

# The Hermite transform : a survey

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# **The Hermite Transform: A Survey**

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With this survey on the Hermite transformation we want to pursue the following two goals. First, we want to provide a comprehensive and up-to-date description of the Hermite transformation, its underlying philosophy, and its most important properties and their implications for applications. As so often when publications and development go hand-in-hand, new insights have led to changes in or generalizations of already published results, and not all of these changes have been considered sufficiently substantial to be published separately. As a consequence, the existing publications on the Hermite transformation do not fully reflect our most recent insights, and the current paper intends to remedy this. Second, we also want to share some new results. Two specific new results, that is, partial signal decompositions and intersection curvatures, are therefore treated in more detail than other aspects.

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#### 1. INTRODUCTION

The Hermite transformation was introduced about 15 years agothrough two related publications [1, 2]. It was originally developed to provide a mathematical model for interpreting receptive fields in the early stages of (human) spatial vision [2, 3]. It was however also one of the first instances where an overcomplete signal representation was considered for applications. This early belief in the potential benefits of overcomplete representations, which was motivated by our belief in the superior characteristics of the human visual system, has since been shared by an increasing number of researchers. The acceptance of the Hermite transformation has grown due to demonstrated applications in diverse areas such as image coding [4–6], image fusion [7, 8], motion estimation (optic flow) [9, 10], image processing [11–15], image parameter estimation with applications in image quality prediction [16–19], image analysis [20], image indexing [21, 22], and modelling of the human visual system [23-25]. The Hermite transformation has recently also been extended to more than two dimensions using the framework of Clifford analysis [26].

The Hermite transformation is an example of an overcomplete signal representation, or signal decomposition. It is an invertible mapping of the original signal (specified by its sample values) to an alternative representation. The potential benefit of such alternative representations is that they make different characteristics of the signal explicit and hence accessible for processing or coding. The interest in a specific signal representation is therefore determined by its usefulness in applications. The recent signal-processing literature shows a fast growth, under the impulse of wavelet transformation [27, 28] and frame [29] theories, of available signal decompositions. It seems that there are endless alternatives for analyzing and resynthesizing a signal, and indeed there are. Obviously, the demand of global correspondence, that is, that analysis followed by synthesis results in the original signal, is not very restrictive. There is therefore ample room for imposing additional demands on signal decompositions. In this paper, we specifically consider signal decompositions from the point of view of applications in adaptive image processing (and coding). We argue that these applications are greatly simplified if signal decompositions have the additional property of local correspondence, that is, if signals can be locally recovered from the decomposition, using only the coefficients at the position where processing is required. The Hermite transformation is a specific example of such a local signal decomposition, but one with some very useful extra features.

The adaptive processing (or coding) of images with the help of local image decompositions will be a three-stage process, as shown in Figure 1 [20]. In the analysis stage *A* the image is decomposed into a sum of a priori selected (basic) functions. We assume that these functions span the signal space. They may comprise either a basis, in case they are linearly independent, or a frame, in case they are linearly dependent. The analysis stage is equivalent to a transformation from an image to a set of coefficients, that is, the weights that need to be assigned to the basic functions in order to reconstruct the image. We assume here that all these basic functions are displaced versions of a limited set of templates of



FIGURE 1: Algorithmic structure for adaptive image processing with local image decompositions. In the analysis step *A*, the input image  $l(\mathbf{x})$  is transformed into a set of coefficients  $[l_n(\mathbf{p})]$ , where *n* indexes the different templates and **p** the different positions. The coefficients  $[l]_{\mathbf{p}} = [l_n(\mathbf{p}), \forall n]$  at a given position **p** undergo a transformation  $T(C_{\mathbf{p}})$ . The control variables  $C_{\mathbf{p}}$  at position **p** are derived from  $[l]_{\mathbf{p}}$  and are used to select the desired transformation *T*. The transformed coefficients  $[\hat{l}_n(\mathbf{p})]$  are combined in the synthesis step *S* to obtain the processed image  $\hat{l}(\mathbf{x})$ .

approximately the same size. The extension to basic functions of substantially different sizes can be accomplished by extending the signal decomposition to a multiresolution scheme, that is, by incorporating it in a pyramid structure [1, 11, 28, 30] or in a scale-space setting [31–34]. The weights of the basic functions can be assembled according to template (i.e., the weights for all positions of a given template) or according to position (i.e., the weights for all templates at a given position). In the second stage of the adaptive algorithm of Figure 1, the coefficients are assembled according to position. The original coefficients at a given position are interpreted and mapped (by T) into processed coefficients. Since only coefficients at the same position are combined, this mapping can easily be made to vary across positions. To implement this adaptivity, we derive at each position one or more control variables  $C_{\mathbf{p}}$  that determine which (possibly nonlinear) transformation T has to be applied to the coefficients at that position. These control variables are intended to measure changes in the image itself (e.g., uniform versus edge region, or variable orientation), or changes in the degradation (e.g., level-dependent noise such as Poisson noise), so that an adequate transformation can be selected. An image decomposition that allows the image to be reconstructed locally obviously allows any desired local property to be derived. Some important local features that will be introduced later are the local average, the local energy (or contrast), the local dimensionality, and the local orientation. In the synthesis stage S of the adaptive processing algorithm, the processed coefficients are applied to their respective basic functions. The weighted sum is the processed output image.

In Section 2, we formalize the concept of local signal decompositions. We show how the idea of performing identical measurements at regular intervals leads to the definition of filters and filter banks. Next, in Section 3, we introduce partial signal decompositions as a specific case of local signal decompositions where the number of required filter operations can be substantially reduced. It is shown in Section 4 how the simplest instance of such a partial signal decomposition, called residue-image processing [15], can be applied for noise reduction and contrast enhancement. Starting from Section 5, we focus on one specific case of local decompositions, that is, the Hermite transformation for twodimensional signals [1, 2, 13]. We define the Hermite transformation and show how the mapping between the image and the Hermite coefficients can be implemented. We also discuss some important properties of the Hermite transformation. Next, in Section 6, we show how local coordinate axes that are oriented along an image-dependent direction can be created. Many geometric image features, such as the isophote and flow-line curvatures and the newly-introduced intersection curvatures, are most easily expressed in a coordinate system, that is, oriented along the gradient direction. These geometric features are discussed in Section 7, and it is shown how they can be used to identify conspicuous details (such as corners, junctions, extrema, etc.) in images. Such details are for instance well suited for performing image matching, such as that required for stereo vision and motion estimation. In Section 8 we apply the concept of partial signal decompositions to the Hermite transformation in order to devise an anisotropic noise-reduction algorithm.

#### 2. LOCAL SIGNAL DECOMPOSITIONS

We assume that the input signals  $l(\mathbf{x})$  are defined for a (compact) subset  $\mathbb{F}$  of the *D*-dimensional Euclidean space  $\mathbb{R}^D$ . This subset  $\mathbb{F}$  can be either continuous or discrete. In the case of image processing, the dimension D = 2, for static images, or D = 3, for image sequences. We assume that the image analysis is linear and hence consists of bounded linear functionals [24], that is, linear mappings from the input signal  $l(\mathbf{x})$  to the finite real/complex numbers that can be expressed as the inproduct between a test function  $\phi(\mathbf{x})$  and the input function  $l(\mathbf{x})$ . We moreover assume that this inproduct can be expressed as an integral

$$\langle \phi(\mathbf{x}), l(\mathbf{x}) \rangle = \int_{\mathbb{F}} \phi(\mathbf{x}) \cdot l(\mathbf{x}) d\mathbf{x},$$
 (1)

in case of continuous signals, or as a sum

$$\langle \phi(\mathbf{x}), l(\mathbf{x}) \rangle = \sum_{\mathbb{F}} \phi(\mathbf{x}) \cdot l(\mathbf{x}),$$
 (2)

in case of discrete signals. The space of test functions that we consider for analyzing the image is denoted by V. Test functions are very often Schwartz functions (smooth localized functions) [35]. The signal space contains those signals that result in finite measurements for all test functions in V, and is hence equal to the dual space V' of bounded linear functionals on V.

In a local signal decomposition, we decompose a *D*-dimensional input signal  $l(\mathbf{x})$ , for  $\mathbf{x} \in \mathbb{F}$ , into windowed signals  $l(\mathbf{x}) \cdot w(\mathbf{x} - \mathbf{p})$ , for  $\mathbf{p} \in \mathcal{P}$ . The windowing function



FIGURE 2: Mapping from the input image  $[l(\mathbf{x}), \mathbf{x} \in \mathbb{F}]$  to the coefficients  $[\hat{l}_n(\mathbf{p}), \mathbf{p} \in \mathcal{P}]$  of order *n*, depicted as a two-stage process. The standard notations \* and  $\downarrow$  are used to denote the convolution and (down-)sampling operations, respectively.

 $w(\mathbf{x})$  has limited support, so that  $w(\mathbf{x} - \mathbf{p})$  is zero except in the neighbourhood of sampling position  $\mathbf{p} \in \mathcal{P}$ . The windowed signal  $l(\mathbf{x}) \cdot w(\mathbf{x} - \mathbf{p})$  is therefore localized around position  $\mathbf{p} \in \mathcal{P}$ . The set  $\mathcal{P}$  contains all positions  $\mathbf{p}$  on a regular lattice  $\Lambda$  [36] for which  $w(\mathbf{x} - \mathbf{p})$  overlaps with  $\mathbb{F}$ , that is,  $w(\mathbf{x} - \mathbf{p}) \neq 0$  for at least one coordinate  $\mathbf{x} \in \mathbb{F}$ .

Each of the windowed signals  $l(\mathbf{x}) \cdot w(\mathbf{x} - \mathbf{p})$  can be decomposed into a sum of orthonormal basis functions  $\varphi_n(\mathbf{x} - \mathbf{p}) \cdot w(\mathbf{x} - \mathbf{p})$ ,  $n \in \mathcal{N}$ , such that

$$w(\mathbf{x} - \mathbf{p}) \left[ l(\mathbf{x}) - \sum_{n \in \mathcal{N}} l_n(\mathbf{p}) \cdot \varphi_n(\mathbf{x} - \mathbf{p}) \right] = 0, \qquad (3)$$

where the coefficients are obtained via

$$l_n(\mathbf{p}) = \langle a_n(\mathbf{x} - \mathbf{p}), l(\mathbf{x}) \rangle = \int_{\mathbb{F}} a_n(\mathbf{x} - \mathbf{p}) \cdot l(\mathbf{x}) d\mathbf{x}, \quad (4)$$

for  $n \in \mathcal{N}$  and  $\mathbf{p} \in \mathcal{P}$  (equality in (3) actually means convergence in the mean-square sense) [20, 24]. The (discrete) set  $\mathcal{N}$  indexes the basis functions. The function

$$a_n(\mathbf{x}) = \varphi_n(\mathbf{x}) \cdot w^2(\mathbf{x}) \tag{5}$$

is referred to as the *analysis function* of order *n*. The mapping between the original signal  $l(\mathbf{x})$  and the coefficients  $[l_n(\mathbf{p})]$  for all orders *n* and positions **p** specifies the analysis stage *A* of Figure 1.

If measurements are performed at all positions, then (4), taken over all positions  $\mathbf{p} \in \mathbb{R}^D$ , defines a *D*-dimensional *convolution* or *filtering* between the input signal  $l(\mathbf{x})$  and the filter with impulse response  $a_n(-\mathbf{x})$ , or equivalently, a *correlation* between the input signal  $l(\mathbf{x})$  and the analysis function  $a_n(\mathbf{x})$ . Only a limited number of these output values are however required, since  $\mathcal{P} \subset \mathbb{R}^D$ . This operation of selecting a subset of values is called (down-)sampling. The combined operation is depicted in Figure 2. The complete signal analysis *A* contains several of these filter/sampling combinations, that is, one for each index  $n \in \mathcal{N}$ , and is called a *multirate filter bank*. The term multirate refers to the fact that the domain  $\mathbb{F}$  on which the original signal is defined is usually different from the domain  $\mathcal{P}$  on which the filter coefficients are required.

