t);
    return(1.0);
}
```

```
double lanczos3(double t)
{
    if(t < 0) t = -t;
    if(t < 3.0) return(sinc(t) * sinc(t/3.0));
    return(0.0);
}
```

Software (8)

```
double hann4(double t)
{
    int N = 4;           // fixed filter width
    if(t < 0) t = -t;
    if(t < N)
        return(sinc(t) * (.5 + .5*cos(PI*t / N)));
    return(0.0);
}
```