There is no unique way of reconstructing the signal  $l(\mathbf{x})$  from the coefficients  $[l_n(\mathbf{p}); n \in \mathcal{N}, \mathbf{p} \in \mathcal{P}]$  since any weighted summation

$$l(\mathbf{x}) = \sum_{\mathbf{p}\in\mathcal{P}} \sum_{n\in\mathcal{N}} l_n(\mathbf{p}) \cdot r_n(\mathbf{x} - \mathbf{p}), \tag{6}$$

with

$$r_n(\mathbf{x} - \mathbf{p}) = \varphi_n(\mathbf{x} - \mathbf{p}) \frac{w(\mathbf{x} - \mathbf{p})u(\mathbf{x} - \mathbf{p})}{\sum_{\mathbf{q} \in \mathcal{P}} w(\mathbf{x} - \mathbf{q})u(\mathbf{x} - \mathbf{q})}, \quad (7)$$

for which

$$\sum_{\mathbf{p}\in\mathscr{P}} w(\mathbf{x}-\mathbf{p})u(\mathbf{x}-\mathbf{p}) \neq \mathbf{0},\tag{8}$$

for all argument values  $\mathbf{x} \in \mathbb{F}$ , is for instance valid. The reason why there is no unique reconstruction is that the coefficients of the local signal decomposition are not independent if the window functions  $w(\mathbf{x} - \mathbf{p})$  at neighboring positions  $\mathbf{p}$  are overlapping. This means that, although there is a unique set of coefficients  $[l_n(\mathbf{p}); n \in \mathcal{N}, \mathbf{p} \in \mathcal{P}]$  for any signal  $l(\mathbf{x})$ , there are many sets of coefficients that do not correspond to a signal. Hence, the set of physically realizable coefficients is a subspace of the space of all coefficients.

To arrive at a unique solution for  $u(\mathbf{x})$ , we have to introduce an additional condition for the reconstruction. One obvious condition is described in the mathematical theory of frames [29]. If we reconstruct a signal from a set of coefficients that are not physically realizable, and subsequently analyze this signal, then we require the output coefficients to be the closest realizable set of coefficients. In mathematical terms, we impose that synthesis *S* followed by analysis *A* is equivalent to an orthogonal projection  $P = A \cdot S$  onto the set of realizable coefficients. This is a useful property in our application since the processed coefficients in adaptive signal processing cannot be guaranteed to be realizable; it is therefore important to design the synthesis stage such that the coefficients of the output signal are automatically mapped onto the closest realizable coefficients.

To find the unique reconstruction (synthesis) satisfying the above condition for the given set of analysis functions, we have to determine the frame operator  $F_a$  which maps any signal  $l(\mathbf{x})$  to the corresponding signal

$$F_a l(\mathbf{x}) = \sum_{\mathbf{p} \in \mathscr{P}} \sum_{n \in \mathscr{N}} l_n(\mathbf{p}) \cdot a_n(\mathbf{x} - \mathbf{p}), \qquad (9)$$

for  $\mathbf{x} \in \mathbb{F}$ , that arises by also using the analysis functions to perform the signal reconstruction [29]. In the case under consideration, the frame operator reduces to a multiplication by a fixed function, namely,

$$F_a l(\mathbf{x}) = l(\mathbf{x}) \cdot \sum_{\mathbf{p} \in \mathscr{P}} w^2(\mathbf{x} - \mathbf{p}), \qquad (10)$$

for  $\mathbf{x} \in \mathbb{F}$ . The inverse frame operator exists if

$$h^{2}(\mathbf{x}) = \sum_{\mathbf{p}\in\mathscr{P}} w^{2}(\mathbf{x} - \mathbf{p}) \neq 0, \qquad (11)$$

for all  $\mathbf{x} \in \mathbb{F}$ . Mapping the analysis functions through this inverse frame operator provides the required *synthesis functions* 

$$s_n(\mathbf{x}) = \varphi_n(\mathbf{x}) \cdot v^2(\mathbf{x}), \qquad (12)$$



FIGURE 3: Mapping from the coefficients  $[l_n(\mathbf{p}), \mathbf{p} \in \mathcal{P}]$  to the interpolated image  $\hat{l}_n(\mathbf{x})$  of order *n*, depicted as a two-stage process. The standard notations \* and  $\uparrow$  are used to denote the convolution and upsampling operations, respectively.



FIGURE 4: Local signal analysis and synthesis, where N denotes the number of different templates (N may be infinite).

for  $n \in \mathcal{N}$ , where

$$v^{2}(\mathbf{x}) = \frac{w^{2}(\mathbf{x})}{h^{2}(\mathbf{x})} = \frac{w^{2}(\mathbf{x})}{\sum_{\mathbf{p}\in\mathscr{P}}w^{2}(\mathbf{x}-\mathbf{p})}$$
(13)

is a modified window. This latter window has the property that the sum of displaced copies  $v^2(\mathbf{x} - \mathbf{p})$  for all positions  $\mathbf{p}$  on the sampling lattice  $\mathcal{P}$  is equal to one for all positions  $\mathbf{x} \in \mathbb{F}$ .

The resulting synthesis stage S is specified by

$$l(\mathbf{x}) = \sum_{\mathbf{p}\in\mathscr{P}} \sum_{n\in\mathscr{N}} l_n(\mathbf{p}) \cdot s_n(\mathbf{x} - \mathbf{p})$$
  
= 
$$\frac{\sum_{\mathbf{p}\in\mathscr{P}} \left[ \sum_{n\in\mathscr{N}} l_n(\mathbf{p}) \cdot a_n(\mathbf{x} - \mathbf{p}) \right]}{\sum_{\mathbf{p}\in\mathscr{P}} w^2(\mathbf{x} - \mathbf{p})},$$
(14)

for  $\mathbf{x} \in \mathbb{F}$ . This optimal synthesis corresponds to  $u(\mathbf{x}) = w(\mathbf{x})$  in our earlier notation.

If an impulse signal with unit weight at position  $\mathbf{p}$  is input into a filter with impulse response  $s_n(\mathbf{x})$ , then the output is, by definition, equal to  $s_n(\mathbf{x} - \mathbf{p})$ . This implies that the signal

$$l_n(\mathbf{x}) = \sum_{\mathbf{p} \in \mathscr{P}} l_n(\mathbf{p}) \cdot s_n(\mathbf{x} - \mathbf{p})$$
(15)

is the output of this filter in case the input is an array of impulses, where the weight of the impulse at position  $\mathbf{p} \in \mathcal{P}$ is equal to the coefficient  $l_n(\mathbf{p})$ . This operation of creating an array of impulses from a set of coefficients is called *upsampling*. The combined operation of upsampling and filtering is called *interpolation* and is depicted in Figure 3. The complete signal synthesis *S* contains several of these interpolations, that is, one for each index  $n \in \mathcal{N}$ , and is hence a multirate filter bank. The synthesized signal arises by combining all outputs of this multirate filter bank.

The overall image analysis and synthesis is depicted in Figure 4. In case of Figure 4, the coefficients  $[l_n(\mathbf{p})]$  that are derived from the original image are not modified, so that the

output signal satisfies  $\hat{l}(\mathbf{x}) = l(\mathbf{x})$ , for  $\mathbf{x} \in \mathbb{F}$ . If  $\mathbb{F}$  is discrete, then the local signal decomposition can be used to interpolate the input signal, that is, to construct signal values  $\hat{l}(\mathbf{x})$ , for  $\mathbf{x} \notin \mathbb{F}$  [13]. In case of image processing or coding, the original coefficients are mapped to modified coefficients [ $\hat{l}_n(\mathbf{p})$ ] before entering the synthesis stage [1], so that  $\hat{l}(\mathbf{x}) \neq l(\mathbf{x})$ , for  $\mathbf{x} \in \mathbb{F}$ .

Note that a local signal decomposition satisfies not only the global correspondence of (14), that is, synthesis after analysis results in the original signal, but also the local correspondences of (3). The latter correspondences imply that the local description  $[l]_{\mathbf{p}} = [l_n(\mathbf{p}), n \in \mathcal{N}]$  completely specifies the signal  $l(\mathbf{x})$  within the local window  $w(\mathbf{x} - \mathbf{p})$  at position  $\mathbf{p}$ . The windowed signal

$$l_w(\mathbf{x} - \mathbf{p}) = w^2(\mathbf{x} - \mathbf{p}) \cdot l(\mathbf{x}) = \sum_{n \in \mathcal{N}} l_n(\mathbf{p}) \cdot a_n(\mathbf{x} - \mathbf{p}) \quad (16)$$

also expresses the contribution of the coefficients  $[l]_{\mathbf{p}} = [l_n(\mathbf{p}), n \in \mathcal{N}]$  at position  $\mathbf{p} \in \mathcal{P}$  to the overall synthesized signal, in a way that is specified in (14).

#### 3. PARTIAL SIGNAL DECOMPOSITIONS

One obvious disadvantage of local signal decompositions is that a large number of filter operations (i.e., equal to the cardinality of the index set  $\mathcal{N}$ ) may be required in the most general case in order to create the processed signal

$$\hat{l}(\mathbf{x}) = \sum_{\mathbf{p}\in\mathcal{P}} \sum_{n\in\mathcal{N}} \hat{l}_n(\mathbf{p}) \cdot s_n(\mathbf{x} - \mathbf{p}) = \frac{\sum_{\mathbf{p}\in\mathcal{P}} \left[ \sum_{n\in\mathcal{N}} \hat{l}_n(\mathbf{p}) \cdot a_n(\mathbf{x} - \mathbf{p}) \right]}{\sum_{\mathbf{p}\in\mathcal{P}} w^2(\mathbf{x} - \mathbf{p})}.$$
(17)

The specific dependency of the output coefficients  $[\hat{l}]_{\mathbf{p}} = [\hat{l}_n(\mathbf{p}), n \in \mathcal{N}]$  at position  $\mathbf{p}$  on the corresponding coefficients  $[l]_{\mathbf{p}} = [l_n(\mathbf{p}), n \in \mathcal{N}]$  of the input signal may however

allow for substantial simplifications. We discuss an important special case.

We assume that the index set can be subdivided in two subsets, that is,  $\mathcal{N} = \mathcal{N}_o \cup \mathcal{N}_r$ , in such a way that the output coefficients  $[\hat{l}_n(\mathbf{p}), n \in \mathcal{N}_o]$  at position  $\mathbf{p}$  only depend on the corresponding input coefficients  $[l_n(\mathbf{p}), n \in \mathcal{N}_o]$ , and the remaining coefficients at position  $\mathbf{p}$  are all transformed in an identical way, that is,

$$\widehat{l}_n(\mathbf{p}) = \kappa(\mathbf{p}) \cdot l_n(\mathbf{p}), \qquad (18)$$

for  $n \in \mathcal{N}_r$ . We moreover assume that the adaptive weighting factor  $\kappa(\mathbf{p})$  on the coefficients in the index set  $\mathcal{N}_r$  can be derived from  $[l_n(\mathbf{p}), n \in \mathcal{N}_o]$  and the local energy

$$E(\mathbf{p}) = \int_{\mathbb{F}} w^2(\mathbf{x} - \mathbf{p}) \cdot l^2(\mathbf{x}) d\mathbf{x}.$$
 (19)

Because the local basis { $\varphi_n(\mathbf{x} - \mathbf{p})$ ,  $n \in \mathcal{N}$ } was assumed to be orthonormal, this local energy can be split into

$$E_o(\mathbf{p}) = \sum_{n \in \mathcal{N}_o} |l_n(\mathbf{p})|^2, \qquad (20)$$

the local energy in the signal component

$$\sum_{n \in \mathcal{N}_o} l_n(\mathbf{p}) \cdot \varphi_n(\mathbf{x} - \mathbf{p}) w(\mathbf{x} - \mathbf{p})$$
(21)

described by the first index set  $N_o$ , and the *residue energy*  $E_r(\mathbf{p}) = E(\mathbf{p}) - E_o(\mathbf{p})$ , that is, the local energy in the *residue* signal

$$\sum_{n \in \mathcal{N}_r} l_n(\mathbf{p}) \cdot \varphi_n(\mathbf{x} - \mathbf{p}) w(\mathbf{x} - \mathbf{p})$$

$$= \left[ l(x) - \sum_{n \in \mathcal{N}_o} l_n(\mathbf{p}) \cdot \varphi_n(\mathbf{x} - \mathbf{p}) \right] w(\mathbf{x} - \mathbf{p})$$
(22)

that is described by the second index set  $\mathcal{N}_r$ .

Using the above assumptions, it is easily verified that the processed signal can be expressed as

$$\hat{l}(\mathbf{x}) = \sum_{\mathbf{p} \in \mathcal{P}} \sum_{n \in \mathcal{N}_o} \left[ \hat{l}_n(\mathbf{p}) - \kappa(\mathbf{p}) \cdot l_n(\mathbf{p}) \right] \cdot s_n(\mathbf{x} - \mathbf{p}) + l(\mathbf{x}) \cdot \kappa(\mathbf{x}).$$
(23)

This so-called *partial signal decomposition* leads to the modified algorithm depicted in Figure 5. Only a partial transformation involving the indices  $n \in N_o$  is hence required in this case to determine the processed signal. The missing information for the other indices  $n \in N_r$  can be recovered from the original signal. The weighting function

$$\kappa(\mathbf{x}) = \sum_{\mathbf{p}\in\mathscr{P}} \kappa(\mathbf{p}) \cdot v^2(\mathbf{x} - \mathbf{p})$$
(24)

that must be applied to the original signal in order to accomplish this correction is obtained by interpolating the adaptive factors  $[\kappa(\mathbf{p}), \mathbf{p} \in \mathcal{P}]$  with a filter with impulse response equal to the modified window function  $v^2(\mathbf{x})$ . In case of image coding, where the original signal is not available at the receiver side, we obviously need to choose  $\kappa(\mathbf{p}) = 0$ ,  $\forall \mathbf{p} \in \mathcal{P}$ . Nonzero values of  $\kappa(\mathbf{p})$  are however useful in case of image processing.

#### 4. RESIDUE-IMAGE PROCESSING

#### 4.1. Algorithm

The simplest case of a partial signal decomposition arises when the index set  $N_o$  contains only a single component, that is, the local average signal value

$$l_o(\mathbf{p}) = \int_{\mathbb{F}} w^2(\mathbf{x} - \mathbf{p}) \cdot l(\mathbf{x}) d\mathbf{x},$$
 (25)

and the residue signal  $l(\mathbf{x}) - l_o(\mathbf{p})$  is the local deviation from this average value. If we assume that the window function is normalized, that is, if  $\int_{\mathbb{F}} w^2(\mathbf{x}) d\mathbf{x} = 1$ , then this corresponds to a single orthonormal basis vector  $\varphi_o(\mathbf{x}) = 1$  in the index set  $\mathcal{N}_o$ . If we impose that the local average signal value  $l_o(\mathbf{p})$  is not altered by the processing, then the partial signal decomposition can be expressed as

$$\hat{l}(\mathbf{x}) = \sum_{\mathbf{p}\in\mathcal{P}} v^2(\mathbf{x}-\mathbf{p}) \cdot \{l_o(\mathbf{p}) + \kappa(\mathbf{p}) \cdot [l(\mathbf{x}) - l_o(\mathbf{p})]\}$$
  
$$= \sum_{\mathbf{p}\in\mathcal{P}} v^2(\mathbf{x}-\mathbf{p}) \cdot [1 - \kappa(\mathbf{p})] l_o(\mathbf{p}) + l(\mathbf{x}) \cdot \kappa(\mathbf{x}),$$
(26)

where  $\kappa(\mathbf{p})$  denotes the residue amplification factor at position  $\mathbf{p} \in \mathcal{P}$ , and  $\kappa(\mathbf{x})$  is the corresponding interpolated function.

In uniform regions of the image, the residue energy

$$E_{r}(\mathbf{p}) = \int_{\mathbb{F}} w^{2}(\mathbf{x} - \mathbf{p}) \cdot \left[l(\mathbf{x}) - l_{o}(\mathbf{p})\right]^{2} d\mathbf{x}$$
  
$$= \int_{\mathbb{F}} w^{2}(\mathbf{x} - \mathbf{p}) \cdot l^{2}(\mathbf{x}) d\mathbf{x} - l_{o}^{2}(\mathbf{p})$$
(27)

will be small, while in transition regions (i.e., near edges)  $E_r(\mathbf{p})$  will be large. This residue energy can hence be used as a control variable in an adaptive processing algorithm that requires different processing for uniform and transition regions. This residue energy can be derived quite efficiently, since it involves only two filter operations, one on the original signal  $l(\mathbf{x})$ , and one on the squared signal  $l^2(\mathbf{x})$ .

Both noise reduction and contrast enhancement can be achieved using residue-image processing [15]. The relationship between the residue amplification factor  $\kappa(\mathbf{p})$  and the residue energy  $E_r(\mathbf{p})$ , or *residue amplitude*  $A_r(\mathbf{p}) = \sqrt{E_r(\mathbf{p})}$ , makes the adopted strategy explicit. Many different approaches can be imagined, but we discuss only one possibility.

If the residue amplitude is large, say  $A_r(\mathbf{p}) > A_v$ , then we can assume that the contrast is sufficiently high to guarantee clear visibility, and that no enhancement is required, that is,  $\kappa(\mathbf{p}) = 1$  for  $A_r(\mathbf{p}) > A_v$ . If, on the contrary, the residue amplitude is small, say  $A_r(\mathbf{p}) < A_t$ , then the residue signal may be assumed to be noise on a uniform background, and this noise can be removed by setting  $\kappa(\mathbf{p}) = 0$  for  $A_r(\mathbf{p}) < A_t$ . For  $A_t \le A_r(\mathbf{p}) \le A_v$ , the residue signal is judged to contain significant information, but needs to be amplified to become clearly visible. An obvious condition for the processing is that the output amplitude  $\kappa(\mathbf{p}) \cdot A_r(\mathbf{p})$  should be an increasing



FIGURE 5: Algorithmic structure for partial image decompositions. In the analysis step  $A_o$ , the input image  $l(\mathbf{x})$  is transformed into a limited set of coefficients. These coefficients  $[l_n(\mathbf{p}), n \in \mathcal{N}_o]$  at a given position  $\mathbf{p}$  undergo the transformation  $\tilde{T}_o(C_{\mathbf{p}}) = T_o(C_{\mathbf{p}}) - \kappa(\mathbf{p}) \cdot I_o$ , where  $T_o(C_{\mathbf{p}})$  is the intended transformation and  $I_o$  denotes the identity transformation. The control variables  $C_{\mathbf{p}}$  at position  $\mathbf{p}$ , which are used to select the desired transformation  $T_o(C_{\mathbf{p}})$  and the weighting factor  $\kappa(\mathbf{p})$  on the missing coefficients, can be derived from  $[l_n(\mathbf{p}), n \in \mathcal{N}_o]$  and the local energy  $E(\mathbf{p})$ . The transformed coefficients  $[\tilde{l}_n(\mathbf{p}) = \hat{l}_n(\mathbf{p}) - \kappa(\mathbf{p}) \cdot l_n(\mathbf{p}), n \in \mathcal{N}_o]$ , where  $[\hat{l}_n(\mathbf{p}), n \in \mathcal{N}_o]$  are the intended coefficients for the processed image  $\hat{l}(\mathbf{x})$ , are combined in the synthesis step  $S_o$  to obtain part of the output image. The missing part of the processed image is extracted from the original image  $l(\mathbf{x})$  by a weighting function  $\kappa(\mathbf{x})$  that interpolates the factors  $[\kappa(\mathbf{p}), \mathbf{p} \in \mathcal{P}]$  to all positions  $\mathbf{x} \in \mathbb{F}$ .

function of the input amplitude  $A_r(\mathbf{p})$ . Many functions satisfy this condition, and a simple example is

$$\kappa(A_r(\mathbf{p})) \cdot A_r(\mathbf{p}) = s \cdot A_r(\mathbf{p}) + (1-s) \cdot A_v \qquad (28)$$

for  $A_t \le A_r(\mathbf{p}) \le A_\nu$ , where  $0 \le s \le 1$  denotes the slope of the input-output amplitude characteristic. In this case, the maximum value of the amplification factor is

$$\kappa(A_t) = s + (1 - s) \cdot \frac{A_v}{A_t}$$
(29)

for  $A_r(\mathbf{p}) = A_t$ , and decreases to 1 for  $A_r(\mathbf{p}) = A_v$ . Alternative expressions for  $\kappa(A_r(\mathbf{p}))$  that avoid the discontinuity at  $A_r(\mathbf{p}) = A_t$  may also be considered.

In Figure 6, we illustrate residue-image processing on an 8-bit subtractive angiography image (values between 0 and 255). The threshold value  $A_t = 3$  was derived from the residue amplitude histogram (i.e.,  $A_t = 2 \cdot A_m$ , where  $A_m$  is the mode of the residue amplitude histogram, see next section), and the remaining parameters were  $A_v = 15$  and s = 0.1. A separable binomial window function [2] of size 7 (i.e., with coefficients  $w^2 = [1 \ 6 \ 15 \ 20 \ 15 \ 6 \ 1]/64$  in one dimension) was used as the window function.

#### 4.2. Energy histograms and noise estimation

Since images usually contain noise, the residue energy  $E_r(\mathbf{p})$  will not most often be exactly zero, not even for regions that we consider to be uniform. We hence have to decide on a threshold value  $E_t = A_t^2$  for the residue energy measure  $E_r(\mathbf{p})$ , below which image positions are classified as belonging to uniform regions. As part of this process, we need to estimate noise characteristics. This noise estimation may also be interesting in its own right, for example, to objectively characterize the quality of the image acquisition [17, 19, 24].

It has been demonstrated [15, 24] that the first-order statistic, that is, the histogram or probability density function (PDF), of  $E_r(\mathbf{p})$ , at least in uniform regions of the image, can



FIGURE 6: Residue-image processing applied to a subtractive angiography image. The top row shows the images, while the bottom row shows the corresponding residue amplitudes  $(A_r)$ . The left and right columns correspond to the original and the processed image, respectively. The displayed residue amplitude images show  $\log(1 + A_r)$  instead of  $A_r$  in order to allow better discrimination.

be approximated by a  $\chi^2$ -distribution

$$P(E_r) = \frac{1}{E_o(q-1)!} \left(\frac{E_r}{E_o}\right)^{q-1} \exp\left(-\frac{E_r}{E_o}\right),$$
(30)

with 2*q* degrees of freedom. The corresponding PDF for the residue amplitude is

$$P(A_r) = \frac{2}{A_o(q-1)!} \left(\frac{A_r}{A_o}\right)^{2q-1} \exp\left[-\left(\frac{A_r}{A_o}\right)^2\right], \quad (31)$$

where  $E_o = A_o^2$ . The average value and the variance in the energy PDF are equal to  $\mu_e = q \cdot E_o$  and  $\sigma_e^2 = q \cdot E_o^2$ , respectively. The maximum values for the energy and amplitude PDFs occur at energy  $E_m = E_o(q - 1)$  and amplitude



FIGURE 7: The drawn curve shows the histogram of the residue amplitude for the image and processing used in Figure 6. The dotted curve corresponds to optimal fitting of the  $\chi^2$ -distribution (q = 3.83 and  $A_o = 0.71$ , resulting in a peak value at  $A_m = 1.49$ ).

 $A_m = A_o\sqrt{q-0.5}$ , respectively. Energy or amplitude histograms for an actual image can be compared against the above theoretical PDFs in order to find the PDF parameters  $E_o$  (or  $A_o$ ) and q. Actual images of course also contain nonuniform regions. The corresponding responses for these image features will influence the shape of the energy histogram, especially for large energy (or amplitude) values. This implies that only the part of the histogram at low energy (or amplitude) values should be used for noise estimation and curve fitting against the theoretical PDF [24], as shown in Figure 7.

A parametrized PDF such as the  $\chi^2$ -distribution can be used to perform *K*-means clustering on the residue amplitude or energy. For example, if the observed residue amplitude histogram  $h(A_r)$  can be approximated by a sum of two components, that is,

$$h(A_r) = P_o \cdot P(A_r) + [h(A_r) - P_o \cdot P(A_r)], \quad (32)$$

where  $P(A_r)$  satisfies (31) and is associated with uniform regions, then maximum a posteriori classification will classify all positions **p** for which the residue amplitude satisfies  $A_r(\mathbf{p}) < A_t$ , with

$$P_o \cdot P(A_t) = h(A_t) - P_o \cdot P(A_t), \qquad (33)$$

as belonging to uniform regions. This provides us with an automated means of selecting the threshold  $A_t$ .

#### 5. THE HERMITE TRANSFORMATION

#### 5.1. Definition

We will now apply the theory of local signal decompositions to the specific case of a two-dimensional Gaussian window

$$w(x, y; \sigma) = \frac{1}{\sigma\sqrt{\pi}} \exp\left[-\frac{x^2 + y^2}{2\sigma^2}\right]$$
(34)

with standard deviation equal to  $\sigma$ . Note that this window function is both separable, that is, can be written as the product of two Gaussian functions in *x*- and *y*-directions, and *circularly symmetric*, that is, its value only depends on the distance  $r = \sqrt{x^2 + y^2}$  from the origin. Especially the circular symmetry will turn out to be essential for the discussion that follows.

We select the basis functions of the signal decomposition (which were denoted by  $\varphi_n$  in Section 2) to be orthonormal polynomials for the Gaussian window [2]. Because this window is separable, these orthonormal polynomials are also separable, that is,

$$p_{n-m,m}(x,y;\sigma) = g_{n-m}\left(\frac{x}{\sigma}\right) \cdot g_m\left(\frac{y}{\sigma}\right), \qquad (35)$$

with

$$g_n(x) = \frac{1}{\sqrt{2^n n!}} H_n(x), \qquad (36)$$

where the standard notation is used to denote the Hermite polynomial  $H_n(x)$  of order n, for n = 0, 1, ... For the discussion in this paper, we will especially need the polynomials up to degree 2, that is,

$$g_0(x) = 1,$$
  $g_1(x) = \sqrt{2}x,$   $g_2(x) = \frac{2x^2 - 1}{\sqrt{2}}.$  (37)

The resulting image decomposition at the generic<sup>1</sup> lattice position (0,0) satisfies

$$w(x, y; \sigma) \left[ l(x, y) - \sum_{n=0}^{\infty} \sum_{m=0}^{n} l_{n-m,m} \cdot p_{n-m,m}(x, y) \right] = 0,$$
(38)

where equality means convergence in the mean-square sense.

The Hermite coefficients  $[l_{n-m,m}; n = 0, ..., \infty; m = 0, ..., n]$  are derived by filtering the spatial image representation l(x, y) with *derivative-of-Gaussian* filters with impulse responses

$$d_{n-m,m}(x, y; \sigma) = g_{n-m} \left( -\frac{x}{\sigma} \right) g_m \left( -\frac{y}{\sigma} \right) \cdot w^2(x, y; \sigma)$$
  
=  $\frac{1}{\sqrt{2^n (n-m)!m!}} \frac{\partial^n}{\partial (x/\sigma)^{n-m} \partial (y/\sigma)^m} w^2(x, y; \sigma),$  (39)

<sup>&</sup>lt;sup>1</sup> In order to simplify the notation we discuss only the contribution at lattice position (0, 0); the generalization to contributions  $w(x-p, y-q; \sigma) \cdot l(x, y)$  at arbitrary lattice positions  $(p, q) \in \mathcal{P}$  is straightforward.

or filter kernels  $d_{n-m,m}(-x, -y; \sigma)$ , and selecting the filtered outputs at the lattice positions. For the generic lattice position (0, 0) this results in

$$l_{n-m,m} = \int_{-\infty}^{+\infty} dx \int_{-\infty}^{+\infty} dy \, l(x,y) \cdot d_{n-m,m}(-x,-y;\sigma).$$
(40)

The fact that the filter functions in (39) can be expressed as derivatives of Gaussians is a consequence of a well-known property of Hermite polynomials, that is,

$$e^{-x^2}H_n(x) = (-1)^n \frac{d^n}{dx^n} e^{-x^2}.$$
 (41)

This property implies that the Hermite coefficients themselves can also be interpreted as partial derivatives

$$l_{n-m,m} = \frac{1}{\sqrt{2^n (n-m)!m!}} \left[ \frac{\partial^n}{\partial (x/\sigma)^{n-m} \partial (y/\sigma)^m} l(x,y;\sigma) \right]_{x=y=0}$$
(42)

to the smoothed image

$$l(x, y; \sigma) = l(x, y) \star w^2(x, y; \sigma)$$
(43)

that is obtained by applying a Gaussian filter  $w^2(x, y; \sigma)$  with standard deviation equal to  $\sigma/\sqrt{2}$  to the original image l(x, y).

Note that the contribution at the generic lattice position (0,0) to the overall synthesized image is

$$l_{w}(x, y) = w^{2}(x, y; \sigma) \cdot l(x, y)$$
  
=  $\sum_{n=0}^{\infty} \sum_{m=0}^{n} l_{n-m,m} \cdot d_{n-m,m}(-x, -y; \sigma),$  (44)

while the local energy at this position is equal to

$$E = \iint w^2(x, y) \cdot l^2(x, y) dx \, dy = \sum_{n=0}^{\infty} \sum_{m=0}^{n} l_{n-m,m}^2.$$
(45)

The residue energy  $E_r = E - l_{00}^2$  or residue amplitude  $A_r = \sqrt{E_r}$  can again be used to differentiate between uniform and nonuniform regions in the image.

In Figure 8 we show an artificial image containing several interesting features such as edges, lines, corners, Tjunctions, X-junctions, circles, and noise. We will use this image throughout this paper to illustrate different alternative forms of the Hermite transformation because it is much easier to judge the properties of the Hermite transformation on this image than on a natural image.

In Figure 9 we show the residue amplitude and the Hermite coefficients up to order 2 for a Gaussian window with spread  $\sigma = 2.5$  times the sampling distance. The coefficients  $l_{n-m,m}$  for all positions are grouped into subimages. The (fuzzy) derivatives of order n - m along the *x*-direction and order *m* along the *y*-direction are obtained by filtering the original image with the derivative-of-Gaussian filter  $d_{n-m,m}(x, y)$  and (optionally) subsampling the filtered image. No subsampling has been applied in the case of Figure 9. One important advantage of the derivative-of-Gaussian filters is that they are separable. This means that they can be implemented very efficiently by first filtering along the *x*-direction and subsequently along the *y*-direction (or vice versa).



FIGURE 8: Original image used in the Hermite analyses.

#### 5.2. Scale space

Derivative-of-Gaussian filters play a key role in the theory of scale space [32, 37, 38]. Scale space represents a systematic way of studying structure in images. An image is regarded as a surface that can be analyzed at varying degrees of detail. To reduce the level of detail, the image is first passed through a Gaussian filter. The smoothed surface is then described using traditional methods from differential geometry (see also Section 7), a well-studied discipline of mathematics [39]. Many interesting surface properties can be expressed as combinations of the local partial derivatives  $l_{n-m,m}$  of (42) to the smoothed image is the solution to a diffusion equation

$$\frac{\partial}{\partial t}u(x, y, t) = \left[\frac{\partial^2}{\partial x^2} + \frac{\partial^2}{\partial y^2}\right]u(x, y, t), \tag{46}$$

with initial condition

$$u(x, y, 0) = l(x, y)$$
 (47)

that is stopped after time  $t = T = (\sigma/2)^2$ . Fast recursive methods for numerically solving this diffusion equation [40] may be used to determine  $u(x, y, T) = l(x, y; 2\sqrt{T})$  for large values of *T*. In the Hermite transformation the operations of smoothing and differentiating are combined, so that the Hermite transformation corresponds to an analysis at a specified level in scale space.

Very often, the most efficient way for calculating the Hermite coefficients is by combining diffusion and filtering, as shown in Figure 10. Indeed, if the input image l(x, y) is first diffused up to time  $T = (\sigma_D/2)^2$ , and subsequently analyzed by a Hermite transformation with parameter  $\sigma_H$ , then the resulting coefficients at the origin, that is,

$$u_{n-m,m} = \left[d_{n-m,m}(x, y; \sigma_H) \star u(x, y; T)\right]_{x=y=0} = \left(\frac{\sigma_H}{\sigma}\right)^n l_{n-m,m},$$
(48)



FIGURE 9: Hermite coefficients  $l_{n-m,m}$ , for m = 0, ..., n, up to order n = 2 for a Gaussian window with spread  $\sigma = 2.5$  times the sampling distance. The image in the lower-right corner is the residue amplitude  $A_r$ . To relate the mathematical results in this paper to the figures, it should be realized that the *x*-axis points to the right and the *y*-axis downwards. Moreover, the coefficients of order *n*, which are located in the *n*th diagonal, are amplified by the same factor  $\beta_n$  for display purposes. The factors  $\beta_n$  are chosen independently for each order *n*, in such a way that the complete display range is used.



FIGURE 10: The Hermite coefficients for a large standard deviation equal to  $\sigma$  can be obtained by a three-step process. First, a diffusion is applied to the original image u(x, y; 0) = l(x, y) for a time T, resulting in the diffused image  $u(x, y; T) = l(x, y; \sigma_D)$ , where  $\sigma_D = 2\sqrt{T}$ . Second, a Hermite transformation with a small standard deviation equal to  $\sigma_H$ , where  $\sigma_H^2 = \sigma^2 - \sigma_D^2$ , is applied to the diffused image, resulting in the coefficients  $u_{n-m,m}$ , for m = 0, ..., nand n = 0, 1, 2... Third, these coefficients of order n are rescaled by a factor  $(\sigma/\sigma_H)^n$  to obtain the required Hermite coefficients  $l_{n-m,m}$ .

belong to a Hermite transformation with parameter

$$\sigma = \sqrt{\sigma_D^2 + \sigma_H^2} = \sqrt{4T + \sigma_H^2}.$$
 (49)

An upper limit on  $\sigma_H$  allows to limit the size of the derivatives-of-Gaussian filters, since this filter size is proportional to  $\sigma_H$ . The number of computations required in the filtering stage increases (linearly) with this filter size. A lower limit on  $\sigma_H$  (typically,  $\sigma_H$  larger than 1.5 times the sampling distance [13]) is required to guarantee that the derivative operations can be closely approximated for digital images. Most of the time we use a well-designed set of

derivatives-of-Gaussian filters (for  $\sigma_H$  equal to 1.8 times the sampling distance), and vary the diffusion time *T* to accomplish the required value of  $\sigma$ .

#### 5.3. Angular functions

An important step towards obtaining a better insight into the Hermite transformation is to express the filtering that is performed on the input image in the Fourier domain. The Fourier transformations of the filter functions, expressed in polar coordinates, are

$$\widehat{D}_{n-m,m}(\omega,\phi;\sigma) = \widehat{D}_n(\omega;\sigma) \cdot \alpha_{n-m,m}(\phi), \qquad (50)$$

where

$$\widehat{D}_n(\omega;\sigma) = \frac{1}{\sqrt{2^n n!}} (-j\omega\sigma)^n \exp\left[-\frac{(\omega\sigma)^2}{4}\right]$$
(51)

is the 1D Fourier transformation of the *n*th-order Gaussian derivative

$$D_n(r;\sigma) = \frac{1}{\sqrt{2^n n!}} \frac{d^n}{d(r/\sigma)^n} \left[ \frac{1}{\sigma\sqrt{\pi}} \exp\left(-\frac{r^2}{\sigma^2}\right) \right], \quad (52)$$

and

$$\alpha_{n-m,m}(\phi) = \sqrt{\frac{n!}{(n-m)!m!}} \cos^{n-m} \phi \cdot \sin^m \phi, \qquad (53)$$

for m = 0, ..., n, are the *Cartesian angular functions* of order n [2, 20]. The shape of these Cartesian angular functions can be judged very well in Figure 9 from the angular modulation of the coefficients along the edge of the circle image. The Cartesian angular functions of order n = 1 are simply  $\alpha_{10}(\phi) = \cos \phi$  and  $\alpha_{01}(\phi) = \sin \phi$ , while the Cartesian angular functions of order n = 2, that is,  $\alpha_{20}(\phi) = \cos^2 \phi$ ,  $\alpha_{11}(\phi) = \sqrt{2} \cos \phi \sin \phi$ , and  $\alpha_{02}(\phi) = \sin^2 \phi$ , are plotted in Figure 11.

The Hermite coefficients can be derived from the Fourier image representation  $\hat{l}(\omega_x, \omega_y)$  through

$$l_{n-m,m} = \int_{-\infty}^{+\infty} \frac{d\omega_x}{2\pi} \int_{-\infty}^{+\infty} \frac{d\omega_y}{2\pi} \hat{l}(\omega_x, \omega_y) \cdot \hat{d}_{n-m,m}(\omega_x, \omega_y; \sigma)$$
(54)

or, alternatively, by changing to polar coordinates, through

$$l_{n-m,m} = \frac{1}{2\pi} \int_0^{2\pi} d\phi \hat{l}_n(\phi) \cdot \alpha_{n-m,m}(\phi)$$
(55)

for  $m = 0, \ldots, n$ , where

$$\hat{l}_n(\phi) = \frac{1}{2\pi} \int_0^\infty \omega \, d\omega \, \hat{l}(\omega, \phi) \cdot \hat{D}_n(\omega; \sigma) \tag{56}$$

is called the "*image*" angular function of order n [20]. This function is real if the original image l(x, y) is real. It moreover satisfies  $\hat{l}_n(\phi + \pi) = (-1)^n \cdot \hat{l}_n(\phi)$ , so that it is odd or even symmetric, depending on whether the order n is odd or even, respectively. Obviously, the Hermite coefficients  $[l_{n-m,m}, m = 0, ..., n]$  of the same order n can all be derived



FIGURE 11: Cartesian angular functions of order n = 2.

from the angular function  $\hat{l}_n(\phi)$ . One advantage of defining the "image" angular function as an intermediate step in the analysis is that it becomes straightforward to extend many of the results below, that is, those that only depend on the properties of the angular components of the filters, to filter banks in which the radial components of the filters, that is,  $\hat{D}_n(\omega; \sigma)$ , are not derivatives-of-Gaussians.

The Fourier transformation of the local image  $l_w(x, y) = w^2(x, y; \sigma) \cdot l(x, y)$  can be expressed as

$$\hat{l}_{w}(\omega,\phi) = \hat{l}(\omega,\phi) \star \hat{D}_{0}(\omega) = \sum_{n=0}^{\infty} \hat{D}_{n}^{\star}(\omega) \cdot l_{n}(\phi), \quad (57)$$

where

$$l_n(\phi) = \sum_{m=0}^n l_{n-m,m} \cdot \alpha_{n-m,m}(\phi)$$
(58)

is called the "*derivative*" angular function of order n. Since the interest here will be mainly in the derivative angular functions of order n = 1 and n = 2, the following explicit expressions

$$l_{1}(\phi) = l_{10} \cdot \cos \phi + l_{01} \cdot \sin \phi,$$
  

$$l_{2}(\phi) = l_{20} \cdot \cos^{2} \phi + l_{11} \cdot \sqrt{2} \cos \phi \sin \phi + l_{02} \cdot \sin^{2} \phi$$
(59)

are provided.

#### 5.4. One-dimensional images

The Hermite coefficients at the generic lattice position (0, 0) describe the local image  $l_w(x, y) = w^2(x, y; \sigma) \cdot l(x, y)$ . An important specific case of local patterns in 2D are 1D patterns, that is, patterns that vary only in one direction (and

are constant along the orthogonal direction). In natural images, many of the image details that are of prime importance, such as edges and lines, can be locally described as one-dimensional patterns, at least within the analysis window  $w(x, y; \sigma)$ . Since analysis, coding, and processing of such patterns are considered important, it is worthwhile to investigate them separately.

The angular function for a locally one-dimensional (1D) image with normal along direction  $(\cos \phi_o, \sin \phi_o)$ , that is,  $l(x, y) = l(x \cos \phi_o + y \sin \phi_o)$ , has only contributions in the directions  $\phi_o$  and  $\phi_o + \pi$  and is therefore of the form

$$\widehat{l}_n(\phi) = \pi l_n [\delta(\phi - \phi_o) + (-1)^n \delta(\phi - (\phi_o + \pi))], \quad (60)$$

where  $l_n$  is the *n*th-order Hermite coefficient of the 1D signal l(r).

For a 1D image we can express the Hermite coefficients as

$$l_{n-m,m} = l_n \cdot \alpha_{n-m,m}(\phi_o), \tag{61}$$

so that the "derivative" angular function becomes

$$l_n(\phi) = l_n \cdot \sum_{m=0}^n \alpha_{n-m,m}(\phi_o) \cdot \alpha_{n-m,m}(\phi) = l_n \cdot \cos^n (\phi - \phi_o).$$
(62)

#### 6. LOCAL COORDINATE AXES

#### 6.1. Effect on Hermite coefficients

In scale-space theory [32], geometric features of the smoothed surface  $l(x, y; \sigma)$  as a function of the position (x, y) (and scale  $\sigma$ ) are used for image analysis [41, 42]. The wellestablished mathematical discipline of differential geometry [43] is especially concerned with how such geometric features can be derived, and some of the major results of applying differential geometry for image feature extraction will be discussed in Section 7.

Adopting local coordinate axes

$$\begin{bmatrix} \widetilde{x} \\ \widetilde{y} \end{bmatrix} = \begin{bmatrix} \cos \phi_o & \sin \phi_o \\ -\sin \phi_o & \cos \phi_o \end{bmatrix} \cdot \begin{bmatrix} x \\ y \end{bmatrix}$$
(63)

is one of the basic operations in differential geometry because many expressions for important surface properties can be simplified greatly in an adaptive coordinate system. For example, for the one-dimensional patterns introduced above, we will see shortly that aligning the  $\tilde{x}$ -axis with the direction of the pattern has distinct advantages.

We analyze the behavior of the angular functions  $l_n(\phi)$ and  $\hat{l}_n(\phi)$  under rotation of the coordinate axes. If we rotate the coordinate axes clockwise (or the image anticlockwise) through an angle  $\phi_o$  (in the left-handed coordinate system that is used in the figures), that is, if we change  $\hat{l}_n(\phi)$  to  $\hat{l}_n(\phi + \phi_o)$ , then the Hermite coefficients in the local coordinate system ( $\tilde{x}, \tilde{y}$ ) are equal to

$$l_{n-m,m}(\phi_o) = \frac{1}{2\pi} \int_0^{2\pi} d\phi \hat{l}_n(\phi) \cdot \alpha_{n-m,m}(\phi - \phi_o), \qquad (64)$$

for m = 0, ..., n, and are related to the coefficients  $[l_{n-m,m}, m = 0, ..., n]$  in the global coordinate system (x, y) by a linear transformation with parameter  $\phi_o$ . The linear nature of the transformation is a consequence of the fact that both  $\cos(\phi - \phi_o)$  and  $\sin(\phi - \phi_o)$  can be expressed as linear combinations of  $\cos \phi$  and  $\sin \phi$ . Specifically, since

$$\alpha_{n,0}(\phi - \phi_o) = \cos^n (\phi - \phi_o) = \sum_{m=0}^n \alpha_{n-m,m}(\phi_o) \cdot \alpha_{n-m,m}(\phi),$$
(65)

we obtain that the derivative of order *n* along direction  $\phi_o$  is given by  $l_{n,0}(\phi_o) = l_n(\phi_o)$ , which explains the name "derivative" angular function for  $l_n(\phi)$ . Based on the *n* + 1 partial derivatives  $[l_{n-m,m}; m = 0, ..., n]$ , we can hence derive the directional derivative  $l_n(\phi_o)$  of order *n* for any direction  $\phi_o$  [20, 44]. The derivative angular function of order *n* can of course also be expressed in terms of the Hermite coefficients in the rotated coordinate frame, that is,

$$l_{n}(\phi) = \sum_{m=0}^{n} l_{n-m,m}(\phi_{o}) \cdot \alpha_{n-m,m}(\phi - \phi_{o}).$$
(66)

Note that the linear transformation of the Hermite coefficients under rotation of the coordinate system requires that the image be locally approximated by a polynomial within a circularly symmetric window  $w(x, y; \sigma)$ . Rotating the coordinate axes can have a complicated effect on the Hermite coefficients, especially if the order *n* is high. A systematic analysis for arbitrary orders has been performed [20], but is beyond the scope of this paper. In the remainder, we will only need the results for orders n = 1, that is,

$$\begin{bmatrix} l_{10}(\phi_o) \\ l_{01}(\phi_o) \end{bmatrix} = \begin{bmatrix} \cos \phi_o & \sin \phi_o \\ -\sin \phi_o & \cos \phi_o \end{bmatrix} \cdot \begin{bmatrix} l_{10} \\ l_{01} \end{bmatrix}, \quad (67)$$

and n = 2, that is,

$$\begin{bmatrix} l_{20}(\phi_o)\\ l_{11}(\phi_o)\\ l_{02}(\phi_o) \end{bmatrix} = T_2^t \cdot \begin{bmatrix} \cos(2\phi_o) & 0 & \sin(2\phi_o)\\ 0 & 1 & 0\\ -\sin(2\phi_o) & 0 & \cos(2\phi_o) \end{bmatrix} \cdot T_2 \cdot \begin{bmatrix} l_{20}\\ l_{11}\\ l_{02} \end{bmatrix},$$
(68)

where

$$T_{2} = \begin{bmatrix} \frac{1}{\sqrt{2}} & 0 & -\frac{1}{\sqrt{2}} \\ \frac{1}{\sqrt{2}} & 0 & \frac{1}{\sqrt{2}} \\ 0 & 1 & 0 \end{bmatrix}$$
(69)

is a unitary transformation matrix (with determinant equal to -1). These results for orders n = 1, 2 can also be obtained by straighforward manipulation of the goniometric

functions involved in (64), for example,

$$l_{1,0}(\phi_o) = \frac{1}{2\pi} \int_0^{2\pi} d\phi \hat{l}_n(\phi) \cos(\phi - \phi_o)$$
  
$$= \frac{1}{2\pi} \int_0^{2\pi} d\phi \hat{l}_n(\phi) \cos\phi \cdot \cos\phi_o$$
  
$$+ \frac{1}{2\pi} \int_0^{2\pi} d\phi \hat{l}_n(\phi) \sin\phi \cdot \sin\phi_o$$
  
$$= l_{10} \cdot \cos\phi_o + l_{01} \cdot \sin\phi_o.$$
 (70)

Since rotation of an image within a circularly symmetric window does not alter the signal energy contained in this window, we obtain that the local energy up to order N satisfies

$$E_N = \sum_{n=0}^{N} \sum_{m=0}^{n} l_{n-m,m}^2 = \sum_{n=0}^{N} \sum_{m=0}^{n} l_{n-m,m}^2(\phi_o), \qquad (71)$$

for all  $N \ge 0$ . This also explains why the above transformation matrices are unitary matrices.

#### 6.2. Optimizing energy compaction

We now discuss how a suitable orientation for the local coordinate axes can be derived from the Hermite coefficients. For a 1D image with orientation  $\phi_o$ , we obtain that

$$l_{n-m,m} = l_n \cdot \alpha_{n-m,m}(\phi_o) \quad \text{or} \quad l_{n-m,m}(\phi_o) = l_n \cdot \alpha_{n-m,m}(0),$$
(72)

so that  $l_{n,0}(\phi_o) = l_n(\phi_o) = l_n$  and  $l_{n-m,m}(\phi_o) = 0$ , for m = 1, ..., n, in this case. We can hence split the local energy up to order *N* as follows:

$$E_{N} = l_{00}^{2} + E_{N,1d}(\phi_{o}) + E_{N,2d}(\phi_{o})$$
  
=  $l_{00}^{2} + \sum_{n=1}^{N} l_{n,0}^{2}(\phi_{o}) + \sum_{n=1}^{N} \sum_{m=1}^{n} l_{n-m,m}^{2}(\phi_{o}),$  (73)

where  $E_{N,1d}(\phi_o)$  and  $E_{N,2d}(\phi_o)$  are the local 1D and 2D energies up to order N along direction  $\phi_o$ , respectively. We therefore propose to use a strategy in which  $\phi_o$  is selected such that  $E_{N,1d}(\phi_o)$  is maximized.

In Figure 12, we show an adaptive Hermite transformation of order N = 2 in which the local orientation  $\phi_o$  has been selected by optimizing  $E_{N,1d}(\phi_o)$ . Note the perfect energy compaction that is obtained for 1D patterns. The 1Damplitude  $A_{N,1d}(\phi_o) = [E_{N,1d}(\phi_o)]^{1/2}$  and 2D-amplitude  $A_{N,2d}(\phi_o) = [E_{N,2d}(\phi_o)]^{1/2}$  are very useful features, since they can be used to distinguish between 1D and 2D features in an image. The 1D features that are detected in this way can be classified based on the set of 1D Hermite coefficients  $l_{n,0}(\phi_o) = l_n$ , for  $n = 0, 1, \dots$  This energy compaction and 1D pattern classification has been exploited in image coding [5, 6] and image-feature extraction [16, 18, 24]. The extracted features have in turn proven useful for single-ended quality predictions (i.e., quality predictions that do not require the original image) on blurred images [19] and JPEGcoded images [45].

#### 6.3. Gradient direction

In differential geometry, the reference angle  $\phi_o$  is often chosen such that the value of the derivative angular function of order n = 1, that is,

$$l_1(\phi_o) = l_{10} \cdot \cos \phi_o + l_{01} \cdot \sin \phi_o, \tag{74}$$

is maximized. This is equivalent to optimizing the 1D energy up to order N = 1. The above angular function can be rewritten as

$$l_1(\phi_o) = l_1 \cdot \cos\left(\phi_o - \phi_d\right),\tag{75}$$

with  $l_1 = \sqrt{l_{10}^2 + l_{01}^2}$ , provided that the angle  $\phi_d$  satisfies

$$\tan \phi_d = \frac{l_{01}}{l_{10}},\tag{76}$$

or equivalently,  $l_{10} = l_1 \cdot \cos \phi_d$  and  $l_{01} = l_1 \cdot \sin \phi_d$ . The maximum value obviously occurs for  $\phi_o = \phi_d$ . This direction that makes an angle  $\phi_d$  with the horizontal axis is called the *gradient direction*.

In the neighbourhood of the generic lattice position (0,0), the smoothed image can be expressed in both the global coordinate system (x, y) and the local coordinate system

$$\begin{bmatrix} x_d \\ y_d \end{bmatrix} = \begin{bmatrix} \cos \phi_d & \sin \phi_d \\ -\sin \phi_d & \cos \phi_d \end{bmatrix} \cdot \begin{bmatrix} x \\ y \end{bmatrix},$$
(77)

that is,  $l(x, y; \sigma) = l_d(x_d, y_d; \sigma)$  are two equivalent descriptions of the same local image. The derivatives for a Cartesian coordinate system that is aligned with this gradient direction can hence be expressed as

$$l_{n-m,m}^{d} = \frac{\partial^{n}}{\partial x_{d}^{n-m} \partial y_{d}^{m}} l_{d}(x_{d}, y_{d}; \sigma) \Big|_{x_{d}=y_{d}=0}$$

$$= \frac{\sqrt{2^{n}(n-m)!m!}}{\sigma^{n}} l_{n-m,m}(\phi_{d}),$$
(78)

for n = 0, 1, 2, ... and m = 0, ..., n. Selecting the reference angle along the gradient direction guarantees that  $l_{01}^d = 0$ . The value

$$l_{10}^{d} = \frac{\sqrt{2}}{\sigma} l_{1} = \frac{\sqrt{2}}{\sigma} \sqrt{l_{10}^{2} + l_{01}^{2}}$$
(79)

is called the *gradient amplitude*. The following explicit relationships

$$l_{20}^{d} = \frac{2\sqrt{2}}{\sigma^{2}} l_{20}(\phi_{d}), \qquad l_{11}^{d} = \frac{2}{\sigma^{2}} l_{11}(\phi_{d}), \qquad l_{02}^{d} = \frac{2\sqrt{2}}{\sigma^{2}} l_{02}(\phi_{d})$$
(80)

exist between the Hermite coefficients of order 2 and the second-order partial derivatives to the smoothed image  $l_d(x_d, y_d; \sigma)$  for coordinate axes  $(x_d, y_d)$  oriented along the gradient direction.



FIGURE 12: Hermite coefficients  $l_{n-m,m}(\phi_o)$ , for m = 0, ..., n, up to order n = 2 for adaptive coordinate axes and a Gaussian window with spread  $\sigma = 2.5$  times the sampling distance. The image in the lower-right corner codes the local orientation  $\phi_o$ . The image at the right in the middle row is the 1D-amplitude  $A_{2,1d}(\phi_o)$ , while the image at the bottom in the middle column is the 2D-amplitude  $A_{2,2d}(\phi_o)$ .

#### 7. GEOMETRIC FEATURES

#### 7.1. Planar curves

The geometric features that we will derive from the Hermite coefficients will all be characteristics of curves that are in some way defined by the surface function  $l(x, y; \sigma)$ . We will only consider curves that belong to a 2D plane, and therefore start by summarizing some basic properties of such *planar curves*. We denote by  $\mathbf{c}(t) = (u(t), v(t))$  the coordinates traced out by a curve segment in an orthogonal coordinate system (u, v) of a 2D plane (this plane need not be the image plane (x, y)). The parameter *t* is defined for an interval of the real axis that includes 0, while  $\mathbf{c}(0) = (u(0), v(0))$  is the point at which we want to measure the curve characteristics. We assume that the functions u(t) and v(t) that relate the parameter *t* to the curve coordinates are at least twice differentiable.

We will use two well-known properties of planar curves that are both illustrated in Figure 13. First, the tangent line to the curve at position c(t) has the direction indicated by the *tangent vector* 

$$\mathbf{c}_{t}(t) = \frac{1}{\left(\left[u'(t)\right]^{2} + \left[v'(t)\right]^{2}\right)^{1/2}} \left(u'(t), v'(t)\right).$$
(81)

This vector can be used as the first vector in a right-handed orthonormal coordinate system. The second coordinate vector thus defined is called the *normal vector* and can be expressed as follows:

$$\mathbf{c}_{n}(t) = \frac{1}{\left(\left[u'(t)\right]^{2} + \left[v'(t)\right]^{2}\right)^{1/2}} \left(-v'(t), u'(t)\right).$$
(82)

As we move along the curve, the direction of the tangent (and normal) vector will change. The next theorem on planar



FIGURE 13: The tangent vector  $\mathbf{c}_t(t)$  at the curve  $\mathbf{c}(t)$  can be combined with the normal vector  $\mathbf{c}_n(t)$  to obtain a right-handed orthonormal coordinate system. The circle with radius  $1/|\kappa(t)|$  has second-order contact with the curve  $\mathbf{c}(t)$ .

curves tells us how to measure this rate of change. More specifically, there exists a circle that is tangent to the curve at  $\mathbf{c}(t)$  and for which the normal vector to the curve turns at the same rate as the normal vector to the circle when moving in the direction of the tangent vector (i.e., this circle has second-order contact with the curve). This circle has radius  $1/|\kappa(t)|$  and its center is located at position

$$\mathbf{c}(t) + \frac{1}{\kappa(t)}\mathbf{c}_n(t),\tag{83}$$

where

$$\kappa(t) = \frac{u'(t) \cdot v''(t) - u''(t) \cdot v'(t)}{\left(\left[u'(t)\right]^2 + \left[v'(t)\right]^2\right)^{3/2}}$$
(84)

is called the *curvature* at position c(t) [43]. A positive or negative curvature implies that the curve bends towards or away from the normal vector, respectively.

#### 7.2. Isophote curvature

A first planar curve that can be derived from the surface  $z = l_d(x_d, y_d; \sigma)$  is the *isophote* through the origin, defined by

$$l_d(u(t), v(t); \sigma) = l_d(u(0), v(0); \sigma)$$
(85)

for some parameter interval including t = 0. This is the curve that arises if the image surface is intersected with a horizontal plane through the surface at the point of interest. The image value along the isophote does not change as a function of t, so that

$$\frac{d}{dt}l_{d}(u(t), v(t); \sigma) \Big|_{t=0} = l_{10}^{d} \cdot u' + l_{01}^{d} \cdot v',$$

$$\frac{d^{2}}{dt^{2}}l_{d}(u(t), v(t); \sigma) \Big|_{t=0}$$

$$= l_{20}^{d} \cdot (u')^{2} + 2l_{11}^{d} \cdot u'v' + l_{02}^{d} \cdot (v')^{2} + l_{10}^{d} \cdot u'' + l_{01}^{d} \cdot v''$$
(86)

are both equal to zero. Since the coordinate system is aligned along the gradient direction, that is,  $l_{01}^d = 0$ , we obtain that u'(0) = 0 and  $u''(0) = -[v'(0)]^2 \cdot l_{02}^d / l_{10}^d$ , which in turn leads to the expression

$$\kappa_{i} = -\frac{u''(0)}{\left[\nu'(0)\right]^{2}} = \frac{l_{02}^{d}}{l_{10}^{d}} = \frac{2}{\sigma} \frac{l_{02}(\phi_{d})}{l_{10}(\phi_{d})},$$
(87)

for the *isophote curvature* at the origin.

Since isophotes will have a high curvature near a corner, such as the one shown in Figure 14, the isophote curvature can assist in corner detection. More specifically, finding local maxima in the derived *isophote curvature measure* 

$$m_{i} = \left| \kappa_{i} \cdot \left( l_{10}^{d} \right)^{3} \right|^{1/3} = \left| l_{02}^{d} \cdot \left( l_{10}^{d} \right)^{2} \right|^{1/3}, \qquad (88)$$

which also contains contrast information by including the image gradient  $l_{10}^d$ , is a frequently used method for selecting corner candidates. Figure 14 illustrates that, although the position of a maximum in the isophote curvature measure is usually very close to the true corner position, both positions will usually not coincide. This implies that a maximum in the isophote curvature measure can at best be used as an indication for the presence of a corner, and that a more indepth analysis is required, for instance to update the position estimate of the corner feature [32].

The isophote curvature measure for the entire test image of Figure 8 is shown in Figure 15(a).

#### 7.3. Flowline curvature

A second planar curve that can be derived from the surface  $z = l_d(x_d, y_d; \sigma)$  is the *flowline* through the origin, defined by the demand that the tangent to the flowline is equal to the image gradient, that is,

$$u'(t) = \frac{\partial}{\partial x_d} l_d(x_d, y_d; \sigma) \bigg|_{x_d = u(t), y_d = v(t)},$$

$$v'(t) = \frac{\partial}{\partial y_d} l_d(x_d, y_d; \sigma) \bigg|_{x_d = u(t), y_d = v(t)},$$
(89)

for some parameter interval including t = 0. Note that, at the origin, the flowline is orthogonal to the isophote.

Evaluating the second derivatives to the flowline at the origin results in

$$u''(0) = l_{20}^d \cdot u'(0) + l_{11}^d \cdot v'(0) = l_{20}^d \cdot l_{10}^d,$$
  

$$v''(0) = l_{11}^d \cdot u'(0) + l_{02}^d \cdot v'(0) = l_{11}^d \cdot l_{10}^d$$
(90)



FIGURE 14: Curvature measures for a corner feature extracted from Figure 8. Upper left: original image, lower left: smoothed image, upper middle: isophote curvature measure  $|m_i|$ , and lower middle: flowline curvature measure  $|m_f|$ . The rightmost figures use the "spot noise" technique [46] to visualize the isophote vector field (upper) and the flowline (or gradient) vector field that is orthogonal to it (lower).

which immediately leads to the expression

$$\kappa_f = \frac{\nu''(0)}{\left[u'(0)\right]^2} = \frac{l_{11}^d}{l_{10}^d} = \frac{\sqrt{2}}{\sigma} \frac{l_{11}(\phi_d)}{l_{10}(\phi_d)}$$
(91)

for the *flowline curvature* at the origin [32], since  $u'(0) = l_{10}^d$ and  $v'(0) = l_{01}^d = 0$ .

The derived *flowline curvature measure* 

$$m_f = \left| \kappa_f \cdot \left( l_{10}^d \right)^3 \right|^{1/3} = \left| l_{11}^d \cdot \left( l_{10}^d \right)^2 \right|^{1/3}$$
(92)

that also depends on the image gradient can be shown to react especially to junctions, such as the T-junction shown in Figure 16. Again note that the position of the maximum in the flowline curvature measure does not coincide exactly with the true position of the T-junction. The flowline curvature measure for the entire test image of Figure 8 is shown in Figure 15(b).

In conclusion, the feature vector  $(|m_i|, |m_f|)$  is very useful for the (inital) detection and classification of corners and junctions. Candidate corner/junction positions are obtained by finding local maxima (and minima) in these curvature measures. The next section discusses some geometric image features that can be used to find such local extrema in (feature) images.

#### 7.4. Intersection and surface curvatures

#### 7.4.1. Intersection curvatures

A third kind of planar curve that can be derived from the surface  $z = l(x, y; \sigma)$  is the intersection curve with a vertical

plane through the origin in the direction that makes an angle  $\phi$  with the global *x*-axis (or an angle  $\phi - \phi_d$  with the local gradient axis  $x_d$ ). This curve can be parametrized by u(t) = t and

$$v(t) = l(t \cdot \cos \phi, t \cdot \sin \phi; \sigma)$$
  
=  $l_d(t \cdot \cos (\phi - \phi_d), t \cdot \sin (\phi - \phi_d); \sigma).$  (93)

The first-order derivatives for t = 0 are equal to u'(0) = 1and

$$v'(0) = l_1(\phi) \frac{\sqrt{2}}{\sigma},$$
 (94)

while higher-order derivatives satisfy  $u^{(n)}(0) = 0$  and

$$v^{(n)}(0) = l_n(\phi) \frac{\sqrt{2^n n!}}{\sigma^n},$$
 (95)

where  $l_n(\phi)$  is the "derivative" angular function of order *n*. The intersection curve hence has an *intersection curvature* in the direction  $\phi$  equal to

$$\kappa(\phi) = \frac{\sigma l_2(\phi)}{\left[(\sigma/\sqrt{2})^2 + (l_1(\phi))^2\right]^{3/2}}.$$
(96)

The intersection curvature will vary between  $\kappa_{\min} = \kappa(\phi_{\min})$ and  $\kappa_{\max} = \kappa(\phi_{\max})$ . These minimum and maximum curvatures can be used directly as image features or they can be combined into  $\kappa_l = \kappa_{\max} + \kappa_{\min}$ ,  $\kappa_g = \kappa_{\max} \cdot \kappa_{\min}$ , or  $\kappa_r = |\kappa_{\max}^2 - \kappa_{\min}^2|$ . These latter curvature measures are illustrated in the left part of Figure 17. An interpretation of



FIGURE 15: Curvature measures: (a) isophote curvature measure  $|m_i|$  and (b) flowline curvature measure  $|m_f|$ .



FIGURE 16: Curvature measures for a T-junction feature extracted from Figure 8. Upper left: original image, lower left: smoothed image, upper middle: isophote curvature measure  $|m_i|$ , and lower middle: flowline curvature measure  $|m_f|$ . The rightmost figures use the "spot noise" technique [46] to visualize the isophote vector field (upper) and the flowline (or gradient) vector field that is orthogonal to it (lower).

these curvature measures will follow after the discussion on surface curvatures.

The intersection curvature can also be used to define the *intersection curvature measure* 

$$m(\phi) = \kappa(\phi) \left(\frac{\sqrt{2}}{\sigma} l_1(\phi)\right)^3$$
  
=  $\frac{2\sqrt{2}}{\sigma^2} l_2(\phi_o) \cdot \frac{(l_1(\phi))^3}{[(\sigma/\sqrt{2})^2 + (l_1(\phi))^2]^{3/2}},$  (97)

which is approximately equal to  $l_2(\phi) \cdot (2\sqrt{2}/\sigma^2)$  if the first-order directional derivative  $l_1(\phi) \cdot \sqrt{2}/\sigma \gg 1$ . The

intersection curvature measure is hence equal to the secondorder directional derivative if the first-order directional derivative is sufficiently large.

#### 7.4.2. Surface curvatures

Intersection curvatures are especially useful at positions where the gradient is zero, that is, for which  $l_1(\phi) = 0$  for all directions  $\phi$ , while  $l_2(\phi) \neq 0$  for some angles  $\phi$ , or equivalently  $\sum_{m=0}^{2} l_{2-m,m}^2 \neq 0$ . Such positions are called *singular points*, and are important landmarks in an image. They can be subdivided into minima, maxima, and saddle points.



FIGURE 17: Curvature measures. Left from top to bottom: intersection curvatures  $\kappa_l$ ,  $\text{sgn}(\kappa_g)|\kappa_g|^{1/2}$ , and  $|\kappa_r|^{1/2}$ . Right from top to bottom: surface curvatures  $\tilde{\kappa}_l$ ,  $\text{sgn}(\tilde{\kappa}_g)|\tilde{\kappa}_g|^{1/2}$ , and  $|\tilde{\kappa}_r|^{1/2}$ .

In this case, the intersection curvatures are identical to the familiar *surface curvatures* [43], since translations in the image plane correspond one to one to translations in the tangent plane. For a singular point, we can arbitrarily select  $\phi_d = 0$  as the reference angle, and obtain the following expression for the surface curvature in the direction  $\phi$ :

$$\begin{aligned} \widetilde{\kappa}(\phi) &= \frac{2\sqrt{2}}{\sigma^2} l_2(\phi) \\ &= l_{20}^d \cdot \cos^2 \phi + 2l_{11}^d \cdot \cos \phi \sin \phi + l_{02}^d \cdot \sin^2 \phi \\ &= \frac{l_{20}^d + l_{02}^d}{2} + \sqrt{\left(l_{11}^d\right)^2 + \left(\frac{l_{20}^d - l_{02}^d}{2}\right)^2} \cdot \cos\left[2(\phi - \phi_c)\right], \end{aligned}$$
(98)

where the direction  $\phi_c$  of maximum surface curvature

$$\widetilde{\kappa}_{\max} = \frac{l_{20}^d + l_{02}^d}{2} + \sqrt{\left(l_{11}^d\right)^2 + \left(\frac{l_{20}^d - l_{02}^d}{2}\right)^2}$$
(99)

satisfies

$$\tan\left(2\phi_{c}\right) = \frac{2l_{11}^{d}}{l_{20}^{d} - l_{02}^{d}} = \frac{\sqrt{2}l_{11}}{l_{20} - l_{02}}.$$
 (100)

The direction of minimum surface curvature

$$\widetilde{\kappa}_{\min} = \frac{l_{20}^d + l_{02}^d}{2} - \sqrt{\left(l_{11}^d\right)^2 + \left(\frac{l_{20}^d - l_{02}^d}{2}\right)^2}$$
(101)

is  $\phi_c + \pi/2$ . The surface curvatures at a singular point hence belong to the range [ $\tilde{\kappa}_{\min}, \tilde{\kappa}_{\max}$ ], where the sum of the two

extreme curvatures is equal to the Laplacian, that is,

$$\widetilde{\kappa}_l = \widetilde{\kappa}_{\max} + \widetilde{\kappa}_{\min} = l_{20}^d + l_{02}^d = \text{trace } \mathcal{H}, \qquad (102)$$

and the product of the two extrema curvatures is equal to the *Gaussian curvature* 

$$\widetilde{\kappa}_g = \widetilde{\kappa}_{\max} \cdot \widetilde{\kappa}_{\min} = l_{20}^d \cdot l_{02}^d - l_{11}^d \cdot l_{11}^d = \det \mathcal{H}.$$
(103)

Both can be expressed in terms of the Hessian matrix

$$\mathcal{H} = \begin{bmatrix} l_{20}^d & l_{11}^d \\ l_{11}^d & l_{02}^d \end{bmatrix} = \frac{2}{\sigma^2} \begin{bmatrix} \sqrt{2}l_{20} & l_{11} \\ l_{11} & \sqrt{2}l_{02} \end{bmatrix}.$$
 (104)

A large absolute value for the Gaussian curvature indicates isolated singular points, since both extreme curvatures must have a large amplitude in this case. If the Gaussian curvature is positive, then both extreme curvatures have the same sign, which can only occur in case of an extremum point. The sign of the Laplacian  $\tilde{\kappa}_l$  allows to distinguish between minima ( $\tilde{\kappa}_l > 0$ ) and maxima ( $\tilde{\kappa}_l < 0$ ) in this case. A negative value for the Gaussian curvature indicates a saddle point.

If one of the extreme curvatures  $\tilde{\kappa}_{max}$  or  $\tilde{\kappa}_{min}$  is zero, then the Gaussian curvature  $\tilde{\kappa}_g$  is also zero. This occurs in case the singular points are not isolated but align to form extended ridges (line-like structures). The *ridge strength* 

$$\widetilde{\kappa}_{r}^{2} = |\widetilde{\kappa}_{\max}^{2} - \widetilde{\kappa}_{\min}^{2}|^{2} = (l_{20}^{d} + l_{02}^{d})^{2} [4(l_{11}^{d})^{2} + (l_{20}^{d} - l_{02}^{d})^{2}]$$
(105)

can be used to find singular points for which there is a substantial difference between the amplitudes of both extreme curvatures. A large value for the ridge strength  $\tilde{\kappa}_r$ , in combination with a small value for the Gaussian curvature  $\tilde{\kappa}_g$ , is then evidence for the presence of ridge points.

All three mentioned surface curvature measures are illustrated in the right part of Figure 17.

Although the surface curvature measures only have a geometrical interpretation at singular points, they give nonzero responses for most nonuniform regions in an image. The gradient is hence needed alongside these measures in order to guide the interpretation. It is straightforward to derive the relationship

$$\kappa(\phi) = \frac{\widetilde{\kappa}(\phi)}{\left[1 + \left(l_{10}^d \cdot \cos\left(\phi - \phi_d\right)\right)^2\right]^{3/2}}$$
(106)

between intersection and surface curvature at positions where the gradient amplitude is  $l_{10}^d$  and the gradient direction is  $\phi_d$ . Because the gradient amplitude is usually nonzero, except for uniform regions and singular points, the surface curvature differs from the intersection curvature in the sense that contributions for nonsingular points are being suppressed in the latter. This can be observed in Figure 17 by comparing the corresponding intersection and surface curvature measures on the left and right, respectively. This suppression of nonsingular points is expected to be beneficial for subsequent pattern matching or classification. Note that, while edges are nonsingular points of the image itself, they are singular points of the image gradient, so that the focus on singular points can also be used to perform edge detection. Some intersection and surface curvature measures obtained from a gradient image are illustrated in Figure 18. A large value for the ridge strength  $\kappa_r$ , in combination with a small value for the Gaussian curvature  $\kappa_g$ , is then an evidence for the presence of edge points.

#### 8. ANISOTROPIC NOISE REDUCTION

We finish by applying the partial image decomposition technique of Figure 5 to the case of the Hermite transformation in order to create an anisotropic noise reduction and contrast enhancement algorithm. More specifically, we will use a Hermite transformation up to order N, so that the index set  $N_o$ contains the Hermite coefficients  $l_{n-m,m}$ , for n = 0, 1, ..., Nand m = 0, ..., n. The mapping of these Hermite coefficients can be influenced by the following control variables.

(1) As in the case of residue-image processing, the average value  $l_{00}$  and the local energy *E* are used to determine the noise in the image. It was discussed earlier how the noise characteristics can be estimated from the histogram of the residue energy  $E_r = E - l_{00}^2$  or residue amplitude  $A_r = \sqrt{E_r}$ .

(2) The Hermite coefficients up to order *N* can be mapped into a 1D and 2D component. Most often, the 1D energy  $E_{2,1d}(\phi_o)$  in the first- and second-order Hermite coefficients is maximized to determine the direction  $\phi_o$  of the local coordinate axes. The resulting adaptive Hermite transformation with coefficients  $l_{n-m,m}(\phi_o)$ , for n = 0, 1, ..., N and m = 0, ..., n, allows to separate the energy up to order *N* into a 1D and 2D component, that is,  $E_N = E_{N,1d}(\phi_o) + E_{N,2d}(\phi_o)$ . The manpings<sup>2</sup>

The mappings<sup>2</sup>

$$\begin{split} \widetilde{l}_{00}(\phi_o) &= (1-\kappa) \cdot l_{00}(\phi_o), \\ \widetilde{l}_{n0}(\phi_o) &= (\kappa_1 - \kappa) \cdot l_{n0}(\phi_o), \\ \widetilde{l}_{n-m,m}(\phi_o) &= (\kappa_2 - \kappa) \cdot l_{n-m,m}(\phi_o), \quad \text{for } m = 1, \dots, n \end{split}$$

for the adaptive Hermite coefficients of orders n = 0, 1, ..., Nspecify the adaptive algorithm, as depicted in Figure 5. The factors  $\kappa_1$ ,  $\kappa_2$ , and  $\kappa$  are the amplification factors of the 1D Hermite component, the 2D Hermite component, and the residue component, respectively. The case of  $\kappa_1 = \kappa_2 = \kappa$ corresponds to the residue-image processing algorithm of Section 4.

We of course need to make a specific choice for the amplification factors in order to realize our anisotropic noise reduction and/or contrast enhancement algorithm. Again, as in the case of residue-image processing, there are some heuristics involved in making this choice. A simple modification to the residue-image processing algorithm of Section 4 leads to

<sup>&</sup>lt;sup>2</sup> Based on the argumentation in [14, 33], a mapping with  $\kappa_1$  and  $\kappa_2$  replaced by  $\kappa_1^n$  and  $\kappa_2^n$  may possibly constitute a better choice.



FIGURE 18: Curvature measures on gradient image. Left from top to bottom: intersection curvatures  $\kappa_l$ , sgn( $\kappa_g$ ) $|\kappa_g|^{1/2}$ , and  $|\kappa_r|^{1/2}$ . Right from top to bottom: surface curvatures  $\tilde{\kappa}_l$ , sgn( $\tilde{\kappa}_g$ ) $|\tilde{\kappa}_g|^{1/2}$ , and  $|\tilde{\kappa}_r|^{1/2}$ .

the following suggestion:

$$\kappa = \kappa_2 = s + (1 - s) \frac{A_v}{A_r}, \qquad \kappa_1 = s_1 + (1 - s_1) \frac{A_v}{A_r}$$
 (108)

with

$$s_1 = s \cdot \left[ 1 - (1 - \alpha) \cdot \left( \frac{E_{N,1d}(\phi_o)}{E_N} \right)^{\beta} \right], \qquad (109)$$

for  $A_t \leq A_r \leq A_v$ . The slope  $s_1$  varies from s, in case  $E_{N,1d}(\phi_o) = 0$ , to  $\alpha \cdot s$ , with  $0 \leq \alpha \leq 1$ , for purely 1D features with  $E_{N,1d}(\phi_o) = E_N$ . The power  $\beta \geq 0$  controls the steepness of this transition. No anisotropic processing is applied outside the range  $[A_t, A_v]$ , that is,  $\kappa = \kappa_1 = \kappa_2 = 0$  for  $A_r < A_t$  and  $\kappa = \kappa_1 = \kappa_2 = 1$  for  $A_r > A_v$ .

In the upper-left corner of Figure 19, we show a detail of an original image. The results of anisotropic processing are illustrated in the bottom row of Figure 19 for some fairly arbitrary choices of the parameters. Specifically,  $\alpha = 0$ ,  $\beta = 0.2$ for the image in the lower left, and  $\alpha = 0$ ,  $\beta = 5$  for the image in the lower right. The value  $A_t = 3$  is at twice the mode of the residue amplitude histogram, while the parameters controlling the amplification factors are fixed at  $A_v = 30$ and s = 0.5. The image in the upper right-hand corner has been included for comparison. It is obtained using the residue-image processing algorithm of Section 4 (with the same parameter values) and corresponds to isotropic image processing. As can also be observed in the figure, the differences between isotropic and anisotropic filtering are often very subtle, so that a formal image-quality comparison [24]



FIGURE 19: Results of anisotropic image processing with the Hermite transformation of order N = 2 and  $\sigma = 1.5$  times the sampling distance. The original image is in the upper left-hand corner. The image in the upper right-hand corner is obtained through isotropic residue-image processing. The two images in the bottom row are obtained with the anisotropic image processing algorithm.

is probably needed to determine if the extra effort required for anisotropic filtering is truly worthwhile.

#### 9. CONCLUSION

With this survey we have attempted to give an overview of the main characteristics of the Hermite transformation and their implications for applications. In summary, we can state that the characteristics of the two-dimensional Hermite transformation at a single scale (i.e., for one value of the Gaussian standard deviation  $\sigma$ ) are by now well understood. In our view, the main challenges for the future lie in the extension to higher dimensions [26] and in a better exploitation of the characteristics of the Hermite transformation across multiple scales (i.e., more advanced than the simple pyramid extension used most of the time today). We need a better understanding of how the Hermite coefficients relate across scales, and how this relationship can be exploited more profitably for image coding, image processing, and image analysis [33, 42]. An improved link between the two alternative ways of interpreting multiscale Gaussian derivatives, that is, as analysis functions of the Hermite transformation or within the mathematical theory of scale space, may potentially also profit both approaches.

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# Special Issue on OFDMA Architectures, Protocols, and Applications

### **Call for Papers**

Orthogonal frequency-division multiple access (OFDMA) technologies are currently attracting intensive attention in wireless communications to meet the ever-increasing demands arising from the explosive growth of Internet, multimedia, and broadband services. OFDMA-based systems are able to deliver high data rate, operate in the hostile multipath radio environment, and allow efficient sharing of limited resources such as spectrum and transmit power between multiple users. OFDMA has been used in the mobility mode of IEEE 802.16 WiMAX, is currently a working specification in 3GPP Long Term Evolution downlink, and is the candidate access method for the IEEE 802.22 "wireless regional area networks." Clearly, recent advances in wireless communication technology have led to significant innovations that enable OFDMA-based wireless access networks to provide better quality-of-service (QoS) than ever with convenient and inexpensive deployment and mobility.

However, regardless of the technology used, OFDMA networks must not only be able to provide reliable and high quality broadband services, but also be implemented costeffectively and be operated efficiently. OFDMA presents many of the advantages and challenges of OFDM systems for single users, and the extension to multiple users introduces many further challenges and opportunities, both on the physical layer and at higher layers. These requirements present many challenges in the design of network architectures and protocols, which have motivated a significant amount of research in the area. Also, many critical problems associated with the applications of OFDMA technologies in future wireless systems are still looking for efficient solutions. The aim of this special issue is to present a collection of high-quality research papers that report the latest research advances in this field from physical and network layers to practical applications. Original papers are solicited in all aspects of OFDMA techniques including physical layer issues, architectures, protocol designs, enabling technologies, theoretical studies, practical applications, and experimental prototypes. Topics of interest include, but are not limited to:

- Adaptive coding and modulation
- Signal processing for OFDMA
- Interference control techniques

- Bandwidth and resources allocation
- Efficient MAC protocol development
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# Special Issue on Active Antennas for Space Applications

## **Call for Papers**

Over the last years, many journal articles appeared on the principles, analysis, and design of active and active integrated antennas (AAs and AIAs). An AA is a single system comprising both a radiating element and one or more active components which are tightly integrated. This gives clear advantages in terms of costs, dimensions, and efficiency. In the case of an AIA, both the active device and the radiator are integrated on the same substrate. Both options lead to very compact, lowloss, flexible antennas, and this is very important especially at high frequencies, such as those typical of a satellite link. As microwave integrated-circuit and the microwave monolithic integrated-circuit technologies have ripened, AA and AIA applications have become more and more interesting, not only at a scientific level but also from a commercial point of view, up to the point that they have recently been applied to phased array antennas on board moving vehicles for satellite broadband communication systems.

The goal of this special issue it to present the most recent developments and researches in this field, with particular attention to space-borne applications, as well as to enhance the state of the art and show how AAs and AIAs can meet the challenge of the XXI century telecommunications applications.

Topics of interest include, but are not limited to:

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# Special Issue on Applications of Signal Processing Techniques to Bioinformatics, Genomics, and Proteomics

# **Call for Papers**

The recent development of high-throughput molecular genetics technologies has brought a major impact to bioinformatics, genomics, and proteomics. Classical signal processing techniques have found powerful applications in extracting and modeling the information provided by genomic and proteomic data. This special issue calls for contributions to modeling and processing of data arising in bioinformatics, genomics, and proteomics using signal processing techniques. Submissions are expected to address theoretical developments, computational aspects, or specific applications. However, all successful submissions are required to be technically solid and provide a good integration of theory with practical data.

Suitable topics for this special issue include but are not limited to:

- Time-frequency representations
- Spectral analysis
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# Special Issue on Iterative Decoding and Cross-Layering Techniques for Multimedia Broadcasting and Communications

### **Call for Papers**

The explosive growth of multimedia applications over the Internet and the ever-increasing users' demands over commercial terrestrial digital multimedia broadcasting all over the world call for efficient physical and cross-layer techniques able to mitigate the potential problems limiting broadband services over wireless networks. In this scenario, mobile multimedia is expected to be one of the key services of future wireless mobile networks. Meanwhile, recent advances in digital communications have paved the way to a variety of standards aimed at providing multimedia services over terrestrial broadband networks. To cite but a few, DVB-H, T-DVB, T-DMB, wireless LANs, and wireless MANs are some of the most recent standards enabling such technology.

Iterative decoding techniques for both source, channel, and joint source-channel coding and decoding and crosslayering techniques have proven to be very effective for providing a viable means of achieving capacity-approaching performance at very reduced computational burden.

The main aim of this special issue is to highlight stateof-the-art techniques on the most recent research advances enabling digital multimedia services over broadband wireless networks, focused on physical and cross-layering solutions. Novel contributions, previously unpublished, that are not being submitted to any other journal, are sought.

Topics of interests include, but are not limited to, the following subject categories:

- Iterative decoding techniques for concatenated channel codes (turbo codes and serially concatenated codes)
- Iterative decoding techniques for joint source-channel decoding
- Novel capacity-approaching channel codes: coding strategies and efficient decoding algorithms
- Cross-layer modelling/analysis and optimization techniques
- Standardization activities on digital multimedia broadcasting protocols
- Space-time coding and decoding

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# Special Issue on Fairness in Radio Resource Management for Wireless Networks

### **Call for Papers**

Radio resource management (RRM) techniques (such as admission control, scheduling, sub-carrier allocation, channel assignment, power allocation, and rate control) are essential for maximizing the resource utilization and providing quality of service (QoS) in wireless networks.

In many cases, the performance metrics (e.g., overall throughput) can be optimized if opportunistic algorithms are employed. However, opportunistic RRM techniques always favor advantaged users who have good channel conditions and/or low interference levels. The problem becomes even worse when the wireless terminals have low mobility since the channel conditions become slowly varying (or even static), which might lead to long-term unfairness. The problem of fair resource allocation is more challenging in multihop wireless networks (e.g., mesh and multihop cellular networks).

The performance fairness can be introduced as one of the QoS requirements (e.g., as a condition on the minimum throughput per user). Fair RRM schemes might penalize advantaged users; hence, there should be a tradeoff between the overall system performance and the fairness requirements.

We are soliciting high-quality unpublished research papers addressing the problem of fairness of RRM techniques in wireless communication systems. Topics include (but are not limited to):

- Fairness of scheduling schemes in wireless networks
- Tradeoff between maximizing the overall throughput and achieving throughput fairness
- RRM fairness: problem definition and solution techniques
- Fairness performance in emerging wireless systems (WiMAX, ad hoc networks, mesh networks, etc.)
- Cross-layer RRM design with fairness
- Short-term and long-term fairness requirements
- Adaptive RRM to support fairness
- Fairness in cooperative wireless communications
- Issues and approaches for achieving fairness in multihop wireless networks

- RRM framework and QoS architecture
